

Connecting Avaya IP Office 500 IP-PBX to BroadCloud SIP Trunk using AudioCodes Mediant™ E-SBC

Version 7.0



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Notice

This document describes how to connect the IP-PBX and BroadCloud SIP Trunk using AudioCodes Mediant E-SBC product series.

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1 Introduction

This Configuration Note describes how to set up AudioCodes Enterprise Session Border Controller (hereafter, referred to as *E-SBC*) for interworking between BroadCloud's SIP Trunk and IP-PBX environment.

1.1 Intended Audience

The document is intended for engineers, or AudioCodes and BroadCloud Partners who are responsible for installing and configuring BroadCloud's SIP Trunk and IP-PBX for enabling VoIP calls using AudioCodes E-SBC.

1.2 About AudioCodes E-SBC Product Series

AudioCodes' family of E-SBC devices enables reliable connectivity and security between the Enterprise's and the service provider's VoIP networks.

The E-SBC provides perimeter defense as a way of protecting Enterprises from malicious VoIP attacks; mediation for allowing the connection of any PBX and/or IP-PBX to any service provider; and Service Assurance for service quality and manageability.

Designed as a cost-effective appliance, the E-SBC is based on field-proven VoIP and network services with a native host processor, allowing the creation of purpose-built multiservice appliances, providing smooth connectivity to cloud services, with integrated quality of service, SLA monitoring, security and manageability. The native implementation of SBC provides a host of additional capabilities that are not possible with standalone SBC appliances such as VoIP mediation, PSTN access survivability, and third-party value-added services applications. This enables Enterprises to utilize the advantages of converged networks and eliminate the need for standalone appliances.

AudioCodes E-SBC is available as an integrated solution running on top of its field-proven Mediant Media Gateway and Multi-Service Business Router platforms, or as a software-only solution for deployment with third-party hardware.

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2 Component Information

2.1 IP-PBX Version

Table 2-1: IP-PBX Version

Vendor	Avaya
Model	IP Office 500 V2
Software Version	7.0 (5)
Protocol	SIP/UDP or SIP/TCP
Additional Notes	None

2.2 AudioCodes E-SBC Version

Table 2-2: AudioCodes E-SBC Version

SBC Vendor	AudioCodes
Models	<ul style="list-style-type: none"> ▪ Mediant 500 E-SBC ▪ Mediant 800 Gateway & E-SBC ▪ Mediant 1000B Gateway & E-SBC ▪ Mediant 3000 Gateway & E-SBC ▪ Mediant 2600 E-SBC ▪ Mediant 4000 E-SBC
Software Version	SIP_F7.00A.035.012
Protocol	<ul style="list-style-type: none"> ▪ SIP/UDP (to the BroadCloud SIP Trunk) ▪ SIP/UDP or SIP/TCP (to the IP-PBX)
Additional Notes	None

2.3 BroadCloud SIP Trunking Version

Table 2-3: BroadCloud Version

Vendor/Service Provider	BroadCloud
SSW Model/Service	BroadWorks
Software Version	21
Protocol	SIP/UDP
Additional Notes	None

