



OpenScape Business

Tutorial

Networking OpenScape Business – OpenScape Voice
Configuration Guide

Version: 1.0

Contents

1.1. GENERAL	4
1.1.1. Prerequisites.....	4
1.1.2. Features and Restrictions in Networking	4
1.2. CONFIGURATION FOR OPENSCAPE BUSINESS	5
1.2.1. Basic Settings configuration in OpenScape Business Assistant	5
1.2.2. Trunk configuration in OpenScape Business Assistant.....	8
1.2.2.1. Telephony Server -> Trunks/Routing -> Trunks -> Lan.....	8
1.2.2.2. Telephony Server -> Trunks/Routing -> Route (CO Route).....	8
1.2.2.3. Telephony Server -> Trunks/Routing -> Route (SIP INT2 Route)	10
1.2.2.4. Telephony Server -> Trunks/Routing -> Route -> Routing Parameters (SIP INT2 Route)	12
1.2.3. Voice Gateway configuration in OpenScape Business Assistant	13
1.2.3.1. Expert Mode -> Telephony Server -> Voice Gateway -> Codec Parameters.....	13
1.2.3.2. Expert Mode -> Telephony Server -> Voice Gateway -> SIP Parameters	14
1.2.3.3. Voice Gateway -> SIP Interconnection -> OpenScape Voice (IP Address).....	15
1.2.3.4. Voice Gateway -> SIP Interconnection -> OpenScape Voice (DNS-SRV)	17
1.2.3.5. "Voice Gateway -> SIP Interconnection -> OpenScape Voice (Digest Authentication)"	19
1.2.4. TLS Configuration.....	20
1.2.4.1. "Telephony Server -> Basic Settings -> Date and Time -> SNTP Settings"	20
1.2.4.2. "Telephony Server -> Basic Settings -> System -> System Flags"	21
1.2.4.3. "Telephony Server -> Security -> Signaling and Payload Encryption"	22
1.2.4.4. "Voice Gateway -> Station -> Station Payload Security"	24
1.2.4.5. "Voice Gateway -> SIP Interconnection -> OpenScape Voice (TLS)"	25
1.2.5. Least Cost Routing.....	28
1.2.5.1. Basic settings.....	28
1.2.5.2. Least cost routing to CO (Carrier).....	29
1.2.5.3. Least cost routing to OpenScape Voice.....	37
1.3. CONFIGURATION OF OPENSCAPE VOICE	40
1.3.1. Settings in Common Management Portal.....	40
1.3.1.1. Administration of a new Gateway Private Numbering Plan.....	41
1.3.1.2. Administration of an Endpoint Profile for an OpenScape Business Endpoint	41
1.3.1.3. Creating and configuring an endpoint for OpenScape Business	45
1.3.1.4. Creating "Digest Authentication" access for OpenScape Business.....	49
1.3.1.5. Configuring a Gateway Numbering Plan for Incoming Calls	51
1.3.1.6. Configuring Outgoing Calls.....	53
1.3.1.7. Display Number Modification for OpenScape Voice V7 R1	62
1.3.2. Settings in StartCli	70
1.4. CONFIGURATION OPENBRANCH	71
1.4.1. Network Services	71
1.4.2. VOIP	77

History

Issue	Date	Reason for Changes
1.0	04/12/2013	Initial creation of version 1.0

1.1. General

This section describes the configuration of SIP-Q networking between OpenScape Business and OpenScape Voice. To do this, you will have to configure settings on OpenScape Business Assistant. On OpenScape Voice, settings are made in the Common Management Portal and in StartCli. The Common Management Portal is also used for OpenScape Branch.

Contents

This section covers the following topics:

Section 1.1.2, "Features and Restrictions in Networking"

Section 1.2, "Configuration of OpenScape Business"

Section 1.3, "Configuring OpenScape Voice"

Section 1.4, "Configuring OpenScape Branch"

1.1.1. Prerequisites

- OpenScape Business Version 1
- OpenScape Voice Version 7 R1

1.1.2. Features and Restrictions in Networking

- Only a star networking topology is supported. This means that OpenScape Voice is always the central switch.
- OpenScape Business supports DNS SRV for OpenScape Voice Interworking.
- OpenScape Business is released in a branch environment only with OpenScape Branch proxies and DNS-SRV configuration.
- OpenScape Business Survivability features are only released with DNS-SRV configuration.
- T.38 Fax protocol is supported.
- Two OpenScape Business endpoints cannot be networked directly via IP Network.
- Path replacement is not supported and must be deactivated.
- Path optimization is not supported and must be deactivated.
- Rerouting is not supported and must be deactivated.
- MoH is not available when proxy is in Limited Mode. The connection to OpenScape Business remains silent when on hold.
- SIP registration (dynamic endpoint) configuration is only possible in direct topology (no proxies) using IP address scheme configuration.
- In case of DNS-SRV configuration OpenScape Business must be configured as static endpoint from OpenScape Voice side. OpenScape Voice is configured only as SIP server and not as SIP registrar.

1.2. Configuration for OpenScape Business

Contents

This section covers the following topics:

Section 1.2.1, "Basic Settings configuration in OpenScape Business Assistant"

Section 1.2.2, "Trunk configuration in OpenScape Business Assistant"

Section 1.2.3, "Voice Gateway configuration in OpenScape Business Assistant"

Section 1.2.4, "TLS Configuration"

Section 1.2.5, "Least Cost Routing configuration in OpenScape Business Assistant"

1.2.1. Basic Settings configuration in OpenScape Business Assistant

Procedure

1. Open OpenScape Business Assistant.
2. **Expert Mode ->Telephony Server -> Basic Settings -> System -> System flags.** The window "System Flags" is displayed.
3. In the "System Flags" area, make the following settings:
 - Path optimization: Deactivate the checkbox.
 - E.164 numbering scheme: Activate the checkbox (option available only in ManagerE).
 - If TLS is available:
 - SPE support: Activate the checkbox.
 - SPE advisory tone: Activate the checkbox.
4. In the "Transit permission" area, make the following settings:
 - External traffic transit: Activate the checkbox.
5. Click **Apply**. Your changes are saved.

Expert mode

► Maintenance

▼ Telephony Server

Basic Settings

Expert mode

Expert Mode is intended for advanced configuration by trained technicians of your Service Provider. If you use Expert Mode to

Expert mode - Telephony Server

Basic Settings

▼ System

System Flags

Time Parameters

Display

DISA

Intercept/Attendant/Hotline

LDAP

Texts

Flexible menu

Speed Dials

Service Codes

Gateway

► DynDNS

AF/EF Codepoints

Quality of Service

► Date and Time

Port Management

► Call Charges

Voicemail / Announcement Player

*!Alarm Signaling

*!Phone Parameter Deployment

System Flags

Edit System Flags

System flags

Through-connection for external FWD on:

Call forwarding to main station interface permitted:

Hunting to external call forwarding destination:

Conference tone:

Warning signal for call pickup groups:

Increase volume for optiPoint/OpenStage terminals:

Relocate allowed:

More than 1 external conference member:

Trunk reservation, automatic:

No. redial with a/c code:

Use only default number for MSN :

Path optimization:

DTMF automatic:

- Automatic redial:
- Voice mail Node call number:
- Call Pickup after automatic recall:
- Configurable CLIP:
- Caller list at destination in case of Forward Line:
- Call forwarding after deflect call / single step transfer:
- Follow call management in case of deflect call / single step transfer:
- Calling number in pick-up groups / ringing groups / CFN /RNA:
- SPE support:
- SPE advisory tone:
- SIP Prov. to SIP Prov. transit:
- Transparent dialing of * and # on trunk interfaces:

Open numbering scheme

active:

Node callnumber:

Transit permission

Feature transit:

Tie traffic transit:

External traffic transit :

Restriction for UC calls

Restriction for UC calls:

1.2.2. Trunk configuration in OpenScape Business Assistant

OpenScape Business supports two SIP interconnection trunks. One SIP interface must be assigned to the Osbiz - OSV interconnection. Usage of "SIP Interconnection 2" trunk type is recommended.

Contents

In Expert Mode of OpenScape Business Assistant, changes will be made on the following screens:

Section 1.2.2.1, "Telephony Server -> Trunks/Routing -> Trunks -> Lan"

Section 1.2.2.2, "Telephony Server -> Trunks/Routing -> Route (CO Route)"

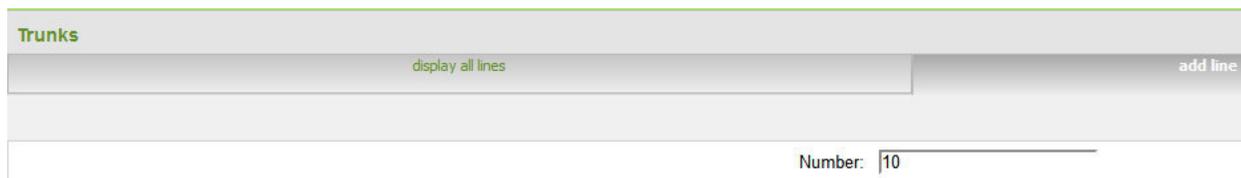
Section 1.2.2.3, "Telephony Server -> Trunks/Routing -> Route (SIP INT2 Route)"

Section 1.2.2.4, "Telephony Server -> Trunks/Routing -> Route -> Routing Parameters (SIP INT2 Route)"

1.2.2.1. Telephony Server -> Trunks/Routing -> Trunks -> Lan

1. **Expert Mode** -> **Telephony Server** -> **Trunks/Routing** -> **Trunks** -> **LAN** -> select SIP Interconnection 2. Information about SIP Interconnection 2 trunk type will be displayed in "display all lines" tab.

2. Select "add line" tab and select the number of SIP Interconnection 2 trunks that will be created (i.e 10).



Trunks

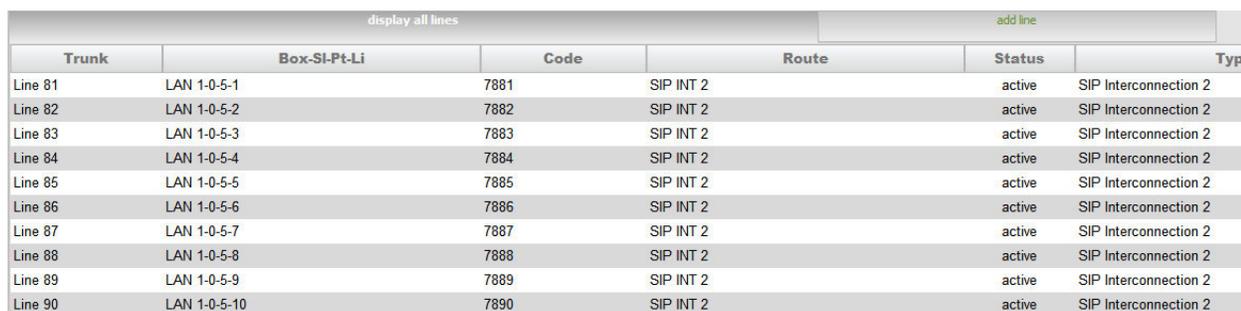
display all lines

add line

Number: 10

3. Click **Apply**. Your changes are saved.

4. Under "display all lines" tab, newly created trunks will be appeared and they will be automatically assigned to SIP INT 2 by default (Trunk Group 11).



Trunk	Box-SI-Pt-Li	Code	Route	Status	Type
Line 81	LAN 1-0-5-1	7881	SIP INT 2	active	SIP Interconnection 2
Line 82	LAN 1-0-5-2	7882	SIP INT 2	active	SIP Interconnection 2
Line 83	LAN 1-0-5-3	7883	SIP INT 2	active	SIP Interconnection 2
Line 84	LAN 1-0-5-4	7884	SIP INT 2	active	SIP Interconnection 2
Line 85	LAN 1-0-5-5	7885	SIP INT 2	active	SIP Interconnection 2
Line 86	LAN 1-0-5-6	7886	SIP INT 2	active	SIP Interconnection 2
Line 87	LAN 1-0-5-7	7887	SIP INT 2	active	SIP Interconnection 2
Line 88	LAN 1-0-5-8	7888	SIP INT 2	active	SIP Interconnection 2
Line 89	LAN 1-0-5-9	7889	SIP INT 2	active	SIP Interconnection 2
Line 90	LAN 1-0-5-10	7890	SIP INT 2	active	SIP Interconnection 2

1.2.2.2. Telephony Server -> Trunks/Routing -> Route (CO Route)

One route to the Carrier must be configured. To do this, proceed as follows:

1. **Expert Mode** -> **Telephony Server** -> **Trunks/Routing** -> **Trunks** -> **Route** -> select Route 1. Configuration options for Route 1 are displayed.

2. For **Route 1**, make the following settings:

- Route Name: **CO**
- PABX number-incoming: Enter the location number in the E.164 format for the country code, local area code and PABX number, for example:30, 210, 650.
- Location number: Activate the checkbox.

4. Click **Apply**. Your changes are saved.

Route	
Change Route	Change Routing Parameters
	Route Name: <input type="text" value="CO"/>
	Seizure code: <input type="text" value="0"/>
	CO code (2nd trunk code): <input type="text"/>
Gateway Location	
	Country code: <input type="text" value="30"/>
	Local area code: <input type="text" value="210"/>
	PABX number: <input type="text" value="650"/>
PABX number-incoming	
	Country code: <input type="text" value="30"/>
	Local area code: <input type="text" value="210"/>
	PABX number: <input type="text" value="650"/>
	Location number: <input checked="" type="checkbox"/>
PABX number-outgoing	
	Country code: <input type="text"/>
	Local area code: <input type="text"/>
	PABX number: <input type="text"/>
	Suppress station number: <input type="checkbox"/>
Overflow route	
	Overflow route : <input type="text" value="None"/>
Digit transmission	
	Digit transmission: <input type="text" value="Digit-by-digit"/>

1.2.2.3. Telephony Server -> Trunks/Routing -> Route (SIP INT2 Route)

SIP INT 2 route to OpenScape Voice must be configured. To do this, proceed as follows:

1. **Expert Mode** -> **Telephony Server** -> **Trunks/Routing** -> **Trunks** -> **Route** -> select SIP INT 2 route (route 11). Configuration options for Route SIP INT 2 are displayed.

2. For **SIP INT 2**, make the following settings:

- Route Name/Name: **SIP INT2**
- 2nd trunk code: Enter **0** for the local PSTN access code
- PABX number-incoming: Enter the location number in the E.164 format for the country code, local area code and system number. This number is generally the same as the number entered for route "CO" (see previous section), in this example: **30, 210, 650**.
- Digit transmission: **en-bloc sending**

4. Click **Apply**. Your changes are saved.

Route	
Change Route	Change Routing Parameters
	Route Name: <input type="text" value="SIP INT 2"/> Seizure code: <input type="text"/> CO code (2nd trunk code): <input type="text" value="0"/>
Gateway Location	Country code: <input type="text" value="30"/> Local area code: <input type="text" value="210"/> PABX number: <input type="text" value="650"/>
PABX number-incoming	Country code: <input type="text" value="30"/> Local area code: <input type="text" value="210"/> PABX number: <input type="text" value="650"/> Location number: <input type="checkbox"/>
PABX number-outgoing	Country code: <input type="text"/> Local area code: <input type="text"/> PABX number: <input type="text"/> Suppress station number: <input type="checkbox"/>
Overflow route	Overflow route : <input type="text" value="None"/>
Digit transmission	Digit transmission: <input type="text" value="en-bloc sending"/>

1.2.2.4. Telephony Server -> Trunks/Routing -> Route -> Routing Parameters (SIP INT2 Route)

You must set routing parameters for the "Hp8k" route created in the previous section. To do this, proceed as follows:

1. **Expert Mode** -> **Telephony Server** -> **Trunks/Routing** -> **Trunks** -> **Route** -> **SIP INT2** -> select **Routing parameters**. The tab "Change Routing Parameters" for SIP INT 2 route is displayed.

