

# Command Line Administration Interface – User Guide

(version 3.1)

#### **Disclaimer**

THE SPECIFICATIONS AND INFORMATION REGARDING THE PRODUCTS IN THIS MANUAL ARE SUBJECT TO CHANGE WITHOUT NOTICE. ALL STATEMENTS, INFORMATION, AND RECOMMENDATIONS IN THIS MANUAL ARE BELIEVED TO BE ACCURATE BUT ARE PRESENTED WITHOUT WARRANTY OF ANY KIND, EXPRESS OR IMPLIED. USERS MUST TAKE FULL RESPONSIBILITY FOR THEIR APPLICATION OF ANY PRODUCTS.

THE SOFTWARE LICENSE AND LIMITED WARRANTY FOR THE ACCOMPANYING PRODUCT ARE SET FORTH IN THE INFORMATION PACKET THAT SHIPPED WITH THE PRODUCT AND ARE INCORPORATED HEREIN BY THIS REFERENCE. IF YOU ARE UNABLE TO LOCATE THE SOFTWARE LICENSE OR LIMITED WARRANTY, CONTACT YOUR WYDE VOICE REPRESENTATIVE FOR A COPY.

IN NO EVENT SHALL WYDE VOICE OR ITS SUPPLIERS BE LIABLE FOR ANY INDIRECT, SPECIAL, CONSEQUENTIAL, OR INCIDENTAL DAMAGES, INCLUDING, WITHOUT LIMITATION LOST PROFITS OR LOSS OR DAMAGE TO DATA ARISING OUT OF THE USE OR INABILITY TO USE THIS MANUAL, EVEN IF WYDE OR ITS SUPPLIERS HAVE BEEN ADVISED OF THE POSSIBILITY OF SUCH DAMAGES.

# Copyright

Except where expressly stated otherwise, the Product is protected by copyright and other laws respecting proprietary rights. Unauthorized reproduction, transfer, and or use can be a criminal, as well as civil, offense under the applicable law.

WYDE Voice and the WYDE Voice logo are registered trademarks of WYDE Voice LLC in the United States of America and other jurisdictions. Unless otherwise provided in this Documentation, marks identified with "R" / ®, "TM" / TM and "SM" are registered marks; trademarks are the property of their respective owners.

For the most current versions of documentation, go to the WYDE support Web site: <a href="http://docs.wydevoice.com/">http://docs.wydevoice.com/</a>

April 23, 2013

# Symbols and Notations in this Manual

The following notations and symbols can be found in this manual.



Denotes any item that requires special attention or care. Damage to the equipment or the operator may result from failure to take note of the noted instructions

**Figure** Denotes any illustration

TableDenotes any table

Text Denotes any text output

Folder/File Denotes any folders (paths) or files names

commands Denotes any commands, attributes and parameters

# **Table of Contents**

Symbols and Notations in this Manual	3
Table of Contents	4
Tables List	12
Figures List	13
Chapter 1: Introduction	16
Assumed Skills	16
Architecture Overview	16
Administration Interfaces	17
Hardware	18
Voice Clients	18
Integration Adapters	18
Definitions, Acronyms and Abbreviations	18
Chapter 2: Command Line Interface	
Prerequisites	
Accessing the Command Line Administration Interface	23
Using the wyde Command Line Utility	23
Using the mf Console	
Subscriber User Management	30
Create a Subscriber	30
View/Modify a Subscriber	
Delete a Subscriber	
Create a Conference Account	33
Configuration of the Conferences	34
View/Modify a Conference Account	35
Delete a Conference Account	
Override a Call Flow Attribute Value for a Conference	36
View Call Flow Attributes Values for a Conference	37
Delete a Conference Call Flow Attribute Redefinition	38
Call Flow and DNIS Management	39
Add a Call Flow	39
View a Call Flow	40
Delete a Call Flow	41
Set a Call Flow Attribute Value	41
View Call Flow Attributes Values	42
Reload All Call Flows	43
Update Call Flow Attributes Definition in a Database	43
Create a DNIS Association	44
View/Modify a DNIS Association	45
Create a DNIS Number Alias	45
View DNIS Number Aliases	46
Delete a DNIS Number Alias	47
Delete a DNIS Association	47
Override a Call Flow Attribute Value for a DNIS	48
View Call Flow Attributes Values for a DNIS	49

Delete a DNIS Call Flow Attribute Redefinition	
Reload All DNISes Caches	
View/Modify Callers Gain Options	
Conferences and Calls Management	
View Conferences and Calls in Progress	
View Conferences in Progress	
View Conference Calls in Progress	
Show Calls that have not placed to Conferences	
Calls Management using mf Console	
Dropping Call Participants	
Change Callers Roles	
Mute Call Participants	
Placing Call Participants on Hold	
Set Custom Name for a Call	
Set Audio Key for a Call	
Conferences Management using <i>mf</i> Console	
Dropping a Conference.	
Dropping a Conferences Hosts, Participants and Listeners	
Conference Mute and Q&A Modes	
Q&A Sessions	
Placing a Conference on Hold	
Activate a Conference for Participants and Listeners	
Making a Conference Secure	
Setting a Conference Job Code	
Recording a Conference	
Playing an Audio File to a Conference.	
Dialing another User	
Move a Call to another Conference	
Move a Call between Sub-Conferences	
Sending Chat Messages to the Conference Callers	
Conferences Broadcasting	
Making a Shunt between Two Conferences	
Polling	
Scheduling	
Maintaining Conferences Maximal Duration	
Set Call Flow Attributes for Started Conferences	
Conferences and Calls Management using asterisk Console	
Using the IVR/asterisk Console	
View Conferences in Progress	
View Conference Calls in Progress	
Show Calls that have not placed to Conferences	
Dropping Conference Call Participants	
Show Session Events Queue	
Show Node Statistics	
Conferences and Calls Management using mp Console	
Using the <i>mp</i> Console	
Show Different <i>mp</i> Information	94

Start and Stop <i>mp</i> Console Timers	96
Dropping Boards and Calls	97
Restarting and Stopping mp Boards and Logs	
Operator Conferences	
Operator Conferences Management using mf Console	100
Samples and Use-Cases of Operator Conference Procedures	105
WYDE Bridge Administration	107
Monitoring	107
WYDE Bridge Settings Management	110
Bridge Configuration Changes	118
Dialout Settings Configuration	119
Recording Settings Configuration	121
Operator Invite Settings Configuration	
AGI Server Pool Configuration for Processes Handlers Requests	123
Bridge Configurations Reloading	124
WYDE Bridge Configuration Save and Restore	125
Nodes Administration	126
Distributed Conferencing Administration	129
Peers Management	137
Calls Transferring	138
Audio Prompts Management	138
Max Number of Calls and Licensing	141
Authorization Adapters and Methods	142
Add an Authorization Adapter	
Delete an Authorization Adapter	
View Authorization Adapters	
Add an Authorization Method	
Delete an Authorization Method	
Modify an Authorization Method	
View Authorization Methods	
Billing	
Add a Billing Adapter	
Delete a Billing Adapter	149
Modify a Billing Adapter	
View Billing Adapters	
Add a Billing Rule	
Delete a Billing Rule	
Modify a Billing Rule	
View Billing Rules	
Database Administration	
Asterisk Administration	
SIP Listening Host IP and Port Configuration	
MF Connection on Non-default Host IP and Port Configuration	
Enable and Disable Audio Codecs	
Chapter 3: Command Reference	
wyde Command Reference	160

ast-status (Show WYDE asterisk Status)	160
auth-adapter-add (Add auth Adapter)	160
auth-adapter-del (Delete auth Adapter)	160
auth-adapter-show (Show auth Adapters)	160
auth-method-add (Add auth Method)	160
auth-method-del (Delete auth Method)	161
auth-method-set (Set auth Method)	161
auth-method-show (Show auth Methods)	161
auth-reload (Reload auth configuration)	161
billing-adapter-add (Add Billing Adapter)	161
billing-adapter-del (Delete Billing Adapter)	161
billing-adapter-set (Set Billing Adapter Properties)	161
billing-adapter-show (Show Billing Adapters)	162
billing-reload (Reload Billing Configuration)	162
billing-rule-add (Add Billing Rule)	162
billing-rule-del (Delete Billing Rule)	162
billing-rule-set (Set Billing Rule)	162
billing-rule-show (Show Billing Rules)	163
bridge-add (Add WYDE Bridge)	163
bridge-del (Delete WYDE Bridge)	163
bridge-list-load (Load List of Bridges from the File)	163
bridge-list-save (Save List of Bridges to the File)	163
bridge-reload (Reload DC Bridges Configuration)	164
bridge-set (Set WYDE Bridge Properties)	164
bridge-show (Show WYDE Bridges)	164
caller-gain-set (Set Gain Options)	164
caller-gain-show (Show Gain Options)	165
callflow-add (Add Call Flow)	165
callflow-attr-set (Set Call Flow Attribute)	165
callflow-attr-show (Show Call Flow Attributes)	165
callflow-attr-update-db (Update Call Flow Attributes Definition in a	
Database)	
callflow-del (Delete Call Flow)	
callflow-reload (Reload All Call Flows)	166
callflow-show (Show Call Flows Table)	
conference-attr-del (Remove Conference Attribute Redefinition)	166
conference-attr-set (Set Conference Attribute)	
conference-attr-show (Show Conference Attributes)	166
config-restore (Restore WYDE Configuration)	
config-save (Save WYDE Configuration)	167
confuser-add (Add Conference User)	167
confuser-del (Delete Conference User)	167
confuser-show (Show Conference Users Table)	167

db (Connect to WYDE Main, i.e. <i>dnca</i> , Database)	167/
db-bil (Connect to WYDE Billing, i.e. dnca_calls, Database)	167
db-drop (Drop of WYDE Database)	168
db-init (Initialize of WYDE Database)	168
db-patch (Apply Last Patches for Databases)	168
did-add (Add DNIS/DID Number)	168
did-alias-add (Add DNIS/DID Number Alias)	168
did-alias-del (Delete DNIS/DID Number Alias)	168
did-alias-show (Show DNIS/DID Number Aliases)	169
did-attr-del (Remove DNIS/DID Attribute Redefinition)	169
did-attr-set (Set DNIS/DID Attribute)	169
did-attr-show (Show DNIS's Attributes)	169
did-del (Delete DNIS/DID Number)	169
did-reload (Reload all DNISes/DIDs Caches)	169
did-show (Show DNISes/DIDs Table)	169
drop-call (Drop Call)	170
drop-conf (Drop Conference)	170
help (Show Help Page and Exit)	170
ivr (Connect to IVR/asterisk Console)	170
node-add (Add Node to the WYDE Bridge)	170
node-del (Delete Node from the WYDE Bridge)	171
node-reload (Reload Nodes Configuration)	171
node-set (Set Node Properties)	171
node-show (Show Nodes of WYDE Bridge)	171
peer-add (Add Peer to the WYDE Bridge)	172
peer-del (Delete Peer from the WYDE Bridge)	172
peer-set (Set Peer Properties)	172
peer-show (Show Peers of WYDE Bridge)	172
register-license (Register License)	172
reload-all (Reload All System Configuration)	172
set-email (Change Email Address)	173
set-ip (Change IP Address)	173
settings-edit (Edit WYDE System Settings)	173
settings-reload (Reload System Settings)	173
settings-show (Show WYDE Settings)	173
settings-update (Update WYDE Settings with Default Values)	173
show-conf (Show Conference or Conferences List)	174
sounds-reload (Reload Sound Files)	174
status (Show WYDE Status)	174
subscriber-add (Add Subscriber)	174
subscriber-del (Delete Subscriber)	
subscriber-set (Set Subscriber Properties)	175
subscriber-show (Show Subscribers Table)	

	transfer (Transfer Calls)	175
	version (Show WYDE Version)	175
	watch (Watch WYDE Status)	176
m	f Console Command Reference	177
	activities-show (Enable/Disable Printing Activities on the mf Console)	177
	billing-reload (Reload Billing Configuration)	177
	call-associate (Set Bundle for the Call)	177
	call-custom-name (Set Custom Name for the Call)	177
	call-drop (Drop Call in the Conference)	178
	call-gain-set (Gain Control)	178
	call-hold (Hold Call)	178
	call-move (Move Call to Other Conference)	178
	call-mute (Mute Call)	179
	call-qa-request (Start/Stop Q&A Request for the Call)	179
	call-qa-talk (Enable/Disable Q&A Session for the Call in the Queue)	179
	call-role (Set Role for the Call)	180
	callflow-reload (Reload Call Flows)	180
	cmdcount-show (Display Values of Command Counters)	180
	conf-active (Start/Stop Conference Active Mode for Participants and Listene	ers)
	conf-attributes (Set Call Flow Attributes for the Conference)	
	conf-broadcast-proposal (Broadcasting Proposal)	
	conf-broadcast-reject (Reject Broadcasting)	
	conf-broadcast-start (Start Broadcasting)	
	conf-chat-send (Send Chat Message to the Specified Sessions or Groups)	
	conf-drop (Drop Conference)	
	conf-drop-group (Drop Group)	
	conf-hold-group (Hold Group)	
	conf-jobcode (Set Job Code for the Conference)	
	conf-mute-group (Mute Group)	
	conf-play-file (Manage of Playing File to the Conference)	
	conf-polling (Conference Polling)	
	conf-qa-mode (Manage Q&A Sessions)	
	conf-qa-mute (Mute/Unmute Active Q&A Session)	
	conf-qa-talk (Enable Q&A Session for the First Call in the Queue)	
	conf-recording (Start/Stop Conference Recording)	187
	conf-schedule-extend (Extend Scheduled Conference Duration)	
	conf-schedule-incsize (Resize Scheduled Conference Subscription)	
	conf-secure (Secure Conference)	
	conf-shunt (Make/Drop Shunt between Two Conferences)	
	confcount-show (Display Values of confcount Counters)	
	dc-reload (Reload DC Bridges Configuration)	
	dc-set-log-level (Set DC Logger Level)	189

dc-show-bridges (Show Known DC Bridges)	190
dc-show-links (Show DC Links)	190
dialout (Do Dialout)	190
dialout-attr (Show Dialout Attributes for Specified Conference)	190
did-reload (Reload DID Entries)	190
errcount-show (Display Values of Error Counters)	190
freenumbers-show (Show Free Number Leases)	191
help (Show Help for Console Commands)	191
node-reload (Reload MF Cluster Nodes List)	191
node-show (Show MF Cluster Nodes List)	191
op-call-move (Move User that Currently Talking with Operator to other	
Conference)	191
op-dialout (Initiate Dialout from Operator's Console)	192
op-listen (Listen Conference)	192
op-message (Control Operator Message)	193
op-message-to-conf (Send Operator's Message to Conference)	193
op-queue (Display Operator Calls Queue)	194
op-reject (Reject Request from Operator's Queue)	194
op-scan (Scan Conferences)	194
op-show (Display Operators)	195
op-talk (Operator Talk to User)	195
opcount-show (Display Operator's Counters)	195
partcount-show (Display Values of partcount Counters)	196
peer-reload (Reload Peers)	196
quit (Quit Console)	196
set-log-level (Set Logger Level)	197
settings-reload (Reload System Settings)	197
show (Show Conferences and Calls)	197
sounds-reload (Reload Sounds)	197
subconf (Manage Sub-conferences)	197
transfer (Transfer Calls)	198
version (Show WYDE.MF Version)	198
asterisk Console Command Reference	199
wyde drop session (Drop Session)	
wyde show conferences (Show Active Conferences)	199
wyde show conference (Show Conference Members)	
wyde show session queue (Show Session Events Queue)	199
wyde show sessions (Show Sessions not Attached to Conference)	
wyde show statistic (Show Statistic)	199
mp Console Command Reference	
drop (Drop Boards and Calls)	200
kill (Stop Specific Timer or All Timers)	200
restart (Restart mp Boards and Logs)	200

show (Show Different mp Statistics)	201
stop (Stop mp Components)	201
timer (Start Timer Running show Command)	201
Appendix A: Support Resources	202
Support Documentation	
Web Support	
Telephone Support	
Email Support	

# Tables List

Table 1: wyde Command Line Utility Available Commands	24
Table 2: mf Console Utility Available Commands	27
Table 3: Show Conferences Columns	54
Table 4: Show Conference Calls Columns	56
Table 5: Q&A Sessions Management Samples Using mf Console Commands	66
Table 6: asterisk Console Utility Available wyde Commands	
Table 7: mp Console Utility Available Commands	94
Table 8: Operator Conference Management Samples Using mf Console Commands	

# Figures List

Figure 1: The WYDE Bridge Architecture	. 17
Figure 2: wyde help Command Output	
Figure 3: wyde help version and wyde version Commands Output Sample	
Figure 4: mf Console, help Command Output	
Figure 5: mf Console help quit and quit Commands Output Sample	
Figure 6: wyde help subscriber-add and wyde subscriber-add Commands Output Sample	
Figure 7: wyde help subscriber-set and wyde subscriber-set Commands Output Sample	
Figure 8: wyde subscriber-show Command Output Sample	
Figure 9: wyde help confuser-add and wyde confuser-add Commands Output Sample	
Figure 10: wyde help confuser-show and wyde confuser-show Commands Output Sample	
Figure 11: wyde help conference-attr-set and wyde conference-attr-set Commands Outpu	
Sample	
Figure 12: wyde conference-attr-show Command Output Sample	.38
Figure 13: wyde help callflow-add and wyde callflow-add Commands Output Sample	
Figure 14: wyde callflow-show Command Output Sample	
Figure 15: wyde help callflow-attr-set and wyde callflow-attr-set Commands Output	
Sample	. 42
Figure 16: wyde callflow-attr-show Command Output Sample	.43
Figure 17: wyde help did-add and wyde did-add Commands Output Sample	
Figure 18: wyde did-show Command Output Sample	
Figure 19: wyde help did-alias-add and wyde did-alias-add Commands Output Sample	
Figure 20: wyde did-alias-show Command Output Sample	
Figure 21: wyde help did-attr-set and wyde did-attr-set Commands Output Sample	
Figure 22: wyde did-attr-show Command Output Sample	
Figure 23: wyde help caller-gain-set and wyde caller-gain-set Commands Output Sample	
Figure 24: wyde caller-gain-show Command Output Sample	
Figure 25: wyde help show-conf and wyde show-conf Commands Output Sample	
Figure 26: mf Console help show and show Commands Output Sample	
Figure 27: wyde show-conf Commands Output Sample for the Specific Conferences	. 55
Figure 28: mf Console show Commands Output Sample for the Specific Conferences	.56
Figure 29: wyde show-conf Command Output Sample for the Calls that have not placed t	O
Conferences	.57
Figure 30: mf Console show Command Output Sample for the Calls that have not placed	to
Conferences	
Figure 31: mf Console call-associate Commands Output Sample	
Figure 32: mf Console conf-play-file Commands Output Sample	
Figure 33: mf Console dialout-attr and dialout Commands Output Sample	
Figure 34: mf Console call-move Command Output Sample	
Figure 35: mf Console subconf Command Output Sample	
Figure 36: mf Console conf-chat-send Command Output Sample	
Figure 37: mf Console conf-broadcast-proposal, conf-broadcast-start, and conf-broadca	
reject Commands Output Sample	
Figure 38: Starting asterisk Console	
Figure 39: asterisk Console, help wyde Command Output	

Figure 40: asterisk Console quit Command Output Sample	90
Figure 41: asterisk Console wyde show conferences and its help Commands Output Sam	ple
	91
Figure 42: asterisk Console wyde show conference and its help Commands Output Samp	
	91
Figure 43: asterisk Console wyde show sessions and its help Commands Output Sample.	92
Figure 44: asterisk Console wyde show session queue and its help Commands Output	
Sample	93
Figure 45: asterisk Console wyde show statistic and its help Commands Output Sample.	
Figure 46: Starting mp console, Available Commands	
Figure 47: mp Console show Commands Output Sample	
Figure 48: mp Console timer and kill Commands Output Sample	
Figure 49: <i>mp</i> Console <i>drop</i> Commands Output Sample	
Figure 50: mp Console restart Commands Output Sample	
Figure 51: mf Console op-show and op-queue Commands Output Sample	
Figure 52: mf Console freenumbers-show Command Output Sample	
Figure 53: <i>mf</i> Console <i>activities-show</i> Command Output Sample	
Figure 54: wyde ast-status Command Output Sample	
Figure 55: wyde status Command Output Sample	
Figure 56: wyde help settings-edit and wyde settings-edit Commands Output Sample	
Figure 57: wyde help settings-show and wyde settings-show Commands Output Sample.	
Figure 58: wyde help peer-add and wyde peer-add Commands Output Sample	
Figure 59: wyde peer-show Command Output Sample	
Figure 60: wyde help config-save and wyde config-save Commands Output Sample	
Figure 61: wyde node-show and mf Console node-show Commands Output Sample	
Figure 62: <i>mf</i> Console <i>show</i> Conference – Different Nodes Calls Command Output Sam	
	129
Figure 63: wyde bridge-show Command Output Sample on WYDE5 Bridge	
Figure 64: wyde bridge-show Command Output Sample on WYDE31 Bridge	
Figure 65: Show Distributed Conference on WYDE5 Bridge	
Figure 66: Show Distributed Conference on WYDE31 Bridge	135
Figure 67: mf Console dc-show-bridges Command Output Sample on WYDE5 Bridge	136
Figure 68: mf Console dc-show-bridges Command Output Sample on WYDE31 Bridge.	136
Figure 69: <i>mf</i> Console <i>dc-show-links</i> Command Output Sample on <i>WYDE5</i> Bridge	137
Figure 70: mf Console dc-show-links Command Output Sample on WYDE31 Bridge	137
Figure 71: Audio File Conversion using <i>transcoder.x</i> Utility Sample	140
Figure 72: Audio File Conversion using <i>convert.sh</i> Utility Sample	140
Figure 73: Register New/Updated Licenses on the Bridge	141
Figure 74: wyde help auth-adapter-add and wyde auth-adapter-add Commands Output	
Sample	143
Figure 75: wyde auth-adapter-show Command Output Sample	144
Figure 76: wyde help auth-method-add and wyde auth-method-add Commands Output	
Sample	145
Figure 77: wyde help auth-method-set and wyde auth-method-set Commands Output	
Sample	
Figure 78: wyde auth-method-show Command Output Sample	147

Figure 79: wyde help billing-adapter-add and wyde billing-adapter-add Comn	nands Output
Sample	148
Figure 80: wyde help billing-adapter-set and wyde billing-adapter-set Comma	nds Output
Sample	150
Figure 81: wyde billing-adapter-show Command Output Sample	150
Figure 82: wyde help billing-rule-add and wyde billing-rule-add Commands O	utput Sample
	151
Figure 83: wyde help billing-rule-set and wyde billing-rule-set Commands Out	put Sample
	153
Figure 84: wyde billing-rule-show Command Output Sample	153
Figure 85: Drop and Initialize the Billing Database	155
Figure 86: Drop and Initialize the Main Database	156
Figure 87: Changing SIP Bind Address and Port	158
Figure 88: Changing IVR Bind Address and Port	159

# **Chapter 1: Introduction**

This is the Administration guide for the WYDE conferencing bridges (like SB-HD100, SB-HD1000, and SB-HD10000). Within this guide you will learn how to perform the basic day to day administration tasks for these units using command line interface.

### **Assumed Skills**

This administration guide assumes you have a working knowledge of the following technologies and skills:

- PC usage
- System administration
- Linux/CentOS basics
- VOIP basics
- TCP/IP networking
- Web Administration Interface User Guide (recommended)

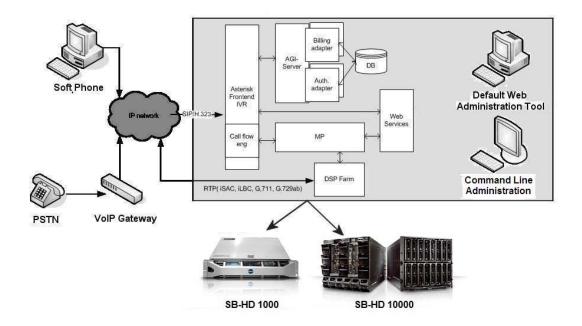
### Architecture Overview

The WYDE architecture is made up of both hardware as well as software services (as shown in Figure 1) that work together to provide the best carrier-class, wideband conferencing available.

WYDE services is not only turnkey software solution, it is the component that can be easily integrated into other products. The WYDE Bridge can be controlled either using web services or using real-time interface. Web services send requests to the bridge and receive information about status of the bridge. The real time interface makes call to the bridge using special client, perform SIP call to send and receive commands and exchange information about the conferences.

## CLIENTS

### WYDE SERVICES



## HARDWARE PLATFORM

Figure 1: The WYDE Bridge Architecture

#### **Administration Interfaces**

There are two basic interfaces to the WYDE conference bridges that allow an administrator to make changes and administrate the operation of the system. These interfaces include:

- The WYDE Command Line Administration Interface
- The WYDE Web Administration Interface

Each interface uses different methods to administrate the various functions of the WYDE conference bridges. It should be noted that there are several places where you can only perform a function in one of the interfaces and there is no corresponding function in the other interface.

This Guide describes WYDE Command Line Administration Interface only. If you need WYDE Web administration interface documentation, please see "Web Administration Interface – User Guide".

#### Hardware

There are three primary WYDE platforms – the SB-HD100, the SB-HD1000 and the SB-HD10000. They are designed to be attached to your network in appliance-fashion. They are, for all intents and purposes, servers that can be installed and managed as such.

For more information on the WYDE hardware and its specifications – please visit www.wydevoice.com.

#### **Voice Clients**

There are three possible clients for WYDE bridge: soft phones, IP phones, and regular (PSTN) telephones. The most famous soft phones are Counterpath X-Lite, Flaphone, SJ Labs SJphone; also you can use Skype as your soft phone. The most famous manufacturers of IP phones are Audio Codes (with Audio Codes 320HD), Polycom HD Voice (with Polycom 550HD), and Snom (with Snom 320). In case of the regular telephone – there is a need to have a separate unit – a voice over IP gateway in order to use the phones, for example Cisco AS 5400 or similar.

## **Integration Adapters**

WYDE can be integrated into an enterprise infrastructure through the set of adapters. There are three points of integration:

- **CDR storage** A storage location for the individual call records. This can be SQL database or something else.
- **Authentication service** This allows the WYDE software to integrate into the enterprise authentication systems. This could be a SQL database, RADIUS, LDAP, or other.
- Call/Conference management This is the ability to manage conference calls, exposed through the Web API for integration with enterprise web sites.

# Definitions, Acronyms and Abbreviations

In order to discuss the Command Line Administration Interface effectively, we need to have a common set of terminology. For this purpose, we should definite the dictionary for the terms you will see throughout this guide:

- **VoIP Voice over Internet Protocol**, a term that refers to the capture/playback of audio streams and their transmission over IP based networks.
- End Point (EP) A generic term used to denote the application running on end-user machines in a VoIP.
- **Public Switched Telephone Network (PSTN)** the traditional phone system.
- **Bridge** A server that hosts voice conferences. Participants can use PSTN or VoIP connections to connect to the bridge. It is responsible for mixing the signals and sending the result back to the participants.
- **Gateway** A gateway server between PSTN and VoIP, i.e. a server that terminates end point connections and routes VoIP data between an end point and the bridge.
- **Node** A computer with the *asterisk* service installed and running. The *asterisk* is being installed in *Frontend* components installation. If you are performing cluster

- installation you can have multiple nodes, i.e. multiple *asterisk* computers in your WYDE bridge environment.
- Conference User A user in a conference. Each connection to the conference bridge is associated with exactly one conference user. An end point can be associated with any number of conference users. A conference user may or may not be associated with an end point. The conference user can have one of the roles: host, participant or listener.
- Conference An audio meeting hosted on a bridge and consisting of PSTN and/or VoIP participants. A data structure is used to describe ongoing conference on the bridge. Objects of this type are only created by server. User may fetch these objects by calling appropriate function. When conference is over the conference object is deleted by the server.
- Conference Number A unique external conference number. Conference number is the property of conference account. If the conference accounts have the same conference number all these accounts determine one single conference. For instance the user can create one conference account record that determine host role, another conference account record that determine participant role, and another conference account record that determine listener role all these records should have the same conference number to determine one unique conference.
- Conference ID A unique conference ID that represents the instance of a conference. When any conference is being started it receives unique conference ID, and all calls to this conference have the same conference ID; if this conference has been completed and another conference is being started that conference will receive another conference ID. Conference ID is normally not exposed to users, unless on the reports.
- Session A data structure represents a single ongoing call on the server. User can not directly create this object. When the call is over server automatically deletes this object. Normally this data structure is used to get information about call attributes like calling/called number etc., or do something with the call, for instance mute, hang, hold etc.
- **Session ID** The unique identifier generated by the bridge for each session (connection, VoIP as well as PSTN) established between a conference user and the bridge. The session id is unique within a given conference.
- Audio Key A key sequence that is used to group different calls from the same conference in a bundle to manage these calls using real-time or another external interface. Audio key is short identifier generated externally and provided to the bridge at the time of joining a conference. Audio key is being generated by real-time application, for instance Moderator-Console, the user can enter the same audio key on his DTMF keypad, usually as #audio key#, these calls (the call from real-time application and the user call to the conference) are being grouped together and the real-time application can manage this user call (the call with the same audio key), for instance mute the call, etc.
- **Distributed Conference** A conference that is taken place on the different bridges simultaneously. That means that the calls are being made to the different bridges, but these calls are participating in the same conference.
- **Subscriber** A real person, he has a name, phone number, e-mail address, etc. The subscriber can have conference accounts, he does not have access codes, but access codes are properties of conference accounts that have subscribers. Note that non-admin

- (non-operator) subscribers can see only "own" information, i.e. his information and information that belongs to subscribers created by him, he can see only their calls, conferences, the reports will show only their data, etc.
- **PIN** The login ID for the subscriber (must be unique). It can be used either as login in Web Administration Interface (in this case it can be either number or alpha-numeric) or as login for some call flows (in this case must be numeric) for participants authorization.
- Conference Account The element of subscriber conferences configuration. Conference accounts always belong to subscriber. It is being used to define a person in a conference with a particular role (e.g. host, participant, listener, etc.), the DNIS number that should be used to call to the conference, and the access code that should be entered by the user that called to the conference DNIS to determine his role. A subscriber could be a host user in one conference and a listener in another. Conference accounts with the same conference number represent single conference setup.
- Call Flow A unique conference service setup, the logic that is used to process the conference calls. This is the process a call goes through from call setup to, to processing, to call tear down. It includes the logic, DTMF key-presses used, functions, and the recorded prompts. There are two basic call flow categories: call flows without authentication and call flows with authentication.
- Attribute In terms of WYDE web services API, a data structure is used to carry attributes for call flow, DNIS and conference account (user). The attributes skeleton is defined by call flow; other attributes can only override some of them, so for instance when a user called in to the conference DNIS it gets attributes exposed by the call flow, but some of these attributes can be already altered by the DNIS. Each attribute has name, type, value, and role.
- **DNIS** A unique set of numbers that is outpulsed by a phone carrier that indicates the intended destination for a particular call. It can be any length digits (although usually 10 digits). DNIS is the property of the conference account, but different DNIS numbers can be used to connect to the same conference.
- Access Code A numeric code unique for DNIS that allows a host or participant or
  listener access to a conference call. When users call to DNIS number they being asked
  to enter their access code. The access code determines the conference and the user role
  in the conference. Different access codes can determine the same conference, for
  instance one access code can determine the connected user has host role, another access
  code can determine that connected user has participant role, and another access code
  can determine that connected user has listener role.
- Host A user in the conference call that can make changes to the system while the
  conference call is in progress. Like change the security setting, change who can talk or
  answer, etc. Sometimes the host user is called moderator. This user role is defined in
  conference account. This is the most privileged role in a conference. By default,
  connections in this role can send and receive RTP data (i.e. the corresponding
  participant is allowed to speak and listen). They also are allowed to execute control
  actions on all connections and roles.
- **Participant** A person in the conference who can actively participate in a call by both talking and listening. This user role is defined in conference account. Connections in this role must be allowed to send and receive RTP data by default. They can execute

- mute and un-mute commands on their own connections (associated with the same audio key); but not on other connections. They are allowed to drop connections within the same bundle (except where the audio key = 0).
- **Listener** A person in the conference who can hear the conference call, but cannot speak. Their audio path is one way only (receive). This user role is defined in conference account. Connections in this role must not have the privilege to speak. They are allowed to send RTP packets to provide feedback for bandwidth adaptively on the stream sent by the bridge. They are allowed to drop connections that are within the same bundle (except where the audio key = 0). Note: users in listener role can be unmuted to enable them to talk; however, the listener group as a whole will never be unmuted.

# **Chapter 2: Command Line Interface**

The WYDE Command Line Administration Interface allows an administrator to make changes and administrate the operation of the system of the WYDE conference bridges. It consists of the following utilities:

- wyde command;
- *mf* console;
- mp console;
- asterisk (IVR) console.

The WYDE Command Line Administration Interface provides direct access to the embedded Linux subsystems of the WYDE conferencing appliances. The command line interface is the powerful tool to administer the bridge – using it you have more administration options at your disposal. In addition, you have direct access to the configuration of the various services that make up the WYDE architecture.

The WYDE Command Line Administration Interface is included with the WYDE conference bridges to allow you administrating of the WYDE bridges. Using *wyde* command you can manage users/subscribers, change call flow/DNIS information for various types of calls, and configure the WYDE bridge. Using *mf* console you can manage conferences and individual calls on-the-fly, i.e. control the conferences and calls that currently are in progress on the bridge.

As subscriber management you can create/view/delete them – manage their PIN, passwords, define subscriber properties, such as first and last names, password, email. In addition you can create/view/delete subscribers' conference accounts – define their conferences, used DNISes and access codes.

As DNIS management you can create/modify/delete DNIS associations of the actual inbound DNIS numbers and call flows that are used to service these numbers. In addition the command line interface allows you managing of call flows and their attributes.

As conference management you can manage the conferences and calls that currently are in progress, you can view started conferences and calls, as well as set and change some of their modes — mute the conferences and participants, placing the call on hold, making the call secure, record the call, etc.

In addition you can use command line administration interface to change system preferences, edit WYDE bridge parameters, perform configuration of the bridge, and monitor WYDE bridge activity.



It is possible to render your system inoperable if you are not familiar with the administration of Linux systems. Changing any of the bridge preferences, unless instructed to do so by WYDE technical support, also can render your system inoperable.

# **Prerequisites**

You may use the WYDE command line administration interface to administer various components of the WYDE system. Before you do, however, you must have the following prerequisites met:

- A computer with a network connection to the TCP/IP subnet where the WYDE appliance is connected;
- SSH secure client to access to that computer for instance PuTTY for Windows or SSH for Linux.

# Accessing the Command Line Administration Interface

To access command line interface you should connect to Linux (CentOS) computer with the WYDE conferencing bridge software installed. You should use SSH secure client and connect to the proper IP address where the WYDE services are running.

# Using the wyde Command Line Utility

The *wyde* command line utility is the tool that can be used to administrate the WYDE services. With it you can view and make changes to the configuration of the various WYDE services, manage subscribers and their conferences, DNISes, call flows, etc. From the command prompt, you can use the utility like so

wyde command [arguments]
where

- command the specific wyde command that describes what kind of task you would like to accomplish;
- arguments optional command arguments, i.e. options of the command that you are using.

To see the list of all available *wyde* command line utility commands you should use the command:

wyde help

You will see the screen similar to shown on Figure 2. All available *wyde* commands are listed in Table 1, the command reference is given in Chapter 3: Command Reference, Section: wyde Command Reference. These commands will be described in more detail later in this Guide.

Figure 2: wyde help Command Output

#### Table 1: wyde Command Line Utility Available Commands

**Commands Description** 

ast-status Show WYDE asterisk status

auth-adapter-add Add auth adapter auth-adapter-del Delete auth adapter auth-adapter-show Show auth adapters auth-method-add Add auth method auth-method-del Delete auth method auth-method-set Setup auth method auth-method-show Show auth methods auth-reload Reload auth configuration billing-adapter-add Add billing adapter billing-adapter-del Delete billing adapter

billing-adapter-set Set billing adapter properties

billing-adapter-show Show billing adapters

billing-reload Reload billing configuration

billing-rule-add Add billing rule
billing-rule-del Delete billing rule
billing-rule-set Set billing rule
billing-rule-show Show billing rules
bridge-add Add WYDE bridge
bridge-del Delete WYDE bridge

bridge-list-load Load list of bridges from file bridge-list-save Save list of bridges to file

bridge-reload Reload DC bridges configuration bridge-set Set WYDE bridge properties

bridge-show Show WYDE bridges caller-gain-set Set gain options caller-gain-show Show gain options callflow-add Add callflow

callflow-attr-set Set callflow attribute Show callflow attributes

callflow-attr-update-db Update callflow attributes in a database

callflow-del Delete callflow
callflow-reload Reload all callflows
callflow-show Show callflows table

conference-attr-del Remove conference attribute redefinition

conference-attr-set Set conference attribute
conference-attr-show Show conference attributes
config-restore Restore WYDE configuration
config-save Save WYDE configuration
confuser-add Add conference user

confuser-add Add conference user
confuser-del Delete conference user
confuser-show Show conference users table

db Connect to WYDE database

db-bil Connect to WYDE billing database

db-drop Drop of WYDE database

**Commands Description** 

db-init Initialize of WYDE database db-patch Apply last patches for database

did-addAdd DID numberdid-alias-addAdd DID number aliasdid-alias-delDelete DID number aliasdid-alias-showShow DID number aliases

did-attr-del Remove DNIS attribute redefinition

did-attr-set
did-attr-show
did-del
did-reload
did-show
Set DNIS attribute
Show DNIS's attributes
Delete DID number
Reload all DIDs caches
Show DIDs table

drop-call Drop call

drop-conf Drop conference

help Show help page and exit ivr Connect to IVR console

node-add Add node to the WYDE bridge node-del Delete node from the WYDE bridge

node-reload Reload nodes configuration

node-set Set node properties

node-show Show nodes of WYDE bridge

peer-add Add peer peer-del Delete peer

peer-set Set node properties
peer-show Show WYDE peers
register-license Register License

reload-all Reload system configuration

set-email Change email address set-ip Change ip address

settings-edit Edit WYDE system settings settings-reload Reload system settings settings-show Show WYDE settings settings-update Update WYDE settings

show-conf Show conference or conferences list

sounds-reload Reload sound files status Show WYDE status subscriber-add Add subscriber subscriber Delete subscriber

subscriber-set Set subscriber properties subscriber-show Show subscribers table

transfer Transfer calls

version Show WYDE version watch Watch WYDE status

If you need detail help about any of these commands you should use the command: wyde help command

#### where

• command – the specific wyde command on which you would like to get help.

For instance, if you would like to get help on *version* command you should use the following command:

wyde help version

If you would like to show current WYDE version you should use the command: wyde version

Sample of both these commands output is shown on Figure 3.

Figure 3: wyde help version and wyde version Commands Output Sample

# Using the mf Console

The *mf* (Multi Frontend Dispatcher) console is the tool that can be used to administrate the conferences and calls that currently are on the WYDE bridge. Once a conference has begun, you can manage it using *mf* console, you can view started conferences and calls, as well as set and change some of their modes – mute the conferences and participants, placing the call on hold, making the call secure, record the call, etc. To enter into this console you should just run from the command prompt the following command:

mf

The current version of WYDE services will be prompted to you one you login into the console.

To implement any *mf* console command you should type *command* [arguments]

once you have entered into console, where

- command the specific mf console command that describes what kind of task you would like to accomplish;
- arguments optional command arguments, i.e. options of the command that you are using.

To see the list of all available *mf* console commands in the console you should type the command:

help

You will see the screen similar to shown on Figure 4. All available *mf* console commands are listed in Table 2, the command reference is given in Chapter 3: Command Reference, Section: mf Console Command Reference. These commands will be described in more detail later in this Guide.

```
Proot@ZILBER:~
                                                                            - | D | X |
root@192.168.1.5's password:
Last login: Thu Mar 25 16:06:16 2010 from 82.207.96.33
[root@ZILBER ~] # mf
WYDE.MF version 2.1.51
WYDE.MF>help
WYDE.MF>
call-associate
                  - Set bundle for the call.
call-custom-name
                    - Set custom name for the call.
call-drop
                    - Drop call or all calls in the conference.
call-hold
                    - Hold call.
```

Figure 4: mf Console, help Command Output

conf-schedule-incsize

# Table 2: mf Console Utility Available Commands

Table 2: mj Console Utility Available Commands	
Description	
Enable/disable printing activities on this console	
Reload billing configuration	
Set bundle for the call	
Set custom name for the call	
Drop call in the conference	
Gain control	
Hold call	
Move call to other conference	
Mute call	
Start/stop Q&A request for the call	
Enable/disable Q&A session for the call in the queue	
Set role for the call	
Reload callflows	
Display values of command counters	
Start/stop active mode	
Set attributes for the conference	
Broadcasting proposal	
Reject broadcasting	
Start broadcasting	
Send chat message to the specified sessions or groups	
Drop conference	
Drop group	
Hold group	
Set JobCode for the conference	
Mute group	
Manage of playing file to the conference	
Conference polling	
Manage Q&A sessions	
Mute/Unmute active Q&A session	
Enable Q&A session for the first call in the queue	
Start/stop conference recording	
Extend scheduled conference duration	

Resize scheduled conference subscription

**Commands Description** conf-secure Secure conference

conf-shunt Make/drop shunt between two conferences confcount-show Display values of confcount counters Reload DC bridges configuration dc-reload

Set DC logs level dc-set-log-level

dc-show-bridges Show known DC bridges

Show DC links dc-show-links dialout Do dialout

dialout-attr Show dialout attributes for specified conference

did-reload Reload DID entries

errcount-show Display values of error counters

Show free number leases freenumbers-show

Show help for console commands help Reload MF cluster nodes list. node-reload node-show Show MF cluster nodes list

Move user that currently talking with operator to other op-call-move

conference

op-dialout Initiate dialout from operator's console

op-listen Listen conference

op-message Send operator's message to queue or to all op-message-to-conf Send operator's message to conference

Display operator calls queue op-queue

op-reject Reject request from operator's queue

op-scan Scan conferences op-show Display operators Operator talk to user op-talk

opcount-show Display operator's counters

partcount-show Display values of partcount counters

peer-reload Reload peers quit Quit console set-log-level Set logger level

settings-reload Reload system settings show Show conferences and calls

sounds-reload Reload sounds

subconf Manage sub-conferences

transfer Transfer calls

version Show WYDEMF version

If you need detail help about any of these commands you should use the command: help command

where

command – the specific mf command on which you would like to get help.

For instance, if you would like to get help on *quit* command you should use the following command:

help quit

If you would like to exit from mf console you should use the command: quit

Sample of both these commands output is shown on Figure 5.