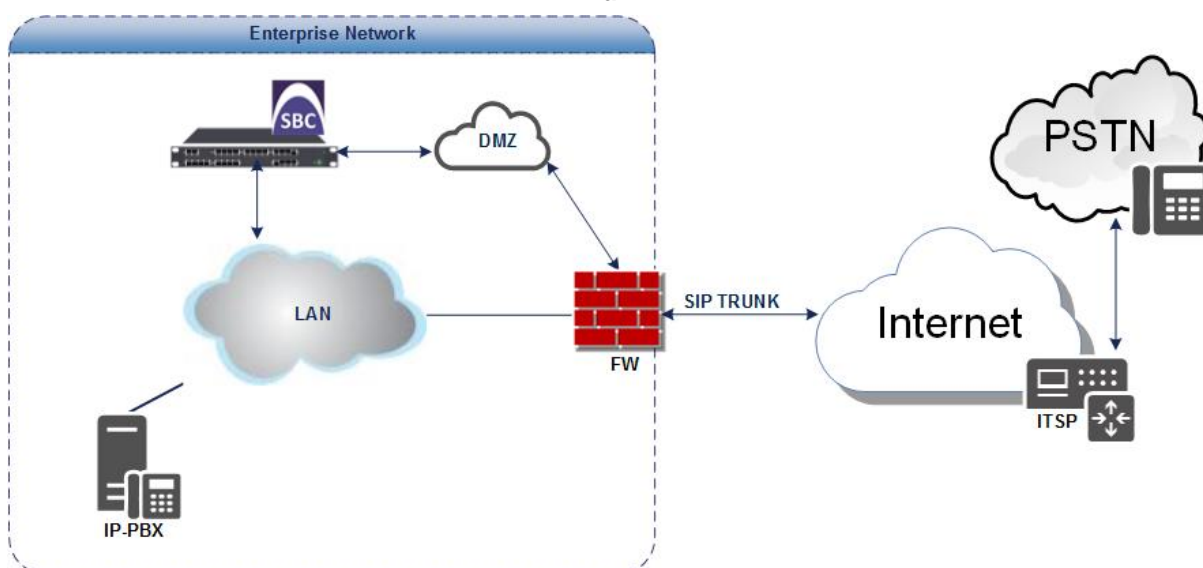
2.4 Interoperability Test Topology

The interoperability testing between AudioCodes E-SBC and BroadCloud SIP Trunk with IP-PBX was done using the following topology setup:

- Enterprise deployed with IP-PBX in its private network for enhanced communication within the Enterprise.
- Enterprise wishes to offer its employees enterprise-voice capabilities and to connect the Enterprise to the PSTN network using BroadCloud's SIP Trunking service.
- AudioCodes E-SBC is implemented to interconnect between the Enterprise LAN and the SIP Trunk.
 - **Session:** Real-time voice session using the IP-based Session Initiation Protocol (SIP).
 - **Border:** IP-to-IP network border between IP-PBX network in the Enterprise LAN and BroadCloud's SIP Trunk located in the public network.

The figure below illustrates this interoperability test topology:

Figure 2-1: Interoperability Test Topology between E-SBC and IP-PBX with BroadCloud SIP Trunk



2.4.1 Environment Setup

The interoperability test topology includes the following environment setup:

Table 2-4: Environment Setup

Area	Setup
Network	<ul style="list-style-type: none"> ▪ IP-PBX is located on the Enterprise's LAN ▪ BroadCloud SIP Trunk is located on the WAN
Signaling Transcoding	<ul style="list-style-type: none"> ▪ IP-PBX operates with SIP-over-UDP transport type ▪ BroadCloud SIP Trunk operates with SIP-over-UDP transport type
Codecs Transcoding	<ul style="list-style-type: none"> ▪ IP-PBX supports G.711A-law, G.711U-law, and G.729 coder ▪ BroadCloud SIP Trunk supports G.711A-law, G.711U-law, and G.729 coder
Media Transcoding	<ul style="list-style-type: none"> ▪ IP-PBX operates with RTP media type ▪ BroadCloud SIP Trunk operates with RTP media type

2.4.2 Known Limitations

There were no limitations observed in the interoperability tests done for the AudioCodes E-SBC interworking between IP-PBX and BroadCloud 's SIP Trunk.

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3 Configuring Avaya IP Office 500 IP-PBX

This chapter describes how to configure basic parameters of the Avaya IP Office 500 IP-PBX to operate with AudioCodes E-SBC.



Note: For more complicated configuration parameters please refer to User Manual of each IP-PBX.

3.1 Basic Configuration Parameters

The figures below display the main parameters which should be configured for the Avaya IP Office 500 IP-PBX to operate with the AudioCodes E-SBC.