2. For "SIP INT 2", make the following settings:

- **Always use DSP**: Activate the checkbox.
- **Route type**: **PABX**
- **No. and type, outgoing**: **Country code**
- **Call number type**: **Internal / DID**
- **Route optimize active**: **No**

4. Click **Apply**. Your changes are saved.

Route

Change Route | Change Routing Parameters

Routing flags

Digit repetition on:

Analysis of second dial tone / Trunk monitoring:

Intercept per direction:

Over. service 3.1 kHz audio:

Add direction prefix incoming:

Add direction prefix outgoing:

Ringback tone to CO:

Segmentation: yes

deactivate UUS per route:

Always use DSP:

Analog trunk seizure: no pause

Trunk call pause: Pause 6 s

Type of seizure: linear

Route type: PABX

No. and type, outgoing: Country code

Call number type: Internal / DID

Rerouting

Change route allowed:

Route optimize active: No

1.2.3. Voice Gateway configuration in OpenScape Business Assistant

This section covers the following topics:

Section 1.2.3.1, "Voice Gateway -> Codec Parameters"

Section 1.2.3.2, "Voice Gateway -> SIP Parameters"

Section 1.2.3.3, "Voice Gateway -> SIP Interconnection -> OpenScape Voice (IP Address)"

Section 1.2.3.4, "Voice Gateway -> SIP Interconnection -> OpenScape Voice (DNS-SRV)"

Section 1.2.3.5, "Voice Gateway -> SIP Interconnection -> OpenScape Voice (Digest Authentication)"

1.2.3.1. Expert Mode -> Telephony Server -> Voice Gateway -> Codec Parameters

Proceed as follows:

1. Go to **Expert Mode -> Telephony Server -> Voice Gateway** and select **Codec Parameters**.
2. Configure **G711 A-law** with **Priority 1** and **20ms** as Frame Size.
3. Configure **G711 μ -law** with **Priority 2** and **20ms** as Frame Size.
4. **T.38 Fax**: **Activate** the checkbox.
5. Use **FillBitRemoval**: **Activate** the checkbox.
6. **Error Correction Used for T.38 Fax (UDP)**: Select **t38UDPRedundancy**
7. **Transmission of Fax/Modem Tones according to RFC2833**: **Activate** the checkbox
8. **Transmission of DTMF Tones according to RFC2833**: **Activate** the checkbox
9. **Payload Type for RFC2833**: Configure value **98**
10. Click **Apply**. Your changes are saved.

Codec Parameters			
Edit Codec Parameters			
Codec	Priority	Voice Activity Detection	Frame Size
G.711 A-law	Priority 1 ▾	VAD: <input type="checkbox"/>	20 ▾ msec
G.711 μ-law	Priority 2 ▾	VAD: <input type="checkbox"/>	20 ▾ msec
G.729A	Priority 4 ▾	VAD: <input type="checkbox"/>	20 ▾ msec
G.729AB	Priority 3 ▾	VAD: <input checked="" type="checkbox"/>	20 ▾ msec
Enhanced DSP Channels			
Use G.711 only <input type="checkbox"/>			
T.38 Fax			
T.38 Fax: <input checked="" type="checkbox"/>			
Use FillBitRemoval: <input checked="" type="checkbox"/>			
Max. UDP Datagram Size for T.38 Fax (bytes): 1472			
Error Correction Used for T.38 Fax (UDP) t38UDPRedundancy ▾			
Misc.			
ClearChannel: <input checked="" type="checkbox"/> Frame Size: 20 ▾ msec			
RFC2833			
Transmission of Fax/Modem Tones according to RFC2833: <input checked="" type="checkbox"/>			
Transmission of DTMF Tones according to RFC2833: <input checked="" type="checkbox"/>			
Payload Type for RFC2833: 98			
Redundant Transmission of RFC2833 Tones according to RFC2198: <input checked="" type="checkbox"/>			

1.2.3.2. Expert Mode -> Telephony Server -> Voice Gateway -> SIP Parameters

Proceed as follows:

1. Go to **Expert Mode -> Telephony Server -> Voice Gateway** and select **SIP Parameters**.

2. Configure the SIP parameters to the default values. These are:

- SIP transport protocol
 - SIP via TCP: **Yes**
 - SIP via UDP: **Yes**
 - SIP via TLS: **Yes**
- SIP Registrar
 - Period of registration (sec): **120**
- RFC 3261 Timer Values
 - Transaction Timeout (msec): **32000**
- SIP session timer
 - RFC 4028 Support: **Yes**
 - Session Expires (sec): **1800**

- Minimal SE (sec): 90
- Provider calls
 - Maximum possible Provider Calls: 0

3. Click **Apply**. Your changes are saved.

SIP Parameters	
Edit SIP Parameters	
SIP Transport Protocol	
SIP via TCP:	Yes
SIP via UDP:	<input checked="" type="checkbox"/>
SIP via TLS:	Yes
SIP Registrar	
Period of registration (sec):	<input type="text" value="120"/>
RFC 3261 Timer Values	
Transaction Timeout (msec):	<input type="text" value="32000"/>
SIP Session Timer	
RFC 4028 support:	<input checked="" type="checkbox"/>
Session Expires (sec):	<input type="text" value="1800"/>
Minimal SE (sec):	<input type="text" value="90"/>
Provider Calls	
Maximum possible Provider Calls:	0

1.2.3.3. Voice Gateway -> SIP Interconnection -> OpenScape Voice (IP Address)

Direct interconnection to OpenScape Voice via SIP Registration is only release with IP Address configuration. No survivability features are possible with this type of interconnection.

Parameters for the OpenScape Voice must be set. To do this, proceed as follows:

1. **Expert Mode -> Telephony Server -> Voice Gateway -> SIP Interconnection** and select **OpenScape Voice**. The tab "Edit SIP Interconnection" is displayed.

2. Make the following settings in the "Edit SIP Interconnection" tab:

- Enable Trunk: **Activate** the checkbox.
- Trunk Identifier in System: Select **SIP-Interconnection2**.
- Remote Domain Name: IP Address of **sipsm1**.
- SIP Server
 - IP Address / Host name: IP Address of **sipsm1**.
 - Port: **5060**.
 - Secure Transport: **Deactivate** the check box.

- SIP Registrar
 - Use Registrar: **Activate** the checkbox.
 - IP Address / Host name: IP Address of **sipsm1**.
 - Port: **5060**.
 - Reregistration Interval (sec): **300**.
- Outbound Proxy (**Check box must be deactivated**)
- Inbound Proxy (**Check box must be deactivated**)

3. Click **Apply**. Your changes are saved.

Voice Gateway	SIP Interconnection		
SIP Parameters	Edit SIP Interconnection	Delete SIP Interconnection	Add SIP Interconnection User
Codec Parameters	Name: <input type="text" value="OpenScopeVoice"/>		
▶ Destination Codec Parameters	Enable Trunk: <input checked="" type="checkbox"/>		
▶ Internet Telephony Service Provider	Trunk Identifier in System: <input type="text" value="SIP-Interconnection2"/>		
▶ Networking	Remote Domain Name: <input type="text" value="192.168.140.231"/>		
▼ SIP Interconnection	SIP Server		
▶ Application Suite	IP Address / Host name: <input type="text" value="192.168.140.231"/>		
▶ HiPath 4000	Port: <input type="text" value="5060"/>		
▶ Native SIP Server trunk	Secure Transport: <input type="checkbox"/>		
▼ OpenScopeVoice	SIP Registrar		
OpenScopeVoice-User	Use Registrar: <input checked="" type="checkbox"/>		
▶ SIPQ Server trunk	IP Address / Host name: <input type="text" value="192.168.140.231"/>		
	Port: <input type="text" value="5060"/>		
	Reregistration Interval (sec) <input type="text" value="300"/>		
	Outbound Proxy		
	Use Outbound Proxy: <input type="checkbox"/>		
	IP Address / Host name: <input type="text" value="0.0.0.0"/>		
	Port: <input type="text" value="0"/>		
	Inbound Proxy		
	Use Inbound Proxy: <input type="checkbox"/>		
	IP Address / Host name: <input type="text" value="0.0.0.0"/>		
	Port: <input type="text" value="5060"/>		
	Apply	Undo	Refresh

1.2.3.4. Voice Gateway -> SIP Interconnection -> OpenScope Voice (DNS-SRV)

DNS SRV interconnection type assumes that: a) a DNS server has been installed and configured in the environment with the appropriate domains and SRV records, b) DNS SRV query for FQDN of OSV (sipsm1) is resolved from DNS server and c) the DNS server is configured in OpenScope Business.

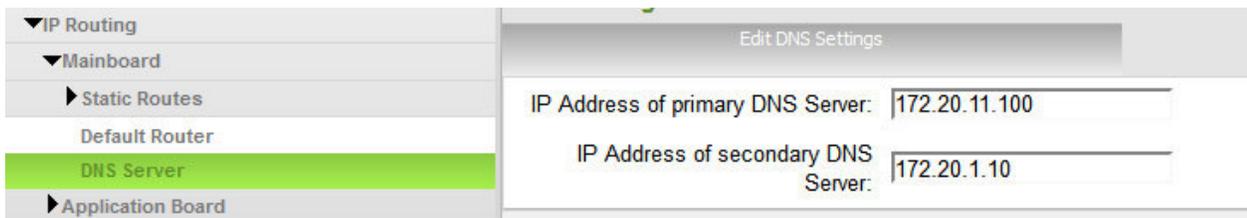
Openscape Business generates the followed type of DNS SRV query: `_sip._tcp.domain_name`. In case of OpenScope Branch configuration, sipsm1 and openscape branch must share the same Fully Qualified Domain Name (i.e. sip.osv.net)

- [-] Queries
 - ⊕ _sip._tcp.sip.osv.net: type SRV, class IN
- [-] Answers
 - ⊕ _sip._tcp.sip.osv.net: type SRV, class IN, priority 10, weight 100, port 5060, target openbranch.osv.net
 - ⊕ _sip._tcp.sip.osv.net: type SRV, class IN, priority 20, weight 100, port 5060, target sipsm1.osv.net

Survivability features are only available with DNS-SRV configuration.

Parameters for the DNS Server must be set. To do this, proceed as follows:

1. **Expert Mode -> Telephony Server -> Routing -> IP Routing -> Mainboard** and select **DNS Server** option. The tab "Edit DNS Settings" is displayed.
2. Make the following settings in the "Edit DNS Settings" tab:
 - IP Address of Primary DNS Server: IP Address of Primary DNS Server.
 - IP Address of Secondary DNS Server: IP Address of Secondary DNS Server (optional).
3. Click **Apply**. Your changes are saved.



Parameters for the OpenScape Voice must be set. To do this, proceed as follows:

1. **Expert Mode -> Telephony Server -> Voice Gateway -> SIP Interconnection** and select **OpenScape Voice**. The tab "Edit SIP Interconnection" is displayed.
2. Make the following settings in the "Edit SIP Interconnection" tab:
 - Enable Trunk: **Activate** the checkbox.
 - Trunk Identifier in System: Select **SIP-Interconnection2**.
 - Remote Domain Name: **Fully Qualified Domain Name** (i.e. sip.osv.net).
 - SIP Server
 - IP Address / Host name: **Fully Qualified Domain Name** (i.e. sip.osv.net).
 - Port: **0**.
 - Secure Transport: **Deactivate** the check box.
 - SIP Registrar (**Check box must be deactivated**)
 - Outbound Proxy (**Check box must be activated in case of OpenScape Brach**)
 - Use Outbound Proxy: **Activate** the checkbox.

- IP Address / Host name: **Fully Qualified Domain Name** (i.e. sip.osv.net).
- Port: **0**.
- Inbound Proxy (Check box must be deactivated)

3. Click **Apply**. Your changes are saved.

SIP Interconnection		SIP Interconnection	
Edit SIP Interconnection	Delete SIP Interconnection	Add SIP Interconnection User	Edit SIP Interconnection
	Delete SIP Interconnection	Add SIP Interconnection User	Delete SIP Interconnection
		Add SIP Interconnection User	Add SIP Interconnection User
Name: OpenScapeVoice		Name: OpenScapeVoice	
Enable Trunk: <input checked="" type="checkbox"/>		Enable Trunk: <input checked="" type="checkbox"/>	
Trunk Identifier in System: SIP-Interconnection2		Trunk Identifier in System: SIP-Interconnection2	
Remote Domain Name: sip.osv.net		Remote Domain Name: sip.osv.net	
SIP Server		SIP Server	
IP Address / Host name: sip.osv.net		IP Address / Host name: sip.osv.net	
Port: 0		Port: 0	
Secure Transport: <input checked="" type="checkbox"/>		Secure Transport: <input checked="" type="checkbox"/>	
SIP Registrar		SIP Registrar	
Use Registrar: <input type="checkbox"/>		Use Registrar: <input type="checkbox"/>	
IP Address / Host name: 0.0.0.0		IP Address / Host name: 0.0.0.0	
Port: 5060		Port: 5060	
Reregistration Interval (sec): 300		Reregistration Interval (sec): 300	
Outbound Proxy		Outbound Proxy	
Use Outbound Proxy: <input type="checkbox"/>		Use Outbound Proxy: <input checked="" type="checkbox"/>	
IP Address / Host name:		IP Address / Host name: sip.osv.net	
Port:		Port: 0	
Inbound Proxy		Inbound Proxy	
Use Inbound Proxy: <input type="checkbox"/>		Use Inbound Proxy: <input type="checkbox"/>	
IP Address / Host name:		IP Address / Host name:	
Port:		Port:	

1.2.3.5. "Voice Gateway -> SIP Interconnection -> OpenScape Voice (Digest Authentication)"

Parameters for the OpenScape Voice must be set. To do this, proceed as follows:

1. **Expert Mode -> Telephony Server -> Voice Gateway -> SIP Interconnection -> OpenScape Voice** and select **OpenScape Voice-User**. The tab "Edit SIP Interconnection User" is displayed.

Make the following settings in the "Edit SIP Interconnection User" window:

- UserId: i.e. **OpenScapeVoice-User**.

- Authorization name: i.e. **OpenScapeVoice-User**.
- Password: **Enter the password.**
- Confirm password: **Re-enter the password to confirm.**

The screenshot shows a web-based configuration interface. On the left is a navigation tree under 'Voice Gateway' with categories like 'SIP Parameters', 'Codec Parameters', 'Destination Codec Parameters', 'Internet Telephony Service Provider', 'Networking', 'SIP Interconnection', 'Application Suite', 'HiPath 4000', 'Native SIP Server trunk', and 'OpenScapeVoice'. The 'OpenScapeVoice' category is expanded, and 'OpenScapeVoice-User' is selected. The main content area is titled 'SIP Interconnection User' and contains two buttons: 'Edit SIP Interconnection User' and 'Delete SIP Interconnection User'. Below these are four input fields: 'UserId' (containing 'OpenScapeVoice-User'), 'Authorization name' (containing 'OpenScapeVoice-User'), 'Password' (masked with dots), and 'Confirm Password' (masked with dots).