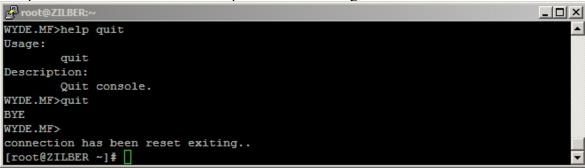


Figure 5: mf Console help quit and quit Commands Output Sample

# Subscriber User Management

One of the administration tasks that you will be called upon frequently to do is to add unique subscribers to the system and allow them to use the system. For terminology's sake we should clarify one main terminology item. A subscriber is a real person – he has a name, phone number, e-mail address, etc. A user is a person in a conference with a particular role (e.g. host, participant, listener, etc). So a subscriber could be a host user in one conference and a listener in another and a subscriber can have different roles in the same the conferences.

#### Create a Subscriber

Whenever a new person needs access to the system, you must create a new subscriber for them so that they can log in to the system. To add subscribers to the system using the command line, you would use the *wyde* command line utility with the *subscriber-add* option. The syntax is as follows:

```
wyde subscriber-add <arguments>
```

Each of the arguments is followed by a space and a value. In *subscriber-add* you can specify the following arguments:

- login <value> PIN for new subscriber, essentially the login ID for the subscriber. It is usually a number so this PIN can be used as a login id by telephone for additional conference features to identify the subscriber, if this is determined by call flow. This login also can be used as login into this Web Administration Interface. If the login is planning to be used as web login only, the PIN can be alpha-numeric. If you don't want to come up with your own numbering scheme for logins, you can omit this argument to generate a random unique ID for the new subscriber.
- name <value> The subscriber's first name.
- lastname <value> The subscriber's last name.
- parent <value> Parent login name, i.e. parent subscriber PIN. A parent is the user account that created this account.
- password <value> This is a password for the subscriber, for instance it can be used to login into Web Administration Interface or Operator Console.
- zip <value> Subscriber's ZIP code.
- email <value> The email address of the subscriber.
- sip phone <value> VoIP phone number of the subscriber.

The arguments can be transferred to this command in any order.

For example you can add new subscriber using the following command (new subscriber attributes are shown in *italic*):

```
wyde subscriber-add login nrobert password rob123 parent admin name Robert lastname Noble email rnoble@rbc.com
```

If the command is successful, the system will not return any errors or messages, in typical Linux fashion, it will just drop you back to the command prompt (#). The sample of the *subscriber-add* command output and the help on this command is shown on Figure 6.

Figure 6: wyde help subscriber-add and wyde subscriber-add Commands Output Sample

# View/Modify a Subscriber

Sometimes you may need to update existing subscribers' information, for example change their password, email address, phone number, etc. To modify existing subscriber on the bridge using the command line, you should use the *wyde* command line utility with the *subscriber-set* option. The syntax is as follows:

```
wyde subscriber-set <arguments>
```

Each of the arguments is followed by a space and a value. In *subscriber-set* you can specify the following arguments:

- login <value> PIN of the existing subscriber, essentially the login ID for the existed subscriber. If the specified subscriber does not exist you receive the error message: "subscriber with same login not exist!".
- name <value> The subscriber's first name that needs to be updated.
- lastname <value> The subscriber's last name that needs to be updated.
- password <value> This is a password for the subscriber that needs to be updated.
- zip <value> Subscriber's ZIP code to update.
- email <value> The email address of the subscriber to update.
- sip phone <value> VoIP phone number of the subscriber to update.

The argument login is required. The arguments can be transferred to this command in any order.

For example you can update VoIP phone number of previously create subscriber *nrobert* and set it to 8345 using the following command (updated subscriber attributes are shown in *italic*):

```
wyde subscriber-set login nrobert sip phone 8345
```

If the command is successful, the system will not return any errors or messages, in typical Linux fashion, it will just drop you back to the command prompt (#). The sample of the *subscriber-set* command output and the help on this command is shown on Figure 7. Note that the WYDE bridge should be configured to perform dialout to the subscribers VoIP phone number; for detail information about how to call to the subscriber's VoIP number and how to configure dialout settings on the bridge see section: Dialout Settings Configuration in this Guide.

```
root@localhost:
                                                                                         [root@localhost ~] # wyde help subscriber-set
Syntax:
    wyde subscriber-set arguments
Arguments:
        * login <value>
name <value>
                                     - Login name.
           name <value>
lastname <value>
password <value>
zip <value>
email <value>
sip_phone <value>
                                      - Subscriber name.
                                     - Subscriber last name.
                                     - Subscriber password.
                                     - Subscriber zip code.
                                     - Subscribe EMail address.
                                     - Subscribe VoIP phone number.
Required arguments marked as '*'.
[root@localhost ~]# wyde subscriber-set login nrobert sip phone 8345
[root@localhost ~]#
```

Figure 7: wyde help subscriber-set and wyde subscriber-set Commands Output Sample

To show a list of the subscribers in the system using the command line, you should use the *wyde* command line utility with the *subscriber-show* option. The syntax is as follows: wyde subscriber-show

This command will output a list of the all existed subscribers on the system, similar to shown on Figure 8. As you can see, the *wyde subscriber-show* command shows the subscribers that have been created in the system as well as their basic properties.

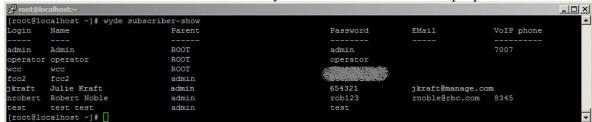


Figure 8: wyde subscriber-show Command Output Sample

#### **Delete a Subscriber**

From time to time, you will need to delete a subscriber.

To delete a subscriber using the *wyde* command line utility you should use *subscriber-del* option. The syntax is as follows:

```
wyde subscriber-del login <login>
where
```

• <login> – login name, i.e. PIN of the subscriber you wish to delete.

For example to delete subscriber *nrobert* (created in previous sample) you should run the command:

```
wyde subscriber-del login nrobert
```

If deletion is successful, you will be returned to the command line with no additional prompts.

#### **Create a Conference Account**

A conference account (conference user) is essentially a set of configuration settings for a particular subscriber for a particular DNIS/call flow. Within a conference account configuration, each given a unique ID#, you specify how a particular subscriber will interact – whether they will be host, participant, or listener, which DNIS/call flow will be used, and what the access code will be used for that conference account.

To add conference accounts (users) to a conference using the command line interface you should use the *wyde* command line utility with the *confuser-add* option. The syntax is as follows:

```
wyde confuser-add <arguments>
```

Each of the arguments is followed by a space and a value (if present). In *confuser-add* you can specify the following arguments:

- accesscode <value> Access code; the value can be any number. If this argument omitted unique access code will be generated.
- did <value> DID (DNIS) number.
- conference <value> Conference number; the value can be any number. If this argument omitted unique access code will be generated.
- conf-accesscode Set the conference number the same as access code. If the conference number argument was transferred together with this argument, the conference number will be ignored, and the conference number will be set equal to the access code.
- role <value> Role in the conference: *Host* or *Participant* or *Listener*. Note: this argument is case-sensitive.
- subscriber <value> Subscriber login name, i.e. subscriber PIN.

Arguments did, role, and subscriber are required. The arguments can be transferred to this command in any order.

For example if you would like to create the conference account for subscriber *nrobert* with DID (DNIS) number (866) 508-0012, conference number 444012, access code 304050 with *Host* role you should run the following command (new conference account attributes are shown in *italic*):

```
wyde confuser-add subscriber nrobert did 8665080012 conference 444012 accesscode 304050 role Host
```

If the command is successful, the system will not return any errors or messages; it will just return you back to the command prompt (#). The sample of the *confuser-add* command output and the help on this command is shown on Figure 9.

```
Proot@ZILBER:
[root@ZILBER ~] # wyde help confuser-add
Syntax:
   wyde confuser-add arguments
Arguments:
         accesscode <value> - Access code.
                               - DID number.
       * did <value>
                              - Conference number.
         conference <value>
         conf-accesscode
                               - Set conference number same sa access code.
         role <value>
                              - Role in conference {Host|Participant}.
       * subscriber <value> - Subscriber login name.
Required arguments marked as '*'.
[root@ZILBER ~]# wyde confuser-add subscriber nrobert did 8665080012 conference
444012 accesscode 304050 role Host
[root@ZILBER ~]#
```

Figure 9: wyde help confuser-add and wyde confuser-add Commands Output Sample

## **Configuration of the Conferences**

Note that it could be different approaches with conferences configuration and setting access codes and roles for the conferences. If you would like to configure the single conference, the conference number must be the same in all conference accounts records that describe this single conference. However the DNIS number and/or subscriber for them could be different; the way that the system determines that it is the single conference is the same conference number.

The first approach to configure the conference is the following: the subscriber creates for himself the conference account records with the same conference number (this determines that it is the single conference configuration) and multiple records with different access codes that are used for different roles. Depending on what access code is used when the user has entered in the conference, the role of this user is different, and the user can be either the host, or the participant or the listener. For example if you need to create for the subscriber *nrobert* the second conference account with participant role and the third conference account with listener role you should execute the following commands: wyde confuser-add subscriber *nrobert* did 8665080012 conference 444012 accesscode 304060 role Participant wyde confuser-add subscriber *nrobert* did 8665080012

conference 444012 accesscode 304070 role *Listener* In this sample for the conference 444012 the host access code is 304050, the participant access code is 304060, the listener access code is 304070; DNIS number in all cases is the same: 8665080012.

The second approach to configure the conference is creating conference accounts for each subscriber that should participate in the conference. The conference number for all these

conference accounts must be the same, and these records will show what access code should use the subscriber and subscriber role in the conference call.

# **View/Modify a Conference Account**

To show a list of the conference accounts (users) in the system using the command line, you should use the *wyde* command line utility with the *confuser-show* option. The syntax is as follows:

```
wyde confuser-show <arguments>
```

Each of the arguments is followed by a space and a value. In *confuser-show* you can specify the following arguments:

- subscriber <value> Subscriber login name, i.e. subscriber PIN; if this argument is transferred this command will return only conference accounts (users) that belong to the specified subscriber.
- accesscode <value> Access code; if this argument is transferred this command will return only conference accounts (users) with the specified access code.

If no arguments were transferred this command lists all conference accounts that exist on the bridge.

# For example the command:

```
wyde confuser-show
```

returns all existed conference accounts, but the command:

```
wyde confuser-show subscriber nrobert
```

returns conference accounts that belong to subscriber *nrobert*, see Figure 10 for details. As you can see, the *wyde confuser-show* command shows the conference accounts that have been created in the system as well as their basic properties: the access codes, DNIS (DID) numbers, call flow names, conference numbers, subscriber's role in that conference, and the first name of the subscriber assigned for each conference account.

```
[root@ZILBER ~] # wyde help confuser-show
Syntax:
   wyde confuser-show arguments
Arguments:
         subscriber <value> - Subscriber login name.
                               - Access code.
         accesscode <value>
Required arguments marked as '*'.
[root@ZILBER ~] # wyde confuser-show subscriber nrobert
AccessCode
               DID
                               CallFlow
                                           Conference
                                                         Role
                                                                     Subscriber
               8665080012 SPECTEL
304070
                                            444012
                                                         Listener
                                                                     Robert
304060
               8665080012
                                SPECTEL
                                            444012
                                                         Participant Robert
304050
                8665080012
                                             444012
                                SPECTEL
                                                         Host
                                                                      Robert
[root@ZILBER ~]#
```

Figure 10: wyde help confuser-show and wyde confuser-show Commands Output Sample

The *wyde* command line utility does not allow changing of conference accounts properties. You can use Web Administration Interface for this purpose (please read "Web Administration Interface – User Guide" if you need assistance in conference accounts modification).

#### **Delete a Conference Account**

If you wish to delete the specific conference account, you can use the *wyde* command line utility with *confuser-del* option. The syntax is as follows:

wyde confuser-del accesscode <access code>
where

• <access code> - the access code of the conference account you wish to delete.

For example to delete the conference account with access code: 304070 (the *listener* role conference account created in previous sample) you should run the command: wyde confuser-del accesscode 304070

If deletion is successful, you will be returned to the command line with no additional prompts.

### Override a Call Flow Attribute Value for a Conference

Call flow attributes can be overridden for a particular subscriber conference definition (conference account). That means that some call flow attributes could be redefined for the specific conference.

To override conference call flow attributes using the command line interface you should use the *wyde* command line utility with the *conference-attr-set* option. The syntax is as follows:

```
wyde conference-attr-set <arguments>
```

Each of the arguments is followed by a space and a value. In *conference-attr-set* you can specify the following arguments:

- number <value> The conference number.
- name <value> This conference call flow attribute name you would like to override.
- value <value> This conference call flow attribute new value.

All these arguments are required. The arguments can be transferred to this command in any order.

The default conference\_exittones (exit tones) call flow attribute value is "on". Let's assume that we need to override its value to "off" for the conference 444012 that we used in previous samples. In this case you should run the following command (the transferred command arguments are shown in *italic*):

```
wyde conference-attr-set number 444012 name conference exittones value off
```

If the command is successful, the system will not return any errors or messages; it will just return you back to the command prompt (#). The sample of the *conference-attr-set* command output and the help on this command is shown on Figure 11.

Figure 11: wyde help conference-attr-set and wyde conference-attr-set Commands Output Sample

If you would like to set new value for the conference call flow attribute that was already overridden, you can just perform another *conference-attr-set* command with new attribute value.

#### **View Call Flow Attributes Values for a Conference**

To show a list of all call flow attributes for the specific conference (including the attributes defined on call flow level and the attributes overridden on DNIS/conference level) using the command line, you should use the *wyde* command line utility with the *conference-attr-show* option. The syntax is as follows:

wyde conference-attr-show number <conference number>
where

• <conference number> - The number of the conference which attributes you would like to view.

For example if you would like to see the list all call flow attributes for the conference 444012 you should use the command:

```
wyde conference-attr-show number 444012
```

This command outputs all call flow attributes for this conference regardless were they defined on call flow level or on DNIS level or on conference level, see Figure 12 for details. As you can see, the *wyde conference-attr-show* command shows the conference call flow attributes names and values; the last column "Override" contains information showing was the attribute overridden either on DNIS (DID) level – in this case this column contains "\*DID" or on conference level – in this case this column contains asterisk "\*".

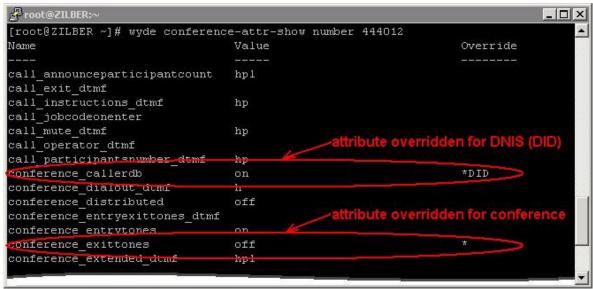


Figure 12: wyde conference-attr-show Command Output Sample

### **Delete a Conference Call Flow Attribute Redefinition**

If you wish to delete the specific conference call flow attribute that was previously overridden, you can use the *wyde* command line utility with *conference-attr-del* option. The syntax is as follows:

```
wyde conference-attr-del <arguments>
```

Each of the arguments is followed by a space and a value. In *conference-attr-show* you can specify the following arguments:

- number <value> The conference number which attribute you wish to delete.
- name <value> This conference call flow attribute name you wish to delete.

Both arguments are required. The arguments can be transferred to this command in any order.

For example to delete the conference 444012 call flow attribute conference\_exittones (overridden in previous sample) you should run the command:

```
wyde conference-attr-del number 444012 name conference exittones
```

If deletion is successful, you will be returned to the command line with no additional prompts.

### Call Flow and DNIS Management

One piece of terminology unique to conference call setup is the idea of a call flow. In other words, the call flow is unique conference call setup, the logic that is used to process the calls; this is the process a call goes through from call setup to, to processing, to call tear down. It includes the logic, DTMF key-presses used, functions, and the recorded prompts. The various system settings and usage parameters are different depending on the call flow used.

The following basic call flows are available in the system:

- CONF
- OPERATOR
- PLAYBACK
- SPECTEL

One of the benefits of the WYDE architecture is the ability to build a complete custom call flow for your organization. So additional custom call flows could be created as customization of the existing call flows. For more information on a custom call flow, please contact your WYDE representative.

A DNIS is the unique set of numbers that is outpulsed by a phone carrier that indicates the intended destination for a particular call. It can be any length digits (although usually less than 10 digits). As DNIS number you can also use the mask of the number, in such mask '\*' – denotes any sequence of digits and '?' – denotes one single digit; in this case for such DNISes the bridge will process all numbers that comply with the DNIS number mask. For the purposes of this guide, a DNIS Association is a combination of a call flow and an actual inbound DNIS (DID) number or number mask.

You can read detail information about call flows and DNISes in "Web Administration Interface – User Guide", Chapter 3: Call Flows and Section: DNIS Management.

Using the command line administration interface you will be able to change parameters of the call flows and DNISes, as it will be described later in this chapter.



Changing any of the call flow attributes, unless instructed to do so by WYDE technical support, can render your system inoperable.

#### Add a Call Flow

Before you register new call flow, you should create the folder for this call flow. This folder should contain the files *callflow.spec* (Asterisk dialplan on AEL language file) and *script.ael* (specification of the scenario file) and optionally subfolder *sounds* for audio (voice) files. The default root folder for call flows is */usr/local/DNCA/callflows/*. You can read "Call Flow Development – Programmer's Guide" for additional information about call flow folder structure and contents.

To add new call flow registration using the command line interface you should use the *wyde* command line utility with the *callflow-add* option. The syntax is as follows: wyde callflow-add *<arguments>* 

Each of the arguments is followed by a space and a value. In *callflow-add* you can specify the following arguments:

- name <value> The name of new call flow that should be registered. This is required argument. This name should be unique, i.e. there should no be any other call flow with the same name on the bridge.
- directory <value> Call flow directory path. If this parameter omitted the current folder will be used as this parameter value.

The arguments can be transferred to this command in any order.

Let's assume that we have created the folder /usr/local/DNCA/callflows/SAMPLE/ for new call flow SAMPLE that contains required files callflow.spec, script.ael and sound subfolder. To add this call flow to the bridge you should use the command:

```
wyde callflow-add name SAMPLE
    directory /usr/local/DNCA/callflows/SAMPLE
```

If the command is successful, the system will not return any errors or messages; it will just return you back to the command prompt (#). The sample of the *callflow-add* command output and the help on this command is shown on Figure 13.

Figure 13: wyde help callflow-add and wyde callflow-add Commands Output Sample

### View a Call Flow

To show a list of all call flows in the system using the command line, you should use the *wyde* command line utility with the *callflow-show* option. The syntax is as follows: wyde callflow-show

This command will output a list of the all existed call flows on the system, similar to shown on Figure 14. As you can see, the *wyde callflow-show* command shows the call flows that have been created in the system as well as their basic properties: call flow name and directory path.

```
[root@ZILBER ~] # wyde callflow-show
                Directory
Name
                                        Description
CONF
                /usr/local/DNCA/callflows/CONF
OPERATOR
                /usr/local/DNCA/callflows/OPERATOR
PLAYBACK
                /usr/local/DNCA/callflows/PLAYBACK
SAMPLE
                /usr/local/DNCA/callflows/SAMPLE
SPECTEL
                /usr/local/DNCA/callflows/SPECTEL
[root@ZILBER ~]#
```

Figure 14: wyde callflow-show Command Output Sample

#### **Delete a Call Flow**

To delete a call flow using the *wyde* command line utility you should use *callflow-del* option. The syntax is as follows:

```
wyde callflow-del name <call flow name>
where
```

• <call flow name> - the name of the call flow you wish to delete.

Note: when you delete the call flow all DNISes associated with this call flow and all this call flow conference accounts also will be deleted.

For example to delete call flow *SAMPLE* (created in previous sample) you should run the command:

```
wyde callflow-del name SAMPLE
```

If deletion is successful, you will be returned to the command line with no additional prompts.

#### Set a Call Flow Attribute Value

To set a call flow attribute value using the command line interface you should use the *wyde* command line utility with the *callflow-attr-set* option. The syntax is as follows:

```
wyde callflow-attr-set <arguments>
```

Each of the arguments is followed by a space and a value. In *callflow-attr-set* you can specify the following arguments:

- callflow <value> The name of the call flow which attribute value you would like to set.
- name <value> This call flow attribute name you would like to update.
- value <value> This call flow attribute new value.

The arguments callflow and name are required; if the argument value is omitted the empty value will be set to this attribute. The arguments can be transferred to this command in any order.

For example if you would like to change <code>call\_instructions\_dtmf</code> (instructions DTMF policy) call flow attribute value to "h", i.e. make the instructions available for hosts only, for the call flow <code>SPECTEL</code>, you should run the following command (the transferred command arguments are shown in <code>italic</code>):

```
wyde callflow-attr-set callflow SPECTEL
  name call_instructions_dtmf value h
```

If the command is successful, the system will not return any errors or messages; it will just return you back to the command prompt (#). The sample of the *callflow-attr-set* command output and the help on this command is shown on Figure 15.

```
Proot@ZILBER:
                                                                          [root@ZILBER ~] # wyde help callflow-attr-set
Syntax:
   wyde callflow-attr-set arguments
Arguments:
        * callflow <value>
                                - Callflow name.
       * name <value>
                                - Attribute name.
         value <value>
                                - Attribute value.
Required arguments marked as '*'.
[root@ZILBER ~] # wyde callflow-attr-set callflow SPECTEL name call instructions
dtmf value h
[root@ZILBER ~]#
```

Figure 15: wyde help callflow-attr-set and wyde callflow-attr-set Commands Output Sample

#### **View Call Flow Attributes Values**

To show a list of all call flow attributes for the specific call flow using the command line, you should use the *wyde* command line utility with the *callflow-attr-show* option. The syntax is as follows:

```
wyde callflow-attr-show callflow <call flow name>
where
```

• <call flow name> - The name of the call flow which attributes you would like to view.

For example if you would like to see the call flow *SPECTEL* attributes you should use the command:

```
wyde callflow-attr-show callflow SPECTEL
```

This command outputs all specific call flow attributes, see Figure 16 for details. As you can see, the *wyde callflow-attr-show* command shows the call flow attributes names, groups, types, values, and allowed values list (if applicable). The group of an attribute can be either "conference" or "dnis"; when the attribute group is "dnis", the attribute can be overridden in DNIS Associations only; when the attribute role is "conference", the attribute can be overridden either in DNIS Associations or in Subscriber Conferences – see "Web Administration Interface – User Guide" for details. The type of an attribute can have one of the following values: "role" (the attribute determines the policy, i.e. the person to whom this option should be available – hosts (h) or participants (p) or listeners (l)), "choice" (only values specified in values list are permitted for this attribute), "string", or "int".

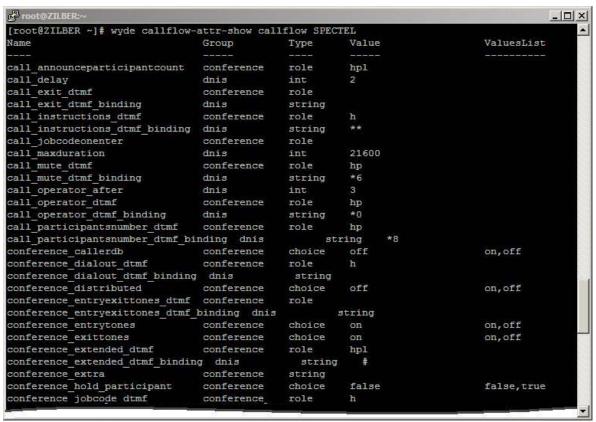


Figure 16: wyde callflow-attr-show Command Output Sample

### **Reload All Call Flows**

If there were changes in the *script.ael* file or sound files were updated in *sounds* subfolder for any of the call flows you should reload <u>all</u> call flows.

To send the signal on the WYDE bridge to the call flow engine to reload the scripts the following command should be executed:

```
wyde callflow-reload
```

If this command is successful, you will be returned to the command line with no additional prompts.

As alternative of this *wyde* command you can use *mf* console command *callflow-reload* for the same purposes. This *wyde* command is basically used in automated scripts and for convenient call flow modification.

# **Update Call Flow Attributes Definition in a Database**

If there were changes in call flow attributes definition and *callflow.spec* file was updated you should update call flow attributes definition in the database.

To send the signal on the WYDE bridge to update the call flow attributes for the specific call flow the following command should be executed:

wyde callflow-attr-update-db callflow <call flow name>
where

• <call flow name> - the name of the call flow which attributes you wish to update.

For example if the definition of call attributes for call flow SAMPLE (described in samples) were updated, i.e. the *callflow.spec* file in the folder /usr/local/DNCA/callflows/SAMPLE/ was changed, you should implement the following command:

wyde callflow-attr-update-db callflow SAMPLE

#### **Create a DNIS Association**

To create new DNIS association of a call flow and an actual inbound DNIS (DID) number using the command line interface you should use the *wyde* command line utility with the *did-add* option. The syntax is as follows:

wyde did-add <arguments>

Each of the arguments is followed by a space and a value. In *did-add* you can specify the following arguments:

- number <value> Dial-in number, i.e. DNIS (DID) number, you want to be associated with this DNIS Association.
- callflow <value> The name of the call flow to be used with this DNIS Association.
- description <value> DNIS (DID) description.

Arguments number and callflow are required. The arguments can be transferred to this command in any order.

For example if you would like to create the DNIS Association of DID (DNIS) number (866) 508-9030 and call flow SPECTEL you should run the following command (new DNIS association properties are shown in *italic*):

```
wyde did-add number 8665089030 callflow SPECTEL
  description "Testing and Samples"
```

Note that to set the description that contains spaces you should use double quotes (").

If the command is successful, the system will not return any errors or messages; it will just return you back to the command prompt (#). The sample of the *did-add* command output and the help on this command is shown on Figure 17.

Figure 17: wyde help did-add and wyde did-add Commands Output Sample

### View/Modify a DNIS Association

To show a list of the DNIS associations in the system using the command line, you should use the *wyde* command line utility with the *did-show* option. The syntax is as follows: wyde did-show

This command will output a list of the all existed DNIS associations on the system, similar to shown on Figure 18. As you can see, the *wyde did-show* command shows the DNIS associations that have been created in the system as well as their basic properties: dial-in number, used call flow, and description of DNIS association.

```
root@ZILBER:

 [root@ZILBER ~] # wyde did-show
Number
                                                                                                                                              CallFlow
                                                                                                                                                                                                                                                                                                  Description
                                                                                                                                               CONF
                                                                                                                                               SPECTEL
16
                                                                                                                                               PLAYBACK
20
                                                                                                                                              OPERATOR
                                                                                                                                                                                                                                                                                                  FLASH SPECTEL
8665080012
                                                                                                                                               SPECTEL
 8665080016
                                                                                                                                               PLAYBACK
                                                                                                                                                                                                                                                                                                  FLASH PLAYBACK
8665089020
                                                                                                                                               SPECTEL
                                                                                                                                                                                                                                                                                                 Scheduling
8665089030
                                                                                                                                                 SPECTEL
                                                                                                                                                                                                                                                                                                   Testing and Samples
[root@ZILBER ~]# [
```

Figure 18: wyde did-show Command Output Sample

The *wyde* command line utility does not allow changing of DNIS associations. You can use Web Administration Interface for this purpose (please read "Web Administration Interface – User Guide" if you need assistance in DNIS associations modification).

### **Create a DNIS Number Alias**

The WYDE conferencing bridge software allows you to create aliases for DNIS (DID) numbers. That means that if you need to have multiple dial-in numbers that should operate the same way and have the same properties, you can create one DNIS association for so called "main" DNIS (DID) number and define other numbers as aliases for that one. In this case the DNIS configuration is being made for this main DNIS number only, and you should not configure additional DNISes for additional DNIS numbers.

To create new DNIS number alias using the command line interface you should use the *wyde* command line utility with the *did-alias-add* option. The syntax is as follows: wyde did-alias-add *<arquments>* 

Each of the arguments is followed by a space and a value. In *did-alias-add* you can specify the following arguments:

- number <value> Main dial-in number, i.e. DNIS (DID) number for which you have created DNIS Association.
- alias <value> The alias of the main dial-in number that you wish to add.
- description <value> The description of DNIS (DID) alias.

Arguments number and alias are required. The arguments can be transferred to this command in any order.

For example if you would like to create the alias number (866) 508-9040 for the DNIS number (866) 508-9030 (that was used in previous samples) you should run the following command (both dial-in numbers are shown in *italic*):

wyde did-alias-add number 8665089030 alias 8665089040 Note that in this sample we do not create the description for the alias.

If the command is successful, the system will not return any errors or messages; it will just return you back to the command prompt (#). The sample of the *did-alias-add* command output and the help on this command is shown on Figure 19.

Figure 19: wyde help did-alias-add and wyde did-alias-add Commands Output Sample

### **View DNIS Number Aliases**

To show a list of DNIS number aliases for the specific DNIS association only using the command line, you should use the *wyde* command line utility with the *did-alias-show* option. The syntax is as follows:

```
wyde did-alias-show [number <DNIS Number>]
where
```

• <DNIS Number> - Main dial-in number, i.e. DNIS (DID) number whose aliases you would like to show.

To show a list of all DNIS number aliases in the system using the command line, you should use the *wyde* command line utility with the same *did-alias-show* option without number *<DNIS Number>* argument. The syntax is as follows: wyde did-alias-show

For example if you would like to see the list of aliases for DNIS number (866) 508-9030 (that was used in previous samples) you should run the following command: wyde did-alias-show number 8665089030

This command outputs all aliases with their descriptions for the specific DNIS number as shown on Figure 20.

Figure 20: wyde did-alias-show Command Output Sample

### **Delete a DNIS Number Alias**

If you wish to delete the specific DNIS number alias, you can use the *wyde* command line utility with *did-alias-del* option. The syntax is as follows:

```
wyde did-alias-del <arguments>
```

Each of the arguments is followed by a space and a value. In *did-alias-del* you can specify the following arguments:

- number <value> Main dial-in number, i.e. DNIS (DID) number which alias you wish to delete.
- alias <value> The alias of the main dial-in number that you wish to delete. Both arguments are required. The arguments can be transferred to this command in any order.

For example to delete the alias (866) 508-9040 of DNIS number association (866) 508-9030 (created in previous sample) you should run the command:

```
wyde did-alias-del number 8665089030 alias 8665089040
```

If deletion is successful, you will be returned to the command line with no additional prompts.

### **Delete a DNIS Association**

If you wish to delete the specific DNIS number association, you can use the *wyde* command line utility with *did-del* option. The syntax is as follows:

```
wyde did-del number <DNIS Number>
where
```

• <DNIS Number> - DNIS (DID) number you wish to delete. This argument is required.

For example to delete the DNIS number association (866) 508-9030 (created in previous sample) you should run the command:

```
wyde did-del number 8665089030
```

If deletion is successful, you will be returned to the command line with no additional prompts.

### Override a Call Flow Attribute Value for a DNIS

Call flow attributes can be overridden for a particular DNIS association. That means that some call flow attributes could be redefined on DNIS level.

To override call flow attributes for a DNIS using the command line interface you should use the *wyde* command line utility with the *did-attr-set* option. The syntax is as follows: wyde did-attr-set <arguments>

Each of the arguments is followed by a space and a value. In *did-attr-set* you can specify the following arguments:

- number <value> DNIS (DID) number.
- name <value> This DNIS call flow attribute name you would like to override.
- value <value> This DNIS call flow attribute new value that you would like to override.

All these arguments are required. The arguments can be transferred to this command in any order.

The default conference\_exittones (exit tones) call flow attribute value for SPECTEL call flow is "on". Let's assume that we need to override its value to "off" for the DNIS (866) 508-9030 that we used in previous samples. In this case you should run the following command (the transferred command arguments are shown in *italic*):

Wyde did-attr-set number 8665089030 pame conference exittones

```
wyde did-attr-set number 8665089030 name conference_exittones
  value off
```

If the command is successful, the system will not return any errors or messages; it will just return you back to the command prompt (#). The sample of the *did-attr-set* command output and the help on this command is shown on Figure 21.

Figure 21: wyde help did-attr-set and wyde did-attr-set Commands Output Sample

If you would like to set new value for the DNIS call flow attribute that was already overridden, you can just perform another *did-attr-set* command with new attribute value.

### **View Call Flow Attributes Values for a DNIS**

To show a list of all call flow attributes for the DNIS association (including the attributes defined on call flow level and the attributes overridden on DNIS level) using the command line, you should use the *wyde* command line utility with the *did-attr-show* option. The syntax is as follows:

wyde did-attr-show number <DNIS Number>
where

• <DNIS Number> - DNIS (DID) number which attributes you would like to view. This argument is required.

For example if you would like to see the list all call flow attributes for the DNIS (866) 508-9030 you should use the command:

```
wyde did-attr-show number 8665089030
```

This command outputs all call flow attributes for this DNIS regardless were they defined on call flow level or on DNIS level, see Figure 22 for details. As you can see, the *wyde did-attr-show* command shows the DNIS call flow attributes names and values; the last column "Override" contains asterisk "\*" if the attribute was overridden for this DNIS (DID).

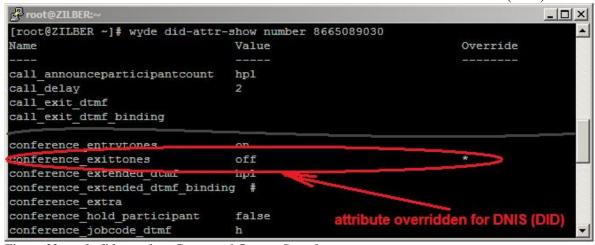


Figure 22: wyde did-attr-show Command Output Sample

#### **Delete a DNIS Call Flow Attribute Redefinition**

If you wish to delete the specific DNIS call flow attribute that was previously overridden, you can use the *wyde* command line utility with *did-attr-del* option. The syntax is as follows:

```
wyde did-attr-del <arguments>
```

Each of the arguments is followed by a space and a value. In *did-attr-show* you can specify the following arguments:

- number <value> DNIS (DID) number which attribute you wish to delete.
- name <value> This DNIS call flow attribute name you wish to delete.

Both arguments are required. The arguments can be transferred to this command in any order.

For example to delete the DNIS (866) 508-9030 call flow attribute conference\_exittones (overridden in previous sample) you should run the command:

```
wyde did-attr-del number 8665089030 name conference_exittones
```

If deletion is successful, you will be returned to the command line with no additional prompts.

### **Reload All DNISes Caches**

*MF* service (Multi Frontend Dispatcher) caches all DNIS information. As the result any DNIS changes (such as changes in call flow attributes of DNISes or changes in DNIS aliases) take effect only in about one minute after the changes have been made. If you would like that these changes take effect immediately you should reload <u>all</u> DNISes caches.

To send the signal on the WYDE bridge *MF* engine to reload all DNISes caches the following command should be executed:

```
wyde did-reload
```

If this command is successful, you will be returned to the command line with no additional prompts.

As alternative of this *wyde* command you can use *mf* console command *did-reload* for the same purposes. This *wyde* command is basically used in automated scripts and for convenient DNIS attributes configuration.

# **View/Modify Callers Gain Options**

WYDE bridge software allows defining specific gain levels for particular DNIS association (called number), caller (calling number) and specific conference. That means that the microphone volume level of the call could be set from the quietest (lowest) sound level to the loudest (highest) sound level or the microphone level could be automatically adjusted by the backend.

To define either received or transmitted audio gain for a caller using the command line interface you should use the *wyde* command line utility with the *caller-gain-set* option. The syntax is as follows:

```
wyde caller-gain-set <arguments>
```

Each of the arguments is followed by a space and a value. In *caller-gain-set* you can specify the following arguments:

- opt\_name <value> The required option name
  {calling number|called number|conf number}:
  - o calling\_number denotes that the caller gain should be set for the calling number (the caller phone number) defined by opt\_value argument;

- o called\_number denotes that the caller gain should be set for the called number (DNIS) defined by opt value argument;
- o conf\_number denotes that the caller gain should be set for the conference number defined by opt value argument.
- opt value <value> The required option value.
- rx\_gain <value> Received audio gain (the sound that goes into the conference); could be from -10 till 10 or 255; it indicates what volume level should be set for the call, 255 denotes that the volume level is being automatically generated by the backend.
- tx\_gain <value> Transmitted audio gain (the sound that goes from the conference); could be from -10 till 10 or 255; it indicates what volume level should be set for the call, 255 denotes that the volume level is being automatically generated by the backend.

The arguments opt\_name and opt\_value are required. For the rx\_gain and tx\_gain arguments -10 is the quietest (lowest) sound level, 10 is the loudest (highest) sound level, 255 denotes that the microphone level is being automatically adjusted by the backend. If rx\_gain and tx\_gain arguments omitted, the gain levels values would be set to default 0. The arguments can be transferred to this command in any order.

For example if you would like

- for the DNIS number (866) 508-0012 to set received audio gain level 4 and transmitted audio gain level 5
- for the conference with number 660660 to set received audio gain level -2 and transmitted audio gain *automatic*

you should run the following commands (the transferred command arguments are shown in *italic*):

```
wyde caller-gain-set opt_name called_number
  opt_value 8665080012 rx_gain 4 tx_gain 5
wyde caller-gain-set opt_name conf_number
  opt_value 660660 rx_gain -2 tx_gain 255
```

If the command is successful, the system will not return any errors or messages; it will just return you back to the command prompt (#). The sample of the *caller-gain-set* command output and the help on this command is shown on Figure 23.

Figure 23: wyde help caller-gain-set and wyde caller-gain-set Commands Output Sample

To show a list of all callers' gains in the system defined either for specific calling number, or specific called number, or conference number using the command line, you should use the *wyde* command line utility with the *caller-gain-show* option. The syntax is as follows: wyde caller-gain-show

This command will output a list of the all callers' gains on the system, similar to shown on Figure 24. As you can see, the *wyde caller-gain-show* command shows the callers gains that have been set in the system as well as their properties: calling number, called number, conference number, received (Rx) audio gain, and transmitted (Tx) audio gain.

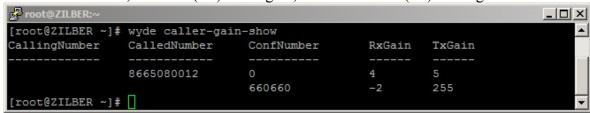


Figure 24: wyde caller-gain-show Command Output Sample

# Conferences and Calls Management

Once a conference has begun, you can manage it using the command line interface. You can manage conferences and calls on-the-fly using the *mf* console. The *wyde* command line utility can be used only for basic tasks, such as view conferences, drop conferences and calls, and it does not provide the possibility to control the conferences and calls. Some of the information can be viewed through *Asterisk* built-in tools, as discussed later.

### **View Conferences and Calls in Progress**

### **View Conferences in Progress**

Any time when there are conferences in progress, you can view them in the command line interface – either via *wyde* command line utility or using *mf* console.

To show a list of all conferences that currently are in process on the bridge using the command line, you should use the *wyde* command line utility with the *show-conf* option: wyde show-conf

You will see the screen similar to shown on Figure 25.

```
root@ZILBER:∼

                                                                      _ | U X
[root@ZILBER ~] # wyde help show-conf
Syntax:
   wyde show-conf arguments
Arguments:
     number <value> - Conference number. Use 'confless' to show cal
ls which hasn't placed to any conference
Required arguments marked as '*'.
[root@ZILBER ~] # wyde show-conf
Number
              Members On IVR On MP
                                          Flags
390008 1 1
667788 3 0
                                0
                                          OB
                                 3
                                          В
4 calls in 2 conferences.
[root@ZILBER ~]#
```

Figure 25: wyde help show-conf and wyde show-conf Commands Output Sample

As alternative you can use *mf* console to show all started conferences. To do so you should run the command *show* from the console:

show

You will see the screen similar to shown on Figure 26.

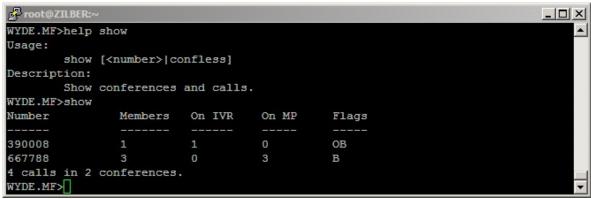


Figure 26: mf Console help show and show Commands Output Sample

As you can see both commands have similar syntax and similar output. This *wyde* command is basically used in automated scripts, *mf* console command can be considered as primary interactive command to show started conferences. All active conferences are listed sorted ascending by conference number. There are several columns of information about each conference. Table 3 details what each column indicates.

**Table 3: Show Conferences Columns** 

Table 5. Show Con	nerences Columns
Column	Description
Number	The unique number assigned to each conference.
	You can use this conference number to view all calls joined to this
	conference
Members	Indicates how many participants are currently joined to this conference
On IVR	Indicates how many conference calls are currently processed by
	the frontend (IVR), for instance these calls could hear music on
	hold, hear any message from some service functions, etc., i.e.
	these calls are not participating in the conference – they could not
	talk and hear the conference
On MP	Indicates how many conference calls are currently processed by
	the Media Processor (MP), i.e. these calls are participating in the
	conversation – they could talk and hear each other
Flags	Indicates the mode of the conference; there could be the following conference modes:
	• O – operator mode;
	• Q – Q&A mode (Q&A mode is switched on for the
	conference);
	<ul> <li>R – the conference is being recorded;</li> </ul>
	• S – secured mode (the conference is secured);
	• T – the conference maintains the real time (RT) protocol;
	Note that the conference could be in multiple modes in the same
	time

Number of calls on IVR plus number of calls on MP makes total number of conference calls shown in *Members* column.

### **View Conference Calls in Progress**

Any time when there are conference calls in progress, you can view them in the command line interface – either via *wyde* command line utility or using *mf* console.

To show a list of all calls that are joined to the specific conference using the command line, you should use the *wyde* command line utility with the *show-conf* option:

```
wyde show-conf number <conference number>
where
```

• <conference number> - The number of the conference which calls you would like to view.

The previous sample shows that there are two conferences on the bridge: 667788 and 390008. If you would like to view the calls joined to these conferences you should run the commands:

```
wyde show-conf number 667788 wyde show-conf number 390008
```

You will see the screen similar to shown on Figure 27.

```
d root@ZILBER:-
                                                                        [root@ZILBER ~] # wyde show-conf number 667788
Conference: number=667788, id=21, flags=B, policy=M(False,False,Strict,False);H(
False, False, False, False)
                                 Location Type Flags
SesId
              Node
                      Role
                                                          AKey
                                           ____
                     Host MP
16777260
                                         PSTN
                                          PSTN
16777261
                     Participant MP
                     Participant MP PSTN
16777262
                                                          0
[root@ZILBER ~]# wyde show-conf number 390008
Conference: number=390008, id=20, flags=OB, policy=M(False,False,False,False);H(
False, False, False, False)
SesId
              Node
                      Role
                                 Location
                                           Type Flags
                                                          AKey
16777259
                      Host
                                  IVR
                                           PSTN
                                                          0
[root@ZILBER ~]# [
```

Figure 27: wyde show-conf Commands Output Sample for the Specific Conferences

As alternative you can use *mf* console to show all calls that are joined to the specific conference. To do so you should run the command *show* from the console: show *<number>* 

#### where

• <number> - The number of the conference which calls you would like to view.