➤ **To configure Avaya IP Office 500 IP-PBX:**

1. In the 'IP Offices' Navigation pane, select **Control Unit** and verify the IP-PBX type and version:

Figure 3-1: Avaya IP Office 500 V2 Version

IP Offices			Control Unit			IP 500 V2	
<ul style="list-style-type: none"> BOOTP (5) Operator (3) 00E0070585A4 System (1) Line (5) Control Unit (3) Extension (33) User (35) HuntGroup (2) Short Code (64) Service (0) RAS (1) Incoming Call Route (3) WanPort (0) Directory (0) Time Profile (0) Firewall Profile (1) IP Route (2) Account Code (0) 	Dev No.	Dev Type	Version	Unit	Device Number	1	
	1	IP 500 V2	7.0 (5)	Unit Type	IP 500 V2		
	2	COMBO6210/ATM4	7.0 (5)	Version	7.0 (5)		
	3	TCM8	7.0 (5)	Serial Number	00e0070585a4		
			Unit IP Address	172.26.241.10			
			Interconnect Number	0			
			Module Number	Control Unit			

2. In the 'IP Offices' Navigation pane, select **Licence** and verify the IP-PBX Licenses:

Figure 3-2: SIP Trunk Licensing

IP Offices			Licence		SIP Trunk Channels	
<ul style="list-style-type: none"> BOOTP (5) Operator (3) 00E0070585A4 System (1) Line (5) Control Unit (3) Extension (33) User (35) HuntGroup (2) Short Code (64) Service (0) RAS (1) Incoming Call Route (3) WanPort (0) Directory (0) Time Profile (0) Firewall Profile (1) IP Route (2) Account Code (0) Licence (22) Tunnel (0) User Rights (9) ARS (1) E911 System (1) 	Licence Type	Status	Licences	Licence Key	NieG@ZglgBmSntzj8@wSoyRUocjIdD8X	
	3rd Party IP Endpoints	Valid	Licence Type	SIP Trunk Channels		
	AUDDX Voicemail	Valid	Licence Status	Valid		
	Avaya IP endpoints	Valid	Instances	5		
	Avaya IP endpoints	Valid	Expiry Date	Never		
	CTI Link Pro	Valid				
	Customer Service Agent	Dorm				
	Customer Service Supervisor	Dorm				
	IP500 Voice Networking Channels	Valid				
	IPSec Tunnelling	Valid				
	Mobile Worker	Valid				
	Office Worker	Valid				
	Phone Manager Pro (per seat)	Valid				
	Phone Manager Pro IP Audio Enabled (users)	Valid				
	Power User	Valid				
	Preferred Edition (VoiceMail Pro)	Valid				
	Receptionist	Valid				
	SIP Trunk Channels	Valid				
	Software Upgrade 4	Valid				
	Teleworker	Valid				
	VMPro Networked Messaging	Valid				
	VMPro TTS (Scansoft)	Valid				
Wave User	Valid					

- In the 'IP Offices' Navigation pane, select **System** and configure the Avaya IP Office local IP address that will be used for the SIP trunks connection:

Figure 3-3: Avaya IP Office local IP

The screenshot shows the configuration page for system 00E0070585A4. The 'LAN2' tab is selected, and the 'SIP Registrar' sub-tab is active. The IP Address field is set to 172.26.100.170, the IP Mask to 255.255.255.0, and the Primary Trans. IP Address to 0.0.0.0. The Firewall Profile is set to <None>, and the RIP Mode is set to None. The 'Enable NAT' checkbox is unchecked. The Number Of DHCP IP Addresses is set to 200. Under the DHCP Mode section, the 'Disabled' radio button is selected, and an 'Advanced' button is visible.