1.2.4. TLS Configuration

This section covers the following topics:

Section 1.2.4.1, "Telephony Server -> Basic Settings -> Date and Time -> SNTP Settings"

Section 1.2.4.2, "Telephony Server -> Basic Settings -> System -> System Flags"

Section 1.2.3.3, "Telephony Server -> Security -> Signaling and Payload Encryption"

Section 1.2.3.5, "Telephony Server -> Station -> Station Payload Security"

Section 1.2.3.6, "Voice Gateway -> SIP Interconnection -> OpenScape Voice (TLS)"

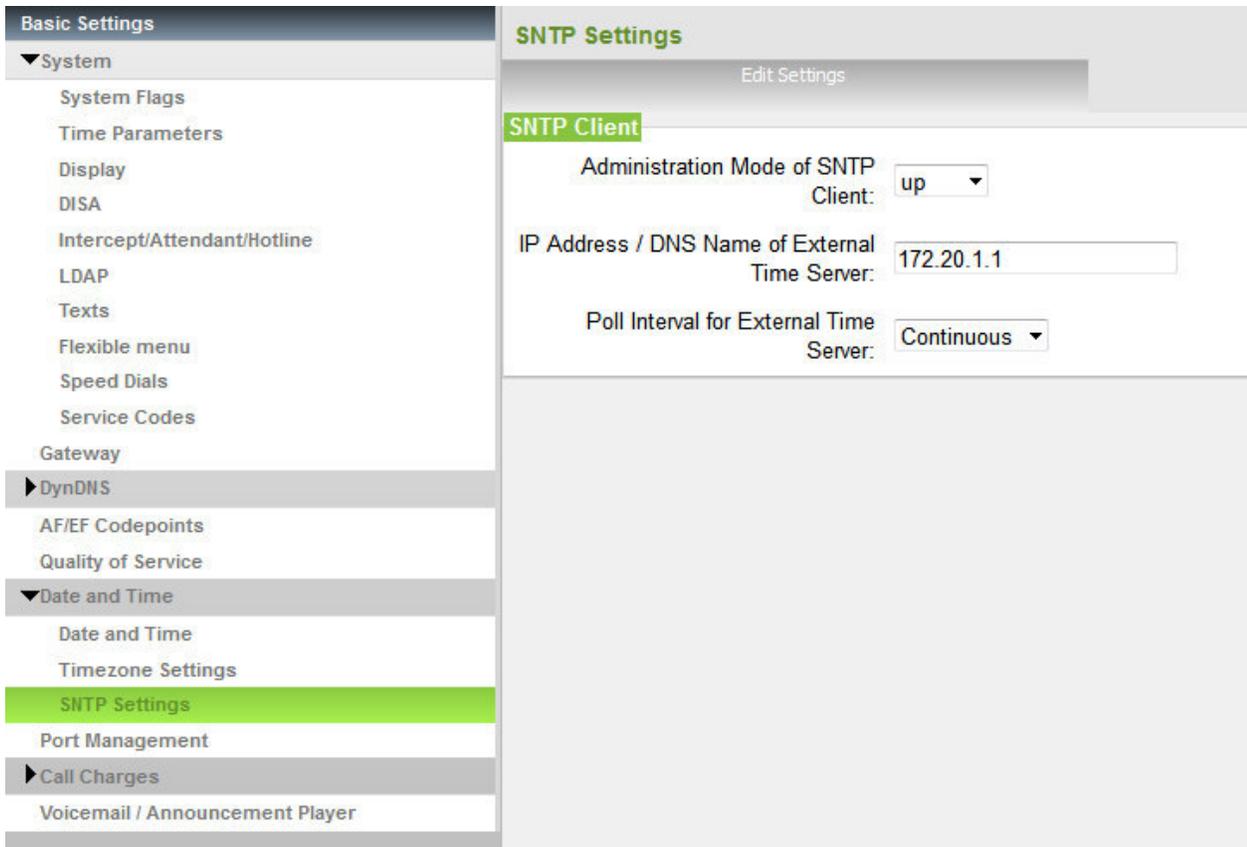
1.2.4.1. "Telephony Server -> Basic Settings -> Date and Time -> SNTP Settings"

Enter the NTP server address. To do this, proceed as follows:

1. **Telephony Server -> Basic Settings -> Date and Time ->** select **SNTP Settings**. The "Edit SNTP Settings" tab is displayed.
2. In the "Edit SNTP Settings" area, make the following settings:
 - Administration Mode of SNTP Client: Select **UP**
 - IP Address / DNS Name of External Time Server: Enter the **IP address** of SNTP Server
 - Poll Interval for External Time Server: Configure "**Continuous**" Polling

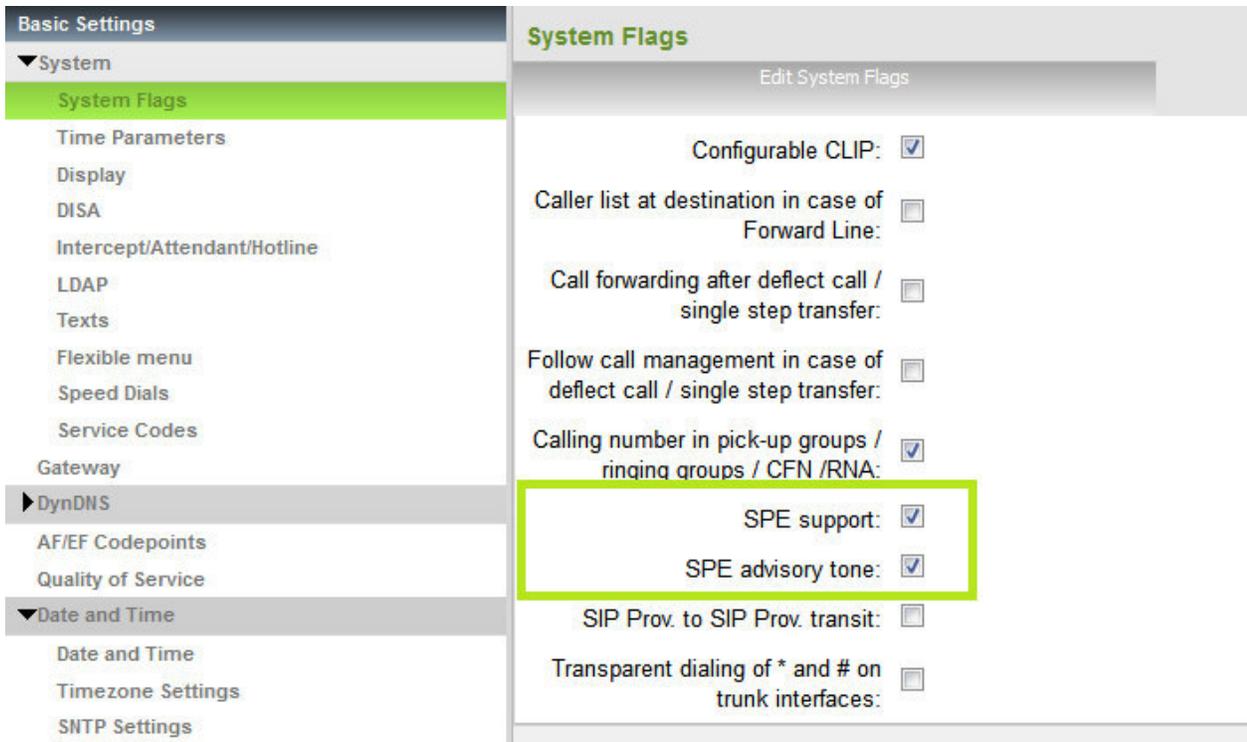
Attention: For TLS connections, the same time must be set for all connections. This is achieved by all components using the same NTP server.

3. Click **Apply**. Your changes are saved.



1.2.4.2. "Telephony Server -> Basic Settings -> System -> System Flags"

1. Go to **Expert Mode** -> **Telephony Server** -> **Basic Settings** -> **System** and select **System flags**. The window "System Flags" is displayed.
2. In the "System Flags" area, make the following settings:
 - SPE support: **Activate** the checkbox.
 - SPE advisory tone: **Activate** the checkbox.
3. Click **Apply**. Your changes are saved.
4. Changes are only applied after system restart.



1.2.4.3. "Telephony Server -> Security -> Signaling and Payload Encryption"

Overview

For Signaling and Payload Encryption via TLS two Certificates must be imported to OpenScape Business (certificate with private key & CA certificate).

Certificates exported from OpenScape Voice must be used (rootcert.pem and client.pem).

client.pem is located in OpenScape Voice in the folder /usr/local/ssl/private

rootcert.pem must be created by using the root.pem.

Copy the file root.pem from OpenScape Voice /usr/local/ssl/certs and rename it to rootcert.pem.

Use an editor to delete RSA-private-key (text from BEGIN RSA PRIVATE KEY to END RSA PRIVATE KEY).

Proceed as follows to configure Signaling and Payload Settings (SPE) and to import the certificates:

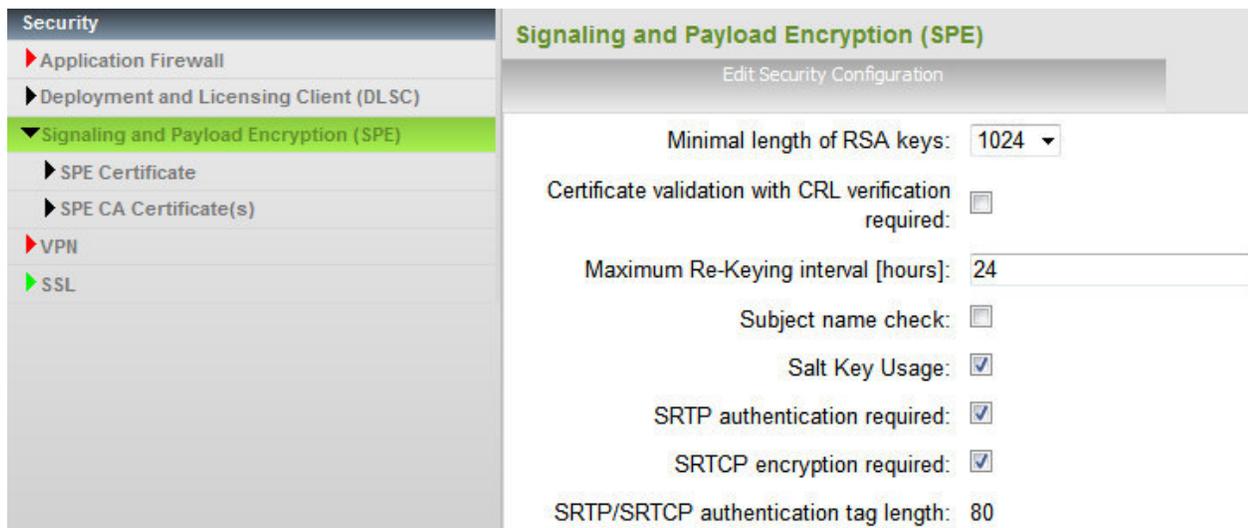
1. Go to **Expert Mode -> Telephony Server -> Security** and select **Signaling and Payload Encryption (SPE)**. The tab "Signaling and Payload Encryption (SPE)" is displayed.

2. In the "Edit Security Configuration" area, make the following settings:

- Minimal length of RSA Keys: Select **1024**.
- Certificate validation with CRL verification required: **Deactivate** the checkbox.
- Maximum Re-Keying interval [hours]: **24** hours (default value).
- Subject name check: **Deactivate** the checkbox.

- Salt Key Usage: **Activate** the checkbox.
- SRTP authentication required: **Activate** the checkbox.
- SRTCP encryption required: **Activate** the checkbox.
- SRTP/SRTCP authentication tag length: **80** (default value).

3. Click **Apply**. Your changes are saved.



Proceed as follows to import the CA Certificate:

1. Go to **Expert Mode ->Telephony Server -> Security -> Signaling and Payload Encryption (SPE)** and select **SPE CA Certificate(s)**. The tab "Import trusted CA Certificate (X.509 file) for SPE" is displayed.
2. In the configuration area, make the following settings:
 - File with certificate (PEM or binary): Click "Browse" and select the **rootcert.pem** file.
3. Press button "View Fingerprint of Certificate". The fingerprint is displayed.
4. Press button "Import Certificate from File". Certificate is imported.



Proceed as follows to import the CA Certificate:

1. Go to **Expert Mode ->Telephony Server -> Security -> Signaling and Payload Encryption (SPE)** and select **SPE Certificate**. The tab "Import SPE certificate plus private key (PKCS#12 file)" is displayed.
2. In the configuration area, make the following settings:
 - Passphrase for decryption: **No passphrase configuration.**
 - File with certificate and private key (PEM or PKCS#12 format): Click "Browse" and select the **client.pem** file.
3. Press button "View Fingerprint of Certificate". The fingerprint is displayed.
4. Press button "Import Certificate from File". Certificate is imported.



1.2.4.4. "Voice Gateway -> Station -> Station Payload Security"

When SPE is activated, payload security option must be configured to "ON" for all supported station types.

1. Go to **Expert Mode ->Telephony Server -> Station -> Station** and select a specific station (e.g UP0 6000). The window "Edit Station Parameter" tab is displayed.
2. In the "Edit Station Parameter" area, make the following setting:
 - Payload Security: Configure **ON**.
3. Click **Apply**. Your changes are saved.

4. Apply the same configuration to all active supported stations.

The screenshot displays a configuration window for a station. On the left, a tree view shows the hierarchy: Station > UP0 Stations > 0 6000 tdm_6000. The main area is titled 'Station' and contains three tabs: 'Edit station parameters' (selected), 'Edit station flags', and 'Edit Group/CFW'. Below the tabs, the configuration for 'Station - 0' is shown:

- Call number: 6000
- Name: tdm_6000
- Direct inward dialing: 6000
- Device Type: OpenStage 40
- Clip/Lin: -
- Access: SLUC8 2-1 Master

Below the station configuration, there is a 'Fax' section with fields for Call number and Direct inward dialing, both set to '-'. The 'Parameter' section contains the following settings:

- Extension Type: Standard
- Language: English
- Call signaling internal: Ring type 1
- Call signaling external: Ring type 1
- Class of service (LCR): 15
- Hotline Mode: Off
- Hotline: None
- Payload Security: On** (highlighted with a green box)
- Operating mode: 7 - SLMO, default template

1.2.4.5. "Voice Gateway -> SIP Interconnection -> OpenScape Voice (TLS)"

In order to configure TLS for the SIP interconnection trunk you have to follow the steps described in sections 1.2.3.3 and 1.2.3.4.

Voice Gateway -> SIP Interconnection -> OpenScape Voice (IP Address)

Parameters for the OpenScape Voice must be set. To do this, proceed as follows:

1. **Expert Mode -> Telephony Server -> Voice Gateway -> SIP Interconnection** and select **OpenScape Voice**. The tab "Edit SIP Interconnection" is displayed.

2. Make the following settings in the "Edit SIP Interconnection" tab:

- Enable Trunk: **Activate** the checkbox.
- Trunk Identifier in System: Select **SIP-Interconnection2**.
- Remote Domain Name: IP Address of **sipsm3**.
- SIP Server
 - IP Address / Host name: IP Address of **sipsm3**.
 - Port: **5061**.
 - Secure Transport: **Activate** the check box.
- SIP Registrar
 - Use Registrar: **Activate** the checkbox.
 - IP Address / Host name: IP Address of **sipsm3**.
 - Port: **5061**.
 - Reregistration Interval (sec): **300**.
- Outbound Proxy (**Check box must be deactivated**)
- Inbound Proxy (**Check box must be deactivated**)

3. Click **Apply**. Your changes are saved.

Voice Gateway		SIP Interconnection		
<ul style="list-style-type: none"> SIP Parameters Codec Parameters ▶ Destination Codec Parameters ▶ Internet Telephony Service Provider ▶ Networking ▼ SIP Interconnection <ul style="list-style-type: none"> ▶ Application Suite ▶ HiPath 4000 ▶ Native SIP Server trunk ▼ OpenScapeVoice <ul style="list-style-type: none"> OpenScapeVoice-User ▶ SIPQ Server trunk 		<div style="display: flex; justify-content: space-between; border-bottom: 1px solid #ccc; padding-bottom: 5px;"> Edit SIP Interconnection Delete SIP Interconnection Add SIP Interconnection User </div>		
		Name: OpenScapeVoice Enable Trunk: <input checked="" type="checkbox"/> Trunk Identifier in System: SIP-Interconnection2 ▼ Remote Domain Name: 192.168.140.232		
		SIP Server IP Address / Host name: 192.168.140.232 Port: 5061 Secure Transport: <input checked="" type="checkbox"/>		
		SIP Registrar Use Registrar: <input checked="" type="checkbox"/> IP Address / Host name: 192.168.140.232 Port: 5061 Reregistration Interval (sec): 300		
		Outbound Proxy Use Outbound Proxy: <input type="checkbox"/> IP Address / Host name: <input type="text"/> Port: <input type="text"/>		
		Inbound Proxy Use Inbound Proxy: <input type="checkbox"/> IP Address / Host name: <input type="text"/> Port: <input type="text"/>		

Voice Gateway -> SIP Interconnection -> OpenScape Voice (DNS-SRV)

For enabling TLS in DNS-SRV interconnection, steps in 1.2.3.4 must be followed. The only difference is that “secure transport” checkbox must be activated and FQDN of sipism3 must be used. No other configuration is necessary from OpenScape Business side.

Openscape Business generates the followed type of DNS SRV query for TLS: `_sips._tcp.domain_name`. In case of OpenScape Branch configuration, sipism3 and openscape branch must share the same Fully Qualified Domain Name (i.e. sip.osvsec.net).

1.2.5. Least Cost Routing

1.2.5.1. Basic settings

Proceed as follows:

1. **Expert Mode** -> **Telephony Server** -> **LCR** and select **LCR Flags**. The "Edit LCR Flags" tab is displayed.
2. Make the following settings:
 - Activate LCR: **Activate** the checkbox.
3. Click **Apply**. Your changes are saved.



4. **Expert Mode** -> **Telephony Server** -> **LCR** and select **Classes of Service**.

6. Make the following settings:

- Class of Service: Configure Class of Service 14 to all endpoints.

The screenshot shows the configuration interface for Classes Of Service. The left navigation menu has "Classes Of Service" selected. The main area is titled "Classes Of Service" and contains a sub-tab "Edit Classes Of Service". Below this, a table titled "Classes of service" is displayed with the following data:

Index	Call number	Name	Class of service
1	6000	tdm_6000	14 ▼
2	6001	tdm_6001	14 ▼
3	6002	tdm_6002	14 ▼

1.2.5.2. Least cost routing to CO (Carrier)

Overview

The first route should lead to the CO. As an option, a second route can be configured for least cost routing to OpenScape Voice. This route is only used to bridge trunk failures. During a trunk failure, calls are routed to OpenScape Voice (with a transition to another OpenScape Voice gateway).

Local calls (Dialed digits 0CZ), national calls (Dialed digits 0C0-Z), and International calls (Dialed digits 0C00-Z), must be handled in 3 LCR rules.