To view the calls for the same conferences 667788 and 390008 using mf console you should run the following commands from the console:

```
show 667788
show 390008
```

You will see the screen similar to shown on Figure 28.

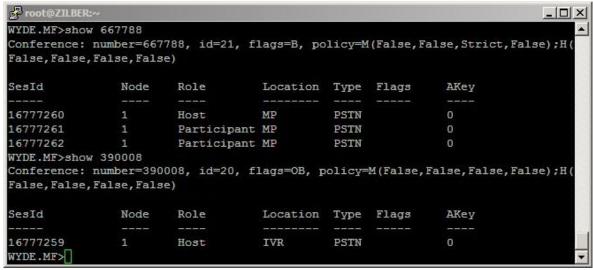


Figure 28: mf Console show Commands Output Sample for the Specific Conferences

As you can see both commands have similar output for conference calls as well. This *wyde* command is basically used in automated scripts, *mf* console command can be considered as primary interactive command to show started conference calls. The information about the conference is shown in the header of the output; it includes number of the conference (*number* field), the conference identifier (id field), the conference flags (flags field), default policy for muting (M) and holding (H) of new participants on connection to the conference (i.e. initial value of mute and hold flags for each role when the call with this role connects to the conference). All active conference calls are listed sorted ascending by session identifier. There are several columns of information about each conference call. Table 4 details what each column indicates.

**Table 4: Show Conference Calls Columns** 

Tuble 1. Show Conference Cums Columns			
Column	Description		
SesId	The session identifier of the call		
Node	The frontend node identifier for the call; you can see the list of all		
	available MF cluster nodes using mf console command node-		
	show (Show MF Cluster Nodes List), see mf Console Command		
	Reference for details		
Role	The subscriber's (call's) role for this conference: Host, Participant or Listener		
Location	Either IVR (the call is being processed by the frontend) or MP		
	(the call is being processed by the Media Processor)		
Type	Either PSTN (phone call) or VoIP (software call that supports real		
	time protocol) or Ctrl (control call)		

# **Column Description**

Flags Indicates the mode and state of the call; there could be the following call modes:

- H the call is placed on hold, possible states: s self, m moderator:
- M the call is muted, possible states: s self, m moderator,
   q Q&A mode is switched on;
- O operator request, possible states: w waiting, t talking;
- Q Q&A request, possible state: w waiting, t talking;
- T real time (RT) mode;

Note that T (real time) mode does not have states, all other modes are being characterized by mode state as well;

Also note that the call could be in multiple modes in the same time, flags (modes) H and M could have multiple states at the same time (for example M (sm) value means that the call has muted itself and the moderator has muted the call as well, M (m) H (m) value means that the call is muted and placed on hold by moderator)

The audio key of the call, 0 - if the audio key is not defined for the call

If you try to show conference calls for the conference that does not started or does not exist the error message "*Error*: Conference not found" will be returned back to you.

### Show Calls that have not placed to Conferences

To show a list of all calls that are not joined to any of the conferences using the command line, you should use the *wyde* command line utility with the *show-conf* option:

wyde show-conf number *confless* 

The keyword *confless* specifies that the command should return the list of calls that have not places to conferences. If you run this command you will see the screen similar to shown on Figure 29.

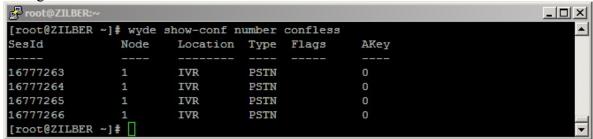


Figure 29: wyde show-conf Command Output Sample for the Calls that have not placed to Conferences

As alternative you can use *mf* console to show all calls that are not joined to the conferences. To do so you should run the command *show* from the console: show *confless* 

The command also uses keyword *confless* to specify that the command should return the list of calls that have not places to conferences. If you run this command you will see the screen similar to shown on Figure 30.

**AKey** 

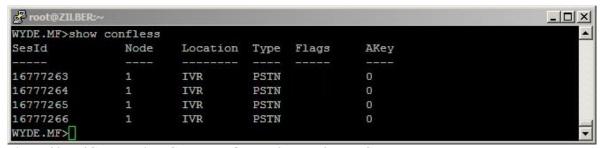


Figure 30: mf Console show Command Output Sample for the Calls that have not placed to Conferences

Once again both commands have similar output for calls that have not placed to conferences. Because these calls have not connected to the conferences this view does not have *Role* column (the call has role only after it is connected to the conference).

### Calls Management using *mf* Console

The basic tool to manage the conference calls is *mf* console; *wyde* command line utility provides only few possibilities to control the calls. This section explains how to manage the calls using *mf* console; the alternative *wyde* command will be noted if applicable.

### **Dropping Call Participants**

If during a call, you wish to cancel a conference call for specific participants, you may use the *call-drop* command of *mf* console to kick someone off the call. To drop someone from a conference call you should run the following *mf* console command:

```
call-drop <conf_number> <ses_id>
where
```

- <conf\_number> The number of the conference which calls you would like to drop.
   You can use *confless* keyword instead of conference number to indicate calls which have not placed to any conference.
- <ses\_id> The session (i.e. call) identifier you wish to drop from the conference. Both arguments are required.

For example if you would like to drop the session 16777261 from the conference 667788 you should run mf console command:

call-drop 667788 16777261

If the command was implemented successfully you will receive the message: "Success"; if the conference not found or not started you will receive the message "Error: Conference not found"; if the session not found you will receive the message "Error: Session not found".

As the alternative to this command to drop call participants you can use the *wyde* command *drop-call*; see wyde Command Reference for details. This *wyde* command is basically used in automated scripts only.

### **Change Callers Roles**

If during a call, you wish to change the role of the specific caller in a conference, you may use the *call-role* command of *mf* console to set new role for the call, i.e. to grant new permissions to this call. To change the role of the caller in a conference you should run the following *mf* console command:

call-role <conf\_number> <ses\_id> {host|participant|listener}
where

- <conf\_number> The number of the conference for which call you wish to set the new role.
- <ses id> The call session identifier which role you would like to change.
- {host:participant|listener} denotes what role (host or participant or listener) should be set for the call, i.e. the specified permissions are being granted to the call.

All arguments are required.

For example if you would like to set *host* permissions to the session *16777261* in the conference *667788* you should run *mf* console command: call-role *667788 16777261 host* 

If the command was implemented successfully you will receive the message: "Success"; if the conference not found or not started you will receive the message "Error: [ivr] Conference not found"; if the session not found you will receive the message "Error: [ivr] Session not found"; if you specified wrong argument, for example made mistake in the role attribute you will receive the message "Error: Wrong argument".

### **Mute Call Participants**

While a conference call is in progress, you can mute and un-mute the specific participants and listeners using the *call-mute* command of *mf* console. That means they will be muted, but they will be able to hear the conversation. To mute someone in a conference call you should run the following *mf* console command:

```
call-mute {false|strict|relaxed} <conf_number> <ses_id>
where
```

- {false|strict|relaxed} The argument false indicates that the call should be un-muted, strict indicates that the call should be muted and participants can not un-mute themselves, relaxed indicates that the call should be muted but participants can un-mute themselves.
- <conf\_number> The number of the conference which call you would like to mute or un-mute.
- <ses\_id> The session (i.e. call) identifier you wish to mute or un-mute. All arguments are required.

For example if you would like to mute the session *16777262* in the conference *667788* without the possibility to un-mute itself you should run *mf* console command: call-mute *strict 667788 16777262* 

If the command was implemented successfully you will receive the message: "Success"; if the conference not found or not started you will receive the message "Error: Conference not found"; if the session not found you will receive the message "Error: Session not found".

### **Placing Call Participants on Hold**

While a conference call is in progress, you can place the specific participants and listeners on hold or take them of hold using the *call-hold* command of *mf* console. That means they will hear music and do not hear the conversation. To place someone on hold or to take someone of hold in the conference call you should run the following *mf* console command: call-hold {true|false} <conf\_number> <ses\_id> where

- {true|false} The argument *true* indicates that the call should be placed on hold, *false* indicates that the call should be taken of hold.
- <conf\_number> The number of the conference which call you would like to place on hold or to take of hold.
- <ses\_id> The session (i.e. call) identifier you wish to place on hold or to take of hold.

All arguments are required.

For example if you would like to place the session 16777262 in the conference 667788 on hold you should run mf console command:

```
call-hold true 667788 16777262
```

If the command was implemented successfully you will receive the message: "Success"; if the conference not found or not started you will receive the message "Error: Conference not found"; if the session not found you will receive the message "Error: Session not found".

#### **Set Custom Name for a Call**

While a conference call is in progress, you can define the custom name for any conference participants using the *call-custom-name* command of *mf* console. To change the custom name for someone in the conference call you should run the following *mf* console command:

```
call-custom-name <conf_number> <ses_id> [<name>]
where
```

- <conf\_number> The number of the conference for which call you wish to define the custom name.
- <ses\_id> The session (i.e. call) identifier for which you wish to define the custom
- <name> new custom name for the call, if this argument is omitted the empty custom name will be set to this call.

Arguments < conf number > and < ses id > are required.

For example if you would like to define the custom name *Guest* for the session *16777262* in the conference *667788* you should run *mf* console command:

call-custom-name 667788 16777262 Guest

If you would like to remove the custom name for this call, i.e. to define the empty custom name for it you should run *mf* console command:

```
call-custom-name 667788 16777262
```

If the command was implemented successfully you will receive the message: "Success"; if the conference not found or not started you will receive the message "Error: Conference not found"; if the session not found you will receive the message "Error: Session not found".

### Set Audio Key for a Call

While a conference call is in progress, you can define the audio key for any conference call using the *call-associate* command of *mf* console. The calls with the same audio key are being grouped into the same bundle. To change the audio key for someone in the conference call you should run the following *mf* console command:

```
call-associate <conf_number> <ses_id> [<audiokey>]
where
```

- <conf\_number> The number of the conference for which call you wish to define the audio key, i.e. the bundle.
- <ses\_id> The session (i.e. call) identifier for which you wish to define the audio key (bundle).
- <audiokey> new audio key for the call, if this argument is omitted the audio key for this call will be set to 0, i.e. the call in this case will not belong to any bundle.

Arguments <conf number> and <ses id> are required.



You can associate the audio key for the voice call only if there is the control call in the conference that already has the same audio key and the roles of both calls (the control call and the voice call) are the same.

For example if you would like to define the audio key 886 for the session 16777287 in the conference 758288 you should run mf console command:

```
call-associate 758288 16777287 886
```

If you would like to remove the audio key for this call, i.e. to exclude the call from any bundle you should run *mf* console command:

```
call-associate 758288 16777287
```

As you can see on Figure 31 after assigning the audio key to the voice call, both calls – the control and the PSTN – have the same audio key; AKey column after the command execution has 886 value for both records; that means both these calls belong to the same bundle. After removing of the audio key from the voice call the AKey column shows  $\theta$  value for the voice call, the control call still have the audio key 886.

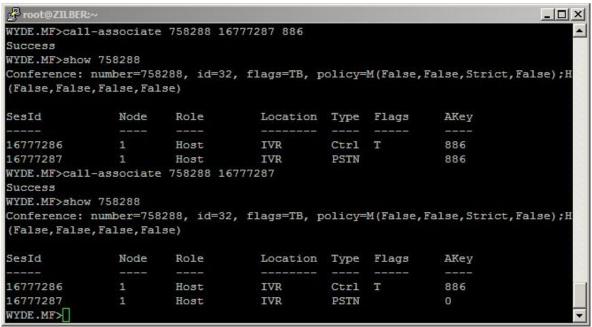


Figure 31: mf Console call-associate Commands Output Sample

If the command was implemented successfully you will receive the message: "Success"; if the conference not found or not started you will receive the message "Error: Conference not found"; if the session not found you will receive the message "Error: Session not found".

# Conferences Management using mf Console

The basic tool to manage the conferences is *mf* console; *wyde* command line utility provides only few possibilities to control them. This section explains how to manage the conferences using *mf* console; the alternative *wyde* command will be noted if applicable.

## **Dropping a Conference**

If during a call, you wish to cancel the entire conference for all users, you may use the *conf-drop* command of *mf* console to end the conference. To drop a conference you should run the following *mf* console command:

```
conf-drop <conf_number>|all
where
```

• <conf\_number> - The number of the conference you would like to drop. You can use *all* keyword instead of conference number to indicate that all conferences started on the bridge should be dropped. This argument is required.

For example if you would like to drop the conference 667788 you should run *mf* console command:

```
conf-drop 667788
```

If the command was implemented successfully you will receive the message: "Success"; if the conference not found or not started you will receive the message "Error: Conference not found".

As the alternative to this command to drop call participants you can use the *wyde* command *drop-conf*; see wyde Command Reference for details. This *wyde* command is basically used in automated scripts only.

Moreover you can drop the individual conference participants as it was previously described in section: Dropping Call Participants.

### **Dropping a Conferences Hosts, Participants and Listeners**

While a conference is in progress, you can drop all hosts and participants and listeners using the *conf-drop-group* command of *mf* console. That means all hosts or all participants or all listeners will be dropped from the conference as the result of this command. To drop all hosts or all participants or all listeners from a conference you should run the following *mf* console command:

conf-drop-group {host|participant|listener} <conf\_number>
where

- {host|participant|listener} Denotes who (all hosts or all participants or all listeners) should be dropped from the conference.
- <conf\_number> The number of the conference which participants or listeners calls you wish to drop (\*).

All arguments are required.

Moreover you can drop the individual conference participants as it was previously described in section: Dropping Call Participants.

#### **Conference Mute and Q&A Modes**

The conference can be in one of the following muting modes that determine how the conference is muted. The conference modes are:

- *open* Anyone can talk and un-mute themselves.
- *relaxed* All participants are muted by default, but they can un-mute themselves, using default \*6 on their DTMF keypad. Hosts are not muted.
- *strict* None of the participants can un-mute themselves. The only way to un-mute a caller is through the command line interface or by the host. In this mode, when participant tries to un-mute using default \*6 the system tells "conference host has muted the conference; this line can not be unmuted".
- question The conference is in Q&A mode. In this mode, if participant presses \*6 the system tells "If you'd like to ask a question press 1". Q&A Sessions will be described later in the next section of this guide.

If during a call, you wish to mute or un-mute all hosts or all participants in the conference you may use the *conf-mute-group* command of *mf* console to mute them. To mute/un-mute the group of callers you should run the following *mf* console command:

```
conf-mute-group {false|strict|relaxed} {host|participant}
  <conf number>
```

#### where

- {false|strict|relaxed}
  - o *false* indicates that the group (hosts or participants) should be un-muted the conference mode will be *open*;
  - o if the next argument is host, strict or relaxed indicates that the hosts should be muted, but they can un-mute themselves;
  - o if the next argument is participant, strict indicates that all participants should be muted and they can not un-mute themselves the conference mode will be *strict*;
  - o if the next argument is participant, relaxed indicates that all participants should be muted but they can un-mute themselves the conference mode will be relaxed.
- {host|participant} Denotes who (all hosts or all participants) should be muted or un-muted.
- <conf\_number> The number of the conference which calls you wish to mute or un-mute.

All arguments are required.

For example if you would like to mute all *participants* in the conference 667788 without the possibility to un-mute themselves you should run *mf* console command: conf-mute-group *strict participant* 667788

If the command was implemented successfully you will receive the message: "Success"; if the conference not found or not started you will receive the message "Error: Conference not found", if you transferred wrong argument (not false, strict, relaxed, host, participant) you will receive the message "Error: Wrong argument".

Moreover you can mute/unmute the individual conference participants as it was previously described in section: Mute Call Participants.

#### **Q&A Sessions**

The WYDE conferencing software has a feature built in known as Q&A (questions and answers). You can set up and manage a Q&A session using the command line interface *mf* console that allows conference participants to, in turn, ask questions and receive answers. When Q&A session is started all participants will be muted and they will be unable to unmute their self, but they will be able to request to speak (to request question). You as the host can allow them to speak using *mf* console. After they have finished speaking you can drop them out of the Q&A queue or you can drop the user from the Q&A queue without allowing them to speak using *mf* console commands.

The Q&A also can be managed using DTMF keypad and the Web Administration Interface as it is described in "Web Administration Interface – User Guide".

To start or stop Q&A mode for the conference and to clear Q&A queue you should run the following *mf* console command:

```
conf-qa-mode {start|stop|clear} <conf number>
```

#### where

- {start|stop|clear} start indicates that Q&A session should be started for the conference, i.e. the conference mode should be *question*; stop indicates that Q&A session should be stopped for the conference; clear indicates that Q&A queue should be cleared for the conference. i.e. all participants should be removed from Q&A queue.
- <conf\_number> The number of the conference for which you would like to manage the O&A session.

All arguments are required.

To request to speak and to cancel this request within Q&A session you should run the following *mf* console command:

```
call-qa-request {start|stop} <conf_number> <ses_id>
where
```

- {start|stop} start indicates that the call should start the request to speak (the request to ask the question), i.e. placed into Q&A queue; stop indicates that the call should stop (cancel) the request to speak, i.e. removed from Q&A queue.
- <conf\_number> The number of the conference where Q&A session is started.
- <ses\_id> The call session identifier that should be placed into Q&A queue or removed from it.

All arguments are required.

To enable Q&A session for the specific call from Q&A queue, i.e. to unmute this call and to disable Q&A session for the specific call, i.e. to mute this call you should run the following *mf* console command:

```
call-qa-talk {enable|disable} <conf_number> <ses_id>
where
```

- {enable|disable} enable indicates that the call from Q&A queue should be unmuted and the participants should be able to speak (to ask his question); disable indicates that the unmuted call should be muted and removed from Q&A queue.
- <conf\_number> The number of the conference where Q&A session is started.
- <ses\_id> The call session identifier that should be unmuted to ask his question or should be muted again and/or removed from Q&A queue.

All arguments are required.

To move to the next questioner from Q&A queue, i.e. to allow speaking (enable Q&A session) for the first call in the queue you should run the following *mf* console command: conf-qa-talk <conf\_number> where

<conf\_number> - The number of the conference where Q&A session is started and where you would like to enable (unmute) the first call in the Q&A queue. The active questioner (if exists) will be removed from Q&A queue, so this command ends Q&A session for the current questioner and starts it for the next one. This argument is required.

To mute and unmute the active Q&A session you should run the following mf console command:

conf-ga-mute {true|false} <conf number>

- $\{true | false\} true indicates that the active Q&A session should be muted;$ false indicates that the active Q&A session should be unmuted (\*). Both these options do not remove the questioner from Q&A queue; the command just temporary allows/disallows the active questioner to speak.
- <conf number> The number of the conference where Q&A session is started and where you would like to mute/unmute the active Q&A session (\*). All arguments are required.

Let's assume that we are maintaining Q&A session in the conference 667788 and there are two participants' sessions in this conference: 16777302 and 16777303. Table 5 shows

samples how to manage this conference Q&A sessions using *mf* console commands. Table 5: O&A Sessions Management Samples Using mf Console Commands

Table 5: Q&A Sessions Management Samples Using mf Console Commands					
Description	mf Console Command	<b>DTMF</b>			
Start Q&A mode for the conference	conf-qa-mode start 667788	*1 1			
Request to speak (question) for	call-qa-request <i>start 667788 16777302</i>	*6			
specific calls, i.e. place the calls to	call-qa-request <i>start 667788 16777303</i>				
Q&A queue					
Cancel the request to speak for	call-qa-request <i>stop 667788 16777303</i>				
specific calls					
Clear Q&A queue in the conference	conf-qa-mode <i>clear 667788</i>	*1 5			
Enable Q&A session for the specific	call-qa-talk <i>enable 667788 16777302</i>				
call from Q&A queue, i.e. to allow					
him to speak					
Disable Q&A session for the specific	call-qa-talk <i>disable 667788 16777302</i>				
call, mute the call, and remove					
participant from Q&A queue					
Enable Q&A session for the first call	conf-qa-talk 667788	*1 2			
in Q&A queue, i.e. move to the next					
questioner					
Mute the active questioner	conf-qa-mute <i>true 667788</i>	*1 4			
Unmute the active questioner	conf-qa-mute false 667788	*1 4			
Stop Q&A mode for the conference	conf-qa-mode <i>stop 667788</i>	*1 3			

### Placing a Conference on Hold

While a conference is in progress, you can place all hosts and participants and listeners on hold or take them of hold (i.e. place them on line) using the conf-hold-group command of mf console. That means they will hear music and do not hear the conversation while hosts have a private discussion. To place all hosts or all participants or all listeners on hold or to take them of hold you should run the following *mf* console command:

```
conf-hold-group {true|false} {host|participant|listener}
    <conf number>
where
```

- {true | false} The argument true that the group (participants or listeners) should be placed on hold, false indicates that the group (participants or listeners) should be taken of hold.
- {host|participant|listener} Denotes who (all hosts or all participants or all listeners) should be placed on hold or taken of hold.
- <conf\_number> The number of the conference which calls you would like to place on hold or to take of hold.

All arguments are required.

For example if you would like to place on hold *participants* in the conference 667788 you should run *mf* console command:

```
conf-hold-group true participant 667788
```

If the command was implemented successfully you will receive the message: "Success"; if the conference not found or not started you will receive the message "Error: Conference not found", if you transferred wrong argument (not true, false, participant, listener) you will receive the message "Error: Wrong argument".

Moreover you can place on hold and take of hold the individual conference participants as it was previously described in section: Placing Call Participants on Hold.

### **Activate a Conference for Participants and Listeners**

While a conference is in progress you may need to start or stop conference active mode for the groups of callers denoted in "Specifies which roles will be hold on the beginning of conference" (conference\_hold\_policy) call flow attribute — by default for participants and listeners. This also could be used if "How conference begins" (conference\_start\_how) call flow attribute value is "explicit"; in this case you should explicitly make the conference active to allow all your participants and/or listeners take an active part in the conference. When the conference active mode is stopped all participants and listeners are on hold, i.e. hear music on hold, and do not hear the conference. To start/stop the conference active mode for participants and listeners you should use the conf-active command of mf console: conf-active {start|stop} <conf\_number> where

- {start|stop} Denotes should or should not the conference be active for the groups of callers denoted in "Specifies which roles will be hold on the beginning of conference" (conference\_hold\_policy) call flow attribute by default for participants and listeners: start indicates that the callers of the specified roles should hear and participate in the conference; stop indicates that all callers of these roles should be on hold and do not hear the conference.
- <conf\_number> The number of the conference you wish to activate/deactivate, i.e. the conference for which you would like to change its active mode.

  All arguments are required.

Let's assume that the "Specifies which roles will be hold on the beginning of conference" (conference\_hold\_policy) call flow attribute value is "p1" (participants and listeners). For example if for the conference 667788 you would like to stop active mode for participants and listeners you should run mf console command:

```
conf-active stop 667788
```

All participants and listeners will be placed on hold as the result of this command execution. If you would like to *start* the conference active mode for them again you should run *mf* console command:

```
conf-active start 667788
```

All participants and listeners will be online as the result of this command execution, i.e. they will hear the conference.

If the command was implemented successfully you will receive the message: "Success"; if the conference not found or not started you will receive the message "Error: Conference not found", if you transferred wrong argument (neither start, nor stop) you will receive the message "Error: Wrong argument".

### **Making a Conference Secure**

While a conference is in progress, from time to time you may want to make a call secure. That is, make it so that no other participants or listeners can join to the conference (hosts still can join to secured conferences). To do this you should use the *conf-secure* command of *mf* console:

```
conf-secure {secure|unsecure} <conf_number>
where
```

- {secure | unsecure} Denotes should the conference be made secured or unsecured.
- <conf\_number> The number of the conference you wish to make secured or unsecured.

All arguments are required.

For example if you would like to make the conference 667788 secured you should run mf console command:

```
conf-secure secure 667788
```

If the command was implemented successfully you will receive the message: "Success"; if the conference not found or not started you will receive the message "Error: Conference not found", if you transferred wrong argument (neither secure, nor unsecure) you will receive the message "Error: Wrong argument".

### **Setting a Conference Job Code**

While a conference is in progress, you can define the conference job code using the *confjobcode* command of *mf* console. When any of the calls ends the current conference job code will be stored in the CDR record of the call; for example this job code can be used in the calls reporting. To set the conference job code you should run the following *mf* console command:

conf-jobcode <conf\_number> [<code>]
where

- <conf\_number> The number of the conference which job code you would like to set. This argument is required.
- <code> New job code for the conference. If this argument is omitted there will be no job code stored in CDR records of the calls.

For example if you would like to set job code 8899 for the conference 667788 you should run *mf* console command:

```
conf-jobcode 667788 8899
```

If the command was implemented successfully you will receive the message: "Success"; if the conference not found or not started you will receive the message "Error: Conference not found".

### **Recording a Conference**

While a conference is in progress, you can record the conference using the *conf-recording* command of *mf* console. The recorded conferences can be playback as it will be described later in this guide. To start/stop the conference recording you should run the following *mf* console command:

### where

- {start|stop} Denotes should the conference recording be started or stopped.
- <conf number> The number of the conference you wish to record.
- <accesscode> Denotes the host access code, i.e. password to the recording server if "Recording method" (recording\_method) call flow attribute value is either "local+pin" or "remote+pin" (this value can be either defined on call flow level or overridden on DNIS level).
- <subscriber\_pin> Denotes the subscriber's pin (that usually could be requested from the user) to perform the recording authorization if "Recording method" (recording\_method) call flow attribute value is either "local+pin" or "remote+pin" (this value can be either defined on call flow level or overridden on DNIS level).

The arguments {start|stop} and <conf\_number> are required. The arguments <accesscode> and <subscriber\_pin> should be used only if "Recording method" call flow attribute value is either "local+pin" or "remote+pin".

For example if you would like to *start* the recording for the conference 667788 you should run *mf* console command:

```
conf-recording start 667788
```

If the command was implemented successfully you will receive the message: "Success"; if the conference not found or not started you will receive the message "Error: Conference

not found", if you transferred wrong argument (neither start, nor stop) you will receive the message "Error: Wrong argument".

Note that the conference recording could be made either locally (i.e. to the computer where your WYDE bridge software installed) or to the external recording server; different recording methods configuration is described later in this Guide in section: Recording Settings Configuration.

## Playing an Audio File to a Conference

While a conference is in progress, you can playback the audio file to the conference using the *conf-play-file* command of *mf* console. Using this command you can either playback previous this conference recordings or playback the files that were previously uploaded via web interface. To playback the file you usually need to use the control call to this conference and play this file via such control call. To manage of playing the audio file to the conference you should run the following *mf* console command:

#### where

- <conf\_number> The number of the conference where you would like to play the audio file.
- <ses\_id> The session identifier (usually the identifier of the control call) which should be used to play the audio file.
- {assign <dir> <filename>|start|stop|seek <offset> <whence>} One of the following arguments should be specified here:
  - o assign <dir> <filename> assign the file for the playback:
    - o <dir> either record for the conference recorded files (i.e. previous this conference recordings) or upload for the uploaded files (i.e. the files uploaded via web);
    - o <filename> the audio file name without extension, this file should be in the conference recording folder (usually /usr/local/DNCA/var/recordings/ folder) subfolder, either record subfolder or upload subfolder for the specific conference.
  - o start Start the playback from the current position.
  - o stop Stop the playback.
  - o seek <offset> <whence> Seek the audio file playback indicator (pointer) on offset seconds relative to the parameter whence:
    - $\circ$  0 starting from the beginning of the file;
    - $\circ$  1 starting from the current position in the file;
    - $\circ$  2 starting from the end of the file.

All arguments are required.

Let's assume that you should implement the following scenario: for the conference 758288 you would like to play the conference recording file 38 (actually this is the conference identifier that is used as conference recording file name), i.e. you need to assign this file to the control call; after that you need to seek the audio file on 5 seconds starting from the

beginning of the file; and finally you need to *start* the playback from the current position. To implement this scenario you should run the following *mf* console commands:

```
conf-play-file 758288 16777297 assign record 38
conf-play-file 758288 16777297 seek 5 0
conf-play-file 758288 16777297 start
```

You will see the screen similar to shown on Figure 32.

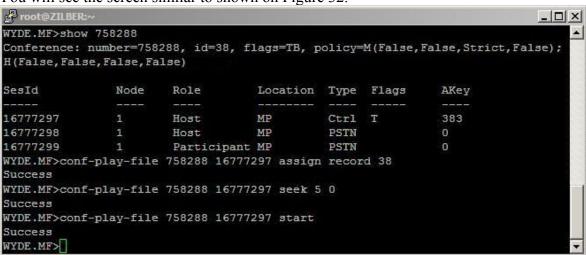


Figure 32: mf Console conf-play-file Commands Output Sample

If the command was implemented successfully you will receive the message: "Success"; if the conference not found or not started you will receive the message "Error: Conference not found"; if the session not found you will receive the message "Error: Session not found".

### **Dialing another User**

Often, there will be a conference call in progress and you will need to call another participant. That is where the dialout feature comes in. If your system supports dialing outside users, you can dial-out another user using the *dialout* command of *mf* console. To dial to another user you should run the following *mf* console command:

```
dialout <peer_number> <did_number> <accesscode>
where
```

- <peer number> The phone number you wish to dial.
- <did\_number> The target conference DNIS (DID) number where the call should be joined after the dial-out is complete.
- <accesscode> The access code that should be used to join to the target conference.

All arguments are required.

Note that this command allows making dial-out and connecting the call to another conference; the target conference is being defined by the pair: access code and role. The access code and role are being verified and the target conference is being defined prior to the call; while the dialout is in progress and when dialout is complete the call is being

joined to the conference defined by access code and role transferred to the *dialout* command as its parameters.

While the conference is in progress if you need to see its dialout attributes, i.e. DNIS numbers and access codes that could be used in *dialout* command and that were used by the calls that already connected to this conference, you can use the following *mf* console command:

```
dialout-attr <originator_conf_number>
where
```

<originator\_conf\_number> - the number of the started conference which
dialout attributes (DNISes and access codes) you would like to show. This argument is
required.

This command returns the list of all used DNIS numbers and access codes with roles for the requested started conference:

For example if you would like

- to show DNIS numbers and access codes for the started conference 666001;
- to make dialout to phone number 9197473626, and connect this call to this conference using DNIS number 8665080012 and access code 6601

you should run *mf* console commands:

```
dialout-attr 666001
dialout 9197473626 8665080012 6601
```

#### See Figure 33 for details.

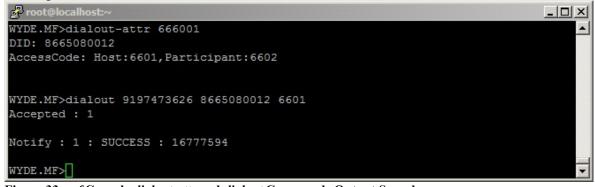


Figure 33: mf Console dialout-attr and dialout Commands Output Sample

The command *dialout* is asynchronous command; because of that *mf* console is unable to return its result immediately after the command was started. When you run this command, the system returns the message: "Accepted: <taskId>", where <taskId> – is identifier for the dialout task. If the command was implemented successfully you will receive the message: "Notify: <taskId>: SUCCESS: <sessionId>", where <taskId> – the same task identifier that was assigned to dialout task and was shown in "Accepted" message and <sessionId> – the dialout call session identifier; if there was any error during the command

execution you will receive the message: "Notify: <taskId>: ERROR: <error message>", where <taskId> – once again the same task identifier and <error message> – the specific error message (for example "DID" or "AccessCode" reporting what is wrong in the command, etc.).

Note that the WYDE bridge should be configured to perform dialout; this dialout configuration is described later in this Guide in section: Dialout Settings Configuration.

### Move a Call to another Conference

From time to time you may need to move the caller from one started conference to another conference. That means that the call is being removed from his current conference and moved to another conference. The target conference could be either started or it could be not started, but it becomes started during this move. To make such move you should run the following *mf* console command:

- <conf\_number> The source conference number whose call you would like to move to another conference.
- <ses id> The call session identifier that you wish to move to another conference.
- <new\_did\_number> New (i.e. target) conference DNIS (DID) number where the call should be moved.
- <new\_accesscode> The access code that should be used to join to new (i.e. target) conference.
- <new\_role> The role that will be granted to the call when it joins to new conference. This argument is applicable only for call flows without authorization, for example *CONF* call flow.

The argument <new\_role> is optional and should be used for call flows without authorization only. All other arguments are required.

Let's review the following scenario: there are two conferences on the bridge – the conference 667788 with the call session 16777325 and the conference 758288 with the call session 16777324; we need to move the call 16777324 from the second conference to the first one; DNIS (DID) number to connect to the conference 667788 is 12 and the access code is 6601 (for hosts). To implement this scenario the following command should be run: call-move 758288 16777324 12 6601

See Figure 34 for details. In addition this figure shows what conferences and calls were on the bridge prior to this command call and after this command call.

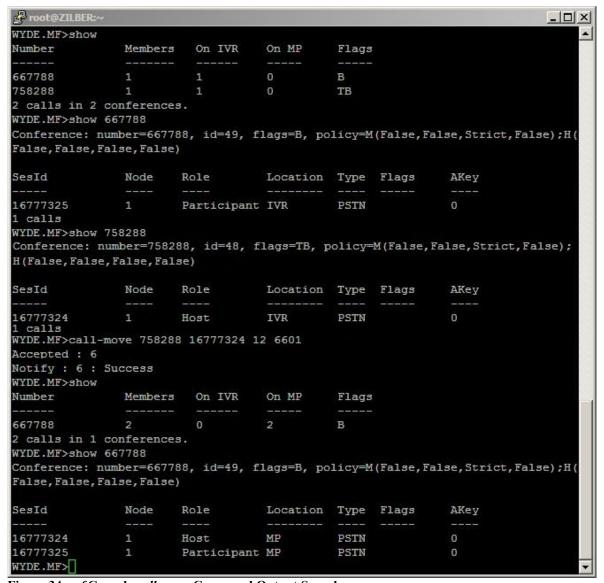


Figure 34: mf Console call-move Command Output Sample

The command *call-move* is asynchronous command; because of that *mf* console is unable to return its result immediately after the command was started. When you run this command, the system returns the message: "Accepted: <taskId>", where <taskId> – is identifier for the call move task. If the command was implemented successfully you will receive the message: "Notify: <taskId>: Success", where <taskId> – the same task identifier that was assigned to the move task and was shown in "Accepted" message; if the was any error during the command execution you will receive the message: "Notify: <taskId>: Error: <error message>", where <taskId> – once again the same task identifier and <error message> – the specific error message (for example "Conference not found", "Session not found", "DID" – if DNIS number specified in <new\_did\_number> argument does not exist or incorrect, "AccessCode" – if the access code in <new\_accesscode> argument does not exist or incorrect, etc.).

#### Move a Call between Sub-Conferences

Sometimes you may need to attach the callers to specific sub-conference or to detach the callers from specific sub-conference or to move the callers between sub-conferences. The callers that are joined to the same sub-conference are having private conversations between each other, but not listening/talking remaining conference callers. To manage sub-conferences, i.e. to attach/detach/move the callers from/to/between sub-conferences you should run the following *mf* console command:

```
 subconf < conf\_number > < ses\_id > \{attach < subconf > | detach < subconf > | move < subconf\_from > < subconf\_to > \} \\
```

#### where

- <conf\_number> The conference number whose call you would like to attach to sub-conference or detach from sub-conference or to move between sub-conferences. This argument is required.
- <ses\_id> The call session identifier that you wish to attach to sub-conference or detach from sub-conference or to move between sub-conferences. This argument is required.
- One of the following arguments should be specified as the next argument of this command:
  - o attach <subconf> denotes the name of the sub-conference up to 16 characters length (only letters and digits are allowed as the name of the sub-conference) to which the call should be attached;
  - o detach <subconf> denotes the name of the existed sub-conference from which the call should be detached;
  - o move <subconf\_from> <subconf\_to> denotes the source subconference name from which the call should be transferred and the target subconference name to which the call should be transferred

Let's review the following scenario: for the started conference 660660 attach its call sessions (for example 16777240 and 16777243) to the sub-conference with name "1". To implement this scenario the following commands should be run:

```
subconf 660660 16777240 attach 1 subconf 660660 16777243 attach 1
```

See Figure 35 for details. In addition this figure shows the conference calls prior to this command call and after this command call. As you can see if the conference has subconferences, the name (or names) of the sub-conference together with the calls that are joined to the sub-conference are listed in *show* conference *mf* command output below the list of calls that are joined to the conference.

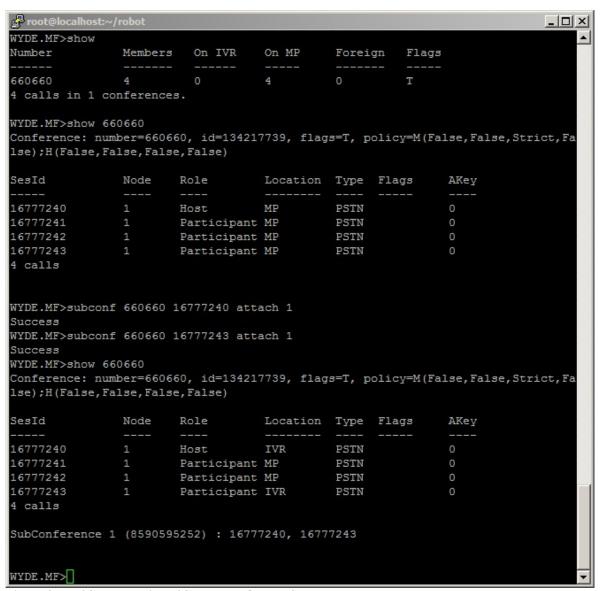


Figure 35: mf Console subconf Command Output Sample

The command *subconf* is asynchronous command; because of that it could take some time after the *subconf* command was run before the call would be actually attached to the subconference or detached from it. If the specified conference not found or not started you receive the error message "*Error*: [ivr] Conference not found: confNumber=<conf\_number>"; if the specified session not found or not belong to the specified conference you receive the error message "*Error*: [ivr] Session not found: confNumber=<conf\_number> sesId=<ses\_id>"; if the command arguments are correct and the command could be implemented you receive the message: "Success".

## **Sending Chat Messages to the Conference Callers**

While the conference is in progress you may need to send text chat messages to the conference callers. These messages could be sent to the specific recipients that will receive the real-time CHAT-MSG notifications about the messages that has been sent (the message

sender session identifier in these notifications is 0); see detailed information in "Real Time Interface – Programmer's Guide". To send text messages to the callers of the conference you should run the following mf console command:

```
conf-chat-send <conf_number> <recipients> <text>
where
```

- <conf\_number> The number of the conference to which callers you would like to send the text chat messages. This argument is required.
- <recipients> The recipients to whom the message should be sent. In this argument you could specify either comma-delimited list of roles, i.e. <host, participant, listener> or comma-delimited list of the caller session identifiers. This argument is required.
- <text> The message that should be sent. This argument must be Base64-encoded, i.e. encoded in MIME's *Base64* scheme. This argument is required.

Below we provide samples of sending text messages to all *participants* of the conference 321451, to *hosts* and *participants* of this conference, to specific single caller in the conference, and to the list of the conference callers.

```
conf-chat-send 321451 participant cGxlYXN1cmUu conf-chat-send 321451 host,participant cGxlYXN1cmUu conf-chat-send 321451 16777670 cGxlYXN1cmUu conf-chat-send 321451 16777670,16777671 cGxlYXN1cmUu
```

See Figure 36 for details. In addition this figure shows the conference calls that are joined to the specified conference.

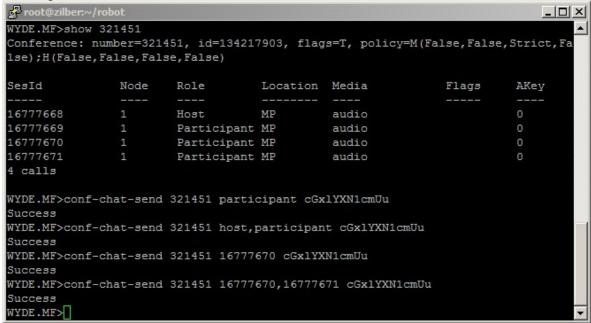


Figure 36: mf Console conf-chat-send Command Output Sample

If the command was implemented successfully you will receive the message: "Success"; if the conference not found or not started you will receive the message "Error: [ivr]

Conference not found'; if the session not found or invalid roles are specified in the feipients> argument you will receive the message "Error: [ivr] Session not found".

# **Conferences Broadcasting**

If the conference is in progress you can also manage broadcasting (e.g. screen sharing meeting, video conferencing) in your conferences. That means that you can send the proposal to start the screen sharing or video broadcasting, to start the presentation, to end the presentation or to reject the proposal to start the presentation using *mf* console commands.

To propose the presentation starting (e.g. screen sharing meeting, video conferencing) to the specific caller of the conference you should run the following *mf* console command: conf-broadcast-proposal <media> <conf\_number> <ses\_id> where

- <media> The comma-delimited list of medias for the presentation, the possible values: <video, screensharing>. This argument is required.
- <conf\_number> The number of the conference to which caller you would like to propose the presentation. This argument is required.
- <ses\_id> The caller session identifier to which proposing to be a presenter. This argument is required.

When this command is executed the bridge sends the real-time NOTIFY-BROADCAST-PROPOSAL notifications about the proposal to start the presentation to all conference calls (the session identifier of the call that is proposed to start broadcasting, i.e. the session that received the proposal is transferred in this notification); see detailed information in "*Real Time Interface – Programmer's Guide*".

To start the presentation (e.g. screen sharing meeting, video conferencing) by the specific caller of the conference you should run the following *mf* console command: conf-broadcast-start <media> <conf\_number> <ses\_id> where

- <media> The comma-delimited list of medias of the starting presentation, the possible values: <video, screensharing>. This argument is required.
- <conf\_number> The number of the conference which caller should start the presentation. This argument is required.
- <ses\_id> The session identifier of the caller who starts the presentation. This argument is required.

Note that this command starts presentation in any case – even if there is another presenter in the conference (in this case the started broadcasting will be forcibly ended). When this command is executed the bridge sends the real-time NOTIFY-BROADCAST-START notifications about the starting presentation (screen sharing or video or both) to all conference calls (the session identifier of the connection that is starting presentation is transferred in this notification); see detailed information in "Real Time Interface – Programmer's Guide".

To reject the started presentation and/or to cancel the proposal to start the presentation of the specific caller of the conference you should run the following *mf* console command: conf-broadcast-reject <reason> <conf\_number> <ses\_id> where

- <reason> The reject reason: Normal is used by presenter (host or participant) to
  end the started presentation, Cancelled is used by the participant received the
  proposal (who received conf-broadcast-proposal) to cancel (refuse) this
  proposal. This argument is required.
- <conf\_number> The number of the conference for which caller the presentation and/or proposal to start the presentation should be rejected. This argument is required.
- <ses\_id> The caller session identifier whose presentation and/or proposal to start the presentation should be rejected. This argument is required.

