

Settings and
configuration guide for
Flexi with MiVO400

Table of content

System Requirements	3
Hardware Requirements	3
Software Requirements	3
License Requirements in MiVO A400.....	3
Configuration	3
Configuration overview MiVO400.....	4
UG configuration.....	5
User configuration	6
Standard SIP Terminal configuration	7
Configuration overview OIP server	7
OIP user configuration	8
Tapiadmin account configuration	8
Used OIP IDL Files	9
Details on Configuration of the Application.....	11
Settings in Datatal CTStack settings.....	11
Good to know.....	19
Scenario 1	19
Problem	19
Solution	19
Scenario 2	19
Problem	19
Solution	19
Scenario 3	19
Scenario 4	20
Problem	20
Setup for MiVO 400 trunk to Flexi.....	21
Overview.....	21
Route configuration	22
Trunk groups configuration.....	23
Datatal CTStack Trunk mode	27
SIP Settings for CTStack	28

System Requirements

Hardware Requirements

The telecommunication interface between the Flexi system and the Aastra PBX can today be either S0 or SIP depending on which Flexi product you want to use.

To get full functionality of your Flexi system requires that there are a number of free S0 ports in your PBX or free SIP licenses (resources). How many S0 ports or SIP licenses required depends on the size of the system (how many users etc.).

Software Requirements

CTI communication between the Flexi system and the Mitel system is via the CORBA interface which means that both systems have to be on the same network.

License Requirements in MiVO A400

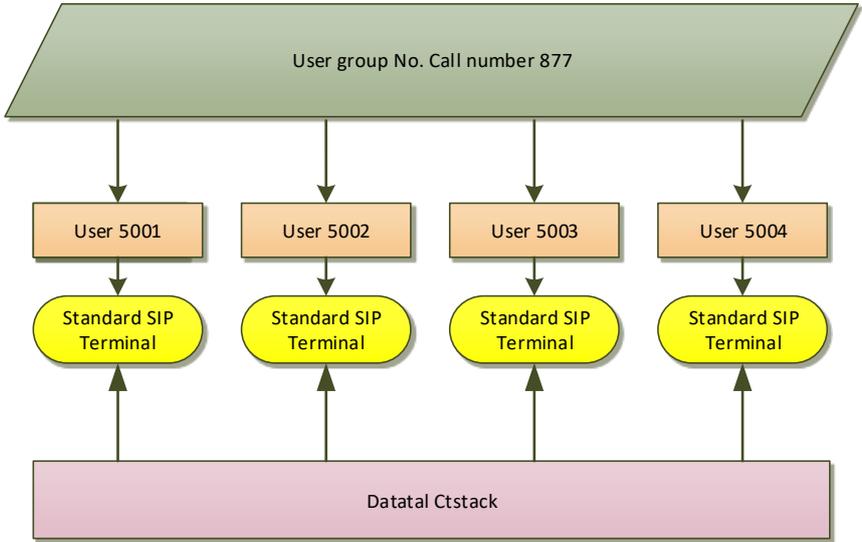
Third party CTI Basic: License "CTI Third Party Basic", art.nr 20327411

ATAS (depends on model CE): License "ATAS Interface 415/430", art.nr 20952021 alt. License "ATAS Interface 470", art.nr 20952022

Standard SIP terminal: License "1 Standard SIP Terminal", art.nr 20329240 one per voice channel to Flexi

Configuration

Configuration overview MiVO400



UG configuration

Apply **Reload**

User group

User group number

Call number

Name

Call distribution 

Main group

Add

-     5003 Flexi ch 3 
-     5004 Flexi ch 4 
-     5005 Flexi ch 5 
-     5006 Flexi ch 6 
-     5007 Flexi ch 7 
-     5008 Flexi ch 8 
-     5001 Flexi ch 1 

Destination and settings

Large user group

General bell

General bell delay

Operator console

Operator console delay

Member group delay

Entry in unanswered call list

Show redirecting user identification

User configuration

[Apply](#)
[Reload](#)
[Back](#)
[Expand all sections](#)

Select Flexi ch 1 (5001)

User

Call number: 5001

Name: Flexi ch 1

PIN: Default PIN Confirm PIN

Windows user name:

Use PIN instead of password:

Password: Password confirmation

E-mail address:

User language: Swedish

Settings

Licence / Role: User 0 - None Go to MiCollab server ...