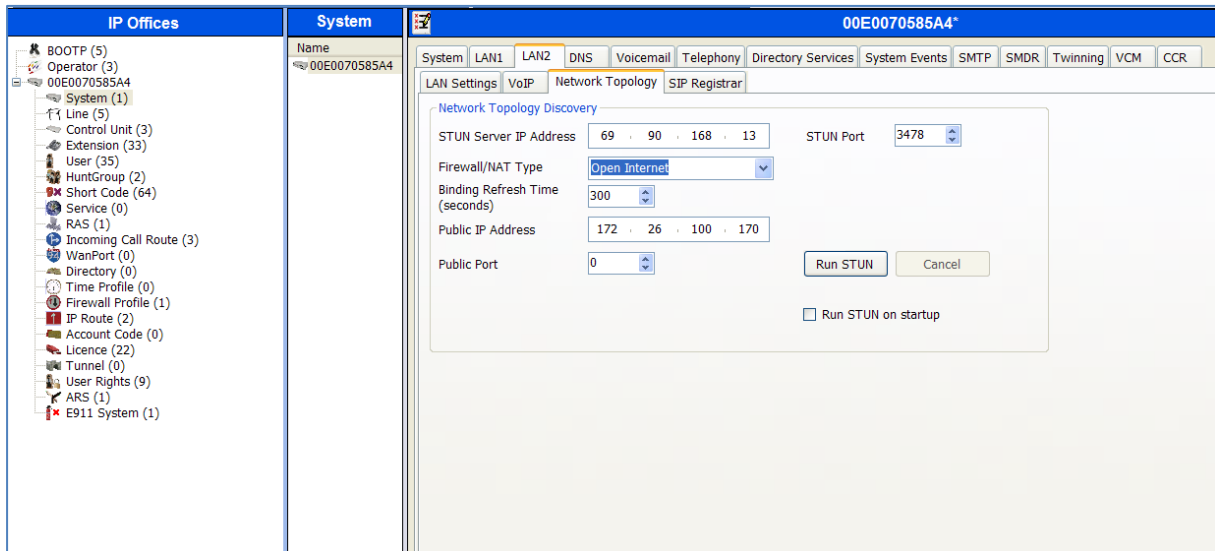
- Select the **LAN2** and **VoIP** tabs and select the **SIP Trunks Enable** check box:

Figure 3-4: Enable SIP Trunks

The screenshot shows the configuration page for system 00E0070585A4 with the 'VoIP' tab selected. The 'SIP Registrar' sub-tab is active. The 'SIP Trunks Enable' checkbox is checked. Other checked options include 'H323 Gatekeeper Enable' and 'SIP Registrar Enable'. Under the 'H323 Auto-create' section, 'H323 Auto-create Extn' and 'H323 Auto-create User' are unchecked. The 'RTP Port Number Range' section shows a Port Range (Minimum) of 49152 and a Port Range (Maximum) of 53246. The 'Enable RTCP Monitoring On Port 5005' checkbox is checked. The 'DiffServ Settings' section shows DSCP (Hex) set to B8, DSCP Mask (Hex) set to FC, and SIG DSCP (Hex) set to 88. Below that, DSCP is set to 46 and DSCP Mask to 63. The 'DHCP Settings' section shows Primary Site Specific Option Number (SSON) set to 176, Secondary Site Specific Option Number (SSON) set to 242, and VLAN set to Not Present. The 1100 Voice VLAN Site Specific Option Number (SSON) is set to 232. The 'RTP Keepalives' section shows Scope set to Disabled and Periodic timeout set to 0.