When this command is executed the bridge sends the real-time NOTIFY-BROADCAST-REJECT notifications informing that the presentation (screen sharing and/or video) has been stopped or that the proposal to start the presentation was cancelled (the session identifier of the connection that stopped the presentation or that cancelled previously received proposal to start the presentation is transferred in this notification); see detailed information in "Real Time Interface – Programmer's Guide".

Below we provide broadcasting workflow logic and samples assuming that the started conference has the following calls: *host* (sessionId = 16777731) and *participant* (sessionId = 16777732):

- *Host* starts the presentation:
  - Host's presentation is being started by *mf* console command:
    - conf-broadcast-start screensharing 321451 16777731
  - o Bridge sends real-time notifications to all calls:
    - NOTIFY-BROADCAST-START 16777731 screensharing
- *Host* ends the presentation:
  - Host's presentation is being ended by *mf* console command:
    - conf-broadcast-reject *Normal 321451 16777731*
  - o Bridge sends real-time notifications to all calls:
    - NOTIFY-BROADCAST-REJECT 16777731 Normal
- The presentation is being proposed to *participant*, but *participant* cancels (refuses) the proposal:
  - The presentation proposal request is being sent by *mf* console command:
    - conf-broadcast-proposal screensharing 321451 16777732
  - Bridge sends real-time notifications to all calls:
    - NOTIFY-BROADCAST-PROPOSAL 16777732 screensharing
  - The presentation proposal request is being rejected by *mf* console command:
    - conf-broadcast-reject Cancelled 321451 16777732
  - o Bridge sends real-time notifications to all calls:
    - NOTIFY-BROADCAST-REJECT 16777732 Cancelled
- *Host*'s presentation is started and *host* proposes the presentation to *participant* and *participant* accepts the proposal, i.e. starts the presentation:
  - The presentation proposal request is being sent by *mf* console command:
    - conf-broadcast-proposal *screensharing 321451 16777732*

- o Bridge sends real-time notifications to all calls:
  - NOTIFY-BROADCAST-PROPOSAL 16777732 screensharing
- o Participant's presentation is being started by *mf* console command:
  - conf-broadcast-start screensharing 321451 16777732
- o Bridge sends real-time notifications to all calls:
  - NOTIFY-BROADCAST-REJECT 16777731 Normal
  - NOTIFY-BROADCAST-START 16777732 screensharing

See Figure 37 for details. In addition this figure shows the conference calls that are joined to the specified conference.

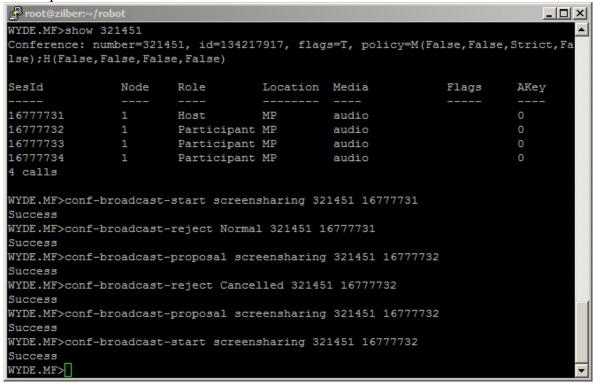


Figure 37: mf Console conf-broadcast-proposal, conf-broadcast-start, and conf-broadcast-reject Commands Output Sample

If these commands were implemented successfully you will receive the message: "Success"; if the conference not found or not started you will receive the message "Error: [ivr] Conference not found"; if the session not found you will receive the message "Error: [ivr] Session not found"; if you transferred wrong number of arguments or invalid media or reason you will receive the message "Error: Wrong argument"; if you are trying to reject not started/not proposed session or sending more than one broadcasting proposal or performing any other invalid actions you will receive the message "Error: [ivr] Invalid usage".

# Making a Shunt between Two Conferences

While two conferences are in progress, you may allow to the callers from one of them hearing the callers from another and vice versa using the *conf-shunt* command of *mf* console. The shunt (cross join bridge) is being created between these two conferences and all callers from these conferences can hear each other, but both these conferences still

remain. To start/stop the shunt between two conferences you should run the following *mf* console command:

#### where

- {start|stop} start denotes that the shunt between two conferences should be made (started). stop denotes that the shunt between two conferences should be dropped (stopped). This argument is required.
  - o <conf\_number> The number of the conference you wish to shunt or unshunt. This argument is required.
  - o conf\_number> The number of the peer conference you wish to
    shunt with the first one. This argument should be specified for start option
    only.

For example if you would like to *start* shunt between two conferences: 667788 and 758288, i.e. to allow these conferences hearing each other, you should run *mf* console command: conf-shunt *start* 667788 758288

If you would like to *stop* shunt between these two conferences you should run *mf* console command:

```
conf-shunt stop 667788
```

If the command was implemented successfully you will receive the message: "Success"; if any of the conferences not found or not started you will receive the message "Error: Conference not found", if you transferred wrong argument (neither start, nor stop) you will receive the message "Error: Wrong argument", if the shunt already exist for the conference you will receive the message "Error: Shunt is already exists for this conference.".

# **Polling**

While a conference is in progress, you can start polling (voting) for the conference using the *conf-polling* command of *mf* console. While the polling is in progress all conference participants can vote, i.e. select one of the available options defined when the polling was started. If you need detail information about the polling process, including information how to get polling results please read "Web Administration Interface – User Guide", section: Polling. To start/stop the conference polling you should run the following *mf* console command:

```
conf-polling <conf_number> {start <keys>|stop}
where
```

- <conf\_number> the number of the conference for which you would like to start or to stop the polling;
- {start <keys>|stop} start indicates that the polling should be started for the conference, stop indicates that the polling should be stopped for the conference;
  - o <keys> available polling options (i.e. digits 1, 2, ..., 9, 0) that should be specified when the polling is started.

All arguments are required.

For example if you would like to *start* the polling for the conference 667788 and options I, 2, and 3 should be available for voting, you should run mf console command:

```
conf-polling 667788 start 123
```

If you would like to *stop* the polling for this conference you should run *mf* console command:

```
conf-polling 667788 stop
```

If the command was implemented successfully you will receive the message: "Success"; if the conference not found or not started you will receive the message "Error: Conference not found", if you transferred wrong argument (neither start, nor stop) or if you do not specify the polling options when you starting the polling you will receive the message "Error: Wrong argument".

# **Scheduling**

Any conference could be scheduled to take place at specific date (date range), time, week day (DOW), with specified number of the conference participants. The following call flow attributes are responsible for the conference scheduling:

- conference\_scheduled (Conference scheduled) denotes is the conference scheduled or not, this attribute value should be equal "on" for the scheduled conferences, if this attribute value is "off" all scheduling attributes are being ignored;
- conference\_schedule (Conference schedule) denotes date (date range), time, day of the week when the conference is scheduled and how many participants can be joined to this scheduled conference, this attribute has the following format: <scheduling 1>[;<scheduling 2>[;<scheduling 3>[;...]]] i.e. this is the list of the conference scheduling records separated with semicolon (;); each scheduling record has the following format: <part\_count>, <days>, <tm\_begin>, <duration>, <date\_begin>, <date\_end> where
  - o part count number of allowed the scheduled conference participants;
  - o days day of weeks when the scheduled conference is allowed: 1 (Sunday), 2 (Monday), ..., 7 (Saturday);
  - o tm begin time when the scheduled conference begins in hhmm format;
  - o duration the scheduled conference duration in minutes:
  - o date\_begin the beginning date in *YYMMDD* format when the conference scheduling is active (i.e. the first date when the scheduling is on);
  - o date\_end the ending date in *YYMMDD* format when the conference scheduling is active (i.e. the last date when the scheduling is on)

For example if you need to schedule the conference for 20 participants on Monday, Wednesday, Thursday, from 10:00 a.m. for 90 minutes (i.e. till 11:30 a.m.), for the date range from November 1<sup>st</sup>, 2009 till March 1<sup>st</sup>, 2010 the attribute value should be: 20, 245, 1000, 90, 091101, 100301

If in addition to previous scheduling you need to schedule the conference for 30 participants on November  $11^{th}$ , 2009 (Wednesday) the attribute value should be: 20, 245, 1000, 90, 091101, 091110; 30, 4, 1000, 90, 091111, 091111; 20, 245, 1000, 90, 091112, 100301

- conference\_schedule\_extend\_dtmf (Schedule conference extend DTMF policy) denotes who (h/p/l, default: h host) can extend duration of the scheduled conference:
- conference\_schedule\_extend\_dtmf\_binding (Schedule conference extend DTMF binding) denotes what keys should be used to extend duration of the scheduled conference, when these keys are pressed the following message will be played: "please press 1 to extend conference duration for 5 minutes, press 2 for 10 minutes, press \* to return to the conference", the total maximal extension time is defined in the attribute conference schedule extend;
- conference\_schedule\_extend (Schedule conference extend maximal time in seconds) denotes to what maximal time in seconds the duration of the scheduled conference can be extended;
- conference\_schedule\_remind (Schedule conference end remind time in seconds) denotes remind time in seconds before the scheduled conference ends when the reminder (warning) message should be played to moderator, for instance if this attribute value is 600 before 10 minutes (600 seconds) till the scheduled conference ending time the remind message will be played to the host;
- conference\_schedule\_hold (Wait before scheduled conference begin) denotes wait time in seconds before the scheduled conference begins when the participants can join; if the participant connected prior to this time his call will be dropped, if he connected after this time but prior the conference begin time music-on-hold will be played to him until the conference starts, for instance if the conference is scheduled to be started at 11:00 and this parameter values is 600 (i.e. 10 minutes) all participants' calls earlier than 10:50 will be dropped with the message that the conference is scheduled to be started at 11:00, the participants' calls from 10:50 till 11:00 will be placed on hold (music on hold will be played) until the conference begins.

The following conference scheduling logic is being implemented:

- if conference\_scheduled attribute is "off", the conference is not scheduled (the processing logic is usual, as for non-scheduled conferences);
- if conference schedule attribute is empty, the conference is not scheduled;
- otherwise, if conference\_scheduled attribute is "on" and conference\_schedule attribute describes conference scheduling, the conferences are being considered as scheduled and the following logic will be applied:
  - o if there are no any scheduled conferences for the nearest 30 days, the call will be dropped with the message "This conference has no schedule. Please check with the organizer. Good bye.";
  - o if for the current day of week all scheduled conferences are already completed, i.e. the current time is more than ending time for the last scheduled conference, the call will be dropped with the message "Your conference is not running or scheduled. Please check its schedule with the organizer. Good bye.";
  - o if till the scheduled conference beginning time (<tm\_begin>) more than hour the message "There is no upcoming conference scheduled within an hour. Please check the schedule with the organizer. Good bye." will be played and the call will be dropped;

- o if till scheduled conference beginning time (<tm\_begin>) less than hour, but more than period defined in conference\_schedule\_hold attribute value the message "Your conference will begin in <NNN> minutes. Please call again later. Good bye." (where <NNN> minutes till the conference beginning time) will be played and the call will be dropped;
- o if till scheduled conference beginning time (<tm\_begin>) less than conference\_schedule\_hold attribute value the message "Please wait. Your conference will begin shortly. This line is now on hold." will be played and the call will be placed on hold (music on hold will be played); at conference beginning time (<tm\_begin>) the message "Your conference is started. Please wait while we place you into the conference." will be played to inform that the conference is ongoing;
- o if the new call exceeds <part\_count> amount the call will be dropped with the message "This conference is already full. Please call again later. Good bye.", the host also receives the warning about that;
- o at conference\_schedule\_remind seconds before the conference ending time of the scheduled conference, the host receives announcement "This conference has <MMM> more minutes to complete. Press <KEY> to request an extension." (where <MMM> minutes till the conference ending time, <KEY> keys defined in conference\_schedule\_extend\_dtmf\_binding call flow attribute value that should be used on DTMF keypad by the host if he wishes to extend the conference duration); the host can extend the conference duration, the maximal extension time is defined in conference\_schedule\_extend attribute value;
- o when the scheduled conference time is over the message "This conference has run out of time. All calls will now end." will be played and all this conference calls will be dropped.

While the scheduled conference is in progress you can use *mf* console to extend conference duration using the *conf-schedule-extend* command and to increase number of allowed participants using the *conf-schedule-incsize* command. To extend the started scheduled conference duration you should run the following *mf* console command with two required arguments:

conf-schedule-extend <conf\_number> <seconds>
where

- <conf\_number> The number of the started scheduled conference which duration you would like to extend.
- <seconds> The increment in seconds that should be added to the allowed duration
  for this scheduled conference. Note that the call flow attribute
  conference\_schedule\_extend (Schedule conference extend maximal time in
  seconds) denotes to what maximal time in seconds the duration of the scheduled
  conference can be extended and you are unable to extend duration more than specified
  in this attribute.

To increase the number of allowed scheduled conference participants that was previously defined in conference schedule (Conference schedule) call flow attribute,

part\_count value for the started conference you should run the following *mf* console command with two required arguments:

conf-schedule-incsize <conf\_number> <count>
where

- <conf\_number> The number of the started scheduled conference where you would like to increase the number of the allowed participants.
- <count> The increment in the count of the conference participants that should be added to the number of maximal allowed the scheduled conference participants.

For example if you would like to extend scheduled conference 999020 duration on 600 seconds (i.e. 10 minutes) you should run mf console command:

conf-schedule-extend 999020 600

If you would like to increase the maximal number of allowed scheduled conference 999020 participants on 2 callers you should run *mf* console command:

conf-schedule-incsize 999020 2

Note that the scheduled conference must be started prior to these commands execution.

If the command was implemented successfully you will receive the message: "Success"; if the conference not found or not started you will receive the message: "Error: Conference not found"; if the specified conference is not scheduled you will receive the message: "Schedule doesn't defined for this conference"; if you are trying to extend the conference duration more than allowed by conference\_schedule\_extend call flow attribute you will receive the message: "Error: Conference cannot be extended".

### **Maintaining Conferences Maximal Duration**

The WYDE bridge software allows you to define the maximal duration time for your conferences. The following call flow attributes are responsible for the conference maximal duration definition:

- conference\_maxduration (Maximal conference duration) denotes the maximal duration of the conference; if this attribute value equal –1 (default) that means that there are no duration limit for the conference, this attribute positive value denotes maximal conference duration in seconds;
- conference\_schedule\_extend\_dtmf (Schedule conference extend DTMF policy) denotes who (h/p/l, default: h host) can extend maximal duration of the conference;
- conference\_schedule\_extend\_dtmf\_binding (Schedule conference extend DTMF binding) denotes what keys should be used to extend maximal duration of the conference, when these keys are pressed the following message will be played: "please press 1 to extend conference duration for 5 minutes, press 2 for 10 minutes, press \* to return to the conference", the total maximal extension time is defined in the attribute conference schedule extend;
- conference\_schedule\_extend (Schedule conference extend maximal time in seconds) denotes to what maximal time in seconds the duration of the conference can be extended, i.e. the maximal conference duration could be conference maxduration + conference schedule extend, default this

attribute value is 1800 seconds (30 minutes), the conference can be extended multiple times by 5 or 10 minutes by the callers in the roles defined in conference\_schedule\_extend\_dtmf attribute using DTMF keys defined in conference\_schedule\_extend\_dtmf\_binding attribute till maximum time defined it this attribute;

• conference\_schedule\_remind (Schedule conference end remind time in seconds) – denotes remind time in seconds before the conference ends when the reminder (warning) message should be played to moderator, for instance if this attribute value is 600 before 10 minutes (600 seconds) till the conference ending time the remind message will be played to the host, default this attribute value is 120 seconds (2 minutes).

Note if the conference is scheduled and it has maximal conference duration defined, the actual maximum conference duration is being defined using the scheduler's parameters.

The following conference maximal duration logic is being implemented:

- if conference\_maxduration attribute value is less than 0, that means that there is no duration limit for the conference, i.e. its duration is unlimited;
- if conference\_maxduration attribute value is equal to 0, that means that the conference will be stopped immediately after its start, the message "This conference has exceeded it's time limit. All calls will now end." will be played;
- otherwise, if conference\_maxduration attribute value is positive, this value defines the conference maximal duration in seconds and the following logic will be applied:
  - o at conference\_schedule\_remind seconds before the conference ending time of the conference with maximal duration defined and conference\_schedule\_extend\_dtmf\_binding attribute value not defined, the host receives announcement "This conference has < MMM> more minutes to complete." (where < MMM> minutes till the conference ending time);
  - o at conference\_schedule\_remind seconds before the conference ending time of the conference with maximal duration defined and conference\_schedule\_extend\_dtmf\_binding attribute value also defined, the host receives announcement "This conference has < MMM> more minutes to complete. Press < KEY> to request an extension." (where < MMM> minutes till the conference ending time, < KEY> keys defined in conference\_schedule\_extend\_dtmf\_binding call flow attribute value that should be used on DTMF keypad by the host if he wishes to extend the conference duration); the host can extend the conference duration, the maximal extension time is defined in conference\_schedule\_extend attribute value;
  - o if hosts or callers in the roles defined in conference\_schedule\_extend\_dtmf attribute press the keys defined by conference\_schedule\_extend\_dtmf\_binding attribute they hear the message "Please press 1 to extend conference duration for 5 minutes, press 2 for 10 minutes, press \* to return to the conference." and are able to extent the conference duration using these keys;

- o after the conference duration has been extended the message "Your conference has been extended for <X> minutes." will be played (where <X> either 5 or 10 minutes);
- o if the conference maximal extend time defined by conference\_schedule\_extend attribute is over the message "We are sorry. Your conference cannot be extended." will be played;
- o when the conference maximal duration time is over the message "*This conference* has exceeded it's time limit. All calls will now end." will be played and all this conference calls will be dropped.

While the conference with defined maximal duration is in progress you can use *mf* console to extend conference duration using the *conf-schedule-extend* command as it was described in the previous section of this guide.

#### **Set Call Flow Attributes for Started Conferences**

While a conference is in progress, you may need to change its call flow attributes "on the fly" using the *conf-attributes* command of *mf* console. The attributes are being changed only for the specified started conference; they do not override call flow attributes for a particular subscriber conference definition (conference account), i.e. they are not stored in the WYDE bridge database and do not affect on the next started conference with this number. To set call flow attributes values for the started conference you should run the following *mf* console command:

#### where

- <conf\_number> The number of the conference which attributes you would like to change.
- name1=value1[, name2=value2[,...]] denotes the pairs of the call flow attributes names and values that need to be updated for the started conference; the following call flow attributes could be updated:
  - o the DTMF policy attributes (i.e. the attributes with the suffix dtmf);
  - o MOH configuration attributes (conference\_moh, conference moh options);
  - o the conference properties (conference\_rollcall, conference\_rollcall\_onenter, conference\_rollcall\_onexit, conference\_entryexittones, conference\_entrytones, conference\_exittones, conference\_start\_how, conference\_start\_wait, conference\_stop\_how, conference\_stop\_wait, recording\_stop\_how, recording\_stop\_wait, conference\_callerdb).

The argument <conf\_number> is required. At least one call flow attribute  $name_i = value_i$  pair should be specified. Note that to set the call flow attribute value that contains spaces you should use double quotes (").

For example if you would like for the started conference 667788 to set call flow attribute <code>call\_instructions\_dtmf</code> (Instructions DTMF policy) equal to <code>hpl</code> (host/participant/listener) and to set call flow attribute <code>conference\_moh</code> (Music on hold) equal to your custom music-on-hold file <code>mymoh</code>, you should run <code>mf</code> console command:

If the command was implemented successfully you will receive the message: "Success"; if the conference not found or not started you will receive the message "Error: [ivr] Conference not found"; if you specified call flow attribute that does not exist or cannot be updated or its value is wrong you will receive the message "Error: [ivr] Wrong argument" with the name and the value that could not be updated; if you transferred wrong argument or if you do not specify call flow attribute name and value pair you will receive the message "Error: Wrong argument".

## Conferences and Calls Management using asterisk Console

Calls and conferences can be also managed not for the entire bridge but for specific *asterisk* node only, i.e. the specific computer where *Frontend* components were installed. See section Nodes Administration for the information how to install and configure additional nodes.

If you are using the single bridge installation where all the WYDE software components were installed *asterisk* console commands will return the same conferences and calls information that analogous *mf* console commands return.

## Using the IVR/asterisk Console

To start IVR, i.e. *asterisk* console you should either use the *wyde* command line utility with the *ivr* option:

```
wyde ivr
```

or you can use the following command to run the *asterisk* console:

```
/opt/DNCA/asterisk/usr/sbin/asterisk -rvvv -C /opt/DNCA/asterisk/etc/asterisk/asterisk.conf
```

You will see the screen similar to shown on Figure 38.

Figure 38: Starting asterisk Console

This console is the standard *asterisk* console where you can execute standard *asterisk* commands. But additionally this console has *wyde* commands extension, i.e. additional *wyde* commands can be executed using this console. This section describes *wyde* commands that were added to the *asterisk* console only and does not describe standard *asterisk* commands.

To see the list of all available *asterisk* console *wyde* commands in the console you should type the command:

```
help wyde
```

You will see the screen similar to shown on Figure 39. All *wyde* commands available in *asterisk* console are listed in Table 6, the command reference is given in Chapter 3:

Command Reference, Section: asterisk Console Command Reference. These commands will be described in more detail later in this Guide.

```
ZILBER*CLI> help wyde

wyde drop session drop session

wyde show conferences show active conferences

wyde show conference show conference members

wyde show session queue show session events queue

wyde show sessions show sessions not attached to conference

wyde show statistic show statistic

ZILBER*CLI>
```

Figure 39: asterisk Console, help wyde Command Output

#### Table 6: asterisk Console Utility Available wyde Commands

Commands	Description
wyde drop session	drop session
wyde show conferences	show active conferences
wyde show conference	show conference members
wyde show session queue	show session events queue
wyde show sessions	show sessions not attached to conference
wyde show statistic	show statistic

If you need detail help about any of these commands you should use the command: help command where

• command – the specific asterisk command on which you would like to get help.

If you would like to exit from *asterisk* console you should use the command: quit

Sample *quit* command output is shown on Figure 40.

```
root@ZILBER:~

ZILBER*CLI> quit

Executing last minute cleanups

[root@ZILBER ~] # □

▼
```

Figure 40: asterisk Console quit Command Output Sample

### **View Conferences in Progress**

Any time when there are conferences in progress, you can view them for the specific node using *asterisk* console.

To show a list of all conferences that currently are in process on the bridge node using the command line, you should use *asterisk* console and run the following command from it: wyde show conferences

You will see the screen similar to shown on Figure 41.

```
ZILBER*CLI> help wyde show conferences
usage: wyde show conferences
      display active conferences.
ZILBER*CLI> wyde show conferences
           Members
                     On IVR
Conf
                               On MP
                                          Flags
667788
          2
                     0
                                2
                                         В
2 calls in 1 conferences
Flags: R - recording, B - broadcast, S - secure, L - lecture mode (strict/relaxe
d), T - realTime
ZILBER*CLI>
```

Figure 41: asterisk Console wyde show conferences and its help Commands Output Sample

All active conferences are listed sorted ascending by conference number. There are several columns of information about each conference. Table 3 details what each column indicates. See section View Conferences in Progress for *mf* console for additional information.

# **View Conference Calls in Progress**

Any time when there are conference calls in progress, you can view them in the command line interface using *asterisk* console.

To show a list of all calls that are joined to the specific conference using *asterisk* console you can use the following command:

```
wyde show conference <number>
where
```

• <number> - The number of the conference which calls you would like to view.

For example if you would like to view the calls that were joined to the conference 667788 you should run the command:

```
wyde show conference 667788
```

You will see the screen similar to shown on Figure 42.

```
ZILBER*CLI> help wyde show conference
usage: wyde show conference <number>
      display conference members.
ZILBER*CLI> wyde show conference 667788
SesId
           Mode
                        Location Channel
                                                            MP Channel
16777674
                        MP
                                  SIP/192.168.1.5-09b60808 SIP/192.168.1.5:5160-09b69070
           Host
           Participant
                        MP
                                  SIP/192.168.1.99-09b026c8 SIP/192.168.1.5:5160-09aeeb28
2 active members in conference
Flags: M - muted, Ms - muted self, H - hold, T - realTime, S - silent (noaudio)
ZILBER*CLI>
```

Figure 42: asterisk Console wyde show conference and its help Commands Output Sample

All active conference calls will be shown. There are several columns of information about each conference call. Table 4 details what each column indicates. In addition to *mf* console *show* command output this command returns two extra columns:

- o *Channel* name/address of the channel from asterisk to the user, for instance it could be the IP address where the call came from;
- o MP Channel name/address of the channel from asterisk to the MP.

If you try to show conference calls for the conference that does not started or does not exist the error message "Conference '<number>' not found" will be returned back to you.

## Show Calls that have not placed to Conferences

To show a list of all calls that are not joined to any of the conferences using *asterisk* console you can use the following command:

wyde show sessions

If you run this command you will see the screen similar to shown on Figure 43.

```
ZILBER*CLI> help wyde show sessions
usage: wyde show sessions
display sessions not attached to conferense.

ZILBER*CLI> wyde show sessions
SesId Channel Flags
---- ------
16777677 SIP/192.168.1.99-09b60808
16777676 SIP/192.168.1.5-09b026c8
2 active sessions in pool
Flags: M - muted, H - hold
ZILBER*CLI>
```

Figure 43: asterisk Console wyde show sessions and its help Commands Output Sample

Because these calls are not connected to any of the conferences this view contains only session identifier, channel and flags columns and it does not contains other columns that are shown in *wyde show conference* command output.

### **Dropping Conference Call Participants**

If during a call, you wish to cancel a conference call for specific participants, you may use the *wyde drop session* command of *asterisk* console to kick someone off the call. To drop someone from a conference call you should run the following *asterisk* console command: wyde drop session *<conf\_number> <session\_id>* where

- <conf\_number> The number of the conference which call you wish to drop (disconnect).
- <session\_id> The call session identifier you wish to drop from the conference. Both arguments are required.

For example if you would like to drop the session *16777673* from the conference *667788* you should run *asterisk* console command:

```
wyde drop session 667788 16777673
```

If the conference not found or not started you will receive the message "Conference '<conf\_number>' not found"; if the session not found you will not receive any error message, but the call will not be dropped; otherwise if all parameters are correct the conference call will be dropped and you may receive asterisk messages in the console about the call hang-up processing.

#### **Show Session Events Queue**

Each call has its own events (commands) queue to *Frontend*. For example when the caller tries to execute any DTMF command this command is being placed into command queue. For debugging purposes you may wish to show the session events queue. This can be made using the following *asterisk* console command:

wyde show session queue <conf\_number> <session\_id>
where

- <conf\_number> the call conference number which events queue you would like to show.
- <session\_id> the call session identifier which events queue you would like to
  show.

Both arguments are required.

For example if you would like to show the events queue for the session 16777831 of the conference 667788, you should run asterisk console command:

wyde show session queue 667788 16777831

You will see the screen similar to shown on Figure 44.

```
ZILBER*CLI> help wyde show session queue

usage: wyde show session queue <conf_number> <session_id>
show session events queue.

ZILBER*CLI> wyde show session queue 667788 16777831

COMMAND call_mute^false|force

COMMAND call_mute^relaxed|force

COMMAND call hold^true

ZILBER*CLI> [
```

Figure 44: asterisk Console wyde show session queue and its help Commands Output Sample

#### **Show Node Statistics**

If you would like to see statistics, i.e. number of conferences and number of calls, for specific *asterisk* node it can be made using the following *asterisk* console command: wyde show statistic

You will see the screen similar to shown on Figure 45.

```
ZILBER*CLI> help wyde show statistic
usage: wyde show statistic queue
show statistic about active calls.
ZILBER*CLI> wyde show statistic
sesCount: 3
confCount: 2
ZILBER*CLI>
```

Figure 45: asterisk Console wyde show statistic and its help Commands Output Sample

Note that information returned by this command displays statistics not for entire bridge, but for selected *asterisk* node only.

# Conferences and Calls Management using mp Console

The *mp* console is the tool that could be used to manage *Backend* components (media processer, *mpw* service) of the WYDE bridge. Calls and conferences that are being managed by this console are being controlled for specific *Backend* only.

### Using the mp Console

The mp console is being installed in Backend installation of the WYDE bridge software.

If you are using the single bridge installation where all the WYDE software components were installed to start *mp* console you should run the command:

```
telnet <Your Bridge IP> 4545
or, for example if you are running this command from your bridge computer:
telnet localhost 4545
```

If you use distributed installation to start mp console you should point to the computer where Backend components were installed:

```
telnet <Your Backend Computer IP> 4545
```

If you enter any unknown command the *mp* console returns you the list of all available commands, see Figure 46 for details. Additionally if you try using any *mp* command incorrectly, the console returns the format of the command you are trying to use. All *mp* console commands are listed in Table 7. These commands will be described in more detail later in this Guide.

```
Welcome to WYDE.MPs admin console 2.1.103 compiled May 20 2010

>help
Unknown command: try show, drop, kill, stop, mute, unmute, restart, remove

>
```

Figure 46: Starting mp console, Available Commands

### Table 7: mp Console Utility Available Commands

Commands	Description
show	Show different mp statistics
timer	Start timer running show command
kill	Stop specific timer or all timers
drop	Drop boards and calls
stop	Stop <i>mp</i> components
restart	Restart mp boards and logs

#### Show Different mp Information

The *mp* console *show* command displays information about conferences, calls, boards, and other statistics. The syntax is as follows:

```
show {sw|conf|conf <conf_number>|calls|stat|boards} where one of the following arguments should be specified:
```

• sw – show the number of available MPw boards, i.e. backend components count;

- conf show all conferences currently running on the mp;
- conf <conf number> show statistics for the specific conference only, including active speaker notifications;
- calls [<id1>-<id2>] show all calls currently running on the mp (the command returns calls identifiers, conference numbers, IP addresses, ports, codecs, status, etc.); if you specify the calls sessions identifiers range, only the calls within this range will be returned by the command;
- stat show overall mp statistics about calls, conferences and MPs boards (now column shows current data, total column shows data from mp start, peak column shows acme/height value);
- boards show detail MPs boards statistics, including the board current state, capacity, CPU load, number of conferences and calls, etc.

Different show commands with different options are shown on Figure 47.

```
ov. Telnet 192.168.1.5
                                                                                                   Welcome to WYDE.MPs admin console 2.1.103 compiled May 20 2010
Syntax error: should be show <[sw];[calls];[conf];[stat];[boards]>
>show sw
Boards: 1
>show conf
Conferences: 1
ConferenceID=667788 size=2
ConferenceID=667788 size=2
                         Subconference 0x9849760 size=2;
Total conferences displayed 1
>show conf 667788
Onferences: 1
Conferences: 1
ConferenceID=667788 AS={ 16778009:8 } size=2
Subconference 0x9849760 size=2;
filtering by ConferenceID 667788
Total conferences displayed 1
>show calls
>show stat
Started: Thu Jun 3 18:29:06 2010
Call: Now=2; Total=43; Peak=3; Last=Wed Jun
Conf: Now=1; Total=22; Peak=1; Last=Wed Jun
Brds: Now=1
                                                          9 16:54:16 2010
9 16:54:16 2010
>show boards
rshow moards
Boards: 1
Serial=127.0.0.1:49301, ON, Capacity=4%, CPU=16%, Calls=2, Conferences=1, garbag
e=0:0
          processed=308, Cores=1, started=Wed Jun 9 16:54:15 <u>2010</u>
```

Figure 47: mp Console show Commands Output Sample

#### Start and Stop mp Console Timers

If it is necessary you can run any *mp* console *show* command automatically in a given interval using *timer* command. The syntax is as follows:

```
timer <interval> show <options>
where
```

- <interval> denotes interval in seconds in which the *show* command should be rerun:
- show <options> denotes specific *show* command with its options that should be repeated in the given *interval*.

These commands will be repeated until the timer will be stopped. The *mp* console *kill* command should be used to stop the specific timer of all started timers. The syntax is as follows:

```
kill \{\langle id \rangle \ [\langle id \rangle] \ \dots | all \} where
```

- <id> the specific timer identifier that should be stopped;
- all denotes that all started timers should be stopped.

For example if you would like to show conference 667788 status each 5 seconds you should run the command:

```
timer 5 show conf 667788
```

To stop this timer you could use the command:

kill all

See Figure 48 for details.

Figure 48: mp Console timer and kill Commands Output Sample

# **Dropping Boards and Calls**

To drop *MPw* boards and calls from *mp* you can use *mp* console *drop* command. The syntax is as follows:

drop {<boards <serial> [<serial>] ...>|<calls <<id> [<id>] ...>| all>} where one of the following arguments should be specified:

- - ✓ this option works as toggle switch if you repeat this command the board will be switched on;
- <calls <<id> [<id>] ...>|all> denotes that the calls should be dropped from mp; use <id> to specify the call sessions identifiers that should be dropped or use all keyword to drop all calls.

For example if you would like to drop the call 16777862 from mp you show run the command:

```
drop calls 16777862

To drop all calls from mp use the command: drop calls all
```

You will see the screen similar to shown on Figure 49.

Figure 49: mp Console drop Commands Output Sample

# Restarting and Stopping mp Boards and Logs

If it is necessary to restart MPw boards or mp logs you can use mp console restart command. The syntax is as follows:

```
restart {<log [<level>]>|<board <serial>|all>}
where
```

- <log [<level>]> denotes that the log should be restarted; <level> specifies what messages should be stored in the log: 3 denotes that only information and error messages should be stored in the log file (default); 9 denotes that debugging mode is switched on and all possible log messages should be added to the log file.
- <board <serial>|all> denotes the MPw boards that should be restarted; use
   all keyword to restart all MPw boards.

For example if you would like to restart the log and include into the log file only information and error messages you show run the following command from mp console: restart log 3

For example if you would like to restart all MPw boards you should run mp console command:

restart board all

You will see the screen similar to shown on Figure 50.

```
Welcome to WYDE.MPs admin console 2.1.59 compiled Apr 2 2010
>restart
Syntax error: please specify what to restart <log> or <br/>
>restart log 3
OK.
>restart board all
Restarting board<s> 127.0.0.1:58714-LOCALHOST
>
```

Figure 50: mp Console restart Commands Output Sample

To stop *mp* processes (*mpw* and *mps* services) you can use *mp* console *stop* command, but this service is being restarted automatically because it is need for the WYDE bridge activity. The syntax is as follows:

stop

# **Operator Conferences**

Operator conferences and call flows are designed to administrate the conferences and assist the users in case of any problems. In other words operator conferences can be used as quick technical support for users. The operator can monitor the conferences, connect to the different conferences, receive requests from the users, attach the users to the different conferences, dialout, etc. When operator connects to his OPERATOR conference he is being notified when new requests are assigned to his queue or he can initiate the conference surveillance and management.

To act as the operator the voice phone call should be connected to the operator conference. There are two possibilities to connect to the operator conference:

- the operator can call to the operator conference, i.e. to call to the OPERATOR conference DNIS number and enter his access code to connect to the conference:
- the WYDE bridge can be configured to call to the operator automatically in case if specific conditions occurred:
  - o there are no operators connected to the bridge but the caller from operators' queue is expecting operator assistance;
  - waiting period for the caller from the operators' queue exceeds specific period of time;

in both cases when the call is made to the operator phone number the operator is being asked to enter his access code; only if the access code is valid, the caller is being joined to his operator conference.

Note that these conditions can be specified not only for entire WYDE bridge, but also for DNIS number or its mask.

See section Operator Invite Settings Configuration for information how to configure the WYDE bridge to call to the operator phone number when it is necessary.

During the conference any call participants can request a call to an operator. To do that they should press \*0 (default, can be overridden); after that the users will be asked to press:

- 1 to speak to an operator
- 2 to pause the request and return to the conference

If the user presses 1 button he will be prompted: "*Please wait and you will connect to an operator*". After that the user will be placed to the operator queue.

Any time while the user is waiting the operator response he can press \*0 again and he will be asked to press:

- 1 -to cancel the request
- 2 to return to the conference

In addition some call flows can provide that if the user is entering access code three times incorrectly (default, can be overridden) this user is being placed to an operator queue automatically. The operator can assist to such user connect to the requested conference.

Operator conference management can be implemented using *mf* console. In addition operator conferences can be managed using DTMF keypad on your phone and the Web Administration Interface as it is described in "Web Administration Interface – User Guide".

# Operator Conferences Management using mf Console

To show started operator conferences, i.e. the operator conferences that are in progress, you should run the following *mf* console command:

```
op-show [<operator_number>]
where
```

- <operator\_number> The operator conference number that you would like to show.
  - If this argument is omitted the brief information about all started operator conferences will be displayed; this information includes operator conference number, used DNIS (DID) number and current operator status (Free, Talk, Listen).
  - o If this argument is specified the detail information about specific operator conference will be shown; this information includes data about this operator conference, such as current operator status (the same as above), flags, and information about users who are talking to the operator: their source conference number, DNIS, session identifier. Operator conference flag is two-digit number MN, where M and N is either 0 or 1; M=1 means that the operator is currently connected to another conference, N=1 means that the operator is currently scanning and listening other conferences. For example status=Talk, flags=01 means that the operator is currently talking to a user and he has activated conference surveillance, when he finishes the conversation the conference monitoring will be automatically resumed.

To show the operator queue you should run the following *mf* console command: op-queue

This command displays all users who are currently waiting the operator assistance in operator queue. This information includes source conference number (if applicable), DNIS and session identifier; if there are no any users in the queue the message "there are 0 calls in the queue" will be returned.

For example to show all started operator conferences you should run *mf* console command: op-show

To show detail information about operator conference 390008 you should run mf console command:

op-show 390008

To show the operator queue you should run mf console command: op-queue

Figure 51 displays both these commands samples when the operator queue is empty, non-empty, and when the operator is talking to the user from another conference.

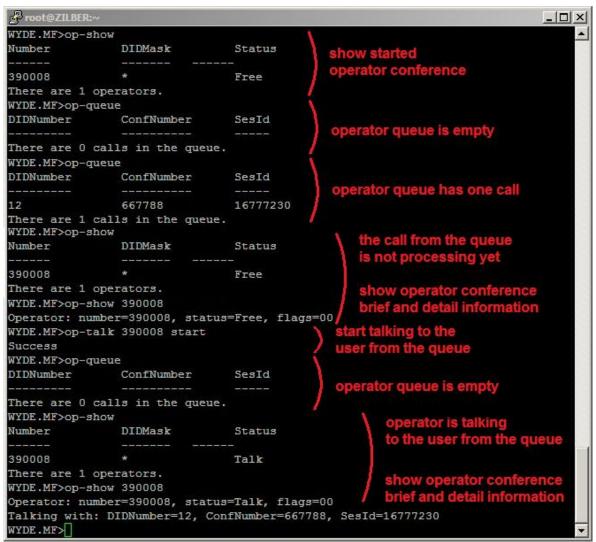


Figure 51: mf Console op-show and op-queue Commands Output Sample

To start or stop conference monitoring (scan), i.e. surveillance call, you should run the following *mf* console command:

```
op-scan <operator_number> {start|stop}
where
```

- <operator\_number> The operator conference number where you would like to
  start or stop conference monitoring (surveillance call).
- {start|stop} start denotes that the monitoring should be started; stop denotes that the monitoring should be stopped.

All arguments are required. When monitoring is started the operator is being switched between all started conferences each 30 seconds until he stops the monitoring or connects to another specific conference or talking to the user from his queue.

To connect and talk to the user, i.e. either to receive the call from the user from the operator queue or to talk to the specific user from any active conference, you should run the following *mf* console command:

#### where

- <operator\_number> The operator conference number where you would like to
  start or stop talking to a user.
- {start|stop} start denotes that the talking should be started; stop denotes that the talking should be stopped. If together with start option the parameters <conf\_number> <ses\_id> omitted the operator starts talking to the first user from his queue, if these parameters are presented the operator starts talking to the user specified by these parameters:
  - o <conf\_number> the number of the conference which caller should be taken
    to talk with the operator (0 for IVR calls);
  - <ses\_id> the session identifier of the caller that should be taken to talk with
     the operator.

All arguments are required. When the talking is stopped the user who was talking to the operator is being returned to his conference or IVR.

To reject the user from the operator queue, i.e. to refuse the conversion with the user and remove the user from the operator queue without starting the conversation with him, you should run the following *mf* console command:

```
op-reject <conf_number> <ses_id>
Arguments:
```

- <conf\_number> the number of the conference which caller should be rejected
  from the operator queue, i.e. it is the conference where the caller came from (0 for
  IVR calls).
- <ses\_id> the session identifier of the caller that should be rejected from the
  operator queue, i.e. the conversion with this user will be refused and the user will be
  removed from the operator queue.

All arguments are required. When the user from the operator is rejected from the operator queue, he is being returned to his conference or IVR.

To connect and listen another conference you should run the following *mf* console command:

```
op-listen <operator_number> {start <conf_number>
      [{directlink|shunt} [mute]]|stop}
where
```

- <operator\_number> the operator conference number that would like to listen
  other users conferences. This argument is required.
- {start|stop} start denotes that the listening should be started; stop denotes that the listening should be stopped. This argument is required. If start is specified the following arguments could be specified here:
  - o <conf\_number> the number of the conference you wish to start listening, must be indicated if start option is specified;
  - o {directlink|shunt} directlink denotes that the operator should be directly connected to the requested conference (in this case only operator can

hear the requested conference, the user connected to the operator can not hear that conference), <code>shunt</code> denotes that between operator conference and requested conference is being made the shunt and both conferences can hear each other (operator and the connected user both can hear the requested conference), if this argument is omitted the <code>directlink</code> mode is being used by default;

o mute – when specified denotes that the operator is muted, so he can only hear the specified conference and he is unable to talk.

Note that op-show command display information about connected conference, i.e. the connection mode (*directlink* or *shunt*) and the number of the conference you are listening.

To attach the user who currently is talking to the operator to a different conference, i.e. move user to a conference, you should run the following *mf* console command:

- <operator\_number> The operator conference number whose current call (i.e. the
  user that currently talking to the operator) you would like to move to another
  conference.
- <new\_did\_number> New (i.e. target) conference DNIS (DID) number where the call should be moved.
- <new\_accesscode> The access code that should be used to join to new (i.e. target) conference.
- <new\_role> The role that will be granted to the call when it joins to new conference (applicable only for call flows without authorization, for example *CONF* call flow).

The argument <new\_role> is optional and should be used for call flows without authorization only. All other arguments are required.

To initiate dial-out from the operator conference you should run the following *mf* console command:

```
op-dialout <operator_number> <peer_number>
where
```

- <operator\_number> The operator conference number where you are initiating
  the dialout.
- <peer number> Denotes the phone number you wish to dial.

All arguments are required. When dial-out is performed the user that was dialed is being connected to the operator conference. After that the operator is able to talk to the connected user, attach him to another conference, etc.