Permission set: 1 Go to permission set

Authorization profile: --- Go to profile

Route: Weblink (1) Go to route

Allow call forwarding on ending lines (key telephone):

Number of private contacts: 50 Go to phone book

Cost centre: None

Connection: Normal

Use for CTI: Not defined

Re-enable user account for Self Service Portal and BluStar 8000i:

Terminals

[Assign](#) [Create and assign](#)

Terminal type	Description	Port	Free seating
SIP terminal	Flexi ch 1		✗

Standard SIP Terminal configuration

Apply Reload Back

Select SIP terminal, 5001 - Flexi ch 1

Settings to terminal interface Standard SIP

Terminal ID	53
Terminal type	SIP terminal
Description	Flexi ch 1
Assigned user/pool	5001 Go to user
Force UDP usage	<input checked="" type="checkbox"/>

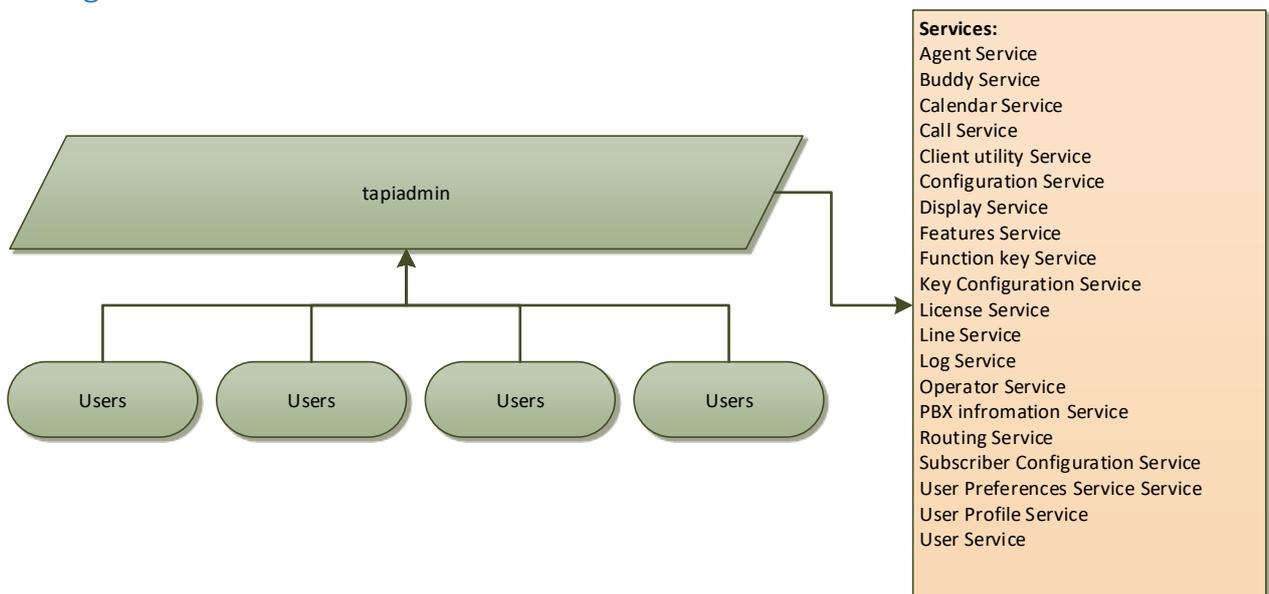
Further settings

Hotline call number	
Hotline delay (s)	0
Number of lines	1
Conference circuit	In communication server
Emergency destinations	None
Emergency location	Inherit
Force call waiting	<input type="checkbox"/>
Special ringing tone	<input type="checkbox"/>
PSTN overflow	No
Region	None

Connection settings

State	Registered
IP address	192.168.255.36
SIP port	5060
MBG controller	None
SIP user name	5001
SIP password	secretpassword
MBG SIP user name	

Configuration overview OIP server



OIP user configuration

3015	Anders Molinder	11	CTI Basic
3017	Erik Kalström	13	CTI Basic
3018	ERIK TEST	170	CTI Basic

TAPI_ADMINISTRATORS configuration

TAPI_ADMINISTRATORS tapiadmin

- ACDLogService / Administrator
- ACDService / Administrator
- AgentService / Administrator
- BuddyService / Administrator
- CalendarService / User
- CallService / Administrator
- DisplayService / Administrator
- FeatureService / User
- FunctionKeyService / User
- JournalService / User
- LicenseService / User
- LineService / User
- Log / User
- OperatorService / User
- PBXInfoService / User
- RoutingService / Administrator
- SubscriberConfig / User
- SystemLogin / User
- UserPreferences / User
- UserProfileService / User
- UserServices / User
- UtilService / User

Tapiadmin account configuration

✖

Apply Undo
Help

General

User name:

Password: Confirm password:

User language: System (System) ▼

Details

User ID: 2

Domain user name:

Microsoft Exchange mailbox address:

Featured in directories:

Login to OIP allowed:

User groups

TAPI_ADMINISTRATORS

+

Add user that tapiadmin account monitoring access right

3015 (Anders Molinder)	Datatal (1)	Monitoring	▼	🗑️
3017 (Erik Kalström)	Datatal (1)	Monitoring	▼	🗑️
3018 (ERIK TEST)	Datatal (1)	Monitoring	▼	🗑️

Used OIP IDL Files

ACDAgentListener.idl
 ACDCfgListener.idl
 ACDdef.idl
 ACDLogDef.idl
 ACDLogListener.idl
 ACDLogService.idl
 ACDQueueListener.idl
 ACDSservice.idl
 AgentDef.idl
 AgentManagementListener.idl
 AgentService.idl
 BaseService.idl
 BuddyClient.idl
 BuddyDef.idl
 BuddySrv.idl
 CalendarClient.idl
 CalendarDef.idl
 CalendarService.idl
 CallClient.idl
 CallDef.idl
 CallService.idl
 Common.idl
 ConfigProfileClient.idl
 ConfigProfileDef.idl
 ConfigProfileService.idl
 ConfigProfileTypes.idl
 Directory.idl
 DirectoryClient.idl
 DirectoryService.idl
 DisplayClient.idl
 DisplayDef.idl
 DisplayService.idl
 DsfConst.idl
 FeatureDef.idl
 FeatureService.idl
 FunctionDef.idl
 FunctionKeyClient.idl

FunctionKeyService.idl
KeyServer.idl
LicenseDef.idl
LicenseService.idl
LineAccess.idl
LineAccessClient.idl
LineAccessDef.idl
LineClient.idl
LineService.idl
MacroSpec.idl
MediaCall.idl
MediaDef.idl
MediaService.idl
PBXInfo.idl
PBXInfoDef.idl
PBXSetupClient.idl
PBXSetupDef.idl
PUMDef.idl
PUMListener.idl
PUMService.idl
RegistrationDef.idl
RegistrationService.idl
RoutingClient.idl
RoutingDef.idl
RoutingManagerDef.idl
RoutingSrv.idl
SubscriberConfiguration.idl
SubscriberConfigurationDef.idl
SystemLogin.idl
UserPreferences.idl
UserProfileConst.idl
UserProfileService.idl
UtilsClient.idl

Details on Configuration of the Application

Settings in Datatal CTStack settings

1. Open browser on Flexi Server, browse to <http://localhost:1339>

Datatal CTStack Configuration

Main menu
→ **CTStack**
Media
SIP

+ Create entity

Commit Revert

CTStack

API
Call Handling

Delay 'idle' event on hangup: ? 0

Server

IP Port: ? 1337

Loopback only: ?

Configuration
Advanced

Advanced mode: ?

Server

IP Port: ? 1339

Loopback only: ?

Service
Current

Date and time: ? 2016-11-23 12:27:10

Priority High: ?

Apply A400(user) profile

The screenshot shows the 'Datatal CTStack Configuration' web interface. On the left is a navigation sidebar with a 'Main menu' containing 'CTStack', 'Media', and 'SIP'. Below it is a 'MiVo400' section with 'Media' and 'SIP' options, and a yellow 'Telephony' button. A '+ Create entity' button is at the bottom of the sidebar. The main content area is divided into two sections: 'Standard' and 'SIP'. The 'Standard' section includes fields for 'BlindCall source mode' (set to 'All'), 'Description' (set to 'string'), 'INVITE expires' (set to '25'), and 'Lines' (set to '0'). The 'SIP' section includes 'Address' (set to 'string'), 'Default domain' (set to 'string'), 'Default SIP URI host' (set to 'string'), 'Default SIP URI port' (set to '5060'), and 'Name' (set to 'string'). Below these is the 'Profile' section, where the 'Apply' dropdown menu is open, showing options: 'None', 'Generic', 'A400 (trunk)', 'A400 (user)', 'MX-ONE (user)', 'Avaya CM (trunk)', and 'Avaya IPO (trunk)'. The 'A400 (user)' option is highlighted in blue. At the bottom of the main area are 'Commit' and 'Revert' buttons, and the 'datatal ab' logo.

Commit changes

The screenshot shows a dialog box with the text '1 change(s) pending' at the top. Below this text are two buttons: a green 'Commit' button and a red 'Revert' button. At the bottom of the dialog is the 'datatal ab' logo.