Dial Plan			
Change Dial Plan			
Dial Plan	Name	Dialed digits	Routing Table
51	OSV Local	0CZ	20 →
52	OSV National	0C0-Z	30 →
53	OSVInternational	0C00-Z	40 →

Local calls

Dialed digits: CO (name), 0CZ (Dialed digits), 20 (Routing table)

Routing table: CO (Route), CO (Dial rule), 14 (min. COS), -, None (Warning)

SIP INT 2 (Route), E.164 Local (Dial rule), 14 (min. COS), -, Display + tone

Routing Table:20

en-bloc sending

Route	Dial Rule	min. COS	Warning
CO	CO	14	None
SIP INT 2	E.164 Local	14	Display + Tone

National calls:

Dialed digits: CO (name), 0C0-Z (Dialed digits), 30 (Routing table)

Routing table: CO (Route), CO (Dial rule), 14 (min. COS), -, None (Warning)

SIP INT 2 (Route), E.164 National. (Dial rule), 14 (min. COS), -, Display + tone

Routing Table:30

en-bloc sending

Route	Dial Rule	min. COS	Warning
CO	CO	15	None
SIP INT 2	E.164 National	15	Display + Tone

International calls:

Dialed digits: CO (name), 0C00-Z (Dialed digits), 40 (Routing table)

Routing table: CO (Route), CO (Dial rule), 14 (min. COS), -, None (Warning)

SIP INT 2 (Route), E.164 Internatio. (Dial rule), 14 (min. COS), -, Display + tone

Routing Table: 40 en-bloc sending

Route	Dial Rule	min. COS	Warning
CO ▾	CO ▾	14 ▾	None ▾
SIP INT 2 ▾	E.164 Internatio ▾	14 ▾	Display + Tone ▾

Dial Rule				
Change Dial Rule				
	Rule Name	Dial rule format	Network access	Type
26	E.164 Local	D4969E2A	Corporate Network ▾	Country code ▾
27	E.164 National	D49E3A	Corporate Network ▾	Country code ▾
28	E.164 Internatio	E3A	Corporate Network ▾	Country code ▾
29	CO	A	Main network supplier ▾	Unknown ▾

Warning!
All outdial rules for OpenScape Voice must be configurable in the international E.164 number format.

Step by step configuration

For local calls

Proceed as follows:

1. **Expert Mode -> Telephony Server -> LCR -> select Dial plan.** The tab "Change Dial plan" is displayed.

2. In a free row configure the dial plan for CO access, named OSV Local:

- Name: **OSV Local**
- Dialed digits: **0CZ**
- Routing table: **20**(will be configured in step 4).

3. **Expert Mode -> Telephony Server -> LCR -> select Dial Rule.** In the "Change Dial Rule" tab configure two new dial rules for CO called **CO** and **E.164 Local**. **CO** is the main CO rule and **E.164 Local** rule is used for rerouting via OSV in case of CO failure.

- Rule name: **CO**
- Dial rule format: **A**
- Network access: **Main network supplier**
- Type of Number (TON): **Unknown**
- Click Apply. The Dial rule wizard is saved.

- Rule name: **E.164 Local**
- Dial rule format: **D4969E2A**
- Network access: **Corporate Network**
- Type of Number (TON): **Country Code**
- Click Apply. The Dial rule wizard is saved.

4. **Expert Mode -> Telephony Server -> LCR -> select Routing Table.** Go to **Routing Table 20:**

First Route (CO route):

- Select the route **CO** in the column "Route".
- In the "Dial rule" column, select **CO** as dial rule.
- min. COS: **14**
- Warning: **none**
- Dedicated Gateway: **No**
- Click Apply. The Routing Table configuration is saved.

Second Route (Rerouting via OSV):

- Select the route **SIP INT 2** in the column "Route".
- In the "Dial rule" column, select **E.164 Local** as dial rule.
- min. COS: **14**
- Warning: **Display + Tone**
- Dedicated Gateway: **No**
- Click Apply. The Routing Table configuration is saved.

5. Click **Apply**. Your changes are saved.

Dial Plan					
Change Dial Plan			Display Dial Plan		
Dial Plan	Name	Dialed digits	Routing Table	Acc. code	Classes of service
51	OSV Local	0CZ	20	<input type="checkbox"/>	<input checked="" type="checkbox"/>

Dial Rule					
Change Dial Rule					
	Rule Name	Dial rule format	Network access	Type	
26	E.164 Local	D4969E2A	Corporate Network	Country code	
29	CO	A	Main network supplier	Unknown	

Change Routing Table					
Routing Table: 20			en-bloc sending		
Index	Route	Dial Rule	min. COS	Warning	Dedicated Gateway
1	CO	CO	14	None	No
2	SIP INT 2	E.164 Local	14	Display + Tone	No

For National calls

Proceed as follows:

1. **Expert Mode -> Telephony Server -> LCR -> select Dial plan.** The tab "Change Dial plan" is displayed.

2. In a free row configure the dial plan for CO access, named OSV National:

- Name: **OSV National**
- Dialed digits: **0C0-Z**
- Routing table: **30**(will be configured in step 4).

3. **Expert Mode -> Telephony Server -> LCR -> select Dial Rule.** In the "Change Dial Rule" tab configure two new dial rules for CO called **CO** and **E.164 National**. **CO** is the main CO rule and **E.164 National** rule is used for rerouting via OSV in case of CO failure.

- Rule name: **CO**
- Dial rule format: **A**
- Network access: **Main network supplier**
- Type of Number (TON): **Unknown**
- Click Apply. The Dial rule wizard is saved.

- Rule name: **E.164 National**
- Dial rule format: **D49E3A**
- Network access: **Corporate Network**
- Type of Number (TON): **Country Code**
- Click Apply. The Dial rule wizard is saved.

4. **Expert Mode -> Telephony Server -> LCR -> select Routing Table.** Go to **Routing Table 30**:

First Route (CO route):

- Select the route **CO** in the column "Route".
- In the "Dial rule" column, select **CO** as dial rule.
- min. COS: **14**
- Warning: **none**
- Dedicated Gateway: **No**
- Click Apply. The Routing Table configuration is saved.

Second Route (Rerouting via OSV):

- Select the route **SIP INT 2** in the column "Route".

- In the "Dial rule" column, select **E.164 National** as dial rule.
- min. COS: **14**
- Warning: **Display + Tone**
- Dedicated Gateway: **No**
- Click **Apply**. The Routing Table configuration is saved.

5. Click **Apply**. Your changes are saved.

Dial Plan

Change Dial Plan
Display Dial Plan

Dial Plan	Name	Dialed digits	Routing Table	Acc. code	Classes of service
51	OSV Local	0CZ	20 →	<input type="checkbox"/>	<input checked="" type="checkbox"/>
52	OSV National	0C0-Z	30 →	<input type="checkbox"/>	<input checked="" type="checkbox"/>
53	OSVInternational	0C00-Z	40 →	<input type="checkbox"/>	<input checked="" type="checkbox"/>

Change Dial Rule

	Rule Name	Dial rule format	Network access	Type
26	E.164 Local	D4969E2A	Corporate Network	Country code
27	E.164 National	D49E3A	Corporate Network	Country code
28	E.164 Internatio	E3A	Corporate Network	Country code
29	CO	A	Main network supplier	Unknown

Change Routing Table

Routing Table: 30 en-bloc sending

Index	Route	Dial Rule	min. COS	Warning	Dedicated Gateway
1	CO	CO	14	None	No
2	SIP INT 2	E.164 National	14	Display + Tone	No

For international calls

Proceed as follows:

1. **Expert Mode -> Telephony Server -> LCR -> select Dial plan.** The tab "Change Dial plan" is displayed.

2. In a free row configure the dial plan for CO access, named OSV National:

- Name: **OSV International**
- Dialed digits: **0C00-Z**
- Routing table: 40(will be configured in step 4).

3. **Expert Mode -> Telephony Server -> LCR -> select Dial Rule.** In the "Change Dial Rule" tab configure two new dial rules for CO called **CO** and **E.164 International**. CO is the main CO rule and E.164 International rule is used for rerouting via OSV in case of CO failure.

- Rule name: **CO**
- Dial rule format: **A**
- Network access: **Main network supplier**
- Type of Number (TON): **Unknown**
- Click Apply. The Dial rule wizard is saved.

- Rule name: **E.164 International**
- Dial rule format: **E3A**
- Network access: **Corporate Network**
- Type of Number (TON): **Country Code**
- Click Apply. The Dial rule wizard is saved.

4. **Expert Mode -> Telephony Server -> LCR -> select Routing Table.** Go to **Routing Table 30:**

First Route (CO route):

- Select the route **CO** in the column "Route".
- In the "Dial rule" column, select **CO** as dial rule.
- min. COS: **14**
- Warning: **none**
- Dedicated Gateway: **No**
- Click Apply. The Routing Table configuration is saved.

Second Route (Rerouting via OSV):

- Select the route **SIP INT 2** in the column "Route".

- In the "Dial rule" column, select **E.164 International** as dial rule.
- min. COS: **14**
- Warning: **Display + Tone**
- Dedicated Gateway: **No**
- Click **Apply**. The Routing Table configuration is saved.

5. Click **Apply**. Your changes are saved.

Dial Plan						
Change Dial Plan				Display Dial Plan		
Dial Plan	Name	Dialed digits	Routing Table	Acc. code	Classes of service	
51	OSV Local	0CZ	20 →	<input type="checkbox"/>	<input checked="" type="checkbox"/>	
52	OSV National	0C0-Z	30 →	<input type="checkbox"/>	<input checked="" type="checkbox"/>	
53	OSVInternational	0C00-Z	40 →	<input type="checkbox"/>	<input checked="" type="checkbox"/>	

Change Dial Rule					
Rule Name	Dial rule format	Network access		Type	
26 E.164 Local	D4969E2A	Corporate Network		Country code	
27 E.164 National	D49E3A	Corporate Network		Country code	
28 E.164 Internatio	E3A	Corporate Network		Country code	
29 CO	A	Main network supplier		Unknown	

Change Routing Table					
Routing Table: 40					
en-bloc sending					
Index	Route	Dial Rule	min. COS	Warning	Dedicated Gateway
1	CO	CO	14	None	No
2	SIP INT 2	E.164 Internatio	14	None	No

1.2.5.3. Least cost routing to OpenScape Voice

Overview

Change Dial Plan			Display Dial Plan		
Dial Plan	Name	Dialed digits	Routing Table	Acc. code	Classes of service
101	OSV_internal	-8XXXX	21		<input checked="" type="checkbox"/>

Change Dial Rule				
Rule Name	Dial rule format	Network access	Type	
51 E.164_internal	D49697007A	Corporate Network	Country code	
52 Rerouting_via_CO	D00049697008E2A	Main network supplier	Unknown	

Routing Table:21						en-bloc sending
Index	Route	Dial Rule	min. COS	Warning	Dedicated Gateway	
1	SIP INT 2	E.164_internal	14	None	No	
2	CO	Rerouting_via_CC	15	None	No	

Dialing a short number in the same location in OpenScape Voice

Dialed digits: **OSV Internal** (name), **-8xxxx** (Dialed digits), **21** (Routing table)
 Routing table: **SIP INT 2** (Route), **E.164 internal** (Dial rule), **14** (min. COS), **-**, **None** (Warning)
CO (Route), **Rerouting_via_CO** (Dial rule), **14** (min. COS), **-**, **Display + tone**

Warning!

All outdial rules for OpenScape Voice must be configurable in the international E.164 numbering format.

Step by step configuration

Dialing a short number in the same location in OpenScape Voice

The first route should lead to OpenScape Voice. As an option, a second route can be configured for least cost routing via the CO. The second route bridges the gap in the case of LAN failure. During LAN failure, calls are routed via the local CO to another OpenScape Voice gateway.

Proceed as follows:

Proceed as follows:

1. **Expert Mode -> Telephony Server -> LCR -> select Dial plan.** The tab "Change Dial plan" is displayed.
2. In a free row configure the dial plan for CO access, named OSV National:
 - Name: **OSV_internal**
 - Dialed digits: **-8xxxx**
 - Routing table: **21**(will be configured in step 4).
3. **Expert Mode -> Telephony Server -> LCR -> select Dial Rule.** In the "Change Dial Rule" tab configure two new dial rules for CO called **Rerouting_via_CO** and **E.164_internal**. CO is the main CO rule and E.164_internal rule is used for outbound calls to OSV internal endpoints.
 - Rule name: **Rerouting_via_CO**
 - Dial rule format: **D00049697008E2A**
 - Network access: **Main network supplier**
 - Type of Number (TON): **Unknown**
 - Click Apply. The Dial rule wizard is saved.

 - Rule name: **E.164_internal**
 - Dial rule format: **D49697007A**
 - Network access: **Corporate Network**
 - Type of Number (TON): **Country Code**
 - Click Apply. The Dial rule wizard is saved.
4. **Expert Mode -> Telephony Server -> LCR -> select Routing Table.** Go to **Routing Table 21:**
First Route (Internal):
 - Select the route **SIP INT 2** in the column "Route".
 - In the "Dial rule" column, select **E.164_internal** as dial rule.
 - min. COS: **14**
 - Warning: **none**

- Dedicated Gateway: **No**
- Click **Apply**. The Routing Table configuration is saved.

Second Route (Rerouting via CO):

- Select the route **CO** in the column "Route".
- In the "Dial rule" column, select **E.164_internal** as dial rule.
- min. COS: **14**
- Warning: **Display + Tone**
- Dedicated Gateway: **No**
- Click **Apply**. The Routing Table configuration is saved.

5. Click **Apply**. Your changes are saved.

1.3. Configuration of OpenScape Voice

Overview

OpenScape Business as a gateway (endpoint) and the OpenScape Voice subscribers shall be located in different Private Numbering Plans but in the same business group.

That means every location gets an own Private Numbering Plan for subscriber and an own Private Numbering Plan for gateways/endpoints (OpenScape Business).

Contents

This section covers the following topics:

Section 1.3.1, "Settings in the Common Management Portal"

Section 1.3.2, "Settings in StartCli"

1.3.1. Settings in Common Management Portal

Procedure

1. Start the Common Management Portal.
2. Follow the remaining steps in sequence.

Contents

This section covers the following topics:

Section 1.3.1.1, "Administration of a new Gateway Private Numbering Plan"

Section 1.3.1.2, "Creating and Configuring an Endpoint Profile for an OpenScape Business Endpoint"

Section 1.3.1.3, "Creating and configuring an endpoint for OpenScape Business"

Section 1.3.1.4, "Creating "Digest Authentication" access for OpenScape Business"

Section 1.3.1.5, "Configuring a Gateway Numbering Plan for Incoming Calls"

Section 1.3.1.6, "Configuring Outgoing Calls"

Section 1.3.1.7, "Display Number Modification for OpenScape Voice V7.1"

1.3.1.1. Administration of a new Gateway Private Numbering Plan

Proceed as follows:

1. **OpenScape Voice** -> select **Business Group**.
2. On the left side of the window, select the following:
 - Available Switches: Select the OpenScape Voice
 - Available Business Groups: Select the business group for which the endpoint profile is to be created, e.g. **bg_sol**.
3. On the left side of the window, select **Private Numbering Plans**. On the right side of the window, a list of all Private Numbering Plans is displayed.
4. To create a new Private Numbering Plan, click the **Add** button. The configuration window is displayed.
5. Name: Enter a name for the Private Numbering Plan, e.g. **NP_br13_gw**.
6. Click **Save**.

1.3.1.2. Administration of an Endpoint Profile for an OpenScape Business Endpoint

Proceed as follows:

1. **OpenScape Voice** -> select **Business Group**.
2. On the left side of the window, select the following:
 - Available Switches: Select the OpenScape Voice
 - Available Business Groups: Select the business group for which the endpoint profile is to be created, e.g. **bg_sol**.
3. On the left side of the window, select **Profiles** -> **Endpoint Profiles**. On the right side of the window, a list of endpoint profiles is displayed.
4. To create a new endpoint profile, click the **Add** button. The configuration window for this endpoint profile is displayed.
5. Make the following settings in the **General** tab
 - Endpoint Profiles
 - Name: Enter a name for the endpoint profile, e.g. **EP_hg1500.br13**.
 - Numbering Plan: Select the numbering plan which was create before, e.g. **NP_br13_gw**.
 - Management Information
 - Class of Service: No setting necessary.
 - Routing Area: No setting necessary.
 - Calling Location: No setting necessary.