To manage the operator message from the operator conference, i.e. to initiate recording and playing back of the operator message or to clear (i.e. switches off) the operator message you should run the following *mf* console command:

```
op-message <operator_number> {start {queue|all} [<message>]|
    stop}
where
```

- <operator\_number> The operator conference number where the message should
  be recorded or listened.
- {start|stop} Determines what action should be performed with the message:
  - o start denotes that the operator message should be recorded, activated and sent either to users from the operator's queue (when queue parameter is specified) or to all users on the bridge (when all parameter is specified); optional message parameter specifies prerecorded audio file that will be played to the callers, if the parameter is omitted the operator will be asked to leave his message if there is no voice call connected to the conference with the specified number the command returns the error "Error: Conference not found", if the operator voice call is connected to the operator conference and message parameter is omitted the operator will be asked to leave his message after the tone and press # when it is done, after the message has been recorded (and # was pressed) this message will be played back to the operator and he will have the options either press 1 to send the recorded message or press \* to cancel (this command works this way regardless there is or there is no previously recorded operator message);
  - o stop denotes that the operator message should be cleared, i.e. playing this message should be stopped.

The <message> argument is optional, all other arguments are required.

To initiate recording and playing back the operator message from the operator conference to the user's conference hosts, participants, or listeners you should run the following *mf* console command:

```
op-message-to-conf <operator_number> <conf_number>
{host|participant|listener} [<message>]
```

#### where

- <operator\_number> The operator conference number where the message should be recorded. This argument is required.
- <conf\_number> The number of the user's conference where the operator message should be played. If 0 is specified as the target conference number the message will be recorded and played to the conference to which the operator is currently connected or to whose user the operator is currently talking. This argument is required.
- {host|participant|listener} The callers' role to which the message should be played. If the message should be played to multiply roles, all these roles could be specified separated by colon (:). This argument is required.
- <message> The prerecorded audio file that will be played to the callers, if the parameter is omitted the operator will be asked to leave his message after the tone and press # when it is done, after the message has been recorded (and # was pressed) this message will be played back to the operator and he will have the options either press 1 to send the recorded message to the conference or press \* to cancel. This argument is optional.

If the operator conference not found or not started or if there is no voice call connected to the conference with the specified number the command returns the error "*Error*: *Conference not found*"; if the operator is not connected to the user's conference, but the

second parameter is omitted, you receive the message "*Error*: *There is not connected conference*"; otherwise if the operator voice call is connected to the operator conference and the target conference is established you receive the message "*Success*". Next if the *message* parameter is specified the message from the file will be played to the conference; if it is omitted the operator will be asked to leave his message.

Let's assume that the operator conference 390008 is stated; another conference that we use to connect to is 667788, its DNIS is 8665080012, its participant access code is 6602; phone number to dial-out 6046880331. Table 8 shows samples what commands could be implemented to assist users via this operator conference using mf console commands.

Table 8: Operator Conference Management Samples Using mf Console Commands

Table 6. Operator Conference Manag	chieft Samples Using my Console Commands	
Description	mf Console Command	<b>DTMF</b>
Start conference monitor (surveillance	op-scan 390008 start	*1
call)		
Stop conference monitor (surveillance	op-scan <i>390008 stop</i>	*1 2
call)		
Connect and talk to the next user from	op-talk 390008 start	*2
the operator queue		
Stop talking to the current user and	op-talk <i>390008 stop</i>	*3
return him to his conference or IVR		
Reject the user from the operator	op-reject <i>667788 16783926</i>	
queue and return him to his conference		
Connect and listen another conference	op-listen <i>390008 start 667788</i>	*4 2
without current user		
(default directlink mode used)		
Connect and listen another conference	op-listen <i>390008 start 667788</i>	*4 2
without current user and muting	directlink mute	
himself		
Connect and listen another conference	op-listen 390008 start 667788 shunt	*4 1
with current user		
(shunt mode used)		
Stop listening another conference	op-listen 390008 stop	*3
Attach current user to a different	op-call-move 390008 8665080012 6602	*5
conference		
Dialing out to another user	op-dialout <i>390008 6046880331</i>	*7
Initiate recording and activation of the	op-message 390008 start all	*8
operator message		
Initiate recording and playing back the	op-message-to-conf 390008 667788	* 9
operator message to specific	participant:listener	
conference		

### **Samples and Use-Cases of Operator Conference Procedures**

This section of the document will describe few samples about how the operator can assist users in their conference calls.

First of all the operator can monitor (hear) the conference to examine if there are any problems where he can assist.

Also for some call flows the operator will receive automatic requests from users who entered incorrect access codes three times (default, can be overridden). In this case the operator asks the user information about who he is and about the conference he wish to connect, connects to the requested conference, asks if this conference expecting this user, and if so, returns and attaches the user to this conference.

Moreover the users can request operator assistance in case of any problems, for instance bad audio quality or noises. In this case operator can connect to the user conference and use Web Administration Interface – Active Speaker notification mode (see "Web Administration Interface – User Guide", section: Active Speaker Notification (ASN) mode) to examine where is the problem and after that inform the user about the reason of the problems.

In addition the operator can dialout to another user and attach the user to the conference. If call flows do not allow making dialout within the conference the operator assistance is needed.

# WYDE Bridge Administration

In addition to the daily conferences and calls administration, subscriber user management, call flow and DNIS management, you will have to administer the WYDE bridge system itself. These system management tasks could be the following:

- bridge monitoring;
- bridge settings management, including their save and restore;
- nodes administration;
- distributed conferencing administration;
- peers management;
- calls transferring management;
- audio prompts management;
- other WYDE bridge administration tasks.



Note that you should be very careful with your bridge administration. It is possible to render your system inoperable if you are not familiar with the WYDE bridge administration. Changing any of the bridge preferences, unless instructed to do so by WYDE technical support, also can render your system inoperable.

# **Monitoring**

The WYDE bridge exposes metrics to allow the monitoring system to measure health and activity on it. Some of these metrics are available via *mf* console commands; some of them are available via web.

To show the number of commands executed on bridge from *mf* starts till now you should run the following *mf* console command:

cmdcount-show

The counters that are being returned by this command are described in cmdcount-show (Display Values of *Command* Counters). You can also use URL:

https://< Wyde bridge domain > /status/cmdcount to expose these counters via web.

To show the break-up of conferences according to the participant size at the moment you should run the following *mf* console command:

confcount-show

The counters that are being returned by this command are described in confcountshow (Display Values of *confcount* Counters). You can also use URL:

https://<Wyde bridge domain>/status/confcount to expose these counters via web.

To show the counters related to the participants at the moment you should run the following *mf* console command:

partcount-show

The counters that are being returned by this command are described in partcountshow (Display Values of *partcount* Counters). You can also use URL:

https://<Wyde bridge domain>/status/partcount to expose these counters via web.

To show the counters related to the operators activity at the moment you should run the following *mf* console command:

```
opcount-show
```

The counters that are being returned by this command are described in opcount-show (Display Operator's Counters). You can also use URL:

https://< Wyde bridge domain > /status/opcount to expose these counters via web.

To show the number of terminated/incomplete calls from *mf* starts till now you should run the following *mf* console command:

```
errcount-show
```

The counters that are being returned by this command are described in errcount-show (Display Values of *Error* Counters). You can also use URL:

https://<Wyde bridge domain>/status/errors to expose these counters via web.

In addition you can use web to show the duration of the conferences that have ended in the last 2 minutes (*https://<Wyde bridge domain>/status/confduration*) and to show the duration of the participants calls that have ended in the last 2 minutes (*https://<Wyde bridge domain>/status/partduration*).

Conference numbers for call flows without authorization (like CONF) are being issued automatically, once the conference begins. To show these conference numbers together with phone number called, access code used, and valid time, you should run the following *mf* console command:

```
freenumbers-show
```

You will see the screen similar to shown on Figure 52. *Key* column contains *<DNIS* number>\_<access code>; *ValidTime* column contains the conference number expiration time in seconds till the end of conference number validity – the calls with the same DNIS number/access code pairs till end of the conference number valid time will get the same conference number.

Figure 52: mf Console freenumbers-show Command Output Sample

If you would like to see all activities (actions) that take place in the conferences and calls you can use the following *mf* console command:

```
activities-show {start|stop}
where
```

• {start|stop} - start indicates that the conferences calls activities (actions) should be outputted to the *mf* console, stop indicates that the activities output should be stopped. This argument is required.

Thus when you wish to start activities output you should run the command:

activities-show start

and to stop this output you should run the command:

activities-show stop

You will see the screen similar to shown on Figure 53.

```
WYDE.MF>activities-show start
Success
WYDE.MF>Action: 16777358 0 0 DTMF 12345#|1
Action: 16777358 134217793 4295608717 CONF_CREATE
Action: 16777358 134217793 4295608717 SES_JOIN Participant
Action: 16777360 134217793 4295608717 SES_DROP Normal
Action: 16777360 134217793 4295608717 SES_DROP Normal
Action: 16777358 134217793 4295608717 SES_DROP Normal
Action: 16777358 134217793 4295608717 CONF_DROP
activities-show stop
Success
WYDE.MF>
```

Figure 53: mf Console activities-show Command Output Sample

WYDE bridge logs *mf* activity into /usr/local/DNCA/log/mf.log file. To set the logger level you should run the following *mf* console command:

```
set-log-level {debug|event|info}
where
```

• info – denotes that only information and error messages should be stored in the log file (default); event – denotes that additionally the log should contain all events logging; debug – denotes that debugging mode is switched on, so additionally debug messages should be added to the log. This argument is required.

You can use the information from this *mf.log* file in case of any problems or if you need to monitor WYDE bridge activity.

To show the *asterisk* process status using the command line interface you should use the *wyde* command line utility with the *ast-status* option. The syntax is as follows: wyde ast-status

You will see the screen similar to shown on Figure 54.



Figure 54: wyde ast-status Command Output Sample

To show the WYDE bridge status using the command line interface you should use the *wyde* command line utility with the *status* option. The syntax is as follows: wyde status

You will see the screen similar to shown on Figure 55.

```
root@ZILBER:∼
[root@ZILBER ~] # wyde status
System load:
cpu idle
                    = 77.97
mem swpd
                    = 288
mem free
                    = 23912
mem buff
                    = 67200
mem cache
                     = 378884
Calls:
calls
conferences
                     = 0
                     = 0
avg conf size
                     = 0
max conf size
                     = 0
min conf size
Errors:
total_terminated = 18
sip terminated = 1
max calls terminated = 0
agi_terminated = 0
mp_terminated = 7
asterisk terminated = 1
max duration terminated = 9
[root@ZILBER ~]#
```

Figure 55: wyde status Command Output Sample

Additionally you can watch the same status permanently and refresh it automatically with the given interval. To do so you should use the *wyde* command line utility with the *watch* option. The syntax is as follows:

```
wyde watch [interval <interval>]
```

• <interval> - The interval in seconds that denotes the period of how often the bridge status should be refreshed. This argument is optional; if it is omitted the 5 seconds interval is used by default.

This command output is similar to *wyde status* command output (see Figure 55); the only difference is that this screen is being refreshed automatically with the given period. To interrupt and exit from this command you should use CTRL+C keystroke. Note that each command execution takes 1-2 seconds, so the real interval is greater than you define.

# **WYDE Bridge Settings Management**

If it is necessary you can change your WYDE bridge settings you can use the command line interface. To set WYDE bridge parameter value using the command line interface you should use the *wyde* command line utility with the *settings-edit* option. The syntax is as follows:

```
wyde settings-edit <arguments>
```

Each of the arguments is followed by a space and a value. In *settings-edit* you can specify the following arguments:

- bridge <value> The name of the bridge which settings you would like to edit.
- name <value> The parameter name that should be edited.
- value <value> New parameter value that should be set.

The argument name is required, all other arguments are optional. If argument bridge is omitted the settings will be edited for the current bridge; the default bridge is defined in /usr/local/DNCA/etc/dnca.conf file:

```
[general]
bridge id = 1
```

If the argument value is omitted the empty value will be set to the parameter. The arguments can be transferred to this command in any order.

For example if you would like to change  $mp\_sip\_addr$  parameter value to 192.168.1.32:5160 (set new IP address) for the current bridge, you should run the following command (the transferred command arguments are shown in *italic*): wyde settings-edit name  $mp\_sip\_addr$  value 192.168.1.32:5160

If the command is successful, the system will return updated parameter name and new value that was set; after that it will return you back to the command prompt (#). The sample of the *settings-edit* command output and the help on this command is shown on Figure 56.

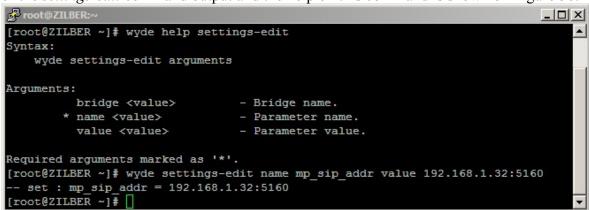


Figure 56: wyde help settings-edit and wyde settings-edit Commands Output Sample

If any of WYDE bridge parameters has been changed you should reload them from the database, from <code>bridge\_settings</code> table. To send the signal on the WYDE bridge <code>MF</code> engine to reload all system settings the following <code>mf</code> console command should be executed: <code>settings-reload</code>

or you should use the *wyde* command line utility with the *settings-reload* option: wyde settings-reload

If the command is successful, the system will not return any errors or messages.

If new system parameter was added it should be added into <code>bridge\_settings</code> table and populated with the default values. Usually you do not need doing it, because it is being run in the WYDE software upgrade installation automatically. However you can update <code>bridge\_settings</code> table in the database using the <code>wyde</code> command line utility with the <code>settings-update</code> option. The syntax is as follows:

```
wyde settings-update <arguments>
```

Each of the arguments is followed by a space and a value. In *settings-update* you can specify the following argument only:

• bridge <value> - The name of the bridge which settings should be updated. This argument is optional; if it is omitted the default bridge is being taken from /usr/local/DNCA/etc/dnca.conf file as it was previously described in this section. This command adds missing bridge parameters populated with default values as well as removes obsolete bridge parameters for the bridge.

If this command is successful, the system will return the message "*Update settings for* <*bridge name*>"; after that it will return you back to the command prompt (#).

To show a list of all or specific WYDE settings, i.e. system parameters and their values, using the command line, you should use the *wyde* command line utility with the *settings-show* option. The syntax is as follows:

```
wyde settings-show <arguments>
```

Each of the arguments is followed by a space and a value. In *settings-show* you can specify the following arguments:

- bridge <value> The name of the bridge which settings you would like to show.
- prefix <value> The prefix of the parameters names that should be shown.

All arguments are optional. If argument <code>bridge</code> is omitted the settings will be shown for the current bridge; the default bridge is being taken from <code>/usr/local/DNCA/etc/dnca.conf</code> file as it was previously described in this section. If argument <code>prefix</code> is omitted all bridge parameters will be shown; if this argument is specified the command shows only parameters that starts with the specified prefix.

For example if you would like to see all current WYDE bridge settings you should use the command:

```
wyde settings-show
```

This command outputs all system parameters for the current bridge, see Figure 57 for details. As you can see, the *wyde settings-show* command shows the bridge name, parameters names and their current values.

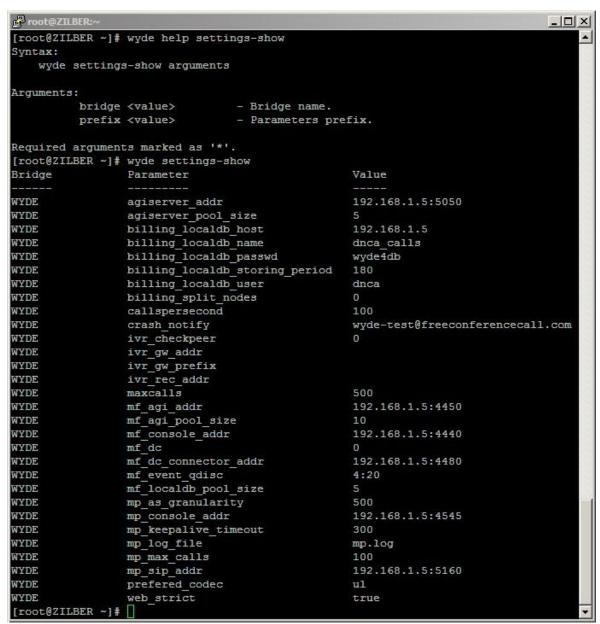


Figure 57: wyde help settings-show and wyde settings-show Commands Output Sample

In order to effectively manage WYDE bridge settings, you may need to have a common knowledge about all configuration parameters; as we mentioned you can see them using the *wyde* command line utility with the *settings-show* option. Below we provide the brief description of each settings parameter that you can control using the *wyde* command line utility with the *settings-edit* option:

• **agiserver\_addr** – denotes the IP address and port on which *agiserver* should listen for requests. The parameter format is <*IP*>:<*port*>; default <*IP*> is IP address of the computer where WYDE software core components are installed and default <*port*> is 5050, for example 192.168.1.30:5050. This parameter IP address is being updated by the *wyde* command line utility with the *set-ip* option.

- - o <static pool size> denotes the static AGI server pool size;
  - ✓ if agiserver\_pool\_size parameter does not have the colon (:) character the static pool is being created, otherwise (if the parameter has colon character) the dynamic pool is being created;
  - o <min\_servers> denotes the minimum number of servers to keep running;
     default value: 5;
  - o <max\_servers> denotes the maximum number of child servers to start; this does not apply to dequeue processes; default value: 50;
  - o <min\_spare\_servers> denotes the minimum number of servers to have waiting for requests; minimum and maximum numbers should not be set to close to each other or the server will fork and kill children too often; default value: 2;
  - o <max\_spare\_servers> denotes the maximum number of servers to have waiting for requests; also see <min\_spare\_servers> above for additional info; default value: 10. For detailed information see section AGI Server Pool Configuration for Processes Handlers Requests in this Guide.
- **billing\_localdb\_driver** denotes the billing database driver, the default value is *pgsql* PostgreSQL, for other possible drivers (for example MySQL or Oracle) please contact WYDE Voice technical support. See detailed parameter description in "WYDE Billing Guide".
- **billing\_localdb\_host** denotes the server (its IP address) where the local billing database is being placed. See detailed parameter description in "WYDE Billing Guide".
- **billing\_localdb\_name** denotes name of the local billing database, the default value is *dnca\_calls*. See detailed parameter description in "WYDE Billing Guide".
- **billing\_localdb\_passwd** denotes the password of the user that is used to access the local billing database. See detailed parameter description in "WYDE Billing Guide".
- **billing\_localdb\_storing\_period** denotes the period in days of how long the billing data are being stored in the local billing database, by default the database contains CDR data for the last 180 days. See detailed parameter description in "WYDE Billing Guide".
- **billing\_localdb\_user** denotes the name of the user that is used to access the local billing database. See detailed parameter description in "WYDE Billing Guide".
- **callspersecond** denotes the maximal number of new calls that could be processed by WYDE software frontend components in one second, default: 100 calls. If during 100ms frontend receives more than callspersecond/10 new calls (SIP INVITE), all calls that exceed this limit will be ignored (i.e. the frontend will not respond on SIP INVITE of these calls).
- **conferencePN** denotes user conference DNIS number (call flow: SPECTEL), default: 12. See detailed parameter description in "WYDE Software Installation Guide".

- **crash\_notify** denotes notification email address for the bridge. For detailed information see section *Bridge Configuration Changes* in this Guide.
- ivr\_checkpeer if you would like to configure the WYDE bridge to accept not all calls, but only the calls from specific IP address you should set this ivr\_checkpeer parameter value equal to 1 (default value is 0, no peers limitation applied) and populate peers table with allowed IP address together with other information. For detailed information see section Peers Management in this Guide.
- ivr\_gw\_addr denotes the gateway address (IP) where INVITE should be sent during the dial-out, if you are making dial-out to the real phone number the specified gateway should have PSTN output, if you are making dial-out to the subscribers VoIP phone number this could be the address of your Kamailio (OpenSER) server. For detailed information see section *Dialout Settings Configuration* in this Guide.
- ivr\_gw\_prefix denotes the prefix that should be added to the beginning of the number during the dial-out or the template for replace. For detailed information see section *Dialout Settings Configuration* in this Guide.
- ivr\_rec\_addr denotes the IP address of the external recording server. For detailed information see section *Recording Settings Configuration* in this Guide.
- mailerFormsTo denotes to-address where service emails should be sent, default: support@wydevoice.com. See detailed parameter description in "WYDE Software Installation Guide".
- mailerFrom denotes your mailer from (reply) address, for example no-reply@localhost.localdomain. See detailed parameter description in "WYDE Software Installation Guide".
- mailerHost denotes your SMTP outgoing mail server address, e.g. usually your host name or IP address, for example *localhost.localdomain*. See detailed parameter description in "WYDE Software Installation Guide".
- mailerSupportmoduleImgUrl denotes the URL to web folder that contains image-files that could be used in support emails, for example <a href="https://192.168.1.30/supportmodule">https://192.168.1.30/supportmodule</a>. See detailed parameter description in "WYDE Software Installation Guide".
- maxcalls denotes the total number of all calls that can be processed by the WYDE bridge. For detailed information see section *Max Number of Calls and Licensing* in this Guide.
- mf\_agi\_addr denotes the IP address and port on which mf service should listen for AGI connections from WYDE software frontend components. The parameter format is <IP>:report>; default <IP> is IP address of the computer where WYDE software core components are installed and default report> is 4450, for example 192.168.1.30:4450.
  This parameter IP address is being updated by the wyde command line utility with the set-ip option.
- **mf\_agi\_pool\_size** denotes the AGI connections pool size for the connections to the single computer with WYDE software frontend components, default: 10.
- **mf\_console\_addr** denotes the IP address and port on which *mf* service should listen for *mf* console connections. The parameter format is <*IP*>:<*port*>; default <*IP*> is IP address of the computer where WYDE software core components are installed and

- default <*port*> is 4440, for example 192.168.1.30:4440. This parameter IP address is being updated by the *wyde* command line utility with the *set-ip* option.
- **mf\_dc** 1 denotes the bridge is participating in distributes conferencing (should be set 1 on all DC bridges). For detailed information see section *Distributed Conferencing Administration* in this Guide.
- mf\_dc\_billing\_rule the billing rule that should be used to process replicated CDR records received from another distributed bridges (any DC bridge that receives replicated CDRs can have its own billing rule to process these received CDR records). For detailed information see section *Distributed Conferencing Administration* in this Guide.
- mf\_dc\_cdr\_replication 1 denotes that the CDR records from the distributed conferences (where call flow attribute *conference\_distributed=on*) should be replicated to all bridges participating in distributes conferencing, even if the conference took place on the single bridge only (should be set 1 on all DC bridges that participate in CDR replication). For detailed information see section *Distributed Conferencing Administration* in this Guide.
- mf\_dc\_connector\_addr denotes the IP address and the port for binding between MF and ASD; it is being used to establish connection between bridges in distributed conferences; if MF sees that the conference is distributed, it sends to this IP address and port the ASD commands that establish SIP connection between distributed conferences bridges. For detailed information see section Distributed Conferencing Administration in this Guide.
- **mf\_dc\_weight** the *weight* of the DC bridge is being calculated as number of links with another bridges plus the value of this *mf\_dc\_weight* system parameter. For detailed information see section *Distributed Conferencing Administration* in this Guide.
- **mf\_did\_adapter** denotes the comma-delimited list of names (or the single name) of the DNIS (DID) adapters that are used to connect to the DNISes database; if the DNIS not found for the first adapter in the list, the next adapter from the list is being used to search the DNIS, etc.; default: *localdb*, (another sample: *localdb,fcc*).
- mf\_event\_qdisc denotes parameters for the commands queue in conferences. To limit the number of commands in the conferences that could be executed in group transactions the WYDE software uses the modified *Token Bucket* algorithm<sup>1</sup>. The parameter format is <rate>:<burst>; where <rate> is number of tickets that could be added to the conference bucket in 10ms, <burst> is the bucket capacity, default parameter value: 4:20.
- mf\_localdb\_pool\_size denotes the local database connections pool size used in SQL queries processing, default: 5.
- mp\_as\_granularity denotes the period in milliseconds of how often the active speaker notifications are being transmitted, default: 500ms.
- **mp\_console\_addr** denotes the IP address and port on which *mp* service should listen for *mp* console connections. The parameter format is <*IP*>:<*port*>; default <*IP*> is IP address of the computer where WYDE software core components are installed and

<sup>&</sup>lt;sup>1</sup> You can view the *token bucket* algorithm description using URL: <a href="http://en.wikipedia.org/wiki/Token\_bucket">http://en.wikipedia.org/wiki/Token\_bucket</a>

default < port> is 4445, for example 192.168.1.30:4445. This parameter IP address is being updated by the wyde command line utility with the set-ip option.

- **mp\_keepalive\_timeout** denotes inactivity timeout in seconds for the calls between the frontend and *mp*, default: 300s.
- mp log file denotes mp log file name, default file name is mp.log.
- mp\_max\_calls denotes the maximal number of the calls (ports) that can be processed by the *mp* services. The parameter value depends on the license that you have, default value without the license is 10.
- mp\_sip\_addr denotes the IP address and port on which mp services should accept (receive) SIP calls. The parameter format is <IP>:<port>; default <IP> is IP address of the computer where WYDE software core components are installed and default <port> is 5160, for example 192.168.1.30:5160. This parameter IP address is being updated by the wyde command line utility with the set-ip option.
- **operator\_invite** denotes when, to what number, from what DNIS the invite call should be made from the bridge to the operator as well as some other options; the parameter has the following format:

```
<phone number>:<operator_did>
   [:<waiting_time>[:<dial_timeout>[:<invite_interval>
    [:<waiting_time_max>:<on_wait_time_max_direct_number>]]]]
where
```

- o <phone number> denotes operator's phone number to dialout, it also could be operator business telephone system / PBX phone number;
- o coperator did> denotes DNIS number of the operator conference;
- o <waiting\_time> denotes the waiting period in seconds how long the caller from the operators' queue will wait until the system performs dialout to the operator; if the parameter not defined the system starts dialout immediately once the user arises in the operator queue;
- o <dial\_timeout> denotes the operator response waiting period in seconds, i.e. the operator dialout call answering period; if the parameter not defined the default dialout timeout is 45 seconds;
- o <invite\_interval> denotes the interval in seconds between subsequent attempts to redial to the operator; if the parameter not defined the default invite interval is 60 seconds;
- o <waiting\_time\_max> denotes the maximal waiting time in seconds for the user in operator's queue, after that time the system starts dialout to the operator direct phone number defined in <on\_wait\_time\_max\_direct\_number> parameter; if the parameter not defined the waiting period is endless so the dialout to the operator direct phone number is not being performed;
- o <on\_wait\_time\_max\_direct\_number> denotes direct phone number to dialing out to the operator; in this case the operator is not being connected to the operator conference, his call is just being directly connected with the user from the operator queue and he has the possibility to talk to this user.

For detailed information see section *Operator Invite Settings Configuration* in this Guide.

- **operatorPN** denotes operator conference DNIS number (call flow: OPERATOR), default: 20. See detailed parameter description in "WYDE Software Installation Guide".
- playbackPN denotes playback conference DNIS number (call flow: PLAYBACK), default: 16. See detailed parameter description in "WYDE Software Installation Guide".
- **prefered\_codec** denotes preferred codec that is being used when the bridge initiates the calls, i.e. dialout calls, calls to the recording server, etc. For detailed information see section *Enable and Disable Audio Codecs* in this Guide.
- **ss\_ip\_addr** denotes the screen sharing media proxy IP address, default is IP address of the computer where WYDE software FreeSee Server components are installed, for example 192.168.1.30. See detailed parameter description in "WYDE Software Installation Guide".
- **webServicesLogin** denotes web services user login, default: *admin*. See detailed parameter description in "*WYDE Software Installation Guide*".
- **webServicesPassword** denotes web services user password, default: *admin*. See detailed parameter description in "*WYDE Software Installation Guide*".
- **webServicesUrl** denotes URL to the WYDE web services WSDL, the default value is *http://127.0.0.1/dnca/jAdmin* and usually it should not be changed. See detailed parameter description in "WYDE Software Installation Guide".
- web\_strict denotes can (true, default) or can not (false) unprivileged user (i.e. neither admin, nor operator user) create users using Web Administration Interface.
- **bridge** id denotes the identifier of the current (default) bridge.

## **Bridge Configuration Changes**

If all WYDE bridge software components are installed on one single computer and the IP address of this computer has been changed you can update this IP address in all configuration parameters using the single *wyde* command line utility run with the *set-ip* option. The syntax is as follows:

wyde set-ip <arguments>

Each of the arguments is followed by a space and a value. In *set-ip* you can specify the following arguments:

- ip <value> New IP address of the bridge. This argument is required.
- bridge <value> The name of the bridge which IP address should be changed. This argument is optional; if it is omitted the default bridge is being taken from /usr/local/DNCA/etc/dnca.conf file as it was previously described in this section.

This command updates the following settings parameters (the parameters from <code>bridge\_settings</code> table): <code>agiserver\_addr</code>, <code>billing\_localdb\_host</code>, <code>mf\_agi\_addr</code>, <code>mf\_console\_addr</code>, <code>mf\_dc\_connector\_addr</code>, <code>mp\_console\_addr</code>, <code>mp\_sip\_addr</code> with new IP address; it also updates <code>ip\_addr</code> field value in <code>bridges</code> table and <code>ivr\_addr</code> field value in <code>nodes</code> table. In addition this command updates <code>PostgreSQL</code> configuration file <code>pg\_hba.conf</code>, <code>tomcat</code> configuration files and the IP information in the following files: <code>opt/tomcat/webapps/wydevoice/index.html</code>, <code>/opt/tomcat/webapps/supportmodule/WEB-INF/classes/supportmodule.properties</code>.

If you would like to change your notification email address using the command line, you should use the *wyde* command line utility with the *set-email* option. The syntax is as follows:

```
wyde set-email <arguments>
```

Each of the arguments is followed by a space and a value. In *set-email* you can specify the following arguments:

- email <value> New notification email address for the bridge. This argument is required.
- bridge <value> The name of the bridge which notification email address should be changed. This argument is optional; if it is omitted the default bridge is being taken from /usr/local/DNCA/etc/dnca.conf file as it was previously described in this section.

This command updates settings parameter <code>crash\_notify</code> (the parameters from <code>bridge\_settings</code> table) and <code>MAILTO</code> parameter in <code>/etc/crontab</code> file.

# **Dialout Settings Configuration**

The following system parameters are responsible and should be configured to perform dialout either to external phone number or to subscriber's VoIP phone number:

- ivr\_gw\_addr denotes the gateway address (IP) where INVITE should be sent during the dial-out, if you are making dial-out to the real phone number the specified gateway should have PSTN output, if you are making dial-out to the subscribers VoIP phone number this could be the address of your Kamailio (OpenSER) server;
- ivr\_gw\_prefix denotes the prefix that should be added to the beginning of the number during the dial-out or the template for replace.

For example if ivr\_gw\_addr="192.168.1.88" and ivr\_gw\_prefix="" (empty) the dialing out to the number 6046880331 actually will send INVITE to the specified gateway with To field equal 6046880331@192.168.1.88. If, for instance, ivr\_gw\_prefix="9" this prefix will be added to the beginning of the number, i.e. To: 96046880331@192.168.1.88. If, for instance, ivr\_gw\_prefix="00:11" 00 in the number will be replaced with 11 and the dialing out to the number 006046880331 will be made as To: 116046880331@192.168.1.88 (00 will be replaced with 11).

To set this gateway address for your bridge you can use the following *wyde* command: wyde settings-edit name *ivr\_gw\_addr* value 192.168.1.88

In addition for testing purposes you can use the gateway *voice.freeconferencecallhd.com*: wyde settings-edit name *ivr\_gw\_addr* value *voice.freeconferencecallhd.com* 

Note after you changed dialout settings you should use the *wyde* command line utility with the *settings-reload* options to reload and apply system configuration changes that are being implemented:

```
wyde settings-reload
```

In addition to general WYDE bridge dialout settings you can also define peers – external gateways that could be used for the same purposes. If it is necessary to add new peer you

can use the command line interface. To put in new peer using the command line interface you should use the *wyde* command line utility with the peer-add option. The syntax is as follows:

```
wyde peer-add <arguments>
```

Each of the arguments is followed by a space and a value. In *peer-add* you can specify the following arguments:

- name <value> The unique name of the peer that should be added.
- ipaddr <value> The unique peer IP address and optionally port. The logic of this peer attribute is similar to ivr gw addr system settings parameter.
- dialout\_prefix <value> The prefix that should be added to the beginning of the number during the dial-out or the template for replace for the peer. The logic of this peer attribute is similar to ivr gw prefix system settings parameter.
- zone <value> The location zone for the peer.

The arguments name and ipaddr are required, all other arguments are optional. The arguments can be transferred to this command in any order.

For example you can add new peer using the following command (new peer attributes are shown in *italic*):

```
wyde peer-add name fcc ipaddr voice.freeconferencecallhd.com
```

If the command is successful, the system will not return any errors or messages, in typical Linux fashion, it will just drop you back to the command prompt (#). The sample of the *peer-add* command output and the help on this command is shown on Figure 58.

Figure 58: wyde help peer-add and wyde peer-add Commands Output Sample

To update the properties of existing WYDE bridge peer using the command line, you should use the *wyde* command line utility with the *peer-set* option. The syntax is as follows:

```
wyde peer-set <arguments>
```

Each of the arguments is followed by a space and a value. In *peer-set* you can specify the following arguments:

- name <value> The name of the peer to update.
- ipaddr <value> The IP address and optionally port of the peer which properties should be updated.

- dialout\_prefix <value> The prefix that should be added to the beginning of the number during the dial-out or the template for replace for the peer;
- zone <value> New location zone for the peer to update.

Argument ipaddr is required. Other arguments should be transferred only if you would like to update the specific peer property. The arguments can be transferred to this command in any order.

For example you can update previously created peer and set its dialout prefix to 9 using the following command:

```
wyde peer-set ipaddr voice.freeconferencecallhd.com
   dialout prefix 9
```

To drop the existing WYDE bridge peer definition using the command line, you should use the *wyde* command line utility with the *peer-del* option. The syntax is as follows:

```
wyde peer-del <arguments>
```

Each of the arguments is followed by a space and a value. In *peer-del* you can specify the following arguments:

• id <value> - The identifier of the peer that should be deleted. Argument id is required.

If you would like to show a list of the peers in the system using the command line, you should use the *wyde* command line utility with the *peer-show* option. The syntax is as follows:

```
wyde peer-show
```

This command will output a list of the all existed peers on the system, similar to shown on Figure 59. As you can see, the *wyde peer-show* command shows the peers that have been created in the system as well as their basic properties.

Figure 59: wyde peer-show Command Output Sample

For detail information how to perform dialout using *mf* console see section Dialing another User in this guide.

## **Recording Settings Configuration**

Conference recording can be made using any of two recording methods:

- *local* the recording is being made directly to the bridge where the conference take place:
  - o the recording files are being placed to the /usr/local/DNCA/var/recordings/ folder.
  - o the frontend components of the WYDE bridge software are responsible for maintaining of the recording,

- o if the cluster configuration where different WYDE software components installed on different computers is used, the /usr/local/DNCA/var/recordings/ folder is placed on the computer where core components are installed and this folder should be accessible through NFS (Network File System) by the computer where frontends components are installed;
- remote the recording is being made to the external recording server:
  - o in this case additional call to the external recoding server is being placed to maintain the conference recording,
  - o the external recording server is the separate computer with the *asterisk* service installed, to maintain the recording it should be able to process INVITE in WYDE format or specific recording server software should be installed on that computer;
    - for detail information about WYDE recording server please contact your WYDE Voice representative,
    - alternatively, if you with, you can create your own recording server, in this case it must be able to process INVITE in WYDE format.

The recording method is being set using recording\_method call flow attribute either for call flow or for DNIS. The *local* value of this attribute is used to make the recording locally, the *remote* value is used to denote that the conference recording should be made to the external recording server.

The following system parameter is responsible and should be configured if you wish to use external recording server for your bridge conferences:

• ivr rec addr – denotes the IP address of the external recording server.

For example, if you are going to use the computer with IP address 192.168.1.81 as your recording server, to set this recording server address for your bridge you can use the following wyde command:

```
wyde settings-edit name ivr rec addr value 192.168.1.81
```

You can override your bridge external recording server settings defined in ivr\_rec\_addr for your call flow or DNIS using the recording server address (recording\_addr) call flow attribute either for call flow or for DNIS. If you use remote recording method and recording\_addr call flow attribute is not empty, this address is being used as your recording server IP; if it is empty, the address from ivr\_rec\_addr system settings parameter is being used as your recording server IP.

### **Operator Invite Settings Configuration**

As it was previously told in the section Operator Conferences the WYDE bridge can be configured to perform dialout to the operator when it is necessary. The following system parameter is responsible and should be configured to perform dial-out to operator:

• operator\_invite - denotes when, to what number, from what DNIS the invite call should be made from the bridge to the operator as well as some other options; the parameter has the following format:

```
<phone number>:<operator_did>
   [:<waiting_time>[:<dial_timeout>[:<invite_interval>
   [:<waiting_time_max>:<on_wait_time_max_direct_number>]]]]
```

### where

- o <phone number> denotes operator's phone number to dialout, it also could be operator business telephone system / PBX phone number;
- o coperator did> denotes DNIS number of the operator conference;
- o <waiting\_time> denotes the waiting period in seconds how long the caller from the operators' queue will wait until the system performs dialout to the operator; if the parameter not defined the system starts dialout immediately once the user arises in the operator queue;
- <dial\_timeout> denotes the operator response waiting period in seconds, i.e. the operator dialout call answering period; if the parameter not defined the default dialout timeout is 45 seconds;
- o <invite\_interval> denotes the interval in seconds between subsequent attempts to redial to the operator; if the parameter not defined the default invite interval is 60 seconds;
- o <waiting\_time\_max> denotes the maximal waiting time in seconds for the user in operator's queue, after that time the system starts dialout to the operator direct phone number defined in <on\_wait\_time\_max\_direct\_number> parameter; if the parameter not defined the waiting period is endless so the dialout to the operator direct phone number is not being performed;
- o <on\_wait\_time\_max\_direct\_number> denotes direct phone number to dialing out to the operator; in this case the operator is not being connected to the operator conference, his call is just being directly connected with the user from the operator queue and he has the possibility to talk to this user.

When the call to the operator is made, the operator is being asked to enter his access code. If the access code is valid, the caller is being joined to his operator conference.

To set the bridge to automatically call to the phone number 6046880331 and connect to the operator DNIS 8665080020 you can use the following wyde command:

```
wyde settings-edit name operator_invite
   value 6046880331:8665080020
```

## **AGI Server Pool Configuration for Processes Handlers Requests**

WYDE bridge provides the possibility to configure either static or dynamic pool for the processes handlers requests. In most cases you can use the static pool. The dynamic pool is recommended if the load of your bridge is high; the primary goal of the dynamic pool is withstand the peak load and release the resources when the load is decreased to normal level. The following system parameter is responsible for the AGI server pool configuration:

• agiserver\_pool\_size - denotes what pool (static or dynamic) with what options is being created for the AGI server; the parameter has the following format:

### where

- o <static pool size> denotes the static AGI server pool size;
  - ✓ if agiserver\_pool\_size parameter does not have the colon (:) character the static pool is being created, otherwise (if the parameter has colon character) the dynamic pool is being created;

- o <min\_servers> denotes the minimum number of servers to keep running;
   default value: 5;
- o <max\_servers> denotes the maximum number of child servers to start; this does not apply to dequeue processes; default value: 50;
- o <min\_spare\_servers> denotes the minimum number of servers to have waiting for requests; minimum and maximum numbers should not be set to close to each other or the server will fork and kill children too often; default value: 2;
- o <max\_spare\_servers> denotes the maximum number of servers to have waiting for requests; also see <min\_spare\_servers> above for additional info; default value: 10.

To set the WYDE bridge static AGI server pool of 5 processes you can use the following *wyde* command:

```
wyde settings-edit name agiserver_pool_size value 5
```

To set the dynamic AGI server pool with the minimum number of servers to keep running equal to 5, the maximum number of child servers to start equal to 20, and default <min\_spare\_servers>(2)/<max\_spare\_servers>(10) options values you can use the following *wyde* command:

```
wyde settings-edit name agiserver pool size value 5:20
```

### **Bridge Configurations Reloading**

From time to time you may need to reload some or all WYDE bridge configuration features. If any changes in the bridge configuration could interfere with normal bridge work, these changes could be applied only manually, i.e. you should manually reload necessary module or sub-system. If the changes could be safely made without interfering in the functioning of the bridge, such changes could be applied automatically.

You can use the *wyde* command line utility with the following options to apply configuration changes that are usually being implemented automatically:

- did-reload reload DNISes (DIDs) cache;
- sounds-reload reload music-on-hold and welcome prompt sound files.

You should always use the *wyde* command line utility with the following options to apply configuration changes that are being implemented only manually:

- auth-reload reload auth configuration;
- billing-reload reload billing configuration;
- bridge-reload-reload DC bridges configuration;
- callflow-reload reload all call flows;
- node-reload reload nodes configuration;
- settings-reload-reload system settings.

All these commands, both automatic and manual, are described in this guide in details with the description where they should be executed.

In addition you have the possibility to reload all configurations, i.e. to run all reload commands described above. To do so you could use the *wyde* command line utility with the *reload-all* option. The syntax is as follows:

wyde reload-all

# WYDE Bridge Configuration Save and Restore

The WYDE bridge software allows you to save the currently installed and configured system. This includes:

- Main database *dnca*;
- Billing database *dnca calls*;
- Contents of the folder /usr/local/DNCA/etc;
- Contents of the folder /usr/local/DNCA/var;
- Configuration files of *asterisk*.

If you would like to save the current bridge settings using the command line, you should use the *wyde* command line utility with the *config2-save* option. The syntax is as follows: wyde config-save *file* <target file name> where

• <target file name> - The file name to save configuration settings. This argument is required.

For example to save the current WYDE bridge configuration into *configsave\_20100512* file you should run the command:

```
wyde config-save file configsave 20100512
```

If the command is successful, you will be informed about each step that performs the command, i.e. you receive the messages when main database is saved, when billing database is saved and when general system settings are saved; after that the command will return you back to the command prompt (#). The sample of the *config-save* command output and the help on this command is shown on Figure 60.

Figure 60: wyde help config-save and wyde config-save Commands Output Sample

To restore previously saved configuration settings from the file using the command line, you should use the *wyde* command line utility with the *config-restore* option. The syntax is as follows:

wyde config-restore <arguments>

Each of the arguments is followed by a space and a value. In *config-restore* you can specify the following arguments:

- file <source file name> The file name to restore configuration settings. This argument is required.
- force <value> Force required {yes | no}:
  - yes denotes that the configuration file should be restored even if the versions of previously saved configuration data and currently installed WYDE software are different;
  - o no denotes that the configuration file should be restored only if the versions of previously saved configuration data and currently installed WYDE software are the same.