2. Scroll down to registrations and click on "ADD"

The screenshot displays the 'Datatal CTStack Configuration' interface. On the left is a sidebar with a 'Main menu' containing 'CTStack' and 'SIP'. Below it is a 'Aastra A400' section with 'Media', 'SIP', and 'Telephony' options, and a '+ Create entity' button. At the bottom of the sidebar are 'Commit' and 'Revert' buttons. The main configuration area is divided into several sections:

- Call Settings:** Includes options for 'Play ring at other calls on MakeCall', 'Terminate local call transfer on INVITE', 'Treat BYE as transfer success', 'Use Remote-Target in Refer-To', and 'Wait for park complete on MakeCall'.
- Registrations:** A section titled 'Users' containing a list of registrations: 192.168.255.40 - 5001 through 192.168.255.40 - 5010. To the right of the list are 'ADD', 'EDIT', and 'REMOVE' buttons.
- SIP Dialogs:** Includes 'Use From header'.
- RFC 3325:** Includes 'P-Identity mode' (set to 'Both') and 'Use P-Asserted-Identity'.
- Transport:** Includes 'Transport' (set to 'TCP').

At the bottom left of the interface is the 'datatalab' logo.

3. Set:

Address: Standard SIP terminal extension

Name: Set a name

Enable: Check

Expires: Timer for re-registrations

Registrar: Server-IP to registrar, usually PBX-IP

Username: Username for Standard SIP terminal

Password: Password for Standard SIP terminal

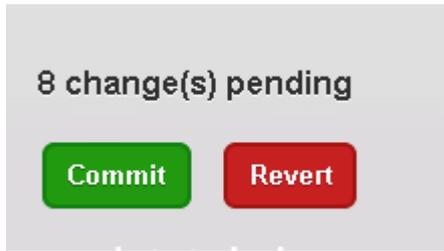
The screenshot shows the Datatal CTStack Configuration web interface. The browser address bar shows 'localhost:1339'. The main content area displays a configuration form for '(empty registration) [disabled]'. The form fields are as follows:

Field	Value
Address:	5001
Name:	Channel 1
Enabled:	<input checked="" type="checkbox"/>
Expires:	600
Local port:	5060
Registrar:	192.168.255.40
Transport:	udp
Username:	5001
Password:	password
Gruu:	<input type="checkbox"/>

At the bottom of the form is a 'Close' button. The sidebar on the left shows a 'Main menu' with options for 'CTStack', 'Media', and 'SIP'. Below that is a 'MiVo400' section with 'Media', 'SIP', and 'Telephony' options, and a '+ Create entity' button. At the bottom of the sidebar, it says '10 change(s) pending' and has 'Commit' and 'Revert' buttons. The 'datatal ab' logo is at the bottom of the sidebar.

Then click Close

4. Commit and commit changes



5. For more registrations, you need to repeat step 6 and 7. Remember to commit after each registrations.



6. Default domain: PBX-IP

Name: Flexi

Port: 5060

Default SIP URI host: PBX-IP

Trunk mode: Checked

Profile: Use A400 user

Lines: Enter number of lines that is order, if entered 20 lines and the license is valid for 16 lines, CTstack will only use 16 lines simultaneous

The screenshot displays the Datatal CTStack Configuration web interface for MiVo400 - Telephony. The interface is divided into a sidebar and a main content area. The sidebar on the left contains a navigation menu with options for Main menu, MiVo400, and Create entity. The main content area is titled "MiVo400 - Telephony" and is divided into two sections: "Line configuration" and "SIP".

Line configuration

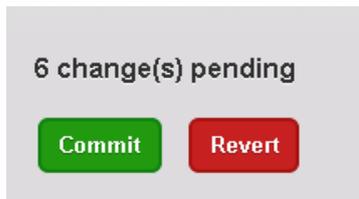
- Standard
- BlindCall source mode: All
- Description: string
- INVITE expires: 30
- Lines: 10