- SIP Privacy Support: **Basic**
- Failed Calls Intercept: **Disabled**
- Language: e.g. **German**

6. Open the **Services** tab. Make the following settings in this tab:

- Voice mail: **Yes**
- Call Transfer: **No**
- Call Forward Invalid Destination: **Yes** and enter station number.
- Toll and Call Restrictions: **No**

8. Click **Save**. Your changes are saved.

The screenshot shows the OpenScope Voice configuration interface. The top navigation bar includes 'Home', 'Operation & Maintenance', 'OpenScope Branch', 'RG8700', 'OpenScope Voice', and 'Users & Resources'. The main navigation tabs are 'General', 'Administration', 'Business Group', 'Global Translation and Routing', and 'Maintenance'. The 'Business Group' tab is selected, showing the configuration for 'klara'.

The left sidebar contains various configuration options, including 'Available Switches' (klara), 'Quick Tasks', 'Business Group Lists', 'Business Groups', 'Available Business Groups' (bg_sol), 'BG Options', 'Profiles', 'Feature Profiles', 'Endpoint Profiles', 'Teams', 'Statistics', 'Branch Office', 'Available Branch Offices' (Main Office), and 'Members'.

The main content area displays the 'Endpoint Profiles' for the selected business group. The title is '[klara] - Endpoint Profiles - bg_sol'. Below the title is a 'List of Endpoint Profiles' section with a filter dropdown set to 'No Filter' and an 'Apply Filter' button. The table below lists the profiles:

<input type="checkbox"/>	Name	Class of Service	Routing Area	Calling Location	Remark	Numbering Plan Name
<input type="checkbox"/>	EP_XPR	netz			No	NP_XPR
<input type="checkbox"/>	EP_comdasys.br11				No	NP_br11_gw
<input type="checkbox"/>	EP_comdasys.br12				No	NP_br12_gw
<input type="checkbox"/>	EP_comdasys.br13				No	NP_br13_gw
<input type="checkbox"/>	EP_dummy.common				No	NP_common
<input type="checkbox"/>	EP_h8k_anton	international			No	NP_h8k_anton
<input type="checkbox"/>	EP_h8k_lisa	international			No	NP_h8k_lisa
<input type="checkbox"/>	EP_h8k_mona	international			No	NP_h8k_mona
<input type="checkbox"/>	EP_h8k_nora	international			No	NP_h8k_nora
<input type="checkbox"/>	EP_hg1500.br12	international			No	NP_br12_gw
<input type="checkbox"/>	EP_hg1500.br13	international			No	NP_br13_gw

OpenScape Voice V7R1

Enter the profile data. Maximum number of allowed blocked number is 10.

Endpoint Profile

Please enter a unique name to identify this profile.

Name: EP_hg1500.br13

Remark:

Numbering Plan: NP_br13_gw

Management Information

Please enter the data for the following fields in the corresponding screens.

Class of Service: international

Routing Area:

Calling Location:

SIP Privacy Support: Basic

Failed Calls Intercept: Disabled

Treatment:

Language: German

OK Cancel

Enter the profile data. Maximum number of allowed blocked number is 10.

General

Name delivery: Yes

Voice mail: Yes

Called name delivery: Yes

Called number delivery: Yes

Call Transfer: No

Call Forward Invalid Destination: Yes 49695113202

Toll and Call Restrictions

International World Zone 1: No

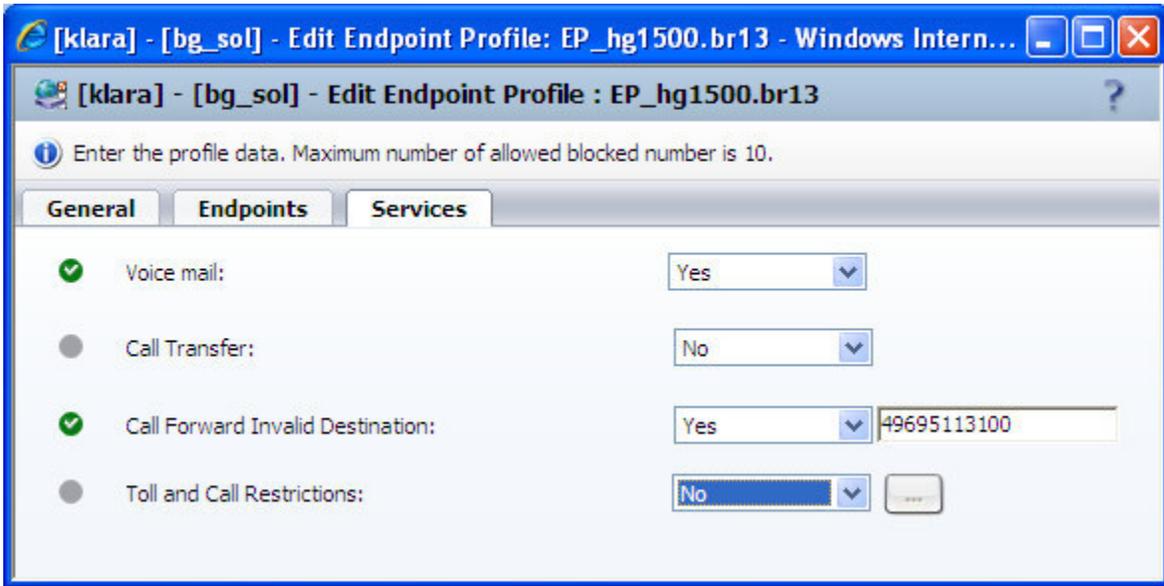
International: No

National: No

Local: No

OK Cancel

OpenScape Voice V7R1



1.3.1.3. Creating and configuring an endpoint for OpenScape Business

Proceed as follows:

1. **OpenScape Voice** -> select **Business Group**.

2. On the left side of the window, select the following:

- Available Switches: Select the OpenScape Voice
- Available Business Groups: Select the business group for which the endpoint profile is to be created, e.g. **bg_sol**.
- Available Branch Offices: Select the branch office for which the endpoint is supposed to be created, e.g. **BR13**.

3. On the left side of the window, select **Members** -> **Endpoints**.

4. To create a new endpoint, click the **Add** button. The configuration window for this endpoint is displayed.

5. Make the following settings in the **General** tab:

- Name: Enter a name for the endpoint, e.g. **hg1500.br13**.
- Profile: Select the endpoint profile selected in the previous step, e.g. **EP_hg1500.br13**.

6. Open the **SIP** tab. Make the following settings in this tab:

SIP Configuration

1. SIP-Q Signaling: Activate the checkbox.
2. for: Select **HiPath 4000/3000**.
3. Transport protocol:
 - Select TCP for traditional network.
 - Select MTLS for secure network.

7. Open the **Attributes** tab. Make the following setting in this tab:

- Activate the "Rerouting Forwarded Calls" checkbox.

8. Open the **Aliases** tab. Make the following settings in this tab:

1. Click the **Add** button and enter the OpenScape Business registration number, e.g. **13310**.
2. Click the **Add** button and enter the OpenScape Business IP address, e.g. **10.22.113.191**.

9. Open the **Aliases** tab. Make the following setting in this tab:

- Accounting Type: PSTN Gateway

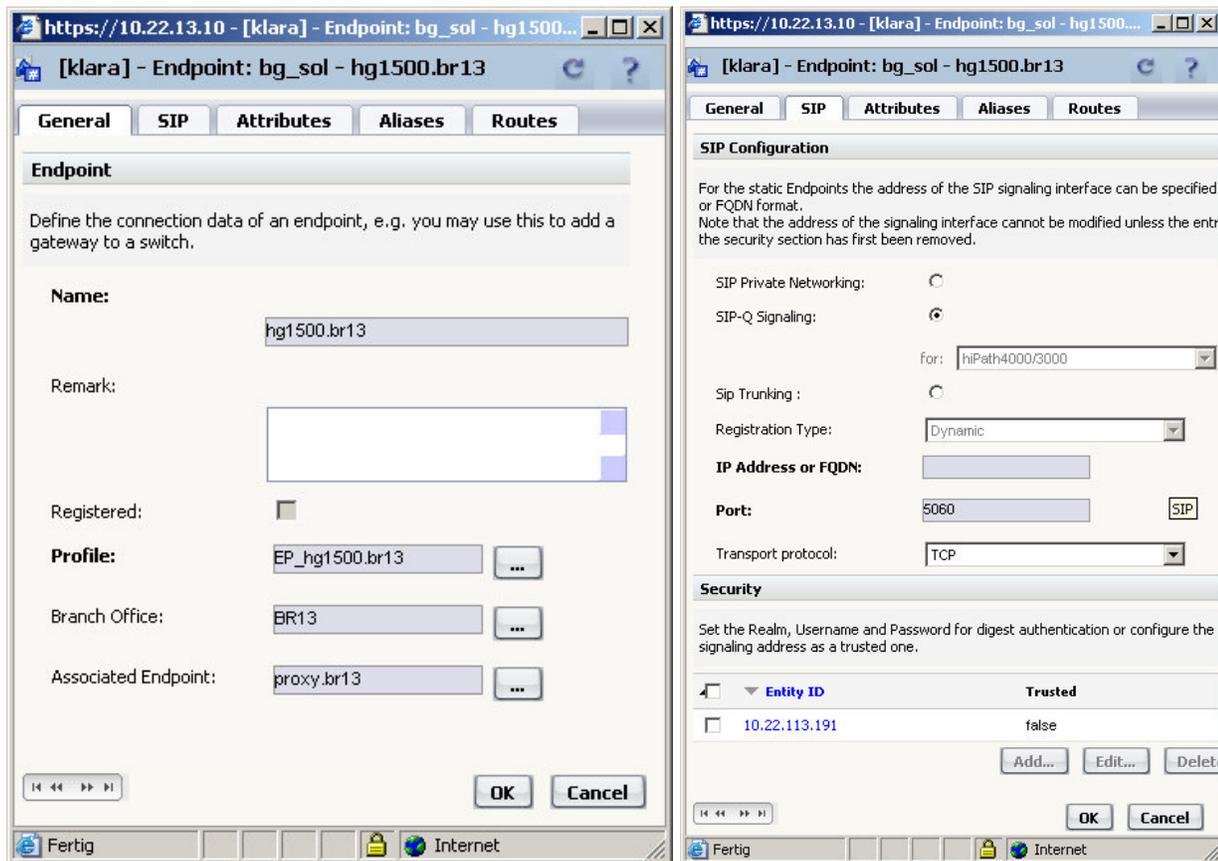
10. Click **Save**. Your changes are saved.

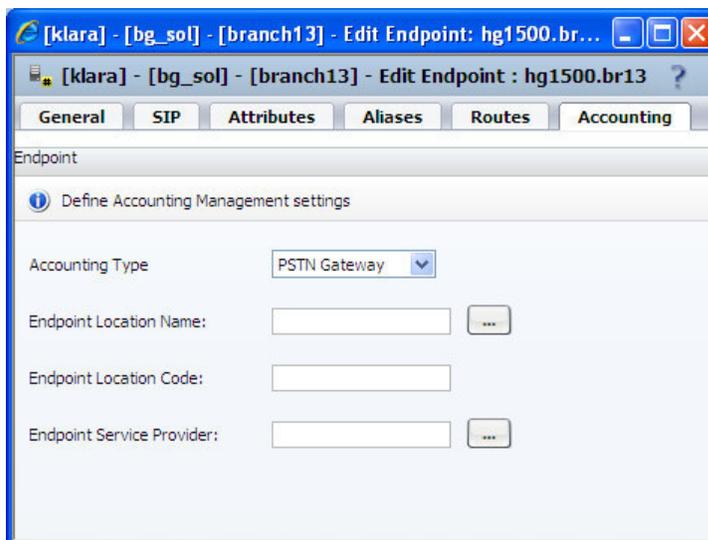
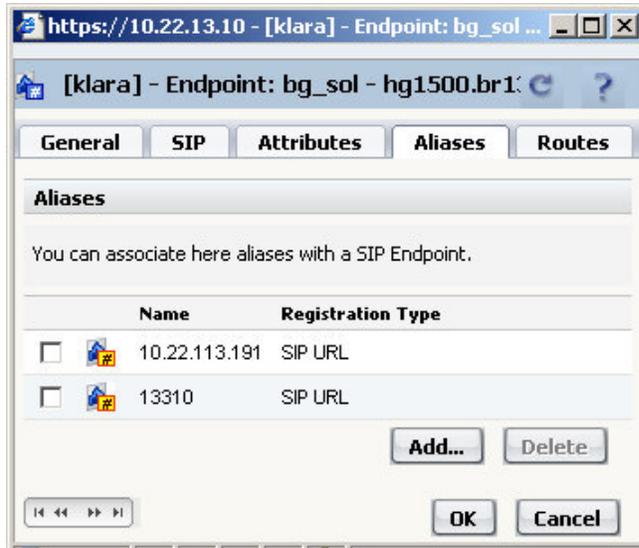
The endpoint is created and configured.

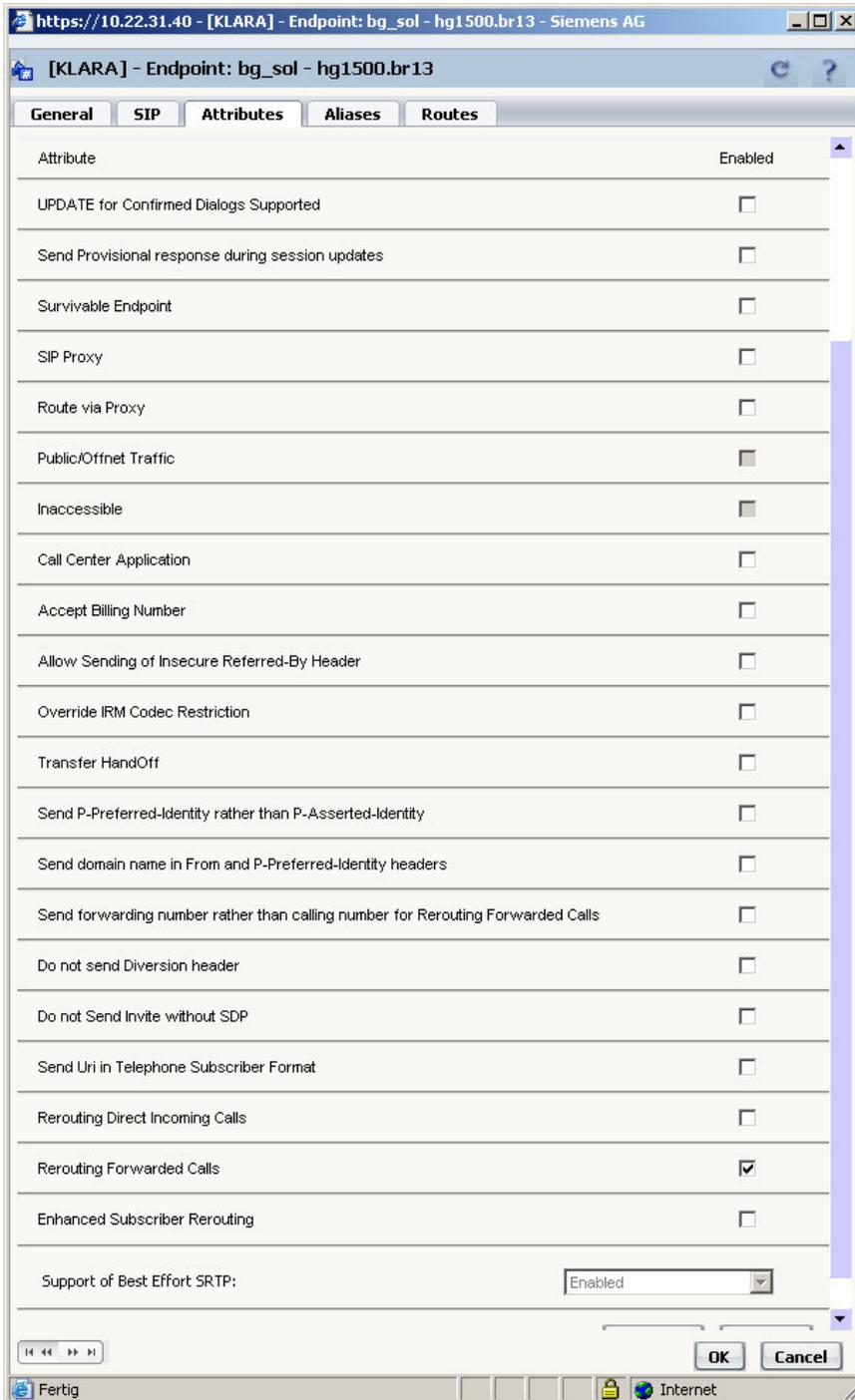
Note:

OpenScope Business supports OSV SIP interconnection with;
direct registration or TCP persistence retransmission.