This argument is optional, if the argument is omitted force no is used by default.

If the versions of previously saved data and currently installed WYDE software are the same this command will restore all saved settings, i.e. databases *dnca* and *dnca\_calls*, folders /usr/local/DNCA/etc and /usr/local/DNCA/var, and asterisk configuration files. If these versions are different and parameter force either omitted or equal no, you will receive the error message about versions mismatch and restore will be canceled. If these versions are different and parameter force equal yes, you will receive the warning message about versions mismatch, but restore still will be processed.

### **Nodes Administration**

In terms of WYDE bridge *MF* service, the *node* is the computer with the *asterisk* service installed and running. The *asterisk* is being installed in *Frontend* components installation. If you are performing cluster installation you can have multiple nodes, i.e. multiple *asterisk* computers in your WYDE bridge environment. Note that the version of the WYDE software must be the same on all nodes of you bridge.

If you are configuring the large scale bridges, i.e. the bridges that should support large amount of simultaneous calls, the single bridge resources may be insufficient. In this case you may consider using multiple nodes for you bridge. Later in this section we will describe how to install and configure additional node for you bridge.

To add new node definition on the bridge using the command line, you should use the *wyde* command line utility with the *node-add* option. The syntax is as follows:

wyde node-add <arguments>

Each of the arguments is followed by a space and a value. In *node-add* you can specify the following arguments:

- bridge <value> The name of the bridge to which you would like to add the node. If this argument is omitted the node will be added for the current bridge.
- name <value> The name of the node that should be added.
- ivr\_addr <value> The node IVR address, i.e. IP address and port of the node.
- sip\_addr <value> The node SIP address, i.e. IP address and port of the node on which *asterisk* services (Frontend) should accept (receive) SIP calls.

- ivr\_maxcalls <value> The maximum number of calls on the IVR for the node. The optimal value of this parameter also depends on your WYDE bridge node hardware configuration (the default value: 500).
- zone <value> The location zone for the node.
- active <value> The node active-flag value (True denotes that the node is active and new calls can go to this node, it is default value for new node, False denotes that the node is inactive and new calls cannot go to this node).

Arguments name, ivr\_addr and sip\_addr are required. The arguments can be transferred to this command in any order.

To update the properties of existing WYDE bridge node using the command line, you should use the *wyde* command line utility with the *node-set* option. The syntax is as follows:

wyde node-set <arguments>

Each of the arguments is followed by a space and a value. In *node-set* you can specify the following arguments:

- bridge <value> The name of the bridge which node properties should be updated. If this argument is omitted the node for the current bridge will be taken.
- name <value> The name of the node which properties should be updated.
- ivr\_addr <value> The node new IVR address, i.e. IP address and port of the node.
- sip\_addr <value> The node new SIP address, i.e. IP address and port of the node on which *asterisk* services (Frontend) should accept (receive) SIP calls.
- ivr\_maxcalls <value> New maximum number of calls on the IVR for the node. The optimal value of this parameter also depends on your WYDE bridge node hardware configuration (the default value: 500).
- zone <value> New location zone for the node.
- active <value> New active-flag value for the node (True denotes that the node is active and new calls can go to this node, False denotes that the node is inactive and new calls cannot go to this node). If you change this flag value from True to False, the calls that were connected to this node while it was active, remain on this node, but new calls will not go to this node.

Argument name is required. Other arguments should be transferred only if you would like to update the specific node property. The arguments can be transferred to this command in any order.

To drop the existing WYDE bridge node definition using the command line, you should use the *wyde* command line utility with the *node-del* option. The syntax is as follows: wyde node-del <a href="mailto:arguments">arguments</a>

Each of the arguments is followed by a space and a value. In *node-del* you can specify the following arguments:

- bridge <value> The name of the bridge which node should be deleted. If this argument is omitted the node will be deleted from the current bridge.
- name <value> The name of the node that should be deleted.

Argument name is required. The arguments can be transferred to this command in any order.

For these *node-add*, *node-set*, and *node-del* commands the default bridge is defined in */usr/local/DNCA/etc/dnca.conf* file:

```
[general]
bridge id = 1
```

If the argument bridge is omitted in these commands the bridge defined in *dnca.conf* file will be taken as default.

If any of these commands is successful, the system will not return any errors or messages; it will just return you back to the command prompt (#).

If you added new node, updated node properties, or deleted existing node you should reload *MF* cluster nodes list, i.e. reload *nodes* table from the database. To send the signal on the WYDE bridge *MF* engine to reload all nodes the following *mf* console command should be executed:

```
node-reload
```

or you should use the *wyde* command line utility with the *node-reload* option: wyde node-reload

If new *asterisk* computer was added this command will connect and turn on this node. If the properties of existing nodes were changed they will be renewed and actualized. If existing node was deleted all conferences and calls on this node will be dropped and the node will be disconnected and turned off.

If you would like to show nodes configuration in the database, you should use the *wyde* command line utility with the *node-show* option. The syntax is as follows:

```
wyde node-show [bridge <bridge>]
where
```

• <br/> <br/> <br/> <br/> <br/> - the name of the bridge which nodes you would like to show. This argument is optional.

If you would like to show current nodes status, you should use the following *mf* console command:

```
node-show
```

Let's assume that you have the bridge with full installation of the WYDE software on IP 192.168.1.5 and you would like to add new node on IP 192.168.1.33. To do that first of all you should install on the 192.168.1.33 computer the *Frontend* components of the WYDE software, i.e. install the *asterisk* service. After that on your 192.168.1.5 computer you should run the following *wyde* command to add this node registration:

```
wyde node-add name AST33 ivr_addr 192.168.1.33:4460 sip addr 192.168.1.33:5065
```

Next you should run *mf* console command *node-reload* to send the signal on the WYDE bridge *MF* engine to reload all nodes.

As soon as these steps have been made you are able to call either on 192.168.1.5 or 192.168.1.33.

Samples of wyde node-show and mf console node-show commands are shown on Figure 61. Please see nodes identifiers in mf console node-show command output: I – for 192.168.1.5 and 6 – for 192.168.1.33; when you are looking the conference information these identifiers are shown in mf Console show, column Node. Figure 62 shows two conference calls, as you can see the host call has been made to the node 1 (192.168.1.5) and the participant call has been made to the node 6 (192.168.1.33).

```
~] # wyde node-show
                 Node
                                   IVR Addr
                                                      SIP Addr
                                                                       MaxCalls
                                                                                          Zone
Bridge
WYDE5
                                   192.168.1.5:4460 192.168.1.5:5065 500
                 AST1
                                                                                                   True
                                   192.168.1.33:4460 192.168.1.33:5065 500
WYDE 5
                 AST33
                                                                                                     True
[root@ZILBER ~] # mf
WYDE.MF version 3.1.116
WYDE.MF>node-show
                    IP Addr
                                                      AGI Pool
d Name
                                                      10
    AST33
                    192.168.1.33:4460
                                          active
```

Figure 61: wyde node-show and mf Console node-show Commands Output Sample

root@ZILBER:~	/install/2.1.1	03-CENTOS5					_IDX
WYDE.MF>show	667788 number=667	788, id=173,	flags=B, p	olicy=	M(False,	False,Strict,F	alse);H
SesId	Node	Role	Location	Type	Flags	AKey	
1.53335.50				DOWN			
16777562	1	Host	MP	PSTN		0	
16777564 WYDE.MF>	6	Participant	MP	PSTN		0	<b>-</b>

Figure 62: mf Console show Conference - Different Nodes Calls Command Output Sample

## **Distributed Conferencing Administration**

In terms of WYDE bridge, the *distributed conference* is the conference that is taking place on the multiple bridges simultaneously. That means that the calls are being made to the different bridges, but these calls are participating in the same conference.

You may need to use distributed conferences if your users are geographically located remotely from each other in different regions or countries and you prefer having the local bridges for them. In this case the users will be able to call to the local bridges, but their calls could be joined into distributed conferences.

This section contains only basic information about distributed conferencing configuration. If you would like to use distributed conferences please contact WYDE Voice technical support for detail instructions and assistance in distributed conferencing configuration for the WYDE Voice conferencing bridge software.

The following steps and conditions should be implemented if you are going to use distributed conferences:

- all necessary bridges that perform distributed conferences must have the same version of the WYDE software installed on them;
- all necessary bridges that perform distributed conferences should be added to the bridges table on all bridges that you use, you can use wyde bridge-add command for these purposes;
- the identifiers of your bridges (*id* field in *bridges* table) should be between 1 and 15:
- the identifiers of your bridges should be the same on all bridges that are performing distributed conferences;
- the system parameter  $mf\_dc$  should be set 1 on all these bridges, see section WYDE Bridge Settings Management for instructions how to set the system parameter value;
- the system parameter mf\_dc\_connector\_addr should be set; it denotes the IP address and the port (for example 192.168.1.5:4480) for binding between MF and ASD; it is being used to establish connection between bridges in distributed conferences; if MF sees that the conference is distributed, it sends to this IP address and port the ASD commands that establishes SIP connection between distributed conferences bridges;
- after that the bridge sends notifications about the conferences to all bridges described in bridges table;
- call flow attribute <code>conference\_distributed</code> (Distributed conferencing) should be equal "on" either on call flow or on DNIS or on conference level for the conference where you are calling;
- if another bridge has the conference with the same conference number, and conference\_distributed call flow attribute for this conference equal "on", and mf\_dc parameter is equal to 1 on that bridge as well, these conferences are being joined into single distributed conference.

Note, if you would like to replicate distributed conferences CDR information from one DC bridge to all other bridges that are participating in distributed conferencing, the system parameter  $mf\_dc\_cdr\_replication$  should be set 1 on all DC bridges whose CDR info you would like to replicate; the system parameter  $mf\_dc\_billing\_rule$  should be set to the billing rule that is used to process replicated CDR records received from another distributed bridge, any such DC bridge can have its own billing rule to process these received CDR records.

The following algorithm is being applied for the distributed conferences:

- the bridge knows identifiers, IP addresses and ports of all bridges, that are participating in distributed conferencing;
- the bridge announces all own conferences each minute and when new conference is being created to all other bridges;
- to send these announcements UDP datagrams are being used, these datagrams have their own specific format;
- if the conference with the same conference number and <code>conference\_distributed</code> attribute equal "on" exists on two or more bridges the master bridge is being selected for this conference,

- o the *weight* entity is being used to determine what bridge is being selected as the master bridge,
- o the system parameter  $mf\_dc\_weight$  is being set for any of distributed bridges (the default value is  $\theta$ ),
- the *weight* of the bridge is being calculated as number of links with another bridges plus *mf* dc weight system parameter value,
- o the master bridge is the bridge with the largest *weight*,
- o if the weights of the bridges are the same, the master bridge is the bridge where the conference was started earlier than on other bridges;
- subordinate bridges make connection with the master bridge using real time protocol;
- real time clients translate all RT commands to another bridges using control calls, by this approach the status of distributed conference is being kept the same on all bridges;
- if during the distributed conference the subordinate bridge RT client loses the master bridge, the existing master bridge is being canceled and new master bridge is being selected again using the same algorithm that was described above.

Note, thus if you wish to select the specific bridge as your master bridge (for example, if it has more powerful hardware, or if the bandwidth for it is higher, or if it is in the center of you star topology network, etc.) you can achieve that using  $mf_dc_weight$  system parameter.

To add new bridge definition you should use the *wyde* command line utility with the *bridge-add* option. The syntax is as follows:

```
wyde bridge-add <arguments>
```

Each of the arguments is followed by a space and a value. In *bridge-add* you can specify the following arguments:

- id <value> The identifier of the bridge that should be added.
- name <value> The name of the bridge that should be added.
- ip addr <value> IP address of the bridge that should be added.
- rx\_gain <value> Received audio gain for the bridge; could be from -10 till 10 or 255; it indicates what volume level should be set for the call, 255 denotes that the volume level is being automatically generated by the backend.
- tx\_gain <value> Transmitted audio gain for the bridge; could be from -10 till 10 or 255; it indicates what volume level should be set for the call, 255 denotes that the volume level is being automatically generated by the backend.

The arguments id, name, and ip\_addr are required. The arguments can be transferred to this command in any order.

If you need to change existing bridge definition you should use the *wyde* command line utility with the *bridge-set* option. The syntax is as follows:

```
wyde bridge-set <arguments>
```

Each of the arguments is followed by a space and a value. In *bridge-add* you can specify the following arguments:

- id <value> The identifier of the bridge which properties should be updated.
- name <value> The name of the bridge that should be updated.
- ip addr <value> IP address of the bridge that should be updated.

- new id <value> New identifier of the bridge if it should be changed.
- rx\_gain <value> Received audio gain for the bridge that should be updated; could be from -10 till 10 or 255; it indicates what volume level should be set for the call, 255 denotes that the volume level is being automatically generated by the backend;
- tx\_gain <value> Transmitted audio gain for the bridge that should be updated; could be from -10 till 10 or 255; it indicates what volume level should be set for the call, 255 denotes that the volume level is being automatically generated by the backend.

The argument id is required. Other arguments should be transferred only if you would like to update the specific bridge property. The arguments can be transferred to this command in any order.

To delete existing bridge definition you should use the *wyde* command line utility with the *bridge-del* option. The syntax is as follows:

wyde bridge-del id <bridge id>
where

If you added new bridge, updated existing bridge properties, or deleted existing bridge you should reload bridge information, i.e. reload *bridges* table from the database, to apply the changes you made. To send the signal on the WYDE bridge to reload all bridges the *wyde* command line utility with the *bridge-reload* option should be used:

wyde bridge-reload

Alternatively you can use *mf* command *dc-reload* to reload DC bridges configuration.

To see all registered WYDE bridges, i.e. the bridges described in bridges table you should use the *wyde* command line utility with the *bridge-show* option. The syntax is as follows:

wyde bridge-show

For each defined bridge this command returns the bridge identifier, name of the bridge, IP address of the bridge, information showing is the bridge reachable or not, received (Rx) audio gain, and transmitted (Tx) audio gain. The *Reachable* column can have the following values:

- Local for your local bridge, i.e. for the bridge installed on the same computer where you are running this *bridge-show* command;
- Yes if the bridge is reachable via network for distributed conferences, i.e. the bridge can participate in distributed conferences;
- No if the bridge is not reachable via network for distributed conferences, i.e. the bridge can not participate in distributed conferences because either distributed conferencing is switched off either for local or for remote bridge (mf\_dc parameter is

equal 0, see below), the remote bridge is inaccessible, there are no such definition on the remote bridge, etc.

Let's assume that you would like to configure distributed conferences between two bridges: 192.168.1.5 (let the name of the bridge is WYDE5 and identifier of the bridge is 1) and 192.168.1.31 (let the name of the bridge WYDE31 and identifier of the bridge is 2). To do that you should run the following wyde commands on the first bridge (i.e. on the bridge with identifier 1, name WYDE5):

```
wyde bridge-set id 1 name WYDE5
wyde bridge-add id 2 name WYDE31 ip_addr 192.168.1.31
wyde bridge-reload
```

Next you should run the following *wyde* commands on the second bridge (i.e. on the bridge with identifier 2, name *WYDE31*):

```
wyde bridge-set id 1 name WYDE31 new_id 2
wyde bridge-add id 1 name WYDE5 ip_addr 192.168.1.5
wyde bridge-reload
```

On the second bridge when you change the identifier for the local bridge the system shows the message "Local bridge has been switched. You should run 'service wyde restart' to apply changes." That means the current local bridge identifier has been updated in the /usr/local/DNCA/etc/dnca.conf file as the result of this command:

```
[general] bridge id = 2
```

as well as <code>bridge\_settings</code> table and nodes configuration have been updated. Next you should run the command:

```
service wyde restart to apply these changes.
```

After that on all these bridges you should switch on the distributed conferences configuration mode and set the system parameter  $mf_dc$  equal 1; the following wyde command should be used for that:

```
wyde settings-edit name mf dc value 1
```

After these changes the content of bridges table, i.e. the configuration of the bridges will be exactly the same on both bridges (see Figure 63 and Figure 64) and both bridges will be fully functional, operational and ready to participate in distributed conferences.

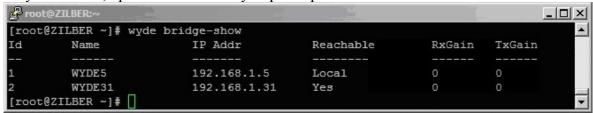


Figure 63: wyde bridge-show Command Output Sample on WYDE5 Bridge

🚱 root	@localhost:~	A. C.			
[root	@localhost ~]# v	wyde bridge-show			
Id	Name	IP Addr	Reachable	RxGain	TxGain
1	WYDE5	192.168.1.5	Yes	0	0
2	WYDE31	192.168.1.31	Local	0	0
[root	@localhost ~]#[				

Figure 64: wyde bridge-show Command Output Sample on WYDE31 Bridge

If you have large quantity of bridges that should participate in distributed conferencing you may consider using *wyde* commands *bridge-list-save* and *bridge-list-load* to simplify bridges configuration process. Using this approach all necessary bridges are being configured on one bridge, next the list of bridges is being saved on this bridge using the *wyde* command line utility with the *bridge-list-save* option, the created file is being copied to other bridges, and the *wyde* command line utility with the *bridge-list-load* option is being run on them to load all necessary bridges.

To save the list of all registered WYDE bridges into the file you should use the *wyde* command line utility with the *bridge-list-save* option. The syntax is as follows: wyde bridge-list-save file <file name> where

• <file name> - The required output file name to save the list of bridges.

This file contains the following comma-delimited columns: the bridge identifier, the name of the bridge, IP address of the bridge, IP port of the bridge.

To load the list of all registered WYDE bridges from the file you should use the *wyde* command line utility with the *bridge-list-load* option. The syntax is as follows: wyde bridge-list-load *file <file name>* where

• <file name> — The required input file name to load the list of bridges.

The structure of the import file should be exactly the same that has the output file created by *bridge-list-save* command.

After that you are able to configure the distributed conference for a subscriber. This conference should have the same conference number on both bridges, other conference properties could be different, but for this sample we configure the conference accounts exactly the same. Let's assume that the conference number is 889900, the subscriber PIN is *nrobert*, the DNIS number to call is 8665080012, host access code is 505051 and participant access code is 505052.

```
wyde confuser-add subscriber nrobert did 8665080012 conference 889900 accesscode 505051 role Host wyde confuser-add subscriber nrobert did 8665080012 conference 889900 accesscode 505052 role Participant wyde conference-attr-set number 889900 name conference_distributed value on
```

Now all calls to both bridges to the conference 889900 are being joined into single distributed conference and all conference participants will be hear and talk to each other.

Figure 65 shows the distributed conference on the first bridge (WYDE5) and Figure 66 shows the distributed conference on the second bridge (WYDE3I). In our sample the first bridge (WYDE5) is being selected as master and the second bridge (WYDE3I) is subordinate bridge. According to the algorithm described above the subordinate bridge (WYDE3I) makes connection with the master bridge (WYDE5) using real time protocol. Because of that you can see on both bridges the third connection with the role DC – distributed conferencing and the flag T – real time (RT) mode.

Proot@ZILBER:~	,						_   ×
WYDE.MF>show							_
Number	Members	On IVR	On MP	Forei	gn Fla	ags	
889900 3 calls in 1	3	0	2	1	TD	(jm)	
WYDE.MF>show		3.					
Conference: n	number=8899	00, id=13421	8735, flag	s=TD (j	m), poli	cy=M(False,Fal	se,Stri
ct, False);H(F	alse,False	,False,False	)				
SesId	Node	Role	Location	Type	Flags	AKey	
16779724	1	Host	MP	PSTN		0	
16779727	1	DC	MP	VoIP	T	0	
33554434	0	Participant	Foreign	PSTN		0	
3 calls							
WYDE.MF>							-

Figure 65: Show Distributed Conference on WYDE5 Bridge

🚱 root@localhos	st:~					:
WYDE.MF>show						
Number	Members	On IVR	On MP	Forei	gn Flag	la
889900	3	0	2	1	TD(	i)
3 calls in 1	conference	s.				
WYDE.MF>show	889900					
Conference:	number=8899	00, id=26843	5458, flag	s=TD(j	), policy	/=M(False,False,Stric
+ Fales .H/F						
t, raise), n(r	alse, ralse,	False, False)				
				_		
	alse, False, Node	False,False) Role	Location	Type	Flags	AKey
SesId	Node	Role			Flags	
SesId  16779724	Node 	Role  Host	Foreign	PSTN	Flags	0
SesId  16779724	Node	Role	Foreign		Flags	
SesId  16779724 33554434 33554435	Node 	Role  Host	Foreign	PSTN PSTN	Flags	0
SesId  16779724 33554434	Node  0 1	Role  Host Participant	Foreign MP	PSTN PSTN		0

Figure 66: Show Distributed Conference on WYDE31 Bridge

If distributed conferencing is switched on you can see all bridges that can participate in distributed conferences using *mf* console command:

```
dc-show-bridges
```

This command returns bridge identifier, name, IP address, flag showing is the bridge reachable for distributed conferences or not, and DC weight of this bridge. At the beginning this list includes other bridges that can participate in distributed conferences. After this list you will see the caption "*This bridge*:" and information about the local bridge. For the local bridge its weight is displayed in the format:

<total bridge weight>(mf dc weight), where mf dc weight—is the base
weight from the system settings parameter defined for this bridge, and
<total bridge weight> is calculated as mf\_dc\_weight plus number of links to
other bridges visible from this bridge.

Let's assume that the bridge *WYDE5* is preferable master bridge. Because of that we increased its weight to 10 using the *wyde* command:

```
wyde settings-edit name mf_dc_weight value 10 the bridge WYDE5 weight remains default 0.
```

Figure 67 shows DC bridges information on the bridge WYDE5, the master bridge in the started conference; the first row shows information about the subordinate bridge WYDE31 and it tells the bridge IP and that it is reachable now and its weight is 1; the last row shows information about the master bridge WYDE5 and it tells the bridge IP and that it is local bridge and its weight is 11 (10), that means that mf\_dc\_weight=10 plus one other bridge (WYDE31) is reachable from this bridge (total weight is 11). Figure 68 shows DC bridges information on the bridge WYDE31, the subordinate bridge in the started conference; it has the first row with information about the master bridge WYDE5 and it tells that the master bridge is also reachable now and its weight is 11; the last row shows information about the subordinate bridge WYDE31 and it tells the bridge IP and that it is local bridge and its weight is 1 (0), that means that mf\_dc\_weight=0 plus one other bridge (WYDE5) is reachable from this bridge (total weight is 1).

```
root@ZILBER:
                                                                                  _ | D | X |
WYDE.MF>dc-show-bridges
                          IP Addr
        Name
                                               Reachable
                                                                 Weight
                          192.168.1.31
        WYDE31
                                               Yes
This bridge:
        WYDE5
                          192.168.1.5
                                               Local
                                                                 11
                                                                          (10)
WYDE.MF>
```

Figure 67: mf Console dc-show-bridges Command Output Sample on WYDE5 Bridge

Proc	ot@zilber:~					_ U ×
WYDE.	.MF>dc-show-bri	dges				
Id	Name	IP Addr	Reachable	Weight		
1	WYDE5	192.168.1.5	Yes	11		
This	bridge:					
2	WYDE31	192.168.1.31	Local	1	(0)	
WYDE.	.MF>					~

Figure 68: mf Console dc-show-bridges Command Output Sample on WYDE31 Bridge

If distributed conferencing is switched on and there are distributed conferences running on the bridge you can see all distributes conferences using *mf* console command: dc-show-links

For all started distributed conferences this command outputs conference number, master bridge name, and comma-separated linked bridges names.

Figure 69 shows output of this command that was run on master bridge (WYDE5), because of that the Master column has its name – WYDE5 and the Links column contains information about the subordinate bridge (WYDE31). Figure 70 shows output of this command that was run on the subordinate bridge (WYDE31), because of that both columns Master and Links contain information about master bridge (WYDE5).

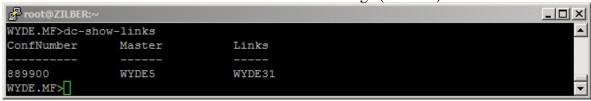


Figure 69: mf Console dc-show-links Command Output Sample on WYDE5 Bridge

p root@localhost;~			×
WYDE.MF>dc-sh	now-links		
ConfNumber	Master	Links	
889900 WYDE.MF>	WYDE5	WYDE5	▼

Figure 70: mf Console dc-show-links Command Output Sample on WYDE31 Bridge

WYDE bridge can log Distributed Conferencing *mf* activity into /*usr/local/DNCA/log/mf.log* file. To set the DC logger level you should run the following *mf* console command:

```
dc-set-log-level {debug|event|info}
where
```

• info – denotes that only DC information and error messages should be stored in the log file (default); event – denotes that additionally the log should contain all DC events logging; debug – denotes that debugging mode is switched on, so additionally debug DC messages should be added to the log. This argument is required.

You can use the information from this *mf.log* file in case of any problems with distributed conferences or if you need to monitor WYDE bridge DC activity.

# **Peers Management**

In terms of WYDE bridge *MF* service, the *peer* is the SIP client from which the bridge accepts the calls. Peers information is being stored in *peers* table. If you would like to configure the WYDE bridge to accept not all calls, but only the calls from specific IP address you should update <code>bridge\_settings</code> table and set <code>ivr\_checkpeer</code> parameter value equal to 1 (default value is 0, no peers limitation applied) and populate <code>peers</code> table with allowed IP address together with other information.

If you added new peers into peers table you should reload them from the database. To send the signal on the WYDE bridge *MF* engine to reload all peers from the peers table the following *mf* console command should be executed:

peer-reload

If the command is successful, the system will not return any errors or messages.

# **Calls Transferring**

If you would like to transfer started conference calls from one WYDE bridge or bridge frontend to another (for instance if you need to switch off the specific frontend or entire bridge), you can make this transfer using the following *mf* console command:

### where

- One of the following arguments should be specified as the first argument of this command:
  - o {node <node\_id>|node all} denotes the node identifier (or all nodes)
    from which the calls should be transferred;
  - o {conference <conf\_number>} denotes the conference number for
    which the calls should be transferred;
  - o {did <did>} denotes the DNIS (DID) number for which the calls should be transferred.
- <destination\_ip> Denotes destination IP address of the bridge or bridge frontend to which the specified calls should be transferred. This argument is required.

You can also use *wyde* command line utility with the *transfer* option for the same purposes. See wyde Command Reference: transfer (Transfer Calls) for detail command information and parameters format.

After this command execution all conference calls that currently are started on the bridge and fit with the criteria that is specified as the first argument of this command will be transferred to the bridge or bridge frontend specified as the second argument of this command

# **Audio Prompts Management**

The WYDE bridge software allows you using your own audio prompts as welcome prompt and music-on-hold prompt in the conferences. If you would like using your own audio files you should perform the steps described in this section of the guide.

Custom audio files should be placed into the following subfolders of /usr/local/DNCA/var folder:

- welcome-prompts (/usr/local/DNCA/var/welcome-prompts) the folder for welcome prompts audio files;
- moh (/usr/local/DNCA/var/moh) the folder for music-on-hold prompts audio files. Thus you should place your audio files in required formats (\*.wav, \*.ul, \*.al, etc.) into these folders.

The audio files for music-on-hold prompts are being searched in specific order:

• if the music-on-hold file name defined in <code>confefence\_moh</code> call flow attribute value is not empty and is not equal to `default`, the system checks the conference recording folder (usually /usr/local/DNCA/var/recordings/ folder) upload subfolder for the specific conference, the files could be uploaded into this folder using Web Control

Console (for example the folder name could be /usr/local/DNCA/var/recordings/673/946673/upload/);

- if the file was not found in the previous location, the system searches the file in /usr/local/DNCA/var/moh folder, the files could be uploaded into this folder using Web Administration Interface;
- otherwise, if the file not found, the system treats <code>confefence\_moh</code> call flow attribute value as equal to `default`, and searches the audio files in /usr/local/DNCA/callflows/<call flow name>/moh/ folder, files musiconhold.<codec> for specific codecs;
- if the audio file was not found in the previous location, the system uses the files *musiconhold*.<*codec*> for specific codecs from */usr/local/DNCA/lib/sounds/moh/* folder.

Note, that if you have file in *wav* format only, you can convert it into other formats on your WYDE server using *transcoder.x* utility:

For example if you would like to convert music-on-hold my\_moh.wav file from wav format into ulaw format you should use the command (the transferred command arguments are shown in italic):

```
/usr/local/DNCA/bin/transcoder.x
-i /usr/local/DNCA/var/moh/my_moh.wav -f wav
-o /usr/local/DNCA/var/moh/my_moh.ul -t ulaw
```

If the command is successful, the system will not return any errors or messages; it will just return you back to the command prompt (#). The sample of the audio file conversion using *transcoder.x* utility and the help on this command is shown on Figure 71.

Figure 71: Audio File Conversion using transcoder.x Utility Sample

To convert the audio files into different codecs you can also use *convert.sh* utility that acts as a wrapper over *transcoder.x* utility:

```
Usage: /usr/local/DNCA/bin/convert.sh <-a|-format|dst_file> <src_file> * Available formats: [ul al g729 g722 g7221 isac ilbc amrwb wav mp3] This utility allows you convert audio files not only from wav format, but from mp3 format as well.
```

For example if you would like to convert music-on-hold my\_moh.wav file from wav format into all supported formats (codecs) you should use the command (the transferred command arguments are shown in italic):

```
/usr/local/DNCA/bin/convert.sh
-a /usr/local/DNCA/var/moh/my moh.wav
```

If your source file is in *mp3* format (*my\_moh.mp3*), you can convert it into all supported formats (codecs) using the command:

```
/usr/local/DNCA/bin/convert.sh
-a /usr/local/DNCA/var/moh/my moh.mp3
```

If the command is successful, the system will not return any errors or messages; it will just return you back to the command prompt (#). The sample of the audio file conversion using *convert.sh* utility and the help on this command is shown on Figure 72.

Figure 72: Audio File Conversion using convert.sh Utility Sample

If you have placed all necessary welcome prompt audio files into *welcome-prompts* subfolder or if you have placed all necessary music-on-hold audio files into *moh* subfolder you should reload them using *mf* console command:

```
sounds-reload
```

or you should use the *wyde* command line utility with the *sounds-reload* option:

wyde sounds-reload

If any of these *mf* commands was implemented successfully you will receive the message: "Success"; as the result of these *wyde* commands implementation you will receive the message: "Reloading sound files.".

Once you reloaded your audio prompts you can use them in your call flows, DNISes and conferences. If you would like to use your own custom welcome prompt you should update <code>dnis\_welcomeprompt</code> (Welcome prompt) call flow attribute value either on call flow level or on DNIS level with your welcome prompt audio file name (without extension). If you would like to use your own custom music-on-hold prompt you should update <code>conference\_moh</code> (Music on hold) call flow attribute value either on call flow level or on DNIS level or conference account level with your music-on-hold prompt audio file name (without extension).

# Max Number of Calls and Licensing

The total number of all calls that can be processed by the WYDE bridge is being defined using system parameter maxcalls. The optimal value of this parameter depends on your WYDE bridge hardware configuration (the default value: 10). This value represents the so-called soft limit of maximal number of allowed calls; if the call that being connected exceeds this parameter value, this call receives the message: "This conference bridge is full. Please call again later." and after that it is being dropped. In addition to the soft max calls limit the bridge constrains number of calls using so-called hard limit; this limit is being calculated as maxcalls system parameter value plus 5% or at least 5 calls more; if the call that being connected exceeds this hard limit it is being immediately dropped.

Your WYDE bridge number of simultaneous calls license limits the number of calls that directly participating in the conferences (i.e. processed by *MP*). By default the WYDE Voice conferencing bridge software has the maximum number of the simultaneous calls on the bridge equal to 10. To support larger number of calls you should obtain the additional licenses. See detail information about how to obtain additional licenses in "WYDE Software Installation Guide", Chapter 3: Licensing.

When you obtain your licenses you receive generated license file *license.bin* (either if it is new license of renewed license). This file should be placed on the bridge into /usr/local/DNCA/etc folder. After that you should register new license using the following command:

wyde register-license

The screen similar to shown on Figure 73 appears next.

```
root@localhost~

[root@localhost ~] # wyde register-license

License has been registered successfully.

Now the amount of simultaneous calls is 200.

Please restart wyde service to update the latest changes.

[root@localhost ~] # □
```

Figure 73: Register New/Updated Licenses on the Bridge

Once you have updated your license the *maxcalls* system parameter value also will be updated with the amount of simultaneous calls from your license as well as the amount of allowed simultaneous calls for the WYDE Voice conferencing bridge software will be set equal to the amount specified by your license file. After that you will be able to have this number of simultaneous calls to your WYDE bridge.

## **Authorization Adapters and Methods**

In terms of WYDE bridge software the *Authorization Adapter* is the component (function) responsible for specifying access rights in the conferences. More formally, "to authorize" is to define access policy, i.e. the right to connect to the conference and specific role (host/moderator/listener) in the conference.

For authorization in the conferences are being used authorization methods. *Authorization Method* determines the specific authorization adapter and if necessary its parameters that are used to perform authorization. The authorization method name should be selected in dnis\_authorizemethod (Authorize method) call flow attribute value either on call flow or on DNIS level.

If you have your own security infrastructure the customized authorization adapter can be written to integrate your security into call flows authorization.

To see detailed information about WYDE bridge authorization process, about authorization adapters and methods see "WYDE Authorization Guide". That guide contains information not only about standard authorization adapters, but also about custom adapters as well as information how to write your own authorization adapter. You can download this guide from the WYDE Voice documentation Web site as noted in Appendix A: Support Resources, Support Documentation section.

## Add an Authorization Adapter

Before you add new authorization adapter, you should create the *Adapter Name*. pm file in the */usr/local/DNCA/lib/Auth/Adapter* folder for this adapter as it was described above.

To add new authorization adapter registration using the command line interface you should use the *wyde* command line utility with the *auth-adapter-add* option. The syntax is as follows:

wyde auth-adapter-add <arguments>

Each of the arguments is followed by a space and a value. In *auth-adapter-add* you can specify the following arguments:

- name <value> The name of the authorization adapter that should be added. This is required argument. This name should be unique, i.e. there should no be any other authorization adapter with the same name on the bridge.
- description <value> The optional description of the authorization adapter that should be added.

The arguments can be transferred to this command in any order.

Let's assume that we have created the file *WYDELdap.pm* in the folder /usr/local/DNCA/lib/Auth/Adapter for new authorization adapter *WYDELdap*. To add this adapter to the bridge you should use the command:

```
wyde auth-adapter-add name WYDELdap
  description "authorization via WYDE LDAP"
```

If the command is successful, the system will not return any errors or messages; it will just return you back to the command prompt (#). The sample of the *auth-adapter-add* command output and the help on this command is shown on Figure 74.

Figure 74: wyde help auth-adapter-add and wyde auth-adapter-add Commands Output Sample

## **Delete an Authorization Adapter**

To delete an authorization adapter using the *wyde* command line utility you should use *auth-adapter-del* option. The syntax is as follows:

wyde auth-adapter-del name <authorization adapter name>
where

• <authorization adapter name> - the name of the authorization adapter you wish to delete.

Note that you can delete only authorization adapters that are not in use, i.e. there should no be any authorization methods that refer to this authorization adapter. If the authorization adapter is used by any authorization method you will receive the error and the deletion will be cancelled.

For example to delete authorization adapter *VSRRadius* you should run the command: wyde auth-adapter-del name *VSRRadius* 

If deletion is successful, you will be returned to the command line with no additional prompts.

### **View Authorization Adapters**

To show a list of all authorization adapters in the system using the command line, you should use the *wyde* command line utility with the *auth-adapter-show* option. The syntax is as follows:

```
wyde auth-adapter-show
```

This command will output a list of the all existed authorization adapters on the system, similar to shown on Figure 75. As you can see, the *wyde auth-adapter-show* command shows the authorization adapters that have been created in the system as well as their basic properties: authorization adapter name and description.

Figure 75: wyde auth-adapter-show Command Output Sample

#### Add an Authorization Method

To create new authorization method for the authorization adapter using the command line interface you should use the *wyde* command line utility with the *auth-method-add* option. The syntax is as follows:

wyde auth-method-add <arguments>

Each of the arguments is followed by a space and a value. In *auth-method-add* you can specify the following arguments:

- name <value> The name of the authorization method that should be added.
- description <value> The description of the authorization method that should be added.
- adapter <value> The authorization adapter name for the authorization method that should be added.
- parameters <value> The list of parameters for the authorization method that should be added. The parameters are specific for authorization adapters that are being used: *Free* and *LocalDb* authorization adapters do not require any parameters; for *WYDELdap* it is the string that defines the list of LDAP servers separated by semicolon (;), and each of these servers is defined as <server IP>: [<server port>]:<password>:<LDAP root DN path> (default port is 389, DN path means distinguished name of the LDAP folder that

(default port is 389, DN path means distinguished name of the LDAP folder that contains conference authorization info); for WYDERadius it is the string that defines the list of RADIUS servers separated by semicolon (;), and each of these servers is defined as separated port>] (default port is 1812).

Arguments name and adapter are required. The arguments can be transferred to this command in any order.

For example if you would like to create the authorization method wydeldap for the authorization adapter WYDELdap with description "Authorization via WYDE LDAP – sample" and parameters "localhost::test:dc=wydevoice,dc=com" you should run the following command (new authorization method properties are shown in italic): wyde auth-method-add name wydeldap adapter WYDELdap description "Authorization via WYDE LDAP - sample" parameters "localhost::test:dc=wydevoice,dc=com"

Note that to set the description that contains spaces and parameters you should use double quotes (").

If the command is successful, the system will not return any errors or messages; it will just return you back to the command prompt (#). The sample of the *auth-method-add* command output and the help on this command is shown on Figure 76.

Figure 76: wyde help auth-method-add and wyde auth-method-add Commands Output Sample

#### **Delete an Authorization Method**

If you wish to delete the specific authorization method, you can use the *wyde* command line utility with *auth-method-del* option. The syntax is as follows:

wyde auth-method-del name <authorization method name> where

• <authorization method name> - The name of the authorization method that should deleted. This argument is required.

Note that you can delete only authorization methods that are not in use. If the method is used by any call flow and/or DNIS you will receive the error: "<authorization method name>: Authorization method is in use and can not be removed.".

For example to delete the authorization method *vsrradius* you should run the command: wyde auth-method-del name *vsrradius* 

If deletion is successful, you will be returned to the command line with no additional prompts.

## **Modify an Authorization Method**

To modify authorization method properties, such as description and parameters, using the command line interface you should use the *wyde* command line utility with the *auth-method-set* option. The syntax is as follows:

```
wyde auth-method-set <arguments>
```

Each of the arguments is followed by a space and a value. In *auth-method-set* you can specify the following arguments:

- name <value> The name of the authorization method that should be changed.
- description <value> New description of the authorization method that should be set.
- parameters <value> New list of parameters for the authorization method that should be set.

The argument name is required; you should specify arguments description and parameters only if you would like to change them. The arguments can be transferred to this command in any order.

For example if you would like to change *wyderadius* authorization method and set its description equal to "*Authorization via WYDE RADIUS server - sample*" and it parameters equal to "*test@192.168.1.41:1812*", you should run the following command (the transferred command arguments are shown in *italic*):

```
wyde auth-method-set name wyderadius
description "Authorization via WYDE RADIUS server - sample"
parameters "test@192.168.1.41:1812"
```

If the command is successful, the system will not return any errors or messages; it will just return you back to the command prompt (#). The sample of the *auth-method-set* command output and the help on this command is shown on Figure 77.

Figure 77: wyde help auth-method-set and wyde auth-method-set Commands Output Sample

#### **View Authorization Methods**

To show a list of all authorization methods in the system using the command line, you should use the *wyde* command line utility with the *auth-method-show* option. The syntax is as follows:

```
wyde auth-method-show
```

This command will output a list of the all existed authorization methods on the system, similar to shown on Figure 78. As you can see, the *wyde auth-method-show* command shows the authorization methods that have been created in the system as well as their basic properties: authorization method name, adapter, parameters, and description.

```
Parameters
                    Adapter
                                                                                     Description
                                                                                      Without authorization
local
                    Local Db
                                                                                      Authorization via local database
                                         localhost::test:dc=wydevoice,dc=com Authorization via WYDE LDAP - sample test@192.168.1.41:1812 Authorization via WYDE RADIUS server
wydeldap
                    WYDELdap
wyderadius
                    WYDERadius
                                         test2@192.168.1.42:1812
                                                                                      Authorization via WYDE RADIUS - sample 2
                    WYDERadius
[root@ZILBER ~]#
```

Figure 78: wyde auth-method-show Command Output Sample

# **Billing**

For billing purposes the WYDE bridge software can store and transmit CDRs (Call Detail Records). Please note, that the WYDE bridge software is not responsible for financial billing; it neither tracks credit cards nor sends invoices to the clients. It only provides CDR data and it is up to you how to use them in your financial billing.

WYDE bridge *MF* service writes CDR information about completed calls into internal journal; such CDRs journal is being created for each billing adapter. That journal could be accessed by the billing adapter to catch and store CDR information.

In terms of WYDE bridge software the *Billing Adapter* is the component (function) responsible for storing billing, i.e. CDR information. Billing adapter processes information from the internal CDRs journal created by *MF* service; in this journal *MF* keeps information about completed calls; billing adapter receives this information, transforms it into required format and stores it in required data carrier. More formally, "billing" is the information about completed conferences and calls also called CDR.

For the conference billing information storing are being used billing rules. *Billing rule* determines the specific billing adapters that are used to save CDR information; this includes comma-separated billing adapters that should be used in the specific billing rule. The billing rule name should be selected in dnis\_billingrule (Billing rule) call flow attribute value either on call flow or on DNIS level.

If you would like to store CDR information in your own database you can create your custom billing adapter that will be responsible for saving CDR data as it is required for your organization.

To see detailed information about WYDE bridge billing, about billing adapters and rules see "WYDE Billing Guide". That guide contains information not only about standard billing adapters, but also about custom adapters as well as information how to write your own billing adapter. You can download this guide from the WYDE Voice documentation Web site as noted in Appendix A: Support Resources, Support Documentation section.

#### Add a Billing Adapter

Before you add new billing adapter, you should create the *Adapter Driver*>.pm driver file in the */usr/local/DNCA/lib/Billing/Adapter* folder for this adapter driver as it was described above.

To add new billing adapter registration using the command line interface you should use the *wyde* command line utility with the *billing-adapter-add* option. The syntax is as follows:

```
wyde billing-adapter-add <arguments>
```

Each of the arguments is followed by a space and a value. In *billing-adapter-add* you can specify the following arguments:

- name <value> The name of the billing adapter that should be added. This name should be unique, i.e. there should no be any other billing adapter with the same name on the bridge.
- description <value> The optional description of the billing adapter that should be added.
- driver <value> The driver name for the billing adapter that should be added. The file <*Adapter Driver Name*>.*pm* should exist in the /*usr/local/DNCA/lib/Billing/Adapter* folder.
- parameters <value> The list of parameters for the billing adapter that should be added.