SIP

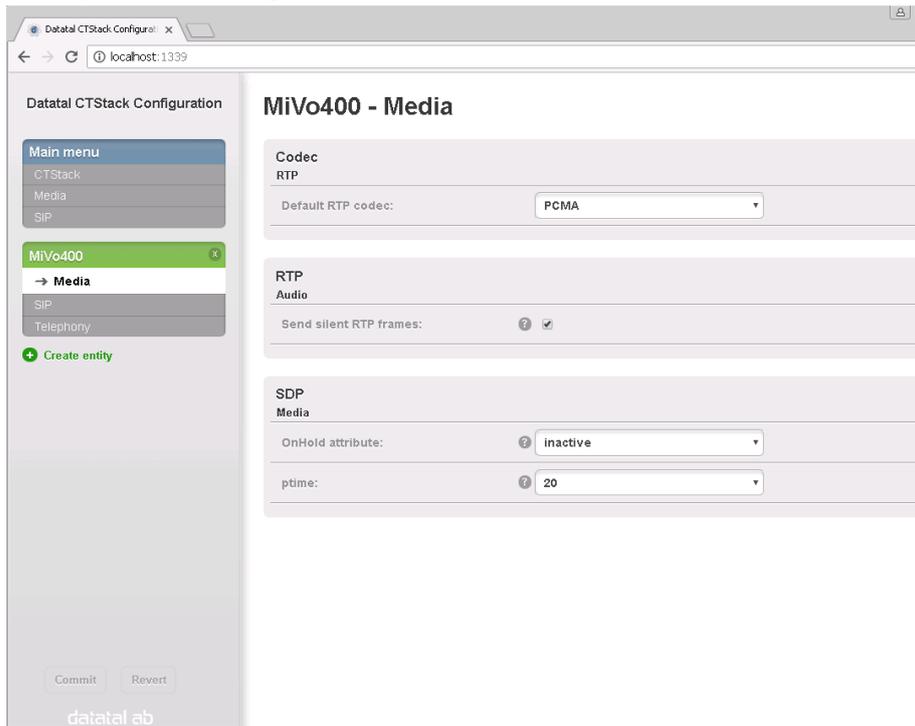
- Address
- Address: string
- Default domain: 192.168.255.40
- Default SIP URI host: 192.168.255.40
- Default SIP URI port: 5060
- Name: Flexi
- Profile
- Apply: None
- Current: A400 (user)

At the bottom of the sidebar, there is a status indicator "5 change(s) pending" and two buttons: "Commit" and "Revert". The Datatal logo is visible at the bottom of the sidebar.

7. Click Commit and Commit changes



8. Media, leave it unchanged



9. STOP and START CTStack and CTENGINE services, when number of lines are change these two service has to be restarted

Services				
Name	Description	Status	Startup Type	Log On As
ASP.NET State Service	Provides s...		Manual	Network S...
Background Intelligent Transfer Service	Transfers f...	Started	Automatic (D...	Local System
Base Filtering Engine	The Base F...	Started	Automatic	Local Service
Certificate Propagation	Copies use...	Started	Manual	Local System
CNG Key Isolation	The CNG k...	Started	Manual	Local System
COM+ Event System	Supports S...	Started	Automatic	Local Service
COM+ System Application	Manages t...	Started	Manual	Local System
Computer Browser	Maintains a...		Disabled	Local System
Credential Manager	Provides s...		Manual	Local System
Cryptographic Services	Provides fo...	Started	Automatic	Network S...
Datatal Datsat Agent	Service so...	Started	Automatic	Local System
Datatal Datsat Engine	Tjänst som...	Started	Automatic	Local System
Datatal Flexi Active State	Componen...	Started	Automatic	Local System
Datatal Flexi ActiveState	Handles Fl...		Disabled	Local System
Datatal Flexi Call Center	Service ma...	Started	Automatic	Local System
Datatal Flexi CC Statistics	Service tha...	Started	Automatic	Local System
Datatal Flexi CSTA III	Service tha...	Started	Automatic	Local System
Datatal Flexi CT Engine	Handle inc...	Started	Automatic	Local System
Datatal Flexi CT Stack	Service tha...	Started	Automatic	Local System
Datatal Flexi dbSyncronizer	Synkronise...	Started	Automatic	Local System
Datatal Flexi Diversion Manager	Hanterar h...	Started	Automatic	Local System
Datatal Flexi Exchange Connector	Service tha...	Started	Automatic	Local System
Datatal Flexi Hvd Proxy	Service wh...	Started	Automatic	Local System
Datatal Flexi Interop Service	Service tha...	Started	Automatic	Local System
Datatal Flexi Log Server	Tjänst som...	Started	Automatic	Local System
Datatal Flexi Maintenance Server	Service tha...	Started	Automatic	Local System
Datatal Flexi Message Manager	Transports...	Started	Automatic	Local System
Datatal Flexi Notification Service	Service tha...	Started	Automatic	Local System
Datatal Flexi Queue Server	Service ma...	Started	Automatic	Local System
Datatal Flexi Skype Connector	This servic...	Started	Automatic	Local System
Datatal Flexi SMS Server	Service wh...	Started	Automatic	Local System
Datatal Flexi Telephony Manager	Service tha...	Started	Automatic	Local System
Datatal Flexi Tid Proxy	Service ma...	Started	Automatic	Local System
Datatal Flexi Tid Server	Service wh...	Started	Automatic	Local System
Datatal Flexi Update Server	Service ma...		Disabled	Local System
Datatal Flexi Voice manager	Generate n...	Started	Automatic	Local System
Datatal License Manager	Uppdatera...	Started	Automatic	Local System
Datatal System Health	Monitor Da...	Started	Automatic	Local System

Extended

Standard

Good to know

Scenario 1

Problem

An incoming call from a customer to Flexi, customers press DTMF 1 for support. Flexi transfer the call to an extension in PBX. Support doesn't answer the call and the call goes back to Flexi after 30 seconds, previous the problem here was that PBX signal the first extension that customer called. That means that if the call was transfer by Flexi it would not reach supports voicemail.