- if set @ OpenScope Business - SIP Interconnection "direct registration" then @ OpenScopeVoice the endpoint should be set as Dynamic
- if set @ OpenScope Business - SIP Interconnection "TCP persistence retransmission" then @ OpenScopeVoice the endpoint should be set as Static



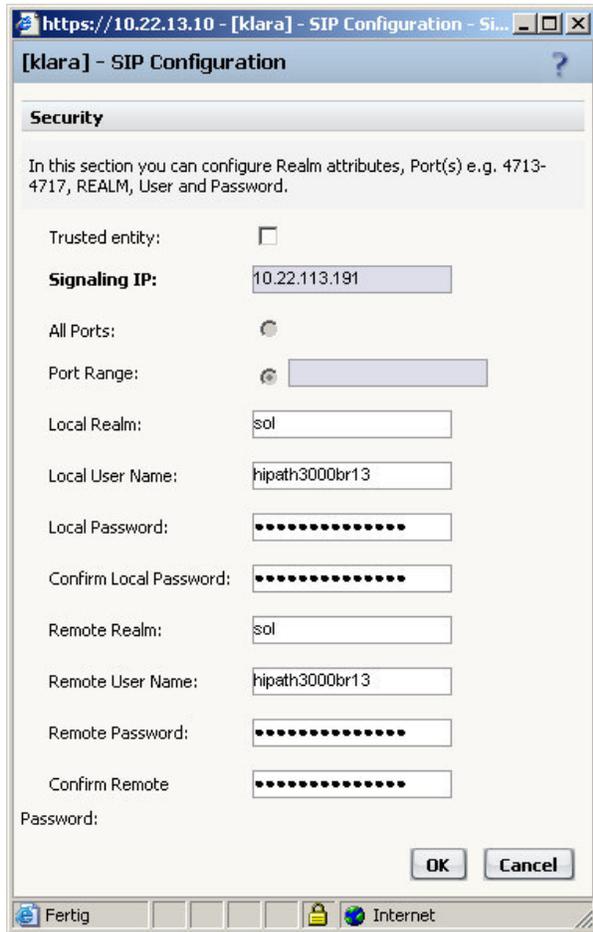




1.3.1.4. Creating "Digest Authentication" access for OpenScape Business

Proceed as follows:

1. **OpenScape Voice** -> select **Administration**.
2. On the left side of the window, select **Signaling Management** -> **Digest Authentication**.
3. Open the **Realms** tab:
4. To create "Digest Authentication" access, click **Add**.
5. Make the following settings:
 - Trusted entity: Deactivate the checkbox.
 - Signaling IP: Enter the OpenScape Business IP address, e.g. **10.22.113.191**.
 - All Ports: Optional field not activated.
 - Port Range: Optional field not activated.
 - Local Realm: Enter the realm, e.g. **sol**.
 - Local User Name: Enter the user name, e.g. **hipath3000br13**.
 - Local Password: Enter the password.
 - Confirm Local Password: Enter the password once more.
 - Remote Realm: Enter the realm, e.g. **sol**.
 - Remote User Name: Enter the user name, e.g. **hipath3000br13**.
 - Remote Password: Enter the password.
 - Confirm Remote Password: Enter the password once more.
6. Click **OK** and the Sip Configuration window is closed.
7. Click **Save**. Your changes are saved and the Digest Authentication window is closed.



1.3.1.5. Configuring a Gateway Numbering Plan for Incoming Calls

This is only a short description how to route an incoming call. This is only a simplified example. A detailed description is available in the routing concept described in the reference architecture.

Proceed as follows:

1. **OpenScape Voice** -> select **Business Group**.
2. On the left side of the window, select the following:
 - Available Switches: Select the OpenScape Voice
 - Available Business Groups: Select the business group for which the endpoint profile is to be created, e.g. **bg_sol**.
 - Available Private Numbering Plan: Select the private numbering plan in which the endpoint was created.
3. On the left side of the window, select **Translation** -> **Destination Codes**.
4. To create new destination code click **Add**.
5. Make the following settings:
 - Destination Code: e.g. **49695113**
 - Nature of Address: **International**
 - Destination Type: **Home**
 - Office Code: e.g. **+49 (69) 5113**
6. Click Save. Your changes are saved.

[klara] - [bg_sol] - [NP_br13_gw] - Add Destination Code ?

General Extensions

Destination Code: ...

Remark:

Country Code:

Nature Of Address: ▾

Traffic Type: ...

Originator Attributes

i Optionally, an additional match is required if the originator of the call belongs to the specified Class of Service and Routing Area.

Class Of Service: ...

Routing Area: ...

NPA:

Destination

i Specify additional parameters to determine how the call will be routed.

Destination Type: ▾

Office Code: ...

Save **Cancel**

1.3.1.6. Configuring Outgoing Calls

This is only a short description how to route an outgoing call. This is only a simplified example. A detailed description is available in the routing concept described in the reference architecture.

Subscriber Numbering Plan - Prefix Access Codes

Proceed as follows:

1. **OpenScape Voice** -> select **Business Group**.
2. On the left side of the window, select the following:
 - Available Switches: Select the OpenScape Voice
 - Available Business Groups: Select the business group for which the endpoint profile is to be created, e.g. **bg_sol**.
 - Available Private Numbering Plan: Select the private numbering plan in which the subscribers are in.
3. On the left side of the window, select **Translation** -> **Prefix Access Codes**.
4. To create new Prefix Access Codes click **Add**. The configuration window for this prefix access code is displayed.
5. Make the following settings:
 - Prefix Access Codes: e.g. **0**
 - Minimum Length: **1**
 - Maximum Length: **30**
 - Digits Position: **1**
 - Digits to insert: **4969**
 - Prefix Type: **On-net Access**
 - Nature of Address: **International**
 - Destination Type: **BG Common Destination**
6. Click **Save**. Your changes are saved.

[klara] - [bg_sol] - [NP_br13_eg] - Edit Prefix Access Code: 0 ?

Identification

i If the dialed digits match this code, the specified modification to these dialed digits is executed.

Prefix Access Code:

Remark:

Minimum Length:

Maximum Length:

Digit Position:

Digits to insert:

Settings

i Specify additional parameters to determine how the call will be routed.

Prefix Type: ▼

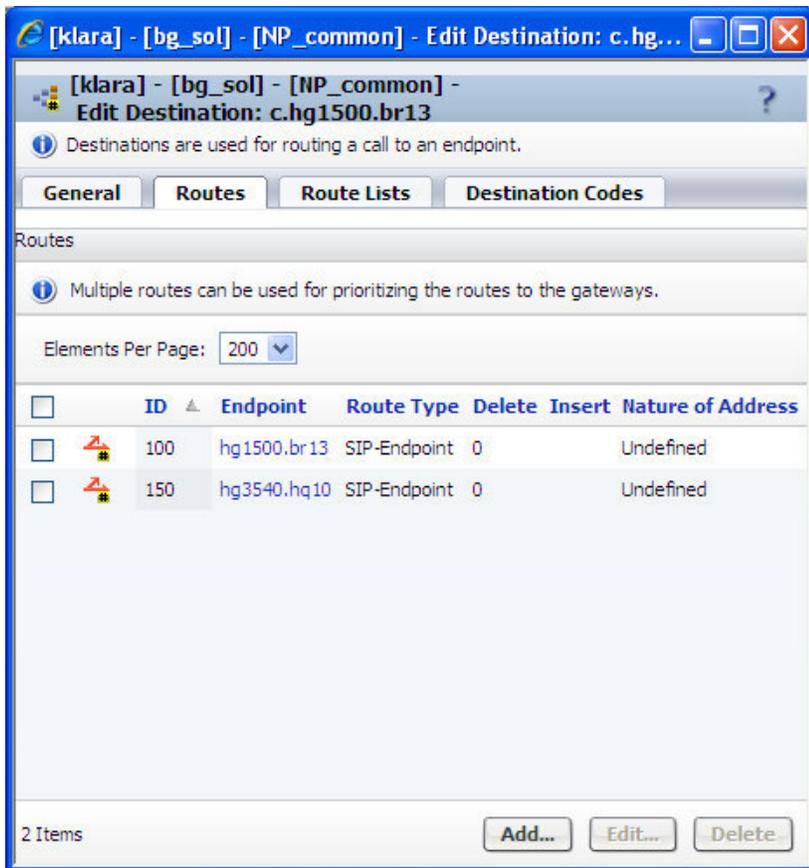
Nature of Address: ▼

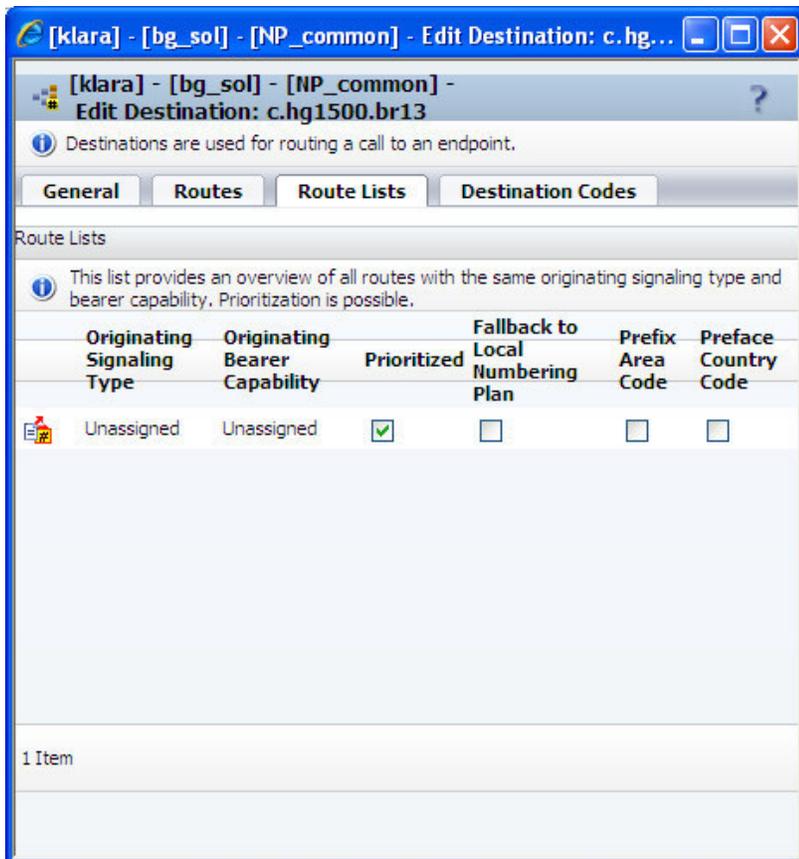
Destination Type: ▼

Destination Name:

Common Numbering Plan - Destination

1. **OpenScape Voice** -> select **Business Group**.
2. On the left side of the window, select the following:
 - Available Switches: Select the OpenScape Voice
 - Available Business Groups: Select the business group for which the endpoint profile is to be created, e.g. **bg_sol**.
 - Available Private Numbering Plan: Select the common numbering plan.
3. On the left side of the window, select **Destinations and Routes** -> **Destinations**.
4. To create a new Destination click **Add**.
5. Make the following setting in the **General** tab:
 - Name: e.g. **c.hg1500.br13**
6. Click Save. Your changes are saved.
7. Open the new Destination to edit.
8. Open the **Routes** tab.
9. To add an endpoint to this destination, click Add.
10. Make the following settings:
 - ID: e.g. **100**
 - Type: **SIP Endpoint**
 - SIP Endpoint: e.g. **hg1500.br13**
 - Nature of Address: **Undefined**
11. Click Save.
12. Add an additional endpoint to have a fallback when OpenScape Business is not in service. Follow the instruction above.
13. Open the **Route Lists** tab.
14. Make the following settings:
 - Prioritized: Activate the checkbox.
 - Fallback to Local Numbering Plan: leave the checkbox unchecked.
15. Click Save. Your changes are saved.





Common Numbering Plan - Prefix Access Codes

1. **OpenScape Voice** -> select **Business Group**.
2. On the left side of the window, select the following:
 - Available Switches: Select the OpenScape Voice
 - Available Business Groups: Select the business group for which the endpoint profile is to be created, e.g. **bg_sol**.
 - Available Private Numbering Plan: Select the common numbering plan.
3. On the left side of the window, select **Translation** -> **Prefix Access Codes**.
4. To create a new Prefix Access Codes click **Add**. The configuration window for this prefix access code is displayed.
5. Make the following settings:
 - Prefix Access Codes: e.g. 4
 - Minimum Length: 1

- Maximum Length: **30**
- Digits Position: **0**
- Digits to insert: leave empty
- Prefix Type: **On-net Access**
- Nature of Address: **International**
- Destination Type: **None**

6. Click Save. Your changes are saved.

[klara] - [bg_sol] - [NP_common] - Edit Prefix Access Code...

[klara] - [bg_sol] - [NP_common] - Edit Prefix Access Code: 4 ?

Identification

i If the dialed digits match this code, the specified modification to these dialed digits is executed.

Prefix Access Code:

Remark:

Minimum Length:

Maximum Length:

Digit Position:

Digits to insert:

Settings

i Specify additional parameters to determine how the call will be routed.

Prefix Type:

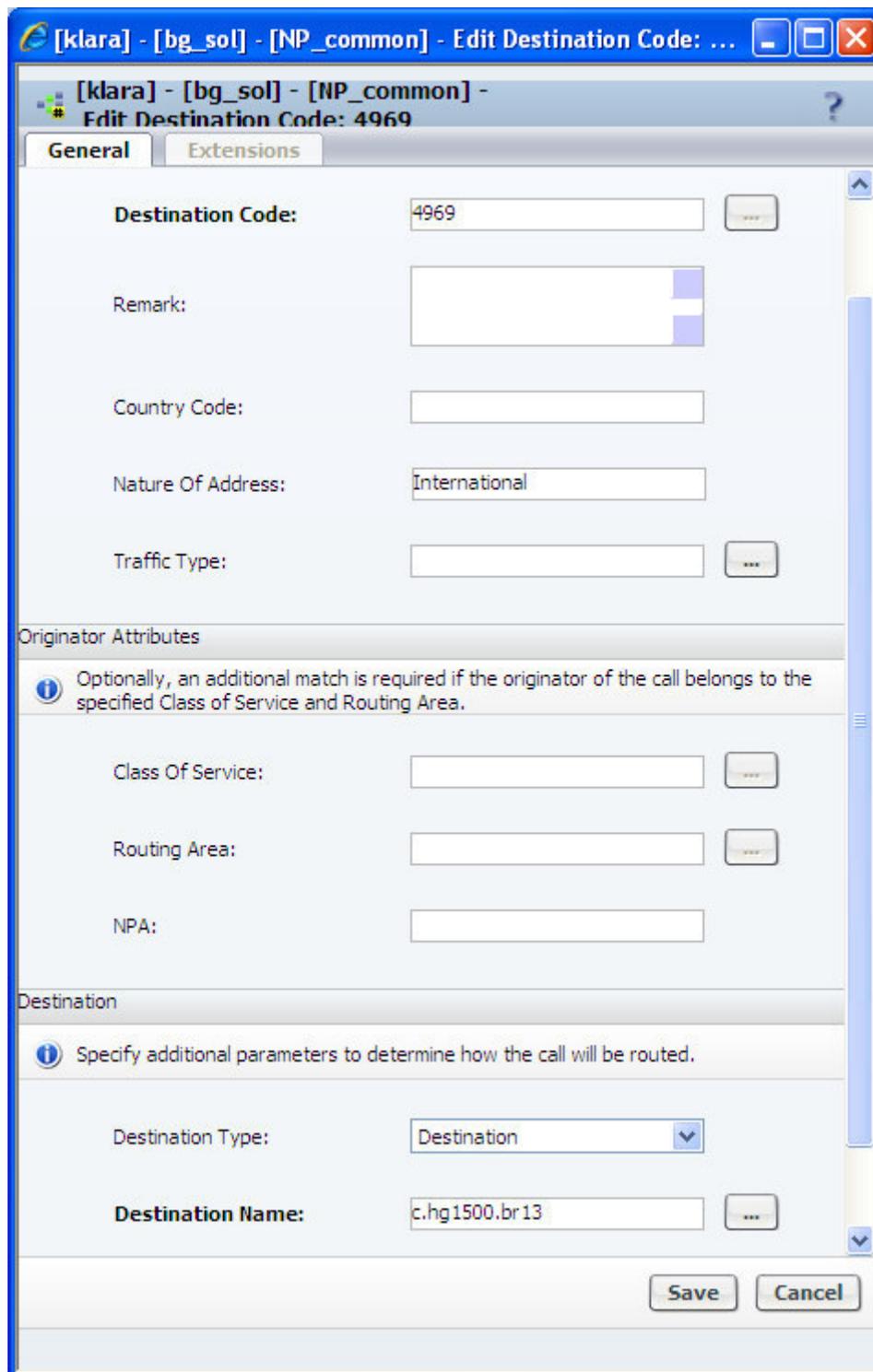
Nature of Address:

Destination Type:

Destination Name:

Common Numbering Plan - Destination Codes

1. **OpenScape Voice** -> select **Business Group**.
2. On the left side of the window, select the following:
 - Available Switches: Select the OpenScape Voice
 - Available Business Groups: Select the business group for which the endpoint profile is to be created, e.g. **bg_sol**.
 - Available Private Numbering Plan: Select the common numbering plan.
3. On the left side of the window, select **Translation** -> **Destination Codes**.
4. To create a new Destination Code click **Add**.
5. Make the following settings:
 - Destination Code: e.g. **4969**
 - Nature Of Address: **International**
 - Destination Type: **Destination**
 - Destination Name: select the destination which was created before e.g. **c.hg1500.br13**
6. Click Save. Your changes are saved.



1.3.1.7. Display Number Modification for OpenScape Voice V7 R1

Definitions

Proceed as follows:

1. **OpenScape Voice** -> select **Business Group**.
2. On the left side of the window, select the following:
 - Available Switches: Select the OpenScape Voice
 - Available Business Groups: Select the business group for which the endpoint profile is to be created, e.g. **bg_sol**.
3. On the left side of the window, select **Display Number Modification** -> **Definitions**.
4. To enter a new Number definition click **Add**.
5. Make the following settings:
 - Business Group: Select the used business group e.g. **bg_sol**
 - Numbering Plan: **ANY**
 - Numbering plan indication: **Public**
 - Country/L2 Code: **49**
 - Area/L1 Code: **69**
 - Local Office/L0 Code: **5113**
 - Number of digits to skip: enter the number length of the Local Office Code e.g. **4**
 - Min. Digits: **8**
 - Max. Digits: **30**
6. Click Save. Your changes are saved.

[klara] - Display Number Definition - Windows Internet Explorer

[klara]- Display Number Definition

Select a business group and/or numbering plan from the list. .

Business Group:

Numbering Plan:

Number Definition

To define a public number, enter country code, area code, local office code and possibly a skip position that defines the numbers of digits to skip in the Local Office Code to create an extension. To define a private number, enter the L2 code, the L1 code, the L0 code and possibly a skip position that defines the number of digits to skip in the L0 code to create an extension. If known, also enter the minimum and maximum number of digits in the fully qualified number definition.

Numbering plan indicator: ▾

Country/L2 Code:

Area/L1 Code:

Local Office/L0 Code:

Number of digits to skip:

Min. Digits:

Max. Digits:

Local Toll

A Local Toll table may define the format of public network numbers as seen by subscribers that match this office code.

Local Toll:

Prefixes

Proceed as follows:

1. **OpenScape Voice** -> select **Business Group**.
2. On the left side of the window, select the following:
 - Available Switches: Select the OpenScape Voice
 - Available Business Groups: Select the business group for which the endpoint profile is to be created, e.g. **bg_sol**.
3. On the left side of the window, select **Display Number Modification** -> **Prefixes**.
4. To enter a global prefix definition click **Add**.
5. Make the following settings:

	Public Network Access Code	Prefix
International	0	00
National	0	0
Subscriber	0	

6. Click Save. Your changes are saved.

[klara] - Edit Display Number Prefix for:ANY - Windows Internet...

[klara]-Edit Display Number Prefix for:ANY ?

Create/Edit display number prefixes and the associated numbering plan, numbering plan indicator, type of number and PNAC

Numbering Plan

Select a numbering plan from the list.

Business Group

Numbering plan

Public Prefix Definition

Change settings for the public numbering plan

	Public Network Access Code	Prefix
International	<input type="text" value="0"/>	<input type="text" value="00"/>
National	<input type="text" value="0"/>	<input type="text" value="0"/>
Subscriber	<input type="text" value="0"/>	<input type="text"/>

Modifications – Gateway Numbering Plan

Proceed as follows:

1. **OpenScape Voice** -> select **Business Group**.
2. On the left side of the window, select the following:
 - Available Switches: Select the OpenScape Voice
 - Available Business Groups: Select the business group for which the endpoint profile is to be created, e.g. **bg_sol**.
3. On the left side of the window, select **Display Number Modification** -> **Modifications**.
4. To enter a new Modification for the Gateway Numbering plan, click **Add**.
5. Make the following settings in Origination Context Setting:
 - Business Group: **ANY**
 - Numbering Plan: **ANY**
6. Make the following settings in Terminating Context Setting:
 - Business Group: Select the used business group e.g. **bg_sol**
 - Numbering Plan: Select the numbering plan of the endpoint e.g. **NP_br13GW**
 - Endpoint: **NONE**
7. Make the following settings in Modification Rule:
 - Input Type Of Number: **ANY**
 - Priority: **1**
 - Output Type Of Number: **International**
 - Number Source: **Input Number**
 - Presentation Restricted: **unselected**
 - Prefix Required: **unselected**
 - Optimize Type Of Number: **None**
8. Click Save. Your changes are saved.

[klara] - Display Number Modification - Windows Intern...

[klara]-Display Number Modification ?

① Create/Edit the 'calling party display number' to a specific format

Originating Context Setting

① Select a business group and/or numbering plan from the list.

Business Group ANY

Numbering Plan ANY

Terminating Context Setting

① Select a business group, numbering plan and/or endpoint from the list.

Business Group bg_sol

Numbering Plan NP_br13_gw

Endpoint NONE

Modification Rule

① Select Input Type of Number, Output Type of number, Number Transmission and define if Optimized possible.

Input Type Of Number: ANY

Priority: 1

Output Type Of Number: International

Number Source: Input Number

Presentation Restricted:

Prefix Required:

Optimize Type Of Number: None

Save Cancel

Modifications – Subscriber Numbering Plan

Proceed as follows:

1. **OpenScape Voice** -> select **Business Group**.
2. On the left side of the window, select the following:
 - Available Switches: Select the OpenScape Voice
 - Available Business Groups: Select the business group for which the endpoint profile is to be created, e.g. **bg_sol**.
3. On the left side of the window, select **Display Number Modification** -> **Modifications**.
4. To enter a new Modification for the Subscriber Numbering plan, click **Add**.
5. Make the following settings in Origination Context Setting:
 - Business Group: **ANY**
 - Numbering Plan: **ANY**
6. Make the following settings in Terminating Context Setting:
 - Business Group: Select the used business group e.g. **bg_sol**
 - Numbering Plan: Select the numbering plan of the subscriber e.g. **NP_br13eg**
 - Endpoint: **NONE**
7. Make the following settings in Modification Rule:
 - Input Type Of Number: **ANY**
 - Priority: **4**
 - Output Type Of Number: **ANY**
 - Number Source: **Input Number**
 - Presentation Restricted: **unselected**
 - Prefix Required: Activate the checkbox.
 - Optimize Type Of Number: **Extension**
8. Click Save. Your changes are saved.

[klara] - Display Number Modification - Windows Internet...

[klara]-Display Number Modification ?

 Create/Edit the 'calling party display number' to a specific format

Originating Context Setting

 Select a business group and/or numbering plan from the list.

Business Group ANY 

Numbering Plan ANY 

Terminating Context Setting

 Select a business group, numbering plan and/or endpoint from the list.

Business Group bg_sol 

Numbering Plan NP_br13_eg 

Endpoint NONE 

Modification Rule

 Select Input Type of Number, Output Type of number, Number Transmission and define if Optimized possible.

Input Type Of Number: ANY 

Priority: 4 

Output Type Of Number: ANY 

Number Source: Input Number 

Presentation Restricted:

Prefix Required:

Optimize Type Of Number: Extension 

1.3.2. Settings in StartCli

In Openscape Voice V4R1 and earlier: the follow CLI parameter must be set

Srx/Main/OutGoingCallingPartyNumberType to Value=0

This needs to send the Calling Party Number with Typ of Number International.

In Openscape Voice V5 and newer: The Parameter above is not more available. Calling Party Number definitions will be made in the Display Number Modification.

1.4. Configuration OpenBranch

Only some OpenScape Business specific information is covered in this guideline. For more detailed information about OpenScape Branch refer to official guide "Administrator Documentation", which is available in SEN E-Docu.

Contents

This section covers the following topics:

Section 1.4.1, "Network Services"

Section 1.4.2, "VoIP"

1.4.1. Network Services

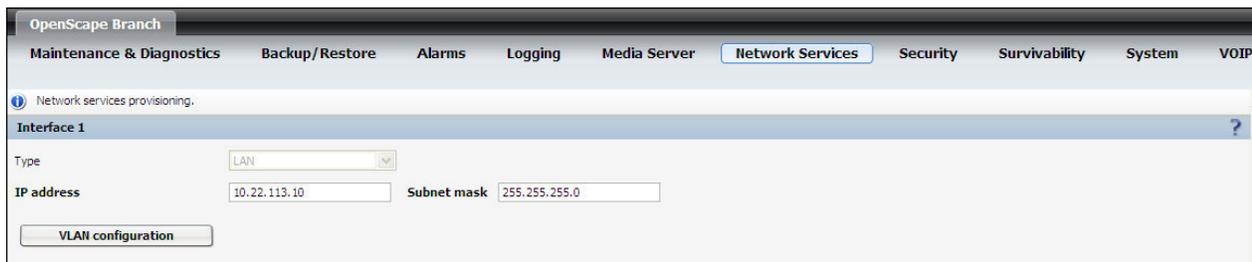
Proceed as follows:

1. **OpenScape Branch** -> select **Network Services**.

2. Make the following settings:

- Interface 1

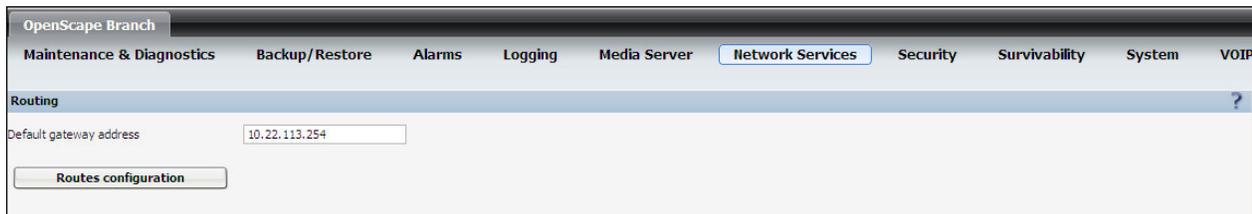
- IP address, Subnet mask: Enter the IP address and subnet mask of the OpenScape Branch proxy server, e.g. 10.22.113.10 and 255.255.255.0.



The screenshot shows the 'OpenScape Branch' configuration interface. The 'Network Services' tab is selected. Under 'Interface 1', the 'Type' is set to 'LAN'. The 'IP address' field contains '10.22.113.10' and the 'Subnet mask' field contains '255.255.255.0'. A 'VLAN configuration' button is visible at the bottom.

- Routing

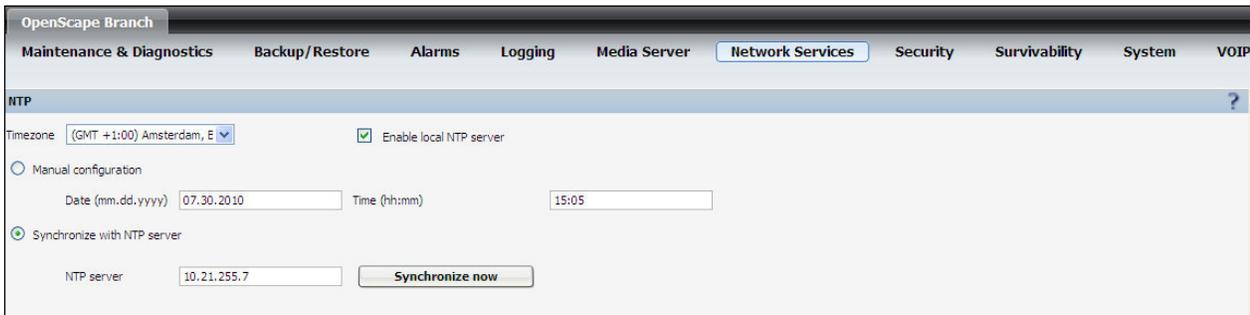
- Default gateway IP address: Enter the IP address of the default gateway, e.g. 10.22.113.254.



The screenshot shows the 'OpenScape Branch' configuration interface. The 'Network Services' tab is selected. Under 'Routing', the 'Default gateway address' field contains '10.22.113.254'. A 'Routes configuration' button is visible at the bottom.

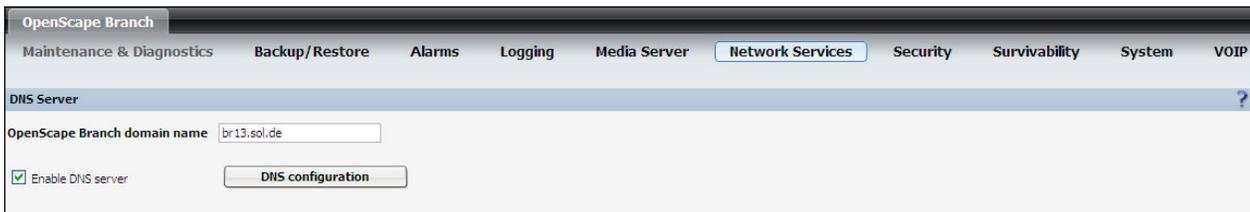
- NTP

- Timezone: e.g. **GMT+1:00**
- Synchronize with NTP server: activate this field.
- NTP server: Enter the NTP server IP address, e.g. **10.21.255.7**.
- Synchronize now: Click this button.



- DNS Server

- OpenScape Branch domain name: Enter the "DNS-SRV" name of the external office (Domain Name System **SERVICE**), e.g. **br13.sol.de**.
- Enable DNS server: Activate the checkbox.
- DNS configuration: Click this button for additional configuration



- DNS configuration

- Zone configuration

To do this proceed as follows:

1. Click **Add**, to add a row to the table.
2. In the "Type" column, select the **slave** type.
3. In the "Zone name" column, enter the name of the DNS zone, e.g. **sol.de**.
4. In the "IP Master/Forward" column, enter the IP address of the customer DNS server.
5. In the "File name" column, enter the file name where the DNS zone data should be saved.

- Forward IP

In the "Forward IP Address list", the IP address of the customer DNS server must be entered. The customer DNS server is required for queries outside of the transmitted zone. To do this proceed as follows:

1. Enter the customer DNS server IP address in the input field, e.g. 10.22.100.100, and click **Add**. The IP address is added to the "Forward IP Address list".
2. Click **Save**. Your changes are saved.

The screenshot shows the "DNS Server" configuration window. At the top, there is a header "DNS Server" with a help icon. Below it, a sub-header "DNS Server provisioning." is visible. The main content area is divided into two sections: "Zone configuration" and "Forward IP".

Zone configuration

Row	Type	Zone name	IP Masters/Forwards	File name
1	slave	sol.de	10.22.100.100	sol.de

Below the table are "Add" and "Delete" buttons.

Forward IP

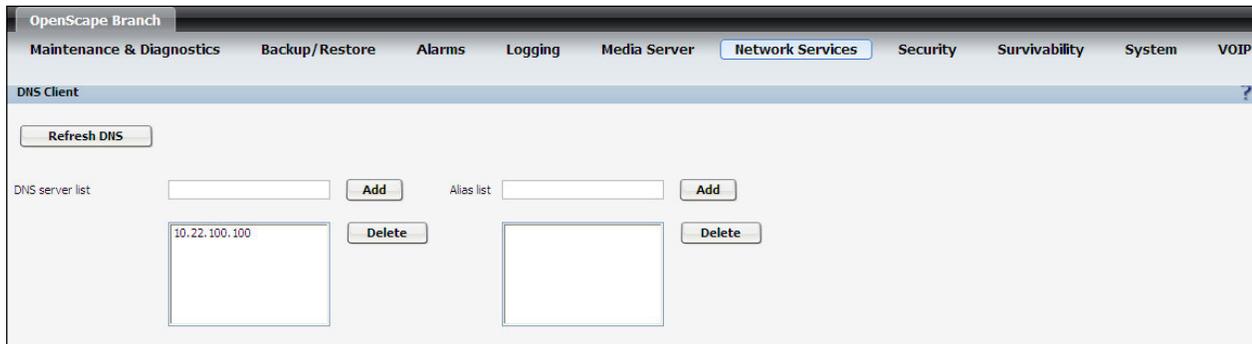
Forward IP Address list: **Add**

Delete

At the bottom of the window, there is a status bar showing "1 Item" and "Save" and "Cancel" buttons.

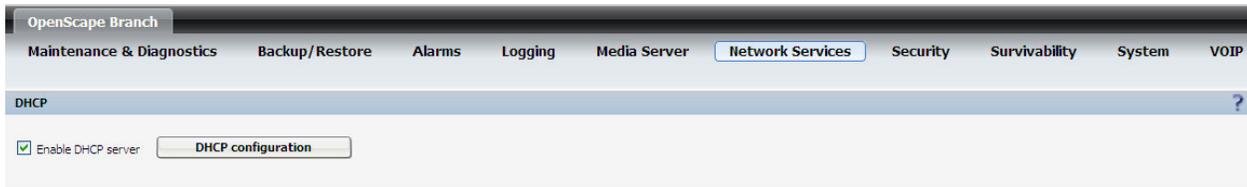
- DNS Client

- Enter the customer DNS server IP address in the input field “DNS server list”, e.g. 10.22.100.100, and click **Add**. The IP address is added to the "DNS server list."



- DHCP

- Enable DHCP server: Activate the checkbox. Enable DNS server: Activate the checkbox.
- DHCP configuration: Click the button to configure the DHCP Server.



- DHCP configuration

Most important setting

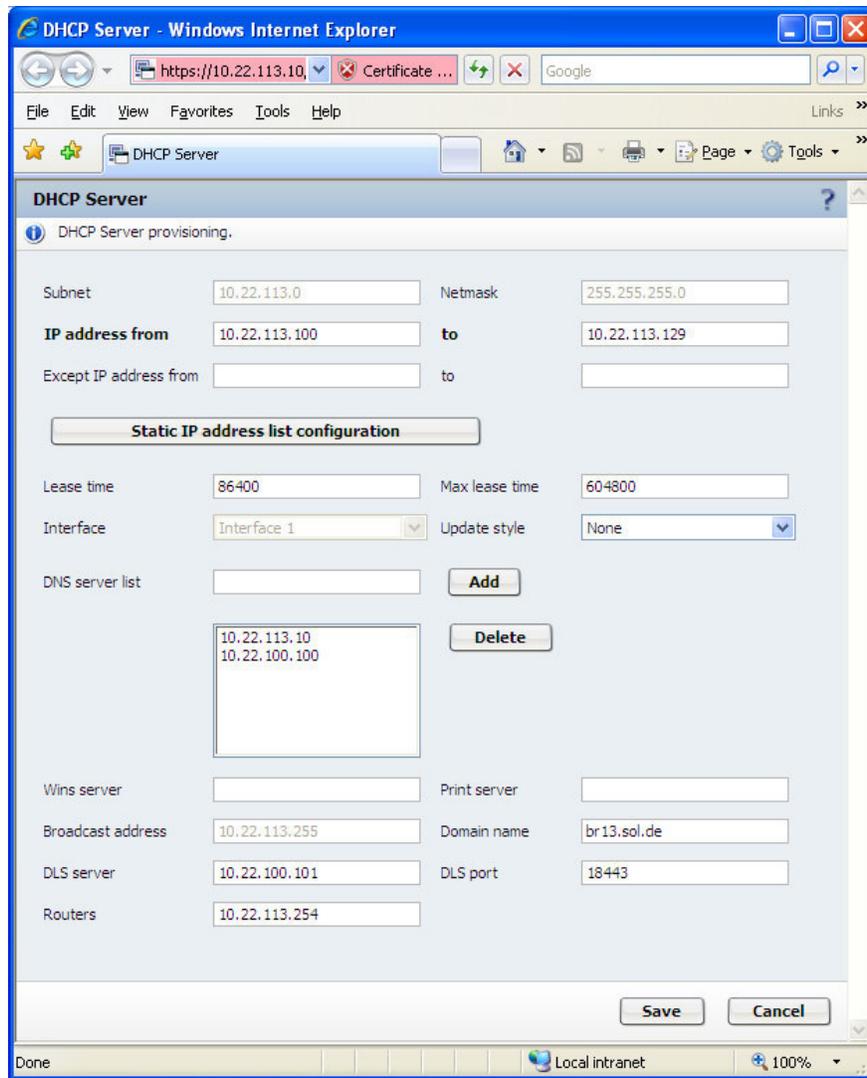
In the "DNS server list" of the "DHCP Server" window, the OpenScape Branch proxy server and customer DNS server IP addresses must be specified. To do this proceed as follows:

1. Enter the OpenScape Branch proxy server IP address in the input field, e.g. **10.22.113.10**, and click **Add**. The IP address is added to the "DNS server list".
2. Enter the customer DNS server IP address in the input field, e.g. **10.22.100.100**, and click **Add**. The IP address is added to the "DNS server list".

Additional settings

In the "DHCP Server" window, make the following settings:

- Subnet: e.g. **10.22.113.0**
- Netmask: e.g. **255.255.255.0**
- IP address from: e.g. **10.22.113.100** to: e.g. **10.22.113.129**
- Static IP address list configuration: Do not click.
- Lease time: **86400**
- Max. lease time: **604800**
- Interface: **Interface 1**
- Update style: **None**
- Broadcast address: e.g. **10.22.113.255**
- Domain name: "DNS-SRV" name of the external office (Domain Name System SERVICE) e.g. **br13.sol.de**
- DLS server: IP of the DLS Server e.g. **10.22.100.101**
- DLS Port: **18443**
- Routers: e.g. **10.22.113.254**



1.4.2. VOIP

Proceed as follows:

1. **OpenScape Branch** -> select **VOIP**.

2. Make the following settings:

- SIP configuration

SIP connectivity to OpenScape Voice will not be described here.

- For more information see the official Openscape Branch Administrator documentation chapter “How to Configure the Communication System”.

OpenScape Branch

Maintenance & Diagnostics Backup/Restore Alarms Logging Media Server Network Services Security Survivability System **VOIP**

SIP configuration

OpenScape Branch mode: Proxy
OpenScape Voice mode: Simplex
Options destination port: 5060

SIP listening ports:
TCP: 5060
UDP: 5060
TLS: 5061

Node 1
Target type: SRV Record
Primary server: Transport: TCP Port: 0
Backup server: Transport: TCP Port: 0
SRV record: node1.klara.sol.de Transport: TCP

Node 2
Target type: Binding
Primary server: Transport: TCP Port: 0
Backup server: Transport: TCP Port: 0
SRV record: Transport: TCP

Outbound SIP Server: Node 1

Enable Far End NAT

Other Trusted Servers Error codes

SIP Manipulation SIP routing

- SIP Manipulation: Click the button to configure SIP headers.

SIP Manipulation

The settings in this table are required for emergency handling. Complete the table as follows:

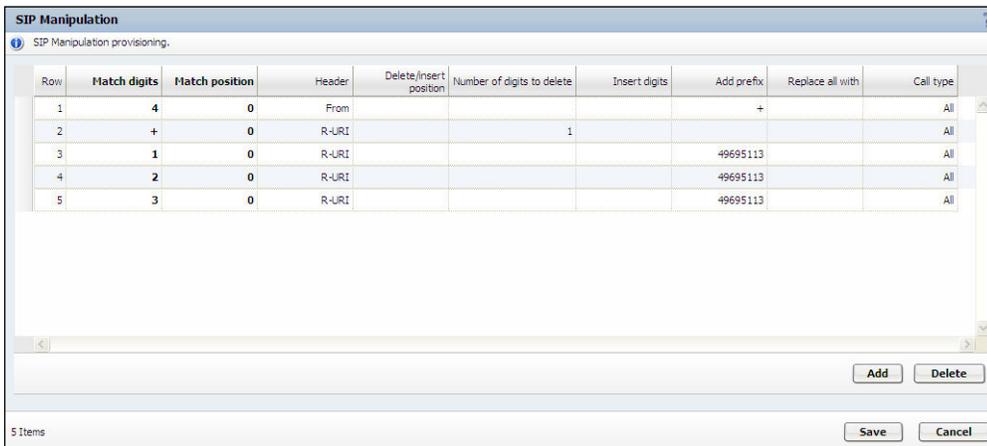
Row	Match digits	Match position	Header	Delete/insert position	Number of digits to delete	Insert digits	Add prefix	Replace all with
1	4	0	From				+	
2	+	0	R-URI		1			
3	1	0	R-URI				e.g. 49695113	
4	2	0	R-URI				e.g. 49695113	
5	3	0	R-URI				e.g. 49695113	

Explanations:

First row: The phone number must have a plus "+" at the beginning so HiPath 3000 can recognize this as an international phone number.

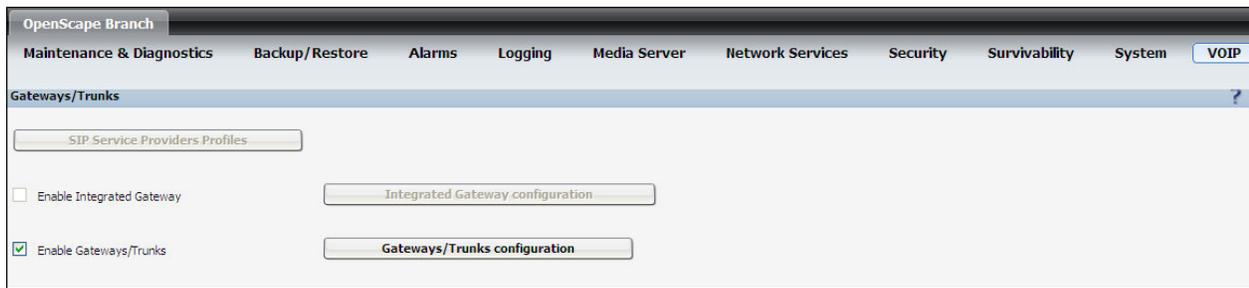
Second row: The plus "+" in the Request-URI (R-URI) must be deleted, as telephones are not registered with a plus "+".

Third to fifth rows: These settings enable phones to dial a short internal number and not international dialing.



- Gateway/Trunks

- Enable Gateways/Trunks: Activate the checkbox.
- Gateways/Trunks configuration: Click this button for adding a gateway



- HG 1500 should be added as a gateway.
- Use the screenshot below as an example. For more information see the official Openscape Branch Administrator documentation chapter “Configuration of Gateways”.

The screenshot shows the 'Gateways/Trunks' provisioning table in the administrator interface. The table has the following columns: Row, IP Address or FQDN, Port, Interface, Transport, Routing prefix/FQDN, Gateway/Trunk type, Functional type, Trunk Profile, Output digit strip, Output digit add, and Priority. There is one row of data:

Row	IP Address or FQDN	Port	Interface	Transport	Routing prefix/FQDN	Gateway/Trunk type	Functional type	Trunk Profile	Output digit strip	Output digit add	Priority
1	10.22.113.191	5060	LAN	TCP	%	3k/4k	All Modes Egress/Ingress	Gateway	0		1

At the bottom right of the table, there are 'Add' and 'Delete' buttons.

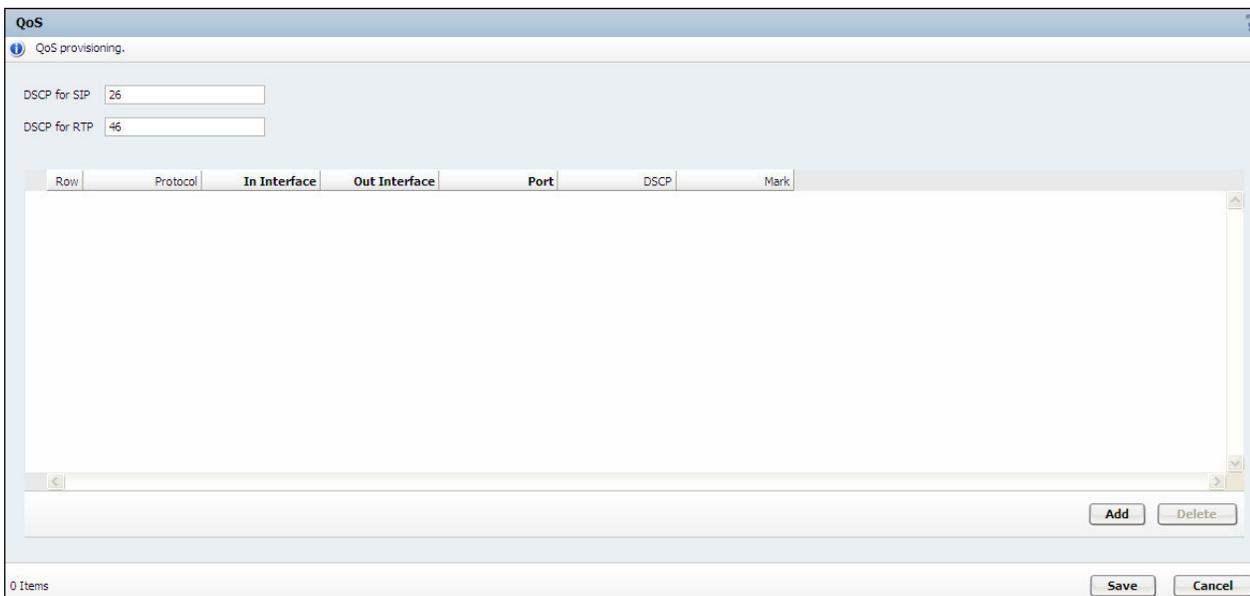
- QoS

- Enable QoS: Activate the checkbox.
- QoS configuration: Click this button to setup the Layer 3 priority.



The DSCP priority must be entered as decimal value.

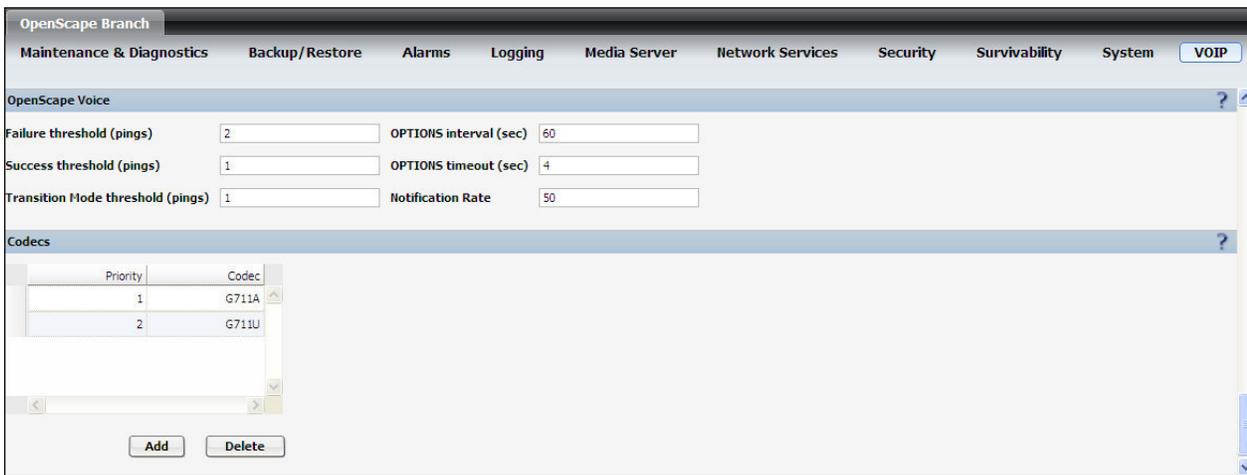
- DSCP for SIP: enter 26 for L3 QoS priority Diffserv AF31.
- DSCP for RTP: enter 46 for L3 QoS priority Diffserv EF.



- Codec

- Set the codec priority as in HG 1500, i.e.:

Priority	Codec
1	G711A
2	G711U



About Unify

Unify is one of the world's leading communications software and services firms, providing integrated communications solutions for approximately 75 percent of the Fortune Global 500. Our solutions unify multiple networks, devices and applications into one easy-to-use platform that allows teams to engage in rich and meaningful conversations. The result is a transformation of how the enterprise communicates and collaborates that amplifies collective effort, energizes the business, and enhances business performance. Unify has a strong heritage of product reliability, innovation, open standards and security.

Unify.com

Copyright © Unify Software and Solutions GmbH & Co. KG 2015
Mies-van-der-Rohe-Str. 6, 80807 Munich/Germany
All rights reserved.

The information provided in this document contains merely general descriptions or characteristics of performance which in case of actual use do not always apply as described or which may change as a result of further development of the products. An obligation to provide the respective characteristics shall only exist if expressly agreed in the terms of contract.

Availability and technical specifications are subject to change without notice.

Unify, OpenScape, OpenStage and HiPath are registered trademarks of Unify Software and Solutions GmbH & Co. KG. All other company, brand, product and service names are trademarks or registered trademarks of their respective holders.

UNIFY Harmonize
your enterprise