Arguments name, driver and parameters are required. The arguments can be transferred to this command in any order.

Let's assume that we have created the file *TFCC.pm* in the folder /usr/local/DNCA/lib/Billing/Adapter for new billing adapter tfcc, parameters that should be used are 192.168.1.45:9000. To add this billing adapter to the bridge you should use the command:

```
wyde billing-adapter-add name tfcc driver TFCC description "Billing using TFCC - sample" parameters 192.168.1.45:9000
```

If the command is successful, the system will not return any errors or messages; it will just return you back to the command prompt (#). The sample of the *billing-adapter-add* command output and the help on this command is shown on Figure 79.

```
_ | U ×
[root@localhost ~] # wyde help billing-adapter-add
    wyde billing-adapter-add arguments
Arguments:
       * name <value>

    Adapter name.

         description <value>
                               - Adapter description.
        * driver <value>
                               - Driver name.
        * parameters <value>
                                - Parameters list.
Required arguments marked as '*'.
[root@localhost ~]# wyde billing-adapter-add name tfcc driver TFCC description
Billing using TFCC - sample" parameters 192.168.1.45:9000
[root@localhost ~]#
```

Figure 79: wyde help billing-adapter-add and wyde billing-adapter-add Commands Output Sample

#### **Delete a Billing Adapter**

To delete a billing adapter using the *wyde* command line utility you should use *billing-adapter-del* option. The syntax is as follows:

wyde billing-adapter-del name <billing adapter name>
where

• <billing adapter name> – the name of the billing adapter you wish to delete. Note that you can delete billing adapters that are not in use only, i.e. there should no be any billing rules that refer to this billing adapter. If the billing adapter is used by any billing rule you will receive the error message: "<br/>
billing adapter name>: Billing adapter is in use and can not be removed." and the deletion will be cancelled.

For example to delete billing adapter *tfcc* (created in previous sample) you should run the command:

```
wyde billing-adapter-del name tfcc
```

If deletion is successful, you will be returned to the command line with no additional prompts.

## **Modify a Billing Adapter**

To modify billing adapter properties, such as description, driver and parameters, using the command line interface you should use the *wyde* command line utility with the *billing-adapter-set* option. The syntax is as follows:

```
wyde billing-adapter-set <arguments>
```

Each of the arguments is followed by a space and a value. In *billing-adapter-set* you can specify the following arguments:

- name <value> The name of the billing adapter that should be changed.
- description <value> New description of the billing adapter that should be set.
- driver <value> New driver name for the billing adapter that should be set.
- parameters <value> New list of parameters for the billing adapter that should be set.

The argument name is required; you should specify other arguments only if you would like to change them. The arguments can be transferred to this command in any order.

For example if you would like to change *tfcc* billing adapter and set its parameters equal to "192.168.1.46:9000", you should run the following command (the transferred command arguments are shown in *italic*):

```
wyde billing-adapter-set name tfcc parameters 192.168.1.46:9000
```

If the command is successful, the system will not return any errors or messages; it will just return you back to the command prompt (#). The sample of the *billing-adapter-set* command output and the help on this command is shown on Figure 80.

Figure 80: wyde help billing-adapter-set and wyde billing-adapter-set Commands Output Sample

# **View Billing Adapters**

To show a list of all billing adapters in the system using the command line, you should use the *wyde* command line utility with the *billing-adapter-show* option. The syntax is as follows:

```
wyde billing-adapter-show
```

This command will output a list of the all existed billing adapters on the system, similar to shown on Figure 81. As you can see, the *wyde billing-adapter-show* command shows the billing adapters that have been created in the system as well as their basic properties: billing adapter name, driver, parameters and description.

Figure 81: wyde billing-adapter-show Command Output Sample

#### Add a Billing Rule

To create new billing rule for the billing adapter using the command line interface you should use the *wyde* command line utility with the *billing-rule-add* option. The syntax is as follows:

```
wyde billing-rule-add <arguments>
```

Each of the arguments is followed by a space and a value. In *billing-rule-add* you can specify the following arguments:

- name <value> The name of the billing rule that should be added.
- description <value> The description of the billing rule that should be added.
- rule <value> The billing rule content that should be added, comma-separated values: {adapter1[,adapter2[,adapter3[,...]]]}:
  - file denotes that CDR information should be stored into /usr/local/DNCA/log/CDR.log file using the billing adapter: file;

- o localdb denotes that CDR information should be stored into *dnca\_calls* local database using the billing adapter: localdb;
- o <adapter\_name>[, . . . ] denotes any other custom billing adapters comma-separated names that should be used to store CDR information.

For example rule argument could be defined as file, localdb or file, localdb, myAdapter.

Arguments name and rule are required. The arguments can be transferred to this command in any order.

For example if you would like to create the billing rule *custom* that defines that CDR records should be stored into file, local *dnca\_calls* database and using *tfcc* billing adapter with description "*Store CDRs in file, local database and via TFCC*" you should run the following command (new billing rule properties are shown in *italic*):

```
wyde billing-rule-add name custom
  description "Store CDRs in file, local database and via TFCC"
  rule "file, localdb, tfcc"
```

Note that to set the description that contains spaces and parameters you should use double quotes (").

If the command is successful, the system will not return any errors or messages; it will just return you back to the command prompt (#). The sample of the *billing-rule-add* command output and the help on this command is shown on Figure 82.

Figure 82: wyde help billing-rule-add and wyde billing-rule-add Commands Output Sample

#### **Delete a Billing Rule**

If you wish to delete the specific billing rule, you can use the *wyde* command line utility with *billing-rule-del* option. The syntax is as follows:

```
wyde billing-rule-del name <billing rule name>
where
```

• <billing rule name> - The name of the billing rule that should deleted. This argument is required.

Note that you can delete only billing rules that are not in use. If the rule is used by any call flow and/or DNIS you will receive the error: "<br/>
billing rule name>: Billing rule is in use and can not be removed." and the deletion will be cancelled.

For example to delete the billing rule *custom* you should run the command: wyde billing-rule-del name *custom* 

If deletion is successful, you will be returned to the command line with no additional prompts.

## Modify a Billing Rule

To modify billing rule properties, such as description and rule content, using the command line interface you should use the *wyde* command line utility with the *billing-rule-set* option. The syntax is as follows:

```
wyde billing-rule-set <arguments>
```

Each of the arguments is followed by a space and a value. In *billing-rule-set* you can specify the following arguments:

- name <value> The name of the billing rule that should be updated.
- description <value> New description of the billing rule that should be set.
- rule <value> New billing rule content that should be set, comma-separated values: {adapter1[,adapter2[,adapter3[,...]]]}:
  - file denotes that CDR information should be stored into /usr/local/DNCA/log/CDR.log file using the billing adapter: file;
  - o localdb denotes that CDR information should be stored into *dnca\_calls* local database e using the billing adapter: localdb;
  - o <adapter\_name>[,...] denotes any other custom billing adapters comma-separated names that should be used to store CDR information.

For example rule argument could be defined as file, localdb or file, localdb, myAdapter.

The argument name is required; you should specify arguments description and rule only if you would like to change them. The arguments can be transferred to this command in any order.

For example if you would like to change *default* billing rule and set its description equal to "*Default billing rule - sample*", you should run the following command (the transferred command arguments are shown in *italic*):

```
wyde billing-rule-set name default
  description "Default billing rule - sample"
```

If the command is successful, the system will not return any errors or messages; it will just return you back to the command prompt (#). The sample of the *billing-rule-set* command output and the help on this command is shown on Figure 83.

Figure 83: wyde help billing-rule-set and wyde billing-rule-set Commands Output Sample

## **View Billing Rules**

To show a list of all billing rules in the system using the command line, you should use the *wyde* command line utility with the *billing-rule-show* option. The syntax is as follows: wyde billing-rule-show

This command will output a list of the all existed billing rules on the system, similar to shown on Figure 84. As you can see, the *wyde billing-rule-show* command shows the billing rules that have been created in the system as well as their basic properties: billing rule name, rule content, and description.

```
| root@ZILBER:~ | # wyde billing-rule-show | A | Name | Rule | Description | Custom | File, localdb, tfcc | Store CDRs in file, local database and via TFCC | default | file, localdb | Default billing rule - sample | Custom | File, localdb | Custom | Custo
```

Figure 84: wyde billing-rule-show Command Output Sample

#### **Database Administration**

The WYDE bridge administration also can be made using the *PostgreSQL* interactive terminal (i.e. database console). This is standard external tool to work with *PostgreSQL* databases.

The following databases are being used by the WYDE bridge software:

- dnca the main database;
- dnca calls the billing database.

To perform initialization of the databases, i.e. database creation, necessary tables creation and population using the command line, you should use the *wyde* command line utility with the *db-init* option. The syntax is as follows:

```
wyde db-init [database {Main|Billing}]
where
```

- database {Main|Billing} Denotes what kind of database should be initialized:
  - o *Main* is being used for *dnca* database.
  - o Billing is being used for dnca calls database.

When the database argument is omitted the main (dnca) database is being initialized.

Before the initialization of any of the databases it should be deleted using the *wyde* command line utility with the *db-drop* option. The syntax is as follows:

wyde db-drop [database {Main|Billing}]

where

- database {Main|Billing} Denotes what kind of database should be dropped:
  - o *Main* is being used for *dnca* database.
  - o Billing is being used for dnca calls database.

When the database argument is omitted the main (dnca) database is being dropped.

Usually you do not need to run the commands wyde db-init and wyde db-drop. If needed these commands are being run automatically during new system installation. However if you need to run them you should use the command wyde db-drop first and wyde db-init next. If you try to initialize the database that already has been initialized you receive the error messages "Main database already exists. If you want to reinitialize the main database you should run 'wyde db-drop database Main' before." or "Billing database already exists. If you want to reinitialize the billing database you should run 'wyde db-drop database Billing' before."

Prior to database removing you should stop WYDE services using the command: service wyde stop

and after that you should start PostgreSQL using the command:

```
service postgresql start
```

Next you can delete necessary databases. If you are going to drop both (main and billing) databases you should drop the billing databases first and the main database next, because the main database contains the billing database configuration parameters. For example to delete and re-initialize the billing database you should run the commands:

```
wyde db-drop database Billing
wyde db-init database Billing
```

See Figure 85 for details. As you can see prior to the database deletion the system asks to confirm that you wish to drop the database.

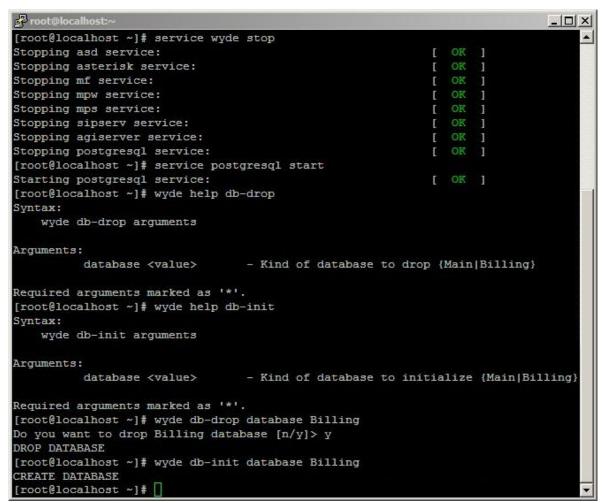


Figure 85: Drop and Initialize the Billing Database

If you drop and initialize the main database after this initialization you should add the local bridge definition using the *wyde* command line utility with the *bridge-add* option, next you should populate default settings for this bridge using the *wyde* command line utility with the *settings-update* option, after that you should add your node using the *wyde* command line utility with the *node-add* option, and next you should add call flows definitions using the *wyde* command line utility with the *callflow-add* option. For example the following commands should be run to drop and re-initialize the main database:

```
wyde db-drop Main
wyde db-init Main
wyde bridge-add id 1 name WYDE31 ip_addr 192.168.1.31
wyde settings-update
wyde node-add bridge WYDE31 name AST31
        ivr_addr 192.168.1.31:4460 sip_addr 192.168.1.31:5060
        ivr_maxcalls 500
wyde callflow-add name CONF
        directory /usr/local/DNCA/callflows/CONF
wyde callflow-add name OPERATOR
        directory /usr/local/DNCA/callflows/OPERATOR
```

```
wyde callflow-add name PLAYBACK
  directory /usr/local/DNCA/callflows/PLAYBACK
wyde callflow-add name SPECTEL
  directory /usr/local/DNCA/callflows/SPECTEL
```

As it was previously told the bridge identifier should be between 1 and 15. Note that the local bridge identifier that you use in *bridge-add* command should be the same with the default bridge defined in /usr/local/DNCA/etc/dnca.conf file:

```
[general]
bridge id = 1
```

See Figure 86 for details.

```
_ | U ×
[root@localhost ~] # wyde db-drop database Main
Do you want to drop Main database [n/y]> y
DROP DATABASE
[root@localhost ~] # wyde db-init database Main
CREATE DATABASE
[root@localhost ~] # wyde bridge-add id 1 name WYDE31 ip addr 192.168.1.31
[root@localhost ~] # wyde settings-update
Update settings for WYDE31
-- add : billing localdb name = dnca calls
-- add : mp sip addr = 192.168.1.31:5160
-- add : ivr rec addr =
 - add : web strict = true
 - add : billing localdb passwd = wyde4db
 - add : mp keepalive timeout = 300
 - add : billing_localdb_storing_period = 180
 - add : mf_agi_addr = 192.168.1.31:4450
 - add : maxcalls = 500
 - add : mf dc = 0
 - add : billing localdb host = 192.168.1.31
 - add : mp max calls = 100
 - add : mf_event_qdisc = 4:20
 - add : callspersecond = 100
 - add : mf console addr = 192.168.1.31:4440
 - add : agiserver addr = 192.168.1.31:5050
 - add : mp_console addr = 192.168.1.31:4545
 - add : mf dc connector addr = 192.168.1.31:4480
 - add : operator invite =
 - add : billing localdb user = dnca
 - add : prefered_codec = ul
 - add : ivr checkpeer = 0
 - add : mf localdb pool size = 5
 - add : mp_as_granularity = 500
 - add : mp_log_file = mp.log
 - add : billing_split_nodes = 0
 - add : ivr gw prefix =
 - add : mf_agi_pool_size = 10
 - add : ivr gw addr =
 - add : agiserver_pool_size = 5
-- add : crash notify = wyde@freeconferencecall.com
[root@localhost ~] # wyde node-add bridge WYDE31 name AST31 ivr addr 192.168.1.31
:4460 sip addr 192.168.1.31:5060 ivr maxcalls 500
[root@localhost ~]#
```

Figure 86: Drop and Initialize the Main Database

To apply the latest patches (changes) to the existing databases using the command line, you should use the *wyde* command line utility with the *db-patch* option. The syntax is as follows:

```
wyde db-patch
```

Usually you also do not need to run this command. This command is being run automatically during the system upgrade installation.

To connect to the WYDE *dnca* (main) database, i.e. to open the PostgreSQL interactive terminal (*psql*) for this database using the command line, you should use the *wyde* command line utility with the *db* option. The syntax is as follows:

wyde db

To connect to the WYDE *dnca\_calls* (billing) database, i.e. to open the PostgreSQL interactive terminal (*psql*) for this database using the command line, you should use the *wyde* command line utility with the *db-bill* option. The syntax is as follows: wyde db-bil

#### Asterisk Administration

In the section Conferences and Calls Management using *asterisk* Console we described how to run the *asterisk* console and how to manage the conferences and calls using this console. This section describes what additional changes and adjustments could be made in the *asterisk* configuration.

# **SIP Listening Host IP and Port Configuration**

By default the SIP is being configured to use 0.0.0.0 as default IP address to bind to and 5060 as default UDP port to bind to. IP address 0.0.0.0 means that *asterisk* binds to all available nodes. This is being configured in the configuration file /opt/DNCA/asterisk/etc/asterisk/sip.conf as follows:

```
[general]
...
bindport=5060 ; UDP Port to bind to (SIP standard port is 5060)
bindaddr=0.0.0.0 ; IP address to bind to (0.0.0.0 binds to all)
...
```

If it is necessary you can configure SIP to listen on non-default IP address and/or port by changing these configuration parameters. When you are changing these configuration parameters you should also update SIP address of your nodes using *wyde node-set* command with *sip addr* argument.

For example if you would like to configure SIP to listen on the port 5061 you should change *sip.conf* file to have the line:

```
bindport=5061
```

After that you should run the following command (we assume that this command is being run for the *asterisk* node *ASTI*):

```
wyde node-set name AST1 sip_addr 192.168.1.5:5061
```

Next you should either restart *asterisk* or run IVR console and reload the SIP using the commands:

wyde ivr
sip reload

See Figure 87 for details.

```
[root@ZILBER ~] # wyde node-set name AST1 sip_addr 192.168.1.5:5061
[root@ZILBER ~] # wyde node-show
                                 IVR Addr
Bridge
                Node
                                                    SIP Addr
                                                                      MaxCalls
                                                                                        Zone
                                  192.168.1.5:4460 192.168.1.5:5061 500
                 AST1
[root@ZILBER ~]# wyde ivr
Asterisk 1.6.2.8, Copyright (C) 1999 - 2010 Digium, Inc. and others.
Created by Mark Spencer <markster@digium.com>
Asterisk comes with ABSOLUTELY NO WARRANTY; type 'core show warranty' for details.
This is free software, with components licensed under the GNU General Public
License version 2 and other licenses; you are welcome to redistribute it under
certain conditions. Type 'core show license' for details.
 == Parsing '/opt/DNCA/asterisk/etc/asterisk/asterisk.conf': == Found
== Parsing '/opt/DNCA/asterisk/etc/asterisk/extconfig.conf': == Found
Connected to Asterisk 1.6.2.8 currently running on ZILBER (pid = 31459)
Verbosity was 0 and is now 3
ZILBER*CLI> sip reload
Reloading SIP
 == Parsing '/opt/DNCA/asterisk/etc/asterisk/sip.conf': == Found
 == Parsing '/opt/DNCA/asterisk/var/spool/asterisk/ivrhelper/peers.conf':
                                                                                == Found
 == Parsing '/opt/DNCA/asterisk/etc/asterisk/users.conf': == Found
 == Parsing '/opt/DNCA/asterisk/etc/asterisk/sip_notify.conf': == Found
ZILBER*CLI> quit
Executing last minute cleanups
[root@ZILBER ~]#
```

Figure 87: Changing SIP Bind Address and Port

## MF Connection on Non-default Host IP and Port Configuration

By default the connection from *mf* service is being expected at 0.0.0.0:4460. But if it is necessary another host IP and/or port can be set in the configuration file /opt/DNCA/asterisk/etc/asterisk/wvde.conf:

```
[general]
bind addr=<ip>:<port>
```

Using this configuration settings *ivrhelper* could be configured to another IP address and port.

If it is necessary you can configure *ivrhelper* to listen on non-default IP address and/or port by changing these configuration parameters. When you are changing them you should also update IVR address of your nodes using *wyde node-set* command with *ivr addr* argument.

For example if you would like to configure IVR to listen on the port 4461 you should change wyde.conf file to have the line:

```
bind addr=0.0.0.0:4461
```

After that you should run the following command (we assume that this command is being run for the *asterisk* node *ASTI*):

```
wyde node-set name AST1 ivr addr 192.168.1.5:4461
```

Next you should restart *asterisk* service using the command:

```
service asterisk restart
```

See Figure 88 for details.

Figure 88: Changing IVR Bind Address and Port

#### **Enable and Disable Audio Codecs**

Audio codecs that are being used by the *asterisk* are listed in the file */opt/DNCA/asterisk/etc/asterisk/sip.conf* as follows:

```
disallow=all
allow=g722
allow=isac
allow=amrwb
allow=siren7
allow=siren14
allow=ulaw
allow=alaw
allow=ilbc
allow=novoice
```

These are the audio codecs that are supported by the WYDE bridge. The codecs *ulaw*, *alaw*, *ilbc* are 8 KHz codecs; the codecs *g722*, *isac*, *amrwb*, *siren7*, *siren14* are 16 KHz HD codecs; the codec *novoice* is the codec for control calls. By default *sip.conf* configuration file disallows all codecs (the line: *disallow=all*); after that *allow* parameters define what audio codecs are allowed by the bridge. In this section of the file you can list the codecs that you are going to use.

If you changed the list of codecs in the *sip.conf* file you should either restart *asterisk* using the command:

```
service asterisk restart or run IVR console and reload the SIP using the commands: wyde ivr sip reload
```

System settings of the WYDE bridge contain *prefered\_codec* parameter; this parameter determines preferred codec that is being used when the bridge initiates the calls, i.e. dialout calls, calls to the recording server, etc. To see your preferred codec settings you can use the command:

```
wyde settings-show prefix prefered codec
```

# **Chapter 3: Command Reference**

# wyde Command Reference

All wyde commands have the following format: wyde command [arguments]

This chapter describes all available *wyde* commands with their arguments. For any of the commands the required arguments marked with asterisk (\*).

## ast-status (Show WYDE asterisk Status)

Syntax:

wyde ast-status

## auth-adapter-add (Add auth Adapter)

Syntax:

wyde auth-adapter-add arguments

Arguments:

name <value> - The name of the authorization adapter that should be added (\*); description <value> - The description of the authorization adapter that should be added.

# auth-adapter-del (Delete auth Adapter)

Syntax:

wyde auth-adapter-del arguments

Arguments:

name <value> - The name of the authorization adapter that should deleted (\*).

# auth-adapter-show (Show auth Adapters)

Svntax:

wyde auth-adapter-show

## auth-method-add (Add auth Method)

Svntax:

 $\label{eq:wyde} \mbox{wyde auth-method-add} \ \ \mbox{arguments}$ 

Arguments:

name <value> - The name of the authorization method that should be added (\*);
description <value> - The description of the authorization method that should
be added;

adapter <value> - The authorization adapter name for the authorization method that should be added (\*);

parameters <value> - The list of parameters for the authorization method that should be added.

# auth-method-del (Delete auth Method)

```
Syntax:
```

wyde auth-method-del arguments

Arguments:

name <value> - The name of the authorization method that should deleted (\*).

## auth-method-set (Set auth Method)

#### Syntax:

wyde auth-method-set arguments

Arguments:

name <value> - The name of the authorization method that should be changed (\*);
description <value> - New description of the authorization method that should
be set:

parameters <value> - New list of parameters for the authorization method that should be set.

## auth-method-show (Show auth Methods)

#### Syntax:

wyde auth-method-show

## auth-reload (Reload auth configuration)

#### Syntax:

wyde auth-reload

# billing-adapter-add (Add Billing Adapter)

#### Syntax:

wyde billing-adapter-add arguments

#### Arguments:

name <value> - The name of the billing adapter that should be added (\*);
description <value> - The description of the billing adapter that should be
added;

driver <value> - The driver name for the billing adapter that should be added (\*);
parameters <value> - The list of parameters for the billing adapter that should
be added (\*).

## billing-adapter-del (Delete Billing Adapter)

#### Syntax:

wyde billing-adapter-del arguments

Arguments:

name <value> - The name of the billing adapter that should deleted (\*).

# billing-adapter-set (Set Billing Adapter Properties)

#### Syntax:

wyde billing-adapter-set arguments

#### Arguments:

name <value> - The name of the billing adapter that should be updated (\*);
description <value> - New description of the billing adapter that should be set;
driver <value> - New driver name for the billing adapter that should be set;
parameters <value> - New list of parameters for the billing adapter that should
be set.

# billing-adapter-show (Show Billing Adapters)

#### Syntax:

wyde billing-adapter-show

# billing-reload (Reload Billing Configuration)

#### Syntax:

wyde billing-reload

# billing-rule-add (Add Billing Rule)

#### Syntax:

wyde billing-rule-add arguments
Arguments:

name <value> - The name of the billing rule that should be added (\*);
description <value> - The description of the billing rule that should be added;
rule <value> - The billing rule content that should be added (\*), comma-separated
 values: {adapter1[,adapter2[,adapter3[,...]]]}:

- o file denotes that CDR information should be stored into /usr/local/DNCA/log/CDR.log file using the billing adapter: file;
- o localdb denotes that CDR information should be stored into *dnca\_calls* local database using the billing adapter: localdb;
- o <adapter\_name>[,...] denotes any other custom billing adapters comma-separated names that should be used to store CDR information.

For example rule argument could be defined as file, localdb or file, localdb, myAdapter.

# billing-rule-del (Delete Billing Rule)

#### Syntax:

wyde billing-rule-del arguments
Arguments:

name <value> - The name of the billing rule that should deleted (\*).

# billing-rule-set (Set Billing Rule)

#### Syntax:

```
wyde billing-rule-set arguments
Arguments:
```

name <value> - The name of the billing rule that should be updated (\*);
description <value> - New description of the billing rule that should be set;

```
rule <value> - New billing rule content that should be set, comma-separated
  values: {adapter1[,adapter2[,adapter3[,...]]]}:
```

- file denotes that CDR information should be stored into /usr/local/DNCA/log/CDR.log file using the billing adapter: file;
- o localdb denotes that CDR information should be stored into *dnca\_calls* local database e using the billing adapter: localdb;
- o <adapter\_name>[,...] denotes any other custom billing adapters comma-separated names that should be used to store CDR information.

For example rule argument could be defined as file, localdb or file, localdb, myAdapter.

## billing-rule-show (Show Billing Rules)

```
Syntax:
```

wyde billing-rule-show

# bridge-add (Add WYDE Bridge)

Syntax:

wyde bridge-add arguments
Arguments:

id <value> - The identifier of the bridge that should be added (\*); name <value> - The name of the bridge that should be added (\*);

ip addr <value> - IP address of the bridge that should be added (\*);

rx\_gain <value> - Received audio gain for the bridge; could be from -10 till 10 or 255; it indicates what volume level should be set for the call, 255 denotes that the volume level is being automatically generated by the backend;

tx\_gain <value> - Transmitted audio gain for the bridge; could be from -10 till 10 or 255; it indicates what volume level should be set for the call, 255 denotes that the volume level is being automatically generated by the backend.

# bridge-del (Delete WYDE Bridge)

Syntax:

wyde bridge-del arguments
Arguments:

id <value> - The identifier of the bridge that should be deleted (\*).

# bridge-list-load (Load List of Bridges from the File)

Syntax:

wyde bridge-list-load arguments
Arguments:

file <value> - The input file name to load the list of bridges (\*).

# bridge-list-save (Save List of Bridges to the File)

Syntax:

wyde bridge-list-save arguments

## Arguments:

file <value> - The output file name to save the list of bridges (\*).

# bridge-reload (Reload DC Bridges Configuration)

## Syntax:

wyde bridge-reload

# bridge-set (Set WYDE Bridge Properties)

#### Syntax:

wyde bridge-set arguments

# Arguments:

id <value> - The identifier of the bridge which properties should be updated (\*);

name <value> - The name of the bridge that should be updated;

ip\_addr <value> - IP address of the bridge that should be updated;

new\_id <value> - New identifier of the bridge if it should be changed;

rx\_gain <value> - Received audio gain for the bridge that should be updated; could be from -10 till 10 or 255; it indicates what volume level should be set for the call, 255 denotes that the volume level is being automatically generated by the backend;

tx\_gain <value> - Transmitted audio gain for the bridge that should be updated; could be from -10 till 10 or 255; it indicates what volume level should be set for the call, 255 denotes that the volume level is being automatically generated by the backend.

## bridge-show (Show WYDE Bridges)

#### Syntax:

wyde bridge-show

# caller-gain-set (Set Gain Options)

#### Syntax:

wyde caller-gain-set arguments
Arguments:

opt name <value> - The option name

{calling number|called number|conf number} (\*):

- o calling\_number denotes that the caller gain should be set for the calling number (the caller phone number) defined by opt value argument;
- o called\_number denotes that the caller gain should be set for the called number (DNIS) defined by opt value argument;
- o conf\_number denotes that the caller gain should be set for the conference number defined by opt\_value argument;

```
opt_value <value> - The option value (*);
```

rx\_gain <value> - Received audio gain; could be from -10 till 10 or 255; it indicates what volume level should be set for the call, 255 denotes that the volume level is being automatically generated by the backend;

tx\_gain <value> - Transmitted audio gain; could be from -10 till 10 or 255; it indicates what volume level should be set for the call, 255 denotes that the volume level is being automatically generated by the backend.

# caller-gain-show (Show Gain Options)

```
Syntax:
```

wyde caller-gain-show

## callflow-add (Add Call Flow)

#### Syntax:

wyde callflow-add arguments
Arguments:
 name <value> - The name of the call flow (\*);
 directory <value> - Call flow directory path.

# callflow-attr-set (Set Call Flow Attribute)

#### Syntax:

wyde callflow-attr-set arguments
Arguments:
 callflow <value> - The name of the call flow (\*);
 name <value> - This call flow attribute name (\*);
 value <value> - This call flow attribute value.

# callflow-attr-show (Show Call Flow Attributes)

#### Syntax:

wyde callflow-attr-show arguments
Arguments:
 callflow <value> - The name of the call flow (\*).

# callflow-attr-update-db (Update Call Flow Attributes Definition in a Database)

#### Syntax:

wyde callflow-attr-update-db arguments
Arguments:
 callflow <value> - The name of the call flow (\*).

#### callflow-del (Delete Call Flow)

#### Syntax:

wyde callflow-del arguments
Arguments:
 name <value> - The name of the call flow (\*).

## callflow-reload (Reload All Call Flows)

```
Syntax:
```

wyde callflow-reload

*Note:* 

This command works the same as *mf* console command: callflow-reload (Reload Call Flows).

## callflow-show (Show Call Flows Table)

```
Syntax:
```

wyde callflow-show

## conference-attr-del (Remove Conference Attribute Redefinition)

```
Syntax:
```

```
wyde conference-attr-del arguments
Arguments:
   number <value> - The conference number (*);
   name <value> - This conference call flow attribute name (*).
```

## conference-attr-set (Set Conference Attribute)

```
Syntax:
```

```
wyde conference-attr-set arguments
Arguments:
   number <value> - The conference number (*);
   name <value> - This conference call flow attribute name (*);
   value <value> - This conference call flow attribute new value (*).
```

#### conference-attr-show (Show Conference Attributes)

Syntax:

```
wyde conference-attr-show arguments
Arguments:
   number <value> - The conference number (*).
```

# config-restore (Restore WYDE Configuration)

Syntax:

```
wyde config-restore arguments
Arguments:
    file <value> - The file name to restore configuration settings (*);
    force <value> - Force required { yes | no } (default no):
```

- yes denotes that the configuration file should be restored even if the versions of previously saved configuration data and currently installed WYDE software are different:
- no denotes that the configuration file should be restored only if the versions of previously saved configuration data and currently installed WYDE software are the same.

# config-save (Save WYDE Configuration)

```
Syntax:
```

wyde config-save arguments

Arguments:

file <value> - The file name to save configuration settings (\*).

## confuser-add (Add Conference User)

#### Syntax:

wyde confuser-add arguments

Arguments:

access code <value> - Access code, if this parameter is omitted unique numeric access code will be generated;

did <value> - DID (DNIS) number (\*);

conference <value> - Conference number, if this parameter is omitted unique numeric access code will be generated;

conf-accesscode – Set conference number the same as access code, if this parameter is transferred the conference number will be set equal to access code (if conference and conf-accesscode parameters both transferred, the conference parameter will be ignored and the conference number will be set equal to the access code);

role <value> - Role in the conference (case-sensitive): Host or Participant or
Listener (\*);

subscriber <value> - Subscriber login name, i.e. subscriber PIN (\*).

## confuser-del (Delete Conference User)

#### Syntax:

```
wyde confuser-del arguments Arguments:
```

accesscode <value> - Access code (\*).

# confuser-show (Show Conference Users Table)

#### Syntax:

```
wyde confuser-show arguments
```

#### Arguments:

```
subscriber <value> - Subscriber login name, i.e. subscriber PIN; accesscode <value> - Access code.
```

# db (Connect to WYDE Main, i.e. dnca, Database)

#### Syntax:

wyde db

# db-bil (Connect to WYDE Billing, i.e. dnca\_calls, Database)

#### Syntax:

wyde db-bil

# db-drop (Drop of WYDE Database)

```
Syntax:
```

```
wyde db-drop arguments
```

## Arguments:

database <value> - Denotes what kind of database should be dropped; the possible values: {Main|Billing} -

- o *Main* is being used for *dnca* database (default);
- o Billing is being used for dnca calls database.

# db-init (Initialize of WYDE Database)

```
Syntax:
```

```
wyde db-init arguments
```

## Arguments:

database <value> - Denotes what kind of database should be initialized; the possible values: {Main|Billing} -

- o *Main* is being used for *dnca* database (default);
- o Billing is being used for dnca calls database.

# db-patch (Apply Last Patches for Databases)

#### Syntax:

wyde db-patch

# did-add (Add DNIS/DID Number)

#### Syntax:

```
wyde did-add arguments
```

#### Arguments:

```
number <value> - Dial-in number (*);
callflow <value> - The name of the call flow (*);
description <value> - DNIS (DID) description.
```

# did-alias-add (Add DNIS/DID Number Alias)

#### Syntax:

```
wyde did-alias-add arguments
Arguments:
```

```
number <value> - Main dial-in number (*);
alias <value> - Alias of the number (*);
description <value> - Description of the alias.
```

# did-alias-del (Delete DNIS/DID Number Alias)

#### *Syntax:*

```
wyde did-alias-del arguments
Arguments:
   number <value> - Main dial-in number (*);
   alias <value> - Alias of the number (*).
```

## did-alias-show (Show DNIS/DID Number Aliases)

```
Syntax:
```

wyde did-alias-show arguments

#### Arguments:

number <value> - Main dial-in number; if the argument omitted show aliases for all DNISes.

## did-attr-del (Remove DNIS/DID Attribute Redefinition)

```
Syntax:
```

wyde did-attr-del arguments

Arguments:

number <value> - DNIS (DID) number (\*);
name <value> - This DNIS call flow attribute name (\*).

#### did-attr-set (Set DNIS/DID Attribute)

#### Syntax:

wyde did-attr-set arguments

Arguments:

number <value>-DNIS (DID) number (\*);

name <value> - This DNIS call flow attribute name (\*); value <value> - This DNIS call flow attribute new value (\*).

## did-attr-show (Show DNIS's Attributes)

Syntax:

wyde did-attr-show arguments

Arguments:

number <value> - DNIS (DID) number (\*).

#### did-del (Delete DNIS/DID Number)

Syntax:

wyde did-del arguments

Arguments:

number <value>-DNIS (DID) number (\*).

## did-reload (Reload all DNISes/DIDs Caches)

Syntax:

wyde did-reload

Note

This command works the same as *mf* console command: did-reload (Reload DID Entries).

# did-show (Show DNISes/DIDs Table)

Syntax:

wyde did-show

## drop-call (Drop Call)

Syntax:

wyde drop-call arguments

Arguments:

conference <value> - the number of the conference which calls you wish to
 drop, use 'confless' keyword instead of conference number to indicate calls
 which have not placed to any conference (\*);

call <value> - the call session ID (\*).

Note:

This command works the same as *mf* console command: call-drop (Drop Call in the Conference).

# drop-conf (Drop Conference)

Syntax:

wyde drop-conf arguments

Arguments:

number <value> - the number of the conference you wish to drop (disconnect), you can use the keyword all instead of conference number to drop all conferences.

Note:

This command works the same as *mf* console command: conf-drop (Drop Conference).

# help (Show Help Page and Exit)

Syntax:

wyde help arguments

Arguments:

command – the specific wyde command on which you would like to get help.

# ivr (Connect to IVR/asterisk Console)

Syntax:

wyde ivr

# node-add (Add Node to the WYDE Bridge)

Syntax:

wyde node-add arguments

Arguments:

bridge <value> - the name of the bridge, if this argument is omitted the node will be added for the current bridge (default bridge is defined in /usr/local/DNCA/etc/dnca.conf file);

name <value> - the name of the node that should be added (\*);

ivr\_addr <value> - the node IVR address, i.e. IP address and port of the node
 (\*);

sip\_addr <value> - the node SIP address, i.e. IP address and port of the node on
which asterisk services (Frontend) should accept (receive) SIP calls (\*);

ivr\_maxcalls <value> - the maximum number of calls on the IVR for the node; zone <value> - the location zone for the node;

active <value> - the node active-flag value (True - denotes that the node is active and new calls can go to this node, False - denotes that the node is inactive and new calls cannot go to this node).

# node-del (Delete Node from the WYDE Bridge)

Syntax:

wyde node-del arguments

Arguments:

bridge <value> – the name of the bridge, if this argument is omitted the node will be deleted from the current bridge (default bridge is defined in /usr/local/DNCA/etc/dnca.conf file);

name <value> - the name of the node that should be deleted (\*).

# node-reload (Reload Nodes Configuration)

Syntax:

wyde node-reload

# node-set (Set Node Properties)

Syntax:

wyde node-set arguments

Arguments:

bridge <value> - the name of the bridge which node properties should be updated, if this argument is omitted the node for the current bridge will be taken (default bridge is defined in /usr/local/DNCA/etc/dnca.conf file);

name <value> - the name of the node which properties should be updated (\*);

ivr\_addr <value> - the node new IVR address, i.e. IP address and port of the node;

sip\_addr <value> - the node new SIP address, i.e. IP address and port of the node on which *asterisk* services (Frontend) should accept (receive) SIP calls;

ivr\_maxcalls <value> - new maximum number of calls on the IVR for the
 node:

zone <value> - new location zone for the node;

active <value> - new active-flag value for the node (True - denotes that the node is active and new calls can go to this node, False - denotes that the node is inactive and new calls cannot go to this node).

# node-show (Show Nodes of WYDE Bridge)

Syntax:

wyde node-show arguments

Arguments:

bridge <value> - the name of the bridge, if this argument is omitted the command shows nodes for all available bridges.

Note:

This command works similar to *mf* console command: node-show (Show *MF* Cluster Nodes List), but it has few differences in format, output and returned data – this *wyde* command shows nodes configuration in the database, *mf* console node-show command shows the current nodes status.

# peer-add (Add Peer to the WYDE Bridge)

```
Syntax:
    wyde peer-add arguments
Arguments:
    name <value> - the unique name of the peer that should be added (*);
    ipaddr <value> - the unique peer IP address and optionally port (*);
    dialout_prefix <value> - the prefix that should be added to the beginning of
        the number during the dial-out or the template for replace for the peer;
    zone <value> - the location zone for the peer.
```

# peer-del (Delete Peer from the WYDE Bridge)

```
Syntax:
    wyde peer-del arguments
Arguments:
    id <value> - the identifier of the peer that should be deleted (*).
```

# peer-set (Set Peer Properties)

```
Syntax:
    wyde peer-set arguments
Arguments:
    name <value> - the name of the peer to update;
    ipaddr <value> - the peer IP address and optionally port that should be updated
        (*);
    dialout_prefix <value> - the prefix that should be added to the beginning of
        the number during the dial-out or the template for replace for the peer;
    zone <value> - the location zone for the peer to update.
```

# peer-show (Show Peers of WYDE Bridge)

```
Syntax:
wyde peer-show

register-license (Register License)

Syntax:
wyde register-license
```

# reload-all (Reload All System Configuration)

```
Syntax: wyde reload-all
```

# set-email (Change Email Address)

```
Syntax:
```

wyde set-email arguments

#### Arguments:

email <value> - new notification email address (\*);

bridge <value> - the name of the bridge which email should be changed, if this argument is omitted the changes will be made for the current bridge (default bridge is defined in /usr/local/DNCA/etc/dnca.conf file).

## set-ip (Change IP Address)

## Syntax:

wyde set-ip arguments

## Arguments:

ip <value> - new IP address of the bridge (\*);

bridge <value> - the name of the bridge which IP address should be changed, if this argument is omitted the changes will be made for the current bridge (default bridge is defined in /usr/local/DNCA/etc/dnca.conf file).

# settings-edit (Edit WYDE System Settings)

#### Syntax:

wyde settings-edit arguments

#### Arguments:

bridge <value> - the name of the bridge which settings you would like to edit, if this argument is omitted the settings will be edited for the current bridge (default bridge is defined in /usr/local/DNCA/etc/dnca.conf file);

name <value> - the parameter name that should be edited (\*);

value <value> - new parameter value that should be set.

# settings-reload (Reload System Settings)

#### Syntax:

wyde settings-reload

# settings-show (Show WYDE Settings)

#### Syntax:

wyde settings-show arguments

#### Arguments:

bridge <value> - the name of the bridge which settings you would like to show, if this argument is omitted the settings will be shown for the current bridge (default bridge is defined in /usr/local/DNCA/etc/dnca.conf file);

prefix <value> - the prefix of the parameters names that should be shown.

# settings-update (Update WYDE Settings with Default Values)

#### Syntax:

wyde settings-update arguments

#### Arguments:

bridge <value> - the name of the bridge which settings should be updated, if this argument is omitted the settings are being updated for the current bridge (default bridge is defined in /usr/local/DNCA/etc/dnca.conf file).

#### *Note:*

This command adds missing and removes obsolete bridge parameters for the bridge.

# show-conf (Show Conference or Conferences List)

#### Syntax:

```
wyde show-conf arguments
Arguments:
```

number <value> - Conference number:

- o if this parameter is omitted the command returns the list of all conferences on the bridge;
- o if the conference number is transferred the command returns the list of all calls joined to the requested conference;
- o use 'confless' keyword instead of conference number to show calls which have not placed to any conference.

#### *Note:*

This command works the same as *mf* console command: show (Show Conferences and Calls).

# sounds-reload (Reload Sound Files)

#### Syntax:

wyde sounds-reload

## status (Show WYDE Status)

#### Svntax:

wyde status

## subscriber-add (Add Subscriber)

#### Syntax:

```
wyde subscriber-add arguments
Arguments:
```

login <value> - Login name, i.e. subscriber PIN, if this parameter is omitted unique numeric subscriber PIN will be generated;

```
unique numeric subscriber PIN will be generated;

name <value> - Subscriber's first name;

lastname <value> - Subscriber's last name;

parent <value> - Parent login name, i.e. parent subscriber PIN;

password <value> - Subscriber's password;

zip <value> - Subscriber's ZIP code;

email <value> - Subscriber's email address;

sip phone <value> - Subscriber's VoIP phone number.
```

## subscriber-del (Delete Subscriber)

```
Syntax:
```

```
wyde subscriber-del arguments
Arguments:
  login <value> - Login name, i.e. subscriber PIN (*).
```

# subscriber-set (Set Subscriber Properties)

```
Syntax:
```

```
wyde subscriber-set arguments
Arguments:
  login <value> - Login name, i.e. subscriber PIN (*);
  name <value> - Subscriber's first name;
  lastname <value> - Subscriber's last name;
  password <value> - Subscriber's password;
  zip <value> - Subscriber's ZIP code;
  email <value> - Subscriber's email address;
  sip phone <value> - Subscriber's VoIP phone number.
```

# subscriber-show (Show Subscribers Table)

#### Syntax:

wyde subscriber-show

# transfer (Transfer Calls)

Syntax:

```
wyde transfer arguments Arguments:
```

One of the following arguments should be specified as the first argument of this command (\*):

- o {node <node\_id>|node all} denotes the node identifier (or all nodes)
  from which the calls should be transferred;
- o {conference <conf\_number>} denotes the conference number for
  which the calls should be transferred;
- o {did <did>} denotes the DNIS (DID) number for which the calls should be transferred:

destination destination IP value> - denotes destination IP address to
 which the specified calls should be transferred (\*).