Solution

Set CFNR destination for internal calls to nothing and goto 2.3.3.2 in AMS configuration. Set Recall time normal to lower than users internal call delay. When no answer is timed PBX will now be signaled that Support didn't answer to Flexi on SIP-diversion-header

Default call forwarding if no answer	
Internal call delay:	30
CFNR destination for internal calls:	
External call delay:	30
CFNR destination for external calls:	
Priority over activated CFNR:	No

Recall / ringing duration	
Recall time normal:	25 s
Recall time for virtual user:	600 s
Internal ringing duration:	10800 s
User group	
User group transfer delay:	3 rings

Scenario 2

Problem

An incoming call to a user, user answer call and start to speak with the customer. User decide to transfer call to Support queue that uses Flexi CC. Before was a recommendation from Datatal that user should transfer blind to Flexi. Reason to that was that Flexi wasn't able to transfer call to an agent when user already has a park call.

Solution

If a user want's to transfer a call to Flexi it's recommended to user "park" first call instead of pressing "new call" on Mitel-phone.

Scenario 3

This is more of a recommendation, if you have a user that works as an operator in Mitel 1560 then Flexi should transfer to extension 9 in PBX.

Scenario 4

Problem

Operator that not uses Mitel 1560 thinks that they have no overview of the queue from Flexi. When a customer press 9 in voicemail to reach operator. It could be a good idea to transfer call to an CDE with an number.

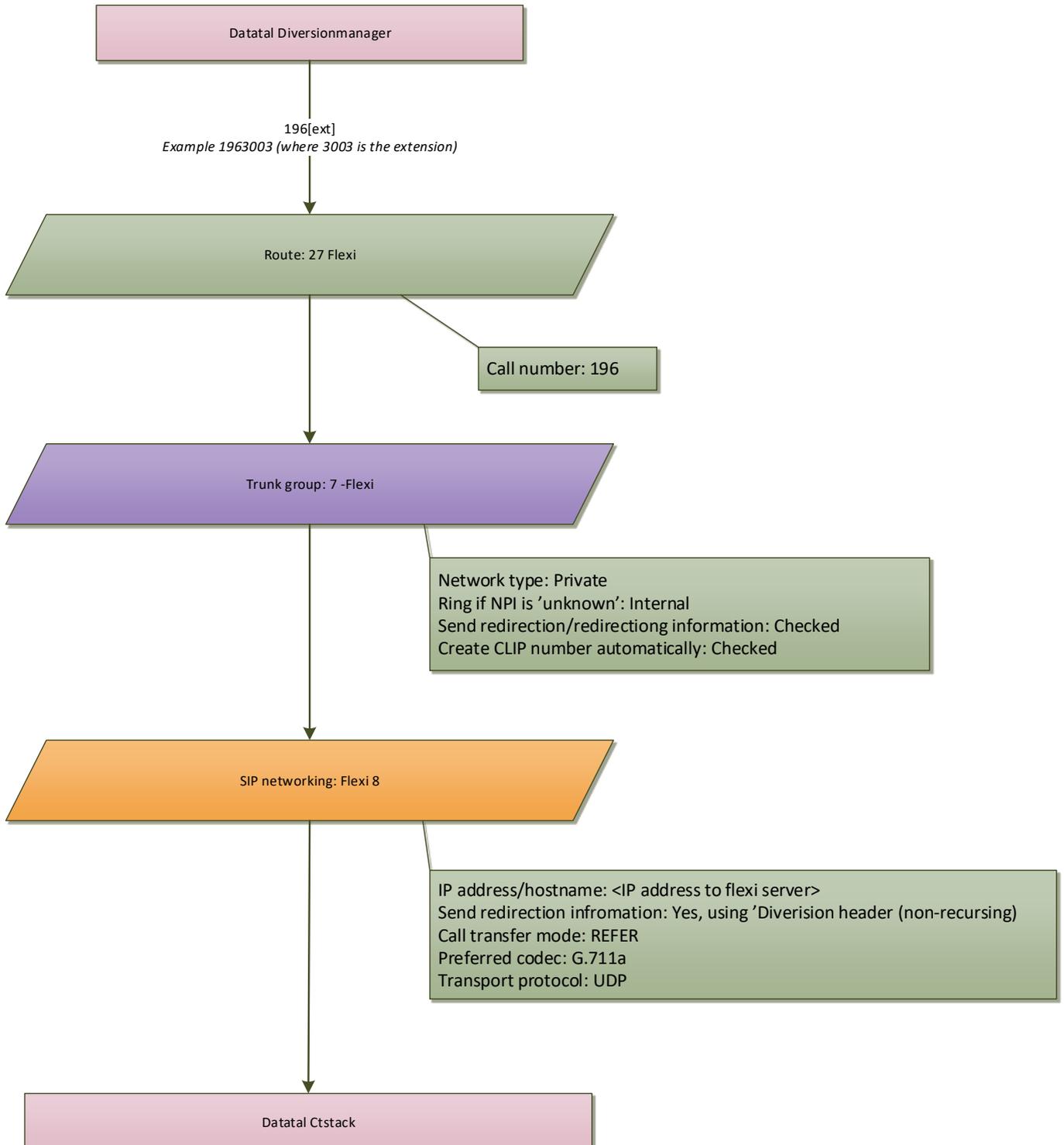
>	27	Flexi-ko	5995	1	A	Yes	Yes	No
27 of 272 record(s)								
CDE: 27 CDE in use: Yes Name: Flexi-ko Force showing the DDI number: No Call number: 5995 CDE if no answer: [up/down] Switch group: 1 CDE forwarding time: 30 Company: A CDE if busy: [up/down] Music on hold: As defined in 'Music on Hold' [CM_5.3] Show forwarding information instead of CDE name: Yes Welcome announcement: [down] Show CDE name at first position: No Enter ICL data: Yes Enable PSTN overflow (AIN): Yes [Delete unused CDE]								

>	27	Flexi-ko	5995	1	A	Yes	Yes	No																																				
27 of 272 record(s)																																												
<table border="1" style="width: 100%;"> <thead> <tr> <th></th> <th>Switch position 1</th> <th>Switch position 2</th> <th>Switch position 3</th> </tr> </thead> <tbody> <tr> <td>Destination</td> <td>KT</td> <td>KT</td> <td>KT</td> </tr> <tr> <td>User group</td> <td>[up/down]</td> <td>[up/down]</td> <td>[up/down]</td> </tr> <tr> <td>User group name</td> <td>[text]</td> <td>[text]</td> <td>[text]</td> </tr> <tr> <td>User</td> <td>[text]</td> <td>[text]</td> <td>[text]</td> </tr> <tr> <td>User name</td> <td>[text]</td> <td>[text]</td> <td>[text]</td> </tr> <tr> <td>Welcome announcement</td> <td>[down]</td> <td>[down]</td> <td>[down]</td> </tr> <tr> <td>Queue</td> <td>[down]</td> <td>[down]</td> <td>[down]</td> </tr> <tr> <td colspan="4" style="text-align: center;">Copy routing settings</td> </tr> </tbody> </table>										Switch position 1	Switch position 2	Switch position 3	Destination	KT	KT	KT	User group	[up/down]	[up/down]	[up/down]	User group name	[text]	[text]	[text]	User	[text]	[text]	[text]	User name	[text]	[text]	[text]	Welcome announcement	[down]	[down]	[down]	Queue	[down]	[down]	[down]	Copy routing settings			
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User group	[up/down]	[up/down]	[up/down]																																									
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Welcome announcement	[down]	[down]	[down]																																									
Queue	[down]	[down]	[down]																																									
Copy routing settings																																												

Terminal data							
Terminal ID	Description	Free seating	Call number	Name	Terminal interface	Terminal subtype	Cor
34	Magnus Backström sidolin	no	3003	Magnus Sidolin	IP	Key telephone	Aastra 5370p
153 of 243 record(s)							
Key number	Key type	Function	Number memory 1	Name memory 1			
Presence	Function	Presence menu					
1	External line			Flexi-ko			
2	External line			Flexi-ko			
3							
4							
5							
6							
7							
8							
9							
10							
11							
12	Personal line						

Setup for MiVO 400 trunk to Flexi

Overview



Route configuration

196 must be configured in Flexi

[Apply](#) [Reload](#)

Route

Route	27
Call number	<input type="text" value="196"/>
Name	<input type="text"/>
Max. outgoing calls	24
Total B channels	30
Send access code	<input type="text"/>
Send delay	0
External digit barring	<input checked="" type="checkbox"/>
Numbering plan identifier (NPI)	E.164
Suppress LCR	<input type="checkbox"/>
Impulse interval for virtual charges (s)	<input type="text"/>

Trunk group allocation

[Add](#)

- 7 IPFlexi

[> Connected users](#)

Trunk groups configuration

Trunk group	
Trunk group	7
Name	PFlexi <input type="text"/>
Type of trunk group	SIP networking
Maximum incoming calls	24 <input type="text"/>
Maximum outgoing calls	30 <input type="text"/>
Maximum simultaneous connections	30 <input type="text"/>
Total B channels	30
Call distribution element	1 <input type="text"/>
DDI(DID) plan	None <input type="text"/>
DDI(DID) cut	0 <input type="text"/>
DDI(DID) lookup	Left to right <input type="text"/>
Trunk line selection mode	Linear <input type="text"/>
Transit route	None <input type="text"/>
Region	None <input type="text"/>
Emergency location	Inherit <input type="text"/>
Emergency location protocol	No location ID <input type="text"/>
Networking	
Network type	Private <input type="text"/>
Ring if NPI is 'Unknown'	Internal <input type="text"/>
Cut CLIP	<input type="text"/>
Overwrite NPI	No <input type="text"/>

Ring back tone for incoming calls	Do not generate	▼
Ring back tone for outgoing calls	Generate	▼
Send immediate release in case of busy	<input type="checkbox"/>	
Early media support for Lync	<input type="checkbox"/>	
▼ Features		
Notification		
Send notifications	<input type="checkbox"/>	
Send redirection/redirecting information	<input checked="" type="checkbox"/>	
ECT information	<input checked="" type="checkbox"/>	
Mobile/external phone integration		
Allow CLIP authentication even if CLIP is not screened	<input checked="" type="checkbox"/>	
Allow enhanced functionality for direct incoming calls	<input type="checkbox"/>	
Blacklist		
Apply blacklist to incoming calls	<input type="checkbox"/>	Go to blacklist
▼ Call identification (CLIP)		
Outgoing CLIP		
Create CLIP number automatically	<input checked="" type="checkbox"/>	
Numbering plan identifier (NPI)	Unknown	▼
CLIP number	<input type="text"/>	
Restrict call identification (CLIR)	<input type="checkbox"/>	
CLIR for redirection	<input checked="" type="checkbox"/>	
Restrict call identification while	<input type="checkbox"/>	

▼ Network interfaces	
Interface	Interface type
8 - IPFlexi	SIP-T

IP address must be the IP to Flexi Server

Apply
Reload
✕

General

SIP node 8

Name

Bandwidth control area 1 - Default Area

Trunk group 7 - IPFlexi

Maximum incoming calls 24

IP addressing

IP address / host name 192.168.255.36

Port 5060

SIP signalling

Use '+' as international prefix

Try to make external calls: Timeout (s) 8

'From' field for CLIR Anonymous (RFC 3261)

Send session refresh (RFC 4028)

Use destination URL from 'To' field

Music on hold

Music on hold: Signalling Automatic

Send redirecting information Yes, using 'Diversion header (non-recurring)'

Call transfer mode REFER

PRACK support (RFC 3262)

Session replacement support

Audio settings

Preferred codec G.711a

Audio settingsPreferred codec G.711a ▼Comfort noise support Off ▼RTCP support Off ▼**NAT**Enable keep alive ALG support Relay RTP data via
communication server (indirect
switching) **Authentication**Local authentication required User name Password Show password**Transport protocol**Transport protocol UDP ▼

Datatal CTStack Trunk mode

Datatal CTStack Configuration

Main menu

- CTStack
- Media
- SIP

MiVo400 x

- Media
- SIP
- Telephony

[+ Create entity](#)

MiVo400 - Telephony

Line configuration

Standard

BlindCall source mode:

Description:

INVITE expires:

Lines:

SIP

Address

Address:

Default domain:

Default SIP URI host:

Default SIP URI port:

Name:

Profile

Apply:

Current:

Trunk

Trunk mode:

SIP Settings for CTStack

- No registrations
- Transport UDP

The screenshot shows the Datatal CTStack Configuration web interface. The browser address bar displays 'localhost:1339'. The interface is divided into a left sidebar and a main content area.

Left Sidebar:

- Datatal CTStack Configuration**
- Main menu**
 - CTStack
 - Media
 - SIP
- MiVo400** (with close icon)
 - Media
 - SIP
 - Telephony
- + Create entity**

Main Content Area:

- Use 'Remote-Target' in 'Refer-To':**
- Wait for park complete on MakeCall:**
- Registrations Users**
 - Registrations:** (empty list with a question mark icon)
 - ADD** (green button)
 - EDIT** (blue button)
 - REMOVE** (red button)
- SIP Dialogs**
 - Use 'From' header:**
 - RFC 3325**
 - P-*Identity mode:**
 - Transport**
 - Transport:**

Bottom of Sidebar:

- 1 change(s) pending
- Commit** (green button)
- Revert** (red button)
- datatal ab