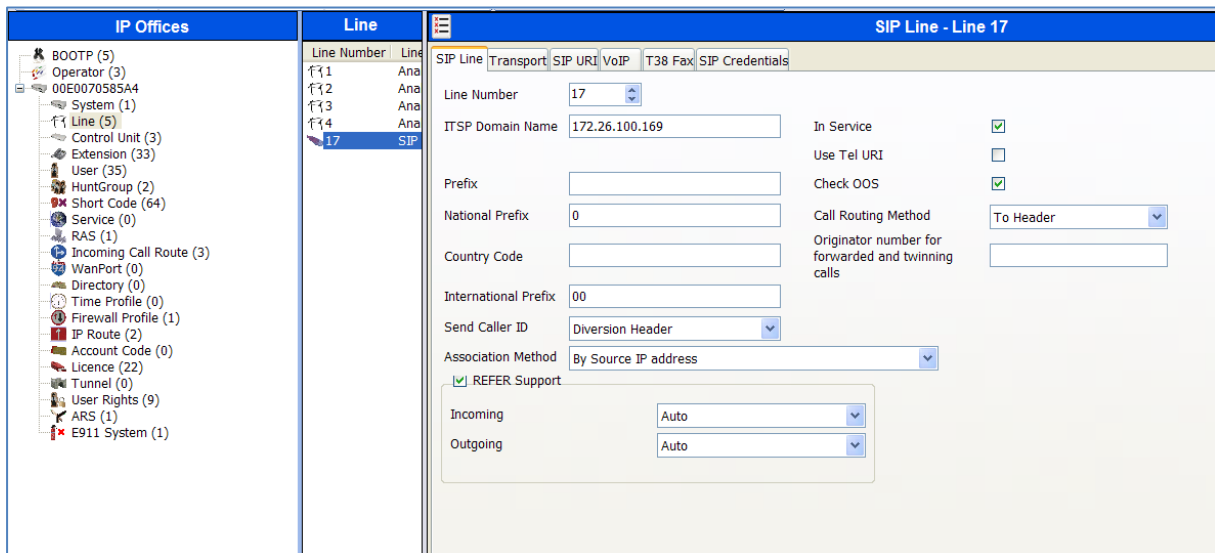
5. Select the **Network Topology** tab and from the 'Firewall/NAT Type' drop-down list, select **Open Internet**:

Figure 3-5: Configure Network Topology



6. In the 'IP Offices' Navigation pane, select **Line** to create a SIP Line.
7. Under the **SIP Line** tab, enter the ITSP Domain name or IP address and ensure that the 'Call Routing Method' is set to **To Header**.

Figure 3-6: Create SIP Line



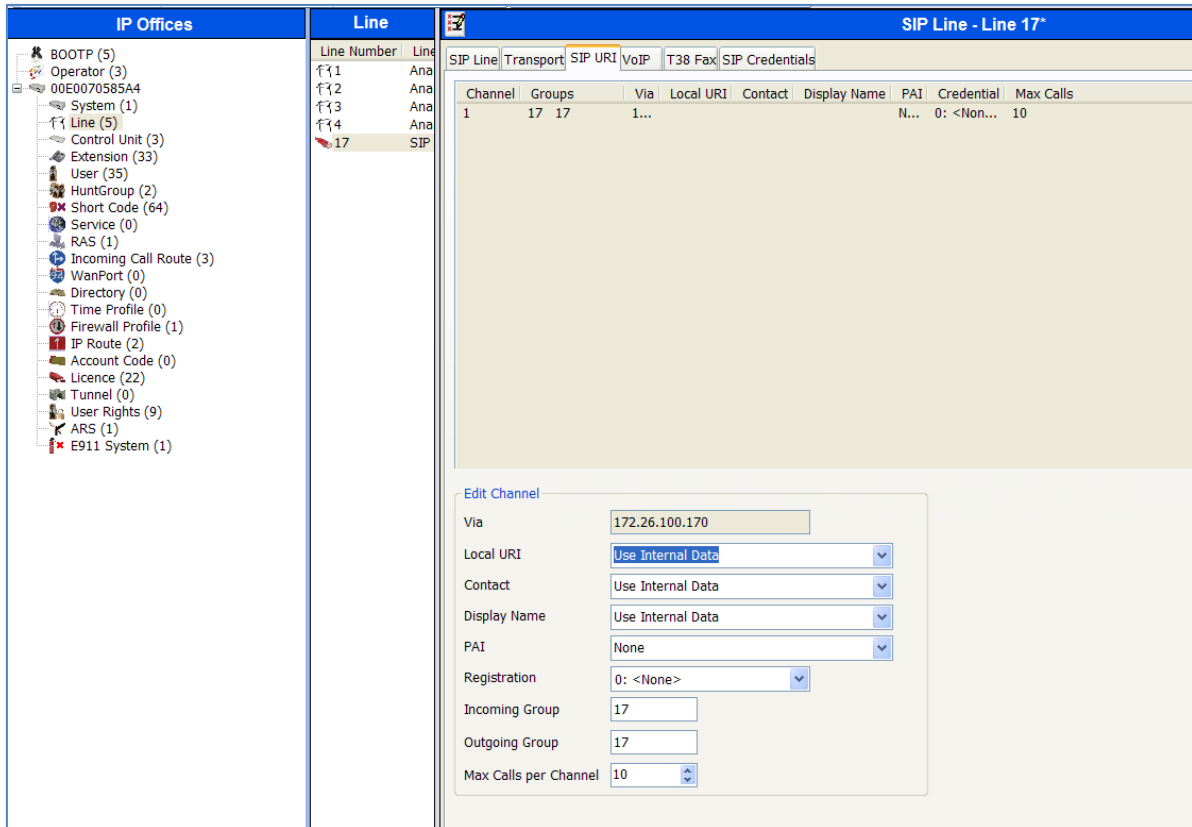
8. Select the **Transport** tab and from the 'Use Network Topology Info' drop-down list, select **LAN 2**:

Figure 3-7: Configuring Transport Type and Destination SIP Port

The screenshot shows the Avaya IP Office configuration interface. On the left, the 'IP Offices' tree is visible, with 'Line (5)' expanded. The 'Line' table shows Line 17 selected. The main configuration area is for 'SIP Line - Line 17'. The 'Transport' tab is active. The 'Use Network Topology Info' dropdown is set to 'LAN 2'. The 'Send Port' and 'Listen Port' are both set to 5060. The 'Calls Route via Registrar' checkbox is checked.

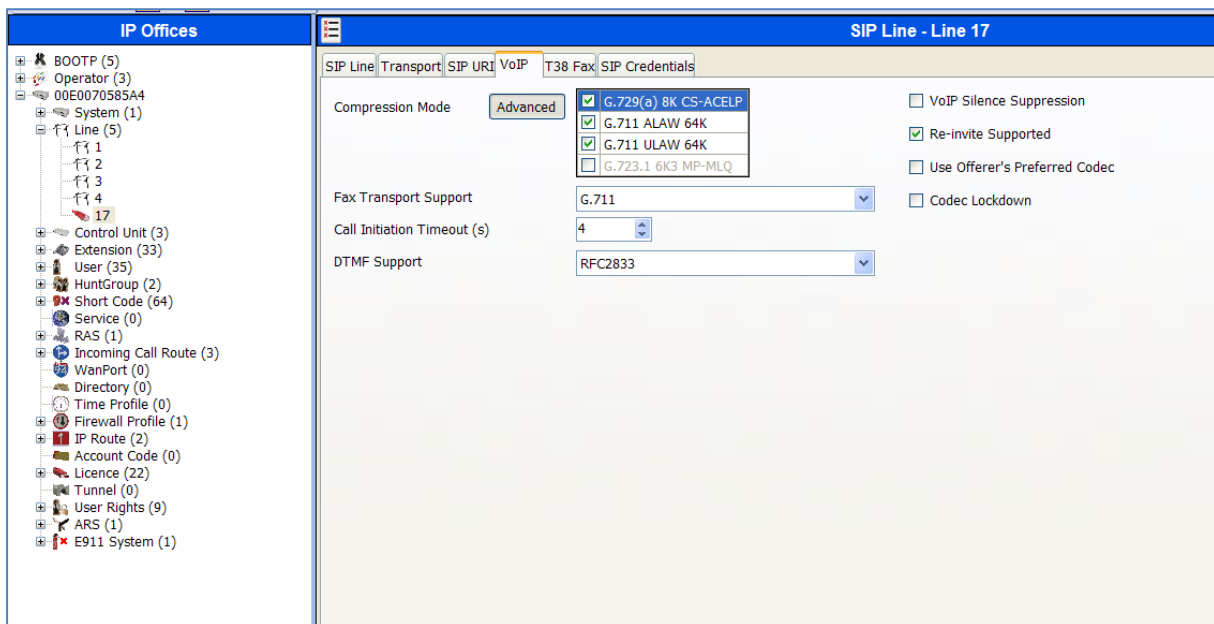
9. Select the **SIP URI** tab and enter the following information:
 - From the 'Local URI' drop-down list, select **Use Internal Data**.
 - From the 'Contact' drop-down list, select **Use Internal Data**.
 - From the 'Display Name' drop-down list, select **Use Internal Data**.
 - From the 'Incoming Group' drop-down list, type the line number created above (see Figure 3-6).
 - From the 'Outgoing Group' drop-down list, type the line number created above (Figure 3-6).

Figure 3-8: Configuring SIP Line Parameters



10. Select the **VoIP** tab, click the **Advanced** button and select the supported coders and DTMF method:

Figure 3-9: Configuring Coders and DTMF Method



- In the 'IP Offices' Navigation pane, select **Short Code** to create an outgoing routing entry to the SIP Line:

Figure 3-10: Configuring Outgoing Routing

The screenshot displays the configuration interface for a Short Code. On the left, a navigation tree under 'IP Offices' shows 'Short Code (64)' selected. The main configuration area is titled 'Short Code' and contains the following fields:

- Code:** 8N;
- Feature:** Dial
- Telephone Number:** N@172.26.100.169
- Line Group Id:** 17
- Locale:** United States (US English)
- Force Account Code:**

The 'Short Code' list on the left includes various patterns such as *20*N#, *21*N#, *29, *30, *31, *32*N#, *33*N#, *34N;, *35*N#, *36, *37*N#, *38*N#, *39, *40, *41, *42, *43, *44, *45*N#, *46, *47, *48, *49, *50, *51, *52, *53*N#, *57*N#, *60*N#, *61*N#, *63*N#, *64, *65, *69, *70*N#, *71*N#, *9000*, *91N;, *92N;, *DSSN, *SDN, *SKN, and *8N;.

12. In the 'IP Offices' Navigation pane, select **Incoming Call Route** to create an incoming routing entry:

Figure 3-11: Configuring Incoming Routing

Line Group Id	Incoming Number	Destination
0		200 Main
0		DialIn
17		.

Standard	Voice Recording	Destinations
Bearer Capability		Any Voice
Line Group Id		17
Incoming Number		
Incoming Sub Address		
Incoming CLI		
Locale		
Priority		1 - Low
Tag		
Hold Music Source		System Source

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4 Configuring AudioCodes E-SBC

This chapter provides step-by-step procedures on how to configure AudioCodes E-SBC for interworking between IP-PBX and the BroadCloud SIP Trunk. These configuration procedures are based on the interoperability test topology described in Section 2.4 on page 10, and includes the following main areas:

- E-SBC WAN interface - BroadCloud SIP Trunking environment
- E-SBC LAN interface - IP-PBX environment

This configuration is done using the E-SBC's embedded Web server (hereafter, referred to as *Web interface*).

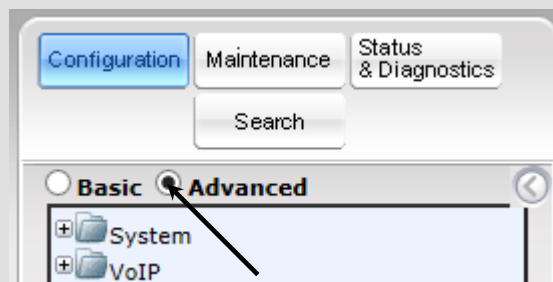
Notes:

- For implementing IP-PBX and BroadCloud SIP Trunk based on the configuration described in this section, AudioCodes E-SBC must be installed with a Software License Key that includes the following software features:

- ✓ **SBC**
- ✓ **Security