#### Note:

This command works the same as *mf* console command: sounds-reload (Reload Sounds).

# version (Show WYDE Version)

#### Syntax:

wyde version

# watch (Watch WYDE Status)

Syntax:

wyde watch arguments

Arguments:

interval <value> - Watch interval in seconds to refresh the WYDE bridge status, if it is omitted the 5 seconds interval is used by default.

# mf Console Command Reference

Once you enter into *mf* console you can run any of the console command just typing the command name and optionally the command arguments:

```
command [arguments]
```

This chapter describes all available mf console commands with their arguments. For any of the commands the required arguments marked with asterisk (\*); optional arguments are shown in square brackets -[...] – in the command syntax.

## activities-show (Enable/Disable Printing Activities on the mf Console)

Syntax:

```
activities-show {start|stop}
Arguments:
    {start|stop} - start indicates that the conferences calls activities (actions)
        should be outputted to the mf console, stop indicates that the activities output
```

# billing-reload (Reload Billing Configuration)

Syntax:

billing-reload

should be stopped (\*).

# call-associate (Set Bundle for the Call)

Syntax:

```
call-associate <conf_number> <ses_id> [<audiokey>]
Arguments:
```

<conf\_number> - the number of the conference for which call you wish to define
the audio key, i.e. the bundle (\*);

<ses id> - the call session identifier (\*);

<audiokey> - new audio key for the call, if this argument is omitted the audio key for this call will be set to 0, i.e. the call in this case will not belong to any bundle.

# call-custom-name (Set Custom Name for the Call)

Syntax:

```
call-custom-name <conf_number> <ses_id> [<name>]
Arguments:
```

<conf\_number> - the number of the conference for which call you wish to define
the custom name (\*);

<ses id> - the call session identifier (\*);

<name> - new custom name for the call, if this argument is omitted the empty custom
name will be set to this call.

```
call-drop (Drop Call in the Conference)
Syntax:
   call-drop <conf number> <ses id> [force [nocdr]]
Arguments:
   <conf number> - the number of the conference which calls you wish to drop
      (disconnect), use 'confless' keyword instead of conference number to indicate
      calls which have not placed to any conference (*);
   <ses id> - the call session identifier (*);
   force – denotes that the command should force call drop;
   nocdr – denotes that CDR record should not be created for this call.
Note:
   This command works similar to wyde command: drop-call (Drop Call).
call-gain-set (Gain Control)
Syntax:
   call-gain-set <conf number> <ses id> <value>
Arguments:
   <conf number> - the number of the conference for which call you wish to change
      the volume level (*);
   <ses id> – the call session identifier for which you would like to change the volume
      level (*);
   <value> - could be from -10 till 10 or 255; it indicates what volume level should
      be set for the call, 255 denotes that the volume level is being automatically
      generated by the backend (*).
call-hold (Hold Call)
Svntax:
   call-hold {true|false} <conf number> <ses id>
Arguments:
   {true | false} - true indicates that the call should be placed on hold, false
      indicates that the call should be taken of hold (*);
   <conf number> - the number of the conference which call you wish to place on
      hold or take of hold (*);
```

## call-move (Move Call to Other Conference)

<ses id> - the call session identifier (\*).

```
Syntax:
```

```
call-move <conf_number> <ses_id> <new_did_number>
  <new_accesscode> [<new_role>]
```

## Arguments:

- <conf\_number> the source conference number whose call you would like to move
  to another conference (\*);
- <ses\_id> the call session identifier that you wish to move to another conference (\*);

- <new\_did\_number> new (i.e. target) conference DNIS (DID) number where the
   call should be moved (\*);
- <new\_accesscode> the access code that should be used to join to new (i.e. target)
  conference (\*);
- <new\_role> the role that will be granted to the call when it joins to new conference
  (applicable only for call flows without authorization, for example CONF call flow).

Note:

This command is asynchronous.

## call-mute (Mute Call)

```
Syntax:
```

```
call-mute {false|strict|relaxed} <conf_number> <ses_id>
Arguments:
```

- {false|strict|relaxed} false indicates that the call should be un-muted, strict indicates that the call should be muted and participants can not un-mute themselves, relaxed indicates that the call should be muted but participants can un-mute themselves (\*);
- <conf\_number> the number of the conference which call you wish to mute or unmute (\*);
- <ses id> the call session identifier (\*).

## call-qa-request (Start/Stop Q&A Request for the Call)

#### Syntax:

```
call-qa-request {start|stop} <conf_number> <ses_id>
Arguments:
```

- {start|stop} start indicates that the call should start the request to speak (the request to ask the question), i.e. placed into Q&A queue, stop indicates that the call should stop (cancel) the request to speak, i.e. removed from Q&A queue (\*);
- <conf number> the number of the conference where Q&A session is started (\*);
- <ses\_id> the call session identifier that should be placed into Q&A queue or removed from it (\*).

Note:

When Q&A session is started the participants can use \*6 (default) on theirs DTMF keypads and confirm that he wants to ask a question. The participants will be placed in a queue in the order that they requested to speak.

## call-qa-talk (Enable/Disable Q&A Session for the Call in the Queue)

Syntax:

```
call-qa-talk {enable|disable} <conf_number> <ses_id>
Arguments:
```

{enable|disable} - enable indicates that the call from Q&A queue should be unmuted and the participants should be able to speak (to ask his question), disable indicates that the unmuted call should be muted and removed from Q&A queue (\*);

<conf number> - the number of the conference where Q&A session is started (\*);

<ses\_id> - the call session identifier that should be unmuted to ask his question or should be muted again (\*).

Note:

When Q&A session is started the hosts can use theirs DTMF keypads to mange Q&A sessions: default \*1 2 can be used to move to the next questioner and \*1 4 can be used to mute or unmute the active questioner.

# call-role (Set Role for the Call)

Svntax:

call-role <conf\_number> <ses\_id> {host|participant|listener}
Arguments:

<conf\_number> - the number of the conference for which call you wish to set the
new role (\*);

<ses\_id> - the call session identifier which role you would like to change (\*);
{host:participant|listener} - denotes what role (host or participant or
listener) should be set for the call, i.e. the specified permissions are being granted to
the call (\*).

## callflow-reload (Reload Call Flows)

Syntax:

callflow-reload

*Note:* 

This command works the same as *wyde* command: callflow-reload (Reload All Call Flows).

# cmdcount-show (Display Values of Command Counters)

Svntax:

cmdcount-show

*Note:* 

Reports the number of commands executed on bridge from *mf* starts till now. The following counters are being returned by this command:

- o cn active number of CONF-ACTIVE commands from web or *mf* console;
- cn\_associate number of ASSOCIATE commands from web or mf console;
- cn\_customname number of SET-CUSTOMNAME commands from web or *mf* console;
- o cn dialout number of DIALOUT commands from web or *mf* console;
- o cn drop number of DROP commands from web or *mf* console;
- o cn dropconf number of CONF-DROP commands from web or *mf* console;
- cn\_dropgroup number of CONF-DROP-GROUP commands from web or mf console;
- o cn gain number of CALL-GAIN-SET commands from web or *mf* console;
- o cn hold number of HOLD commands from web or *mf* console;
- o cn\_holdgroup number of HOLD-GROUP commands from web or *mf* console;

- o cn jobcode number of JOB-CODE commands from web or mf console;
- o cn\_move number of re-attach commands from web or *mf* console;
- o cn mute number of MUTE commands from web or *mf* console;
- o cn\_mutegroup number of MUTE-GROUP commands from web or *mf* console;
- o cn playfile number of PLAY-FILE commands from web or *mf* console;
- o cn qamode number of QA-MODE commands from web or *mf* console;
- cn\_qarequest number of QA-REQUEST commands from web or mf console;
- o on qatalk-number of QA-TALK commands from web or mf console;
- o cn\_recording number of RECORDING commands from web or *mf* console;
- o cn secure number of SECURE commands from web or *mf* console;
- o cn setrole number of SET-ROLE commands from web or *mf* console;
- o cn\_subconf number of SUBCONF commands from web or *mf* console;
- o on transfer number of TRANSFER commands from web or *mf* console;
- o dtmf active number of CONF-ACTIVE commands from DTMF;
- o dtmf associate number of ASSOCIATE commands from DTMF;
- o dtmp dialout number of DIALOUT commands from DTMF;
- o dtmp\_dropgroup number of of CONF-DROP-GROUP commands from DTMF;
- o dtmf gain number of CALL-GAIN-SET commands from DTMF;
- o dtmf hold-number of HOLD commands from DTMF;
- o dtmf holdgroup number of HOLD-GROUP commands from DTMF;
- o dtmf jobcode number of SET-JOBCODE commands from DTMF;
- o dtmf mute number of MUTE commands from DTMF;
- o dtmf mutegroup number of MUTEGROUP commands from DTMF;
- o dtmf gamode number of QA-MODE commands from DTMF;
- o dtmf garequest number of QA-REQUEST commands from DTMF;
- o dtmf gatalk number of QA-TALK commands from DTMF;
- o dtmf\_recording number of START-RECORDING commands from DTMF;
- o dtmf secure number of SECURE commands from DTMF;
- o dtmf subconf number of SUBCONFERENCE commands from DTMF;
- o rt active number of CONF-ACTIVE commands from RT;
- o rt associate number of ASSOCIATE commands from RT;
- rt\_broadcastproposal number of BROADCAST-PROPOSAL commands from RT;
- o  $rt\_broadcaststart-number of BROADCAST-START commands from RT;$
- o rt customname number of SET-CUSTOMNAME commands from RT;
- o rt dialout number of DIALOUT commands from RT;
- o rt drop number of DROP commands from RT;
- o rt dropconf number of CONF-DROP commands from RT;

```
o rt dropgroup – number of DROP-GROUP commands from RT;
o rt gain - number of SET-GAIN commands from RT;
o rt hold-number of HOLD commands from RT;
o rt holdgroup – number of HOLD-GROUP commands from RT;
o rt jobcode - number of SET-JOBCODE commands from RT;
o rt mute – number of MUTE commands from RT;
o rt mutegroup – number of MUTE-GROUP commands from RT;
o rt playfile - number of PLAY-FILE commands from RT;
o rt gamode – number of QA-MODE commands from RT;
o rt gareguest - number of QA-REQUEST commands from RT;
o rt gatalk - number of QA-TALK commands from RT;
o rt recording - number of START-RECORDING commands from RT;
o rt secure – number of SECURE commands from RT;
o rt setrole - number of SET-ROLE commands from RT;
o rt setrole – number of SET-ROLE commands from RT;
o rt subconf - number of SUB-CONFERENCE commands from RT.
```

# conf-active (Start/Stop Conference Active Mode for Participants and Listeners)

```
Syntax:
```

```
conf-active {start|stop} <conf_number>
Arguments:
```

{start|stop} - denotes should or should not the conference be active for the groups of callers denoted in "Specifies which roles will be hold on the beginning of conference" (conference\_hold\_policy) call flow attribute - by default for participants and listeners: start indicates that the callers of the specified roles should hear and participate in the conference; stop indicates that all callers of these roles should be on hold and do not hear the conference;

<conf\_number> - the number of the conference you wish to activate/deactivate, i.e.
the conference for which you would like to change its active mode.

#### conf-attributes (Set Call Flow Attributes for the Conference)

```
Svntax:
```

```
conf-attributes <conf_number>
    name1=value1[,name2=value2[,...]]
```

Arguments.

<conf\_number> - the number of the conference which attributes you would like to
 change;

name1=value1[, name2=value2[, ...]] - denotes the pairs of the call flow attributes names and values that need to be updated for the started conference; the following call flow attributes could be updated:

o the DTMF policy attributes (i.e. the attributes with suffix dtmf);

```
o MOH configuration attributes (conference_moh, conference moh options);
```

```
o the conference properties (conference_rollcall, conference_rollcall_onenter, conference_rollcall_onexit, conference_entryexittones, conference_entrytones, conference_exittones, conference_start_how, conference_start_wait, conference_stop_how, conference_stop_wait, recording_stop_how, recording_stop_wait, conference_callerdb).
```

# conf-broadcast-proposal (Broadcasting Proposal)

Syntax:

```
conf-broadcast-proposal <media> <conf_number> <ses_id>
Arguments:
```

<media> - the comma-delimited list of medias for the presentation, the possible
 values: <video, screensharing> (\*);

<conf\_number> - the number of the conference to which caller you would like to
propose the presentation (\*);

<ses\_id> - the caller session identifier to which proposing to be a presenter (\*).
Note:

When this command is executed the bridge sends the real-time NOTIFY-BROADCAST-PROPOSAL notifications about the proposal to start the presentation to all conference calls (the session identifier of the call that is proposed to start broadcasting, i.e. the session that received the proposal is transferred in this notification); see detailed information in "Real Time Interface – Programmer's Guide".

## conf-broadcast-reject (Reject Broadcasting)

*Syntax:* 

```
conf-broadcast-reject <reason> <conf_number> <ses_id>
Arguments:
```

- <reason> the reject reason: Normal is used by presenter (host or participant) to
   end the started presentation, Cancelled is used by the participant received the
   proposal (who received conf-broadcast-proposal) to cancel (refuse) this
   proposal (\*);
- <conf\_number> the number of the conference for which caller the presentation
  and/or proposal to start the presentation should be rejected (\*);
- <ses\_id> the caller session identifier whose presentation and/or proposal to start
  the presentation should be rejected (\*).

Note:

When this command is executed the bridge sends the real-time NOTIFY-BROADCAST-REJECT notifications informing that the presentation (screen sharing and/or video) has been stopped or that the proposal to start the presentation was cancelled (the session identifier of the connection that stopped the presentation or that

cancelled previously received proposal to start the presentation is transferred in this notification); see detailed information in "Real Time Interface – Programmer's Guide".

## conf-broadcast-start (Start Broadcasting)

Syntax:

```
conf-broadcast-start <media> <conf_number> <ses_id>
Arguments:
```

<media> - the comma-delimited list of medias of the starting presentation, the
 possible values: <video, screensharing> (\*);

<conf\_number> - the number of the conference which caller should start the
presentation (\*);

<ses\_id> - the session identifier of the caller who starts the presentation (\*).

Note:

When this command is executed the bridge sends the real-time NOTIFY-BROADCAST-START notifications about the starting presentation (screen sharing or video or both) to all conference calls (the session identifier of the connection that is starting presentation is transferred in this notification); see detailed information in "Real Time Interface – Programmer's Guide".

# conf-chat-send (Send Chat Message to the Specified Sessions or Groups)

Syntax:

```
conf-chat-send <conf_number> <recipients> <text>
Arguments:
```

<conf\_number> - the number of the conference to which callers you would like to
send the text chat messages (\*);

<recipients> - the recipients to whom the message should be sent; in this
argument you could specify either comma-delimited list of roles, i.e.

<host, participant, listener> or comma-delimited list of the caller
session identifiers (\*);

<text> – the message that should be sent; this argument must be Base64-encoded, i.e. encoded in MIME's *Base64* scheme (\*).

Note:

This command sends the real-time CHAT-MSG notifications to the specified recipients about the messages that has been sent (the message sender session identifier in these notifications is 0); see detailed information in "*Real Time Interface – Programmer's Guide*".

# conf-drop (Drop Conference)

Syntax:

```
conf-drop <conf_number>|all
```

Arguments:

One of the following arguments should be specified (\*):

- o <conf\_number> the number of the conference you wish to drop
   (disconnect);
- o all—the keyword determines that all conferences should be dropped.

Note:

This command works the same as *wyde* command: drop-conf (Drop Conference).

## conf-drop-group (Drop Group)

```
Syntax:
```

conf-drop-group {host|participant|listener} <conf\_number>
Arguments:

{host|participant|listener} - denotes who (all hosts or all participants or all listeners) should be dropped from the conference (\*);

<conf\_number> - the number of the conference which participants or listeners calls
you wish to drop (\*).

## conf-hold-group (Hold Group)

## Syntax:

```
conf-hold-group {true|false} {host|participant|listener}
  <conf_number>
```

## Arguments:

{true|false} - true indicates that the group (participants or listeners) should be placed on hold, false indicates that the group (participants or listeners) should be taken of hold (\*);

{host|participant|listener} - denotes who (all hosts or all participants or all listeners) should be placed on hold or taken of hold (\*);

<conf\_number> - the number of the conference which calls you wish to place on hold or take of hold (\*).

# conf-jobcode (Set Job Code for the Conference)

#### Syntax:

```
conf-jobcode <conf_number> [<code>]
```

# Arguments:

<conf\_number> - the number of the conference which job code you would like to
 set (\*);

<code> - new job code for the conference.

# conf-mute-group (Mute Group)

#### Syntax:

```
conf-mute-group {false|strict|relaxed} {host|participant}
  <conf_number>
```

#### Arguments:

{false|strict|relaxed} - false indicates that the group (hosts or participants) should be un-muted, strict indicates that the group (hosts or participants) should be muted and participants can not un-mute themselves, relaxed indicates that the group (hosts or participants) should be muted but participants can un-mute themselves (\*);

{host|participant} - denotes who (all hosts or all participants) should be muted
 or un-muted (\*);

<conf\_number> - the number of the conference which calls you wish to mute or unmute (\*).

## conf-play-file (Manage of Playing File to the Conference)

## Syntax:

```
conf-play-file <conf_number> <ses_id> {assign <dir>
     <filename>|start|stop|seek <offset> <whence>}
```

#### Arguments:

- <conf\_number> the number of the conference where you would like to play the
   audio file (\*);
- <ses\_id> the session identifier (usually the identifier of the control call) which
  should be used to play the audio file (\*);
- {assign <dir> <filename>|start|stop|seek <offset> <whence>} one of the following arguments should be specified here (\*):
  - o assign <dir> <filename> assign the file for the playback:
    - o <dir>- either record for the conference recorded files (i.e. previous this conference recordings) or upload for the uploaded files (i.e. the files uploaded via web);
    - o <filename> the audio file name without extension, this file should be in the conference recording folder (usually /usr/local/DNCA/var/recordings/ folder) subfolder, either record subfolder or upload subfolder for the specific conference;
  - o start start the playback from the current position;
  - o stop stop the playback;
  - o seek <offset> <whence> seek the audio file playback indicator (pointer) on offset seconds relative to the parameter whence:
    - $\circ$  0 starting from the beginning of the file;
    - 1 starting from the current position in the file;
    - $\circ$  2 starting from the end of the file.

## conf-polling (Conference Polling)

#### Syntax:

```
conf-polling <conf_number> {start <keys>|stop}
Arguments:
```

- <conf\_number> the number of the conference for which you would like to start or
  to stop the polling (\*);
- {start <keys>|stop} start indicates that the polling should be started for the conference, stop indicates that the polling should be stopped for the conference (\*);
  - <a href="keys"> available polling options (i.e. digits 1, 2, ..., 9, 0) that should be specified when the polling is started.

# conf-qa-mode (Manage Q&A Sessions)

#### Syntax:

```
conf-qa-mode {start|stop|clear} <conf number>
```

#### Arguments:

{start|stop|clear} - start indicates that Q&A session should be started for the conference, i.e. the conference should be placed in *question* mode, stop indicates that Q&A session should be stopped for the conference, clear indicates that Q&A queue should be cleared for the conference (\*);

<conf\_number> - the number of the conference for which you would like to manage
the Q&A session (\*).

#### Note:

To implement the same Q&A sessions management hosts can use DTMF keypad on their phones: Q&A session can be started using default \*1 1; Q&A session can be stopped using default \*1 3; Q&A queue can be cleared using default \*1 5.

## conf-qa-mute (Mute/Unmute Active Q&A Session)

#### Syntax:

```
conf-qa-mute {true|false} <conf_number>
Arguments:
```

{true|false} - true indicates that the active Q&A session should be muted, false indicates that the active Q&A session should be unmuted (\*); both these options do not remove the questioner from Q&A queue, the command just temporary allows/disallows the active questioner to speak;

<conf\_number> - the number of the conference where Q&A session is started and
where you would like to mute/unmute the active Q&A session (\*).

#### *Note:*

To implement the same Q&A sessions actions hosts can use DTMF keypad on their phones: to mute or unmute the active questioner in Q&A session default \*1 4 should be used.

# conf-qa-talk (Enable Q&A Session for the First Call in the Queue)

#### Syntax:

```
conf-qa-talk <conf_number>
```

#### Arguments:

<conf\_number> - the number of the conference where Q&A session is started and
where you would like to enable (unmute) the first call in the Q&A queue; the active
questioner (if exists) will be removed from Q&A queue, so this command ends
Q&A session for the current questioner and starts it for the next one (\*).

#### *Note:*

To implement the same Q&A sessions action, i.e. to allow the first questioner to speak, hosts can use DTMF keypad on their phones and press default \*1 2.

# conf-recording (Start/Stop Conference Recording)

#### Syntax:

#### Arguments:

{start|stop} - denotes should the conference recording be started or stopped (\*);

- <conf number> the number of the conference you wish to record (\*);
- <accesscode> denotes the host access code, i.e. password to the recording server
  if "Recording method" (recording\_method) call flow attribute value is
  either "local+pin" or "remote+pin" (this value can be either defined on call
  flow level or overridden on DNIS level);
- <subscriber\_pin> denotes the subscriber's pin (that usually could be requested
  from the user) to perform the recording authorization if "Recording method"
   (recording\_method) call flow attribute value is either "local+pin" or
   "remote+pin" (this value can be either defined on call flow level or overridden
   on DNIS level).

# conf-schedule-extend (Extend Scheduled Conference Duration)

Syntax:

```
conf-schedule-extend <conf_number> <seconds>
Arguments:
```

- <conf\_number> the number of the started scheduled conference which duration
  you would like to extend (\*);
- <seconds> the increment in seconds that should be added to the allowed duration
  for this scheduled conference (\*).

## conf-schedule-incsize (Resize Scheduled Conference Subscription)

Syntax:

```
conf-schedule-incsize <conf_number> <count>
Arguments:
```

- <conf\_number> the number of the started scheduled conference where you would
  like to increase the number of the allowed participants (\*);
- <count> the increment in the count of the conference participants that should be
  added to the number of maximal allowed the scheduled conference participants (\*).

# conf-secure (Secure Conference)

```
Syntax:
```

```
conf-secure {secure|unsecure} <conf_number>
Arguments:
```

{ secure | unsecure } - denotes should the conference be made secured or unsecured (\*);

<conf\_number> - the number of the conference you wish to make secured or unsecured (\*).

# conf-shunt (Make/Drop Shunt between Two Conferences)

#### Syntax:

#### Arguments:

- {start | stop} start denotes that the shunt between two conferences should be made (started), stop denotes that the shunt between two conferences should be dropped (stopped) (\*);
  - o <conf\_number> the number of the conference you wish to shunt or unshunt (\*);
  - o <per\_conf\_number> the number of the peer conference you wish to
    shunt with the first one.

# confcount-show (Display Values of confcount Counters)

Syntax:

confcount-show

*Note:* 

Reports the break-up of conferences according to participants size at the moment. The following counters are being returned by this command:

- o co total count gives the total number of conferences;
- o co 1 count is the total number of 1-person conferences;
- o co 2 count is the total number of 2-person conferences;
- o co 3 count is the total number of conferences with 3 people;
- o co le 10 count is the total number of conferences with 4-10 participants;
- o co\_le\_100\_count gives the total number of conferences with 11-100 participants;
- o co\_gt\_100\_count gives the total number of conferences with 100+ participants;
- o co avg pt size gives the average size of conferences hosted on bridge;
- o co max pt size max size of conference;
- o co min pt size min size of conference;
- o co md pt size median size of conference.

# dc-reload (Reload DC Bridges Configuration)

Syntax:

dc-reload

## dc-set-log-level (Set DC Logger Level)

Syntax:

```
dc-set-log-level {debug|event|info}
```

Arguments:

o info-denotes that only DC information and error messages should be stored in the log file (default); event-denotes that additionally the log should contain all DC events logging; debug-denotes that debugging mode is switched on, so additionally debug DC messages should be added to the log.

Note:

Log file name: /usr/local/DNCA/log/mf.log.

## dc-show-bridges (Show Known DC Bridges)

```
Syntax:
```

dc-show-bridges

## dc-show-links (Show DC Links)

Svntax:

dc-show-links

## dialout (Do Dialout)

#### Syntax:

```
dialout <peer_number> <did_number> <accesscode>
Arguments:
```

<peer number> - denotes the phone number you wish to dial (\*);

<did\_number> - the target conference DNIS (DID) number where the call should be
joined after the dial-out is complete (\*);

<accesscode> - the access code that should be used to join to the target conference (\*).

Note:

This command is asynchronous.

## dialout-attr (Show Dialout Attributes for Specified Conference)

Syntax:

```
dialout-attr <originator_conf_number>
Arguments:
```

<originator\_conf\_number> - the number of the started conference which
 dialout attributes, i.e. DNISes and access codes with roles, you would like to show
 (\*).

## did-reload (Reload DID Entries)

Syntax:

```
did-reload [<did number>]
```

Arguments:

<did\_number> - the conference DNIS (DID) number which caches should be reloaded by the WYDE bridge MF engine.

Note:

This command works the same as *wyde* command: did-reload (Reload all DNISes/DIDs Caches).

# errcount-show (Display Values of Error Counters)

Svntax:

```
errcount-show
```

Note:

Reports the number of terminated/incomplete calls on bridge from *mf* starts till now. The following counters are being returned by this command:

o pt total terminated – gives the total number of terminated calls;

- o pt\_sip\_terminated provides the total number of calls terminated because of incomplete SIP handshakes;
- o pt max calls terminated terminate by call amount limits;
- o pt agi terminated terminated by agiserver;
- o pt mp terminated terminated by mp;
- o pt asterisk terminaned terminated by asterisk;
- o pt\_max\_duration\_terminated provides the total number of participants that exceeded the maximum call duration.

## freenumbers-show (Show Free Number Leases)

Syntax:

freenumbers-show

## help (Show Help for Console Commands)

Syntax:

help command

Arguments:

command – the specific mf console command on which you would like to get help.

## node-reload (Reload MF Cluster Nodes List)

Syntax:

node-reload

## node-show (Show MF Cluster Nodes List)

Syntax:

node-show

*Note:* 

This command works similar to *wyde* command: node-show (Show Nodes of WYDE Bridge), but it has few differences in format, output and returned data – this *mf* console command shows the current nodes status, *wyde* node-show command shows nodes configuration in the database.

# op-call-move (Move User that Currently Talking with Operator to other Conference)

```
Syntax:
```

```
op-call-move <operator_number> <new_accesscode> [<new_role>]
Arguments:
```

<operator\_number> - the operator conference number whose current call (i.e. the
 user that currently talking to the operator) you would like to move to another
 conference (\*):

<new\_did\_number> - new (i.e. target) conference DNIS (DID) number where the
 call should be moved (\*);

<new\_accesscode> - the access code that should be used to join to new (i.e. target)
 conference (\*);

<new\_role> - the role that will be granted to the call when it joins to new conference
(applicable only for call flows without authorization, for example CONF call flow).

Note:

To implement the same operator conference actions, i.e. to attach current user to a different conference, operator conference hosts can use DTMF keypad on their phones and press default \*5.

This command is asynchronous.

# op-dialout (Initiate Dialout from Operator's Console)

Syntax:

```
op-dialout <operator_number> <peer_number>
Arguments:
```

<operator\_number> - the operator conference number where you are initiating the
 dialout (\*);

<peer number> - denotes the phone number you wish to dial (\*).

Note:

To implement the same operator conference actions, i.e. to dial-out to another user, operator conference hosts can use DTMF keypad on their phones and press default \*7. This command is asynchronous.

# op-listen (Listen Conference)

Syntax:

```
op-listen <operator_number> {start <conf_number>
   [{directlink|shunt} [mute]]|stop}
```

Arguments:

<operator\_number> - the operator conference number that would like to listen
 other users conferences (\*);

{start | stop} - start denotes that the listening should be started, stop denotes that the listening should be stopped (\*);

- o <conf\_number> the number of the conference you wish to start listening, must be indicated if start option is specified;
- o {directlink|shunt} directlink denotes that the operator should be directly connected to the requested conference (in this case only operator can hear the requested conference, the user connected to the operator can not hear that conference), shunt denotes that between operator conference and requested conference is being made the shunt and both conferences can hear each other (operator and the connected user both can hear the requested conference), if this argument is omitted the directlink mode is being used by default;
- o mute when specified denotes that the operator is muted, so he can only hear the specified conference and he is unable to talk.

Note:

To connect and listen another conference operator conference hosts can use DTMF keypad on their phones and press default \*4, the operator also has the options to

connect with current user (shunt mode) and without current user (directlink mode).

## op-message (Control Operator Message)

Syntax:

Arguments:

<operator\_number> - the operator conference number where the message should
be recorded (\*);

{start|stop} - start denotes that the operator message should be recorded, activated and sent either to users from the operator's queue (when queue parameter is specified) or to all users on the bridge (when all parameter is specified); optional message parameter specifies prerecorded audio file that will be played to the callers, if the parameter is omitted the operator will be asked to leave his message – if there is no voice call connected to the conference with the specified number the command returns the error, if the operator voice call is connected to the operator conference and message parameter is omitted the operator will be asked to leave his message after the tone and press # when it is done, after the message has been recorded (and # was pressed) this message will be played back to the operator and he will have the options either press 1 to send the recorded message or press \* to cancel (this command works this way regardless there is or there is no previously recorded operator message); stop denotes that the operator message should be cleared, i.e. playing this message should be stopped.

Note:

The command either initiates recording and playing back of the new operator message or playing back of prerecorded message or clears (i.e. switches off) the operator message. The similar actions can be performed by the operator conference hosts using default \*8 on DTMF keypad of their phones.

Prerecorded messages are being stored in the *\$varlib\_dir/sounds/operator-messages* folder; messages recorded by the operator are being temporary stored in the *\$varlib\_dir/recordings/operator-messages* folder.

## op-message-to-conf (Send Operator's Message to Conference)

Svntax:

<operator\_number> - the operator conference number where the message should
be recorded (\*);

<conf\_number> - the number of the user's conference where the operator message
should be played; if 0 is specified as the target conference number the message will
be recorded and played to the conference to which the operator is currently
connected or to whose user the operator is currently talking (\*);

{host|participant|listener} - the callers role to which the message should
be played, if the message should be played to multiply roles they all could be
specified separated by colon(:), for instance participant:listener
argument specifies that the message should be played to participants and listeners
(\*);

<message> – the prerecorded audio file that will be played to the callers, if the parameter is omitted the operator will be asked to leave his message after the tone and press # when it is done, after the message has been recorded (and # was pressed) this message will be played back to the operator and he will have the options either press 1 to send the recorded message or press \* to cancel.

#### Note:

The similar actions can be performed by the operator conference hosts using default \*9 on DTMF keypad of their phones.

Prerecorded messages are being stored in the \$varlib\_dir/sounds/operator-messages folder; messages recorded by the operator are being temporary stored in the \$varlib\_dir/recordings/operator-messages folder.

## op-queue (Display Operator Calls Queue)

Syntax:

op-queue

## op-reject (Reject Request from Operator's Queue)

Syntax:

```
op-reject <conf_number> <ses_id>
Arguments:
```

<conf\_number> - the number of the conference which caller should be rejected
from the operator queue, i.e. it is the conference where the caller came from (0 for
IVR calls) (\*);

<ses\_id> - the session identifier of the caller that should be rejected from the
 operator queue, i.e. the conversion with this user will be refused and the user will be
 removed from the operator queue.

## op-scan (Scan Conferences)

Syntax:

```
op-scan <operator_number> {start|stop}
Arguments:
```

<operator\_number> - the operator conference number where you would like to
 start or stop conference monitoring (surveillance call) (\*);

 $\{ \text{start} \mid \text{stop} \} - \text{start} \text{ denotes that the monitoring should be started}, stop denotes that the monitoring should be stopped (*).$ 

#### Note:

To implement the same operator conference actions, i.e. to start or stop conference monitor (surveillance call) operator conference hosts can use DTMF keypad on their phones and press default \*1.

## op-show (Display Operators)

```
Syntax:
```

```
op-show [<operator_number>]
```

#### Arguments:

<operator\_number> - the operator conference number that you would like to
show.

Note:

To hear the operator conference current status the operators can use DTMF keypad on their phones and press default \*6.

# op-talk (Operator Talk to User)

#### Syntax:

#### Arguments:

<operator\_number> - the operator conference number where you would like to
 start or stop talking to a user (\*);

- {start|stop} start denotes that the talking should be started, stop denotes that the talking should be stopped (\*); if together with start option the parameters <conf\_number> <ses\_id> omitted the operator starts talking to the first user from his queue, if these parameters are presented the operator starts talking to the user specified by these parameters:
  - o <conf\_number> the number of the conference which caller should be taken
    to talk with the operator (0 for IVR calls);
  - <ses\_id> the session identifier of the caller that should be taken to talk with
     the operator.

#### Note:

This command allows operator to start talking not only to the user from the operator queue, but also to any user from any active conference.

To implement the same operator conference actions, i.e. to start talking with the next user from the operator queue or stop talking to the user and return him to his conference or ivr operator conference hosts can use DTMF keypad on their phones and press default \*2 to start talking or \*3 to stop talking.

# opcount-show (Display Operator's Counters)

#### Svntax:

```
opcount-show
```

#### *Note:*

Reports the counters related to the operators activities at the moment. The following counters are being returned by this command:

- o op count gives the total number of active operators;
- o op\_console\_count gives the number of operators with the Operator Console started;
- o op\_voice\_count gives the number of operators with the active voice call;

o op\_queue\_length - gives length of the operator queue, i.e. the number of callers currently waiting the operator assistance in operator queue.

## partcount-show (Display Values of partcount Counters)

Syntax:

partcount-show

*Note:* 

Reports the counters related to the participants at the moment. The following counters are being returned by this command:

- o pt total count gives the total number of participants;
- o pt\_ivr\_count gives the total number of participants that does not joined to any conference;
- o pt 1 count gives number of participants in 1-person conferences;
- o pt 2 count gives number of participants in 2-person conferences;
- o pt 3 count gives number of participants in 3-person conferences;
- o pt\_le\_10\_count gives number of participants from 4 to 10-person
  conferences;
- o pt\_le\_100\_count gives number of participants from 11 to 100-person conferences;
- o pt gt 100 count gives number of participants from 100+ conferences;
- o pt pcmu count gives number of participants in PCMU codec;
- o pt pcmua count gives number of participants in PCMA codec;
- o pt isac count gives number of participants in ISAC codec;
- o pt ilbc count gives number of participants in ILBC codec;
- o pt\_g729a\_count gives number of participants in g729a codec;
- o pt\_g722\_count gives number of participants in g722 codec;
- o pt\_voip\_full\_count gives number of participants that join via voip and have right to speak;
- o pt\_voip\_listen\_count gives number of participants that join via voip and does not have right to speak;
- o pt\_voip\_ctrl\_count gives number of control calls;
- $\verb"opt_voip_rec_count-gives number of recording calls; \\$
- pt\_pstn\_full\_count gives number of participants that join via pstn and have right to speak;
- o pt\_pstn\_listen\_count gives number of participants that join via pstn and does not have right to speak.

# peer-reload (Reload Peers)

Syntax:

peer-reload

# quit (Quit Console)

Syntax:

quit

## set-log-level (Set Logger Level)

Syntax:

set-log-level {debug|event|info}

Arguments:

o info-denotes that only information and error messages should be stored in the log file (default); event-denotes that additionally the log should contain all events logging; debug-denotes that debugging mode is switched on, so additionally debug messages should be added to the log.

*Note:* 

Log file name: /usr/local/DNCA/log/mf.log.

# settings-reload (Reload System Settings)

Syntax:

settings-reload

## **show (Show Conferences and Calls)**

Syntax:

show [<number>|confless]

Arguments:

- o the command without parameters returns the list of all conferences on the bridge;
- <number> the conference number in this case the command returns the list of all calls joined to the requested conference;
- o confless the keyword determines that the command should show the calls which have not placed to any conference.

Note:

This command works the same as *wyde* command: show-conf (Show Conference or Conferences List).

# sounds-reload (Reload Sounds)

Syntax:

sounds-reload

## subconf (Manage Sub-conferences)

Syntax:

Arguments:

<conf\_number> - the conference number whose call you would like to attach to
sub-conference or detach from sub-conference (\*);

<ses\_id> - the call session identifier that you wish to attach to sub-conference or
 detach from sub-conference (\*);

one of the following arguments should be specified as the next argument of this command (\*):

- o attach <subconf> denotes the name of the sub-conference up to 16 characters length (only letters and digits are allowed as the name of the sub-conference) to which the call should be attached;
- o detach <subconf> denotes the name of the sub-conference up to 16 characters length from which the call should be detached;
- o move <subconf\_from> <subconf\_to> denotes the source subconference name from which the call should be transferred and the target subconference name to which the call should be transferred.

#### *Note:*

This command is asynchronous.

## transfer (Transfer Calls)

Syntax:

```
transfer {node <node_id>|all}|{conference <conf_number>}|
    {did <did>} <destination ip>
```

#### Arguments:

One of the following arguments should be specified as the first argument of this command (\*):

- o {node <node\_id>|all} denotes the node identifier (or all nodes) from
  which the calls should be transferred;
- o {conference <conf\_number>} denotes the conference number for
  which the calls should be transferred;
- o {did <did>} denotes the DNIS (DID) number for which the calls should be transferred;

<destination\_ip> - denotes destination IP address to which the specified calls
 should be transferred (\*).

#### Note:

This command works the same as *wyde* command: transfer (Transfer Calls).

#### version (Show WYDE.MF Version)

#### Syntax:

version

# asterisk Console Command Reference

Once you enter into *asterisk* console you can run any of the console command just typing the command name and optionally the command arguments:

```
command [arguments]
```

This chapter describes available *asterisk* console *wyde* commands with their arguments. For any of the commands the required arguments marked with asterisk (\*); optional arguments are shown in square brackets – [...] – in the command syntax.

## wyde drop session (Drop Session)

```
Syntax:
```

```
wyde drop session <conf_number> <session_id>
Arguments:
    <conf_number> - the number of the conference which call you wish to drop
```

(disconnect) (\*);

<session\_id> - the call session identifier you wish to drop (\*).

## wyde show conferences (Show Active Conferences)

Syntax:

wyde show conferences

## wyde show conference (Show Conference Members)

Syntax:

```
wyde show conference <number>
```

Arguments:

<number> - the number of the conference which calls you would like to show (\*).

# wyde show session queue (Show Session Events Queue)

Syntax:

```
wyde show session queue <conf_number> <session_id>
Arguments:
```

<conf\_number> - the call conference number which events queue you would like to
show (\*);

<session\_id> - the call session identifier which events queue you would like to
show (\*).

# wyde show sessions (Show Sessions not Attached to Conference)

Syntax:

```
wyde show sessions
```

# wyde show statistic (Show Statistic)

Syntax:

```
wyde show statistic
```

## mp Console Command Reference

Once you enter into *mp* console you can run any of the console command just typing the command name and optionally the command arguments:

```
command [arguments]
```

This chapter describes available mp console commands with their arguments. For any of the commands the required arguments marked with asterisk (\*); optional arguments are shown in square brackets -[...] – in the command syntax.

## drop (Drop Boards and Calls)

Syntax:

Arguments:

One of the following arguments should be specified as the argument of this command (\*):

- o <boards <serial> [<serial>] ...> denotes MPw boards that should be dropped (placed on hold); use <serial> to specify the boards that should be dropped;
  - ✓ this option works as toggle switch if you repeat this command the board will be switched on:
- o <calls <<id> [<id>] ...>|all> denotes that the calls should be dropped from mp; use <id> to specify the call sessions identifiers that should be dropped or use all keyword to drop all calls.

# kill (Stop Specific Timer or All Timers)

Syntax:

```
kill {<id> [<id>] ...|all}
```

Arguments:

<id> - the specific timer identifier that should be stopped; all - denotes that all started timers should be stopped (\*).

# restart (Restart mp Boards and Logs)

Svntax:

```
restart {<log [<level>]>|<board <serial>|all>}
Arguments:
```

One of the following arguments should be specified as the argument of this command (\*):

<log [<level>]> - denotes that the log should be restarted; <level> specifies what messages should be stored in the log: 3 - denotes that only
 information and error messages should be stored in the log file (default); 9 denotes that debugging mode is switched on and all possible log messages
 should be added to the log file.

o <board <serial>|all> - denotes the *MPw* boards that should be restarted; use all keyword to restart all *MPw* boards.

## show (Show Different mp Statistics)

#### Syntax:

show {sw|conf|conf <conf\_number>|calls|stat|boards}
Arguments:

One of the following arguments should be specified as the argument of this command (\*):

- o sw show the number of available MPw boards, i.e. backend components count;
- o conf show all conferences currently running on the mp;
- o conf <conf\_number> show statistics for the specific conference only, including active speaker notifications;
- o calls [<id1>-<id2>] show all calls currently running on the *mp* (the command returns calls identifiers, conference numbers, IP addresses, ports, codecs, status, etc.);
  - if you specify the calls sessions identifiers range, only the calls within this range will be returned by the command;
- o stat show overall *mp* statistics about calls, conferences and *MPs* boards (*now* column shows current data, *total* column shows data from *mp* start, *peak* column shows acme/height value);
- o boards show detail *MPs* boards statistics, including the board current state, capacity, CPU load, number of conferences and calls, etc.

## stop (Stop mp Components)

Syntax:

stop

## timer (Start Timer Running show Command)

#### Syntax:

timer <interval> show <options>

#### Arguments:

<interval> - denotes interval in seconds in which the show command should be rerun (\*);

show options> - denotes specific show command with its options that should be
repeated in the given interval (\*).

# **Appendix A: Support Resources**

If you have difficulty with this guide and any of the procedures listed herein, please contact us using the following support resources.

# Support Documentation

In addition to this Guide, you may obtain other WYDE Voice documentation from WYDE Voice or from the WYDE Voice documentation Web site: <a href="http://docs.wydevoice.com/">http://docs.wydevoice.com/</a>.

# Web Support

Our support website is available 24 hours a day, 7 days a week, and 365 days a year at <a href="http://www.wydevoice.com">http://www.wydevoice.com</a>. You may download patches, support documentation and other technical support information.

# Telephone Support

For difficulties with any procedures described in this Guide, please contact us at 866-508-9020 during our normal phone support hours of 7:00 am to 6:00 pm Pacific Standard Time (PST). An engineer will respond to your inquiry within 24 hours.

# **Email Support**

You may also email us your questions at <u>support@wydevoice.com</u>. We will respond to your question within 24 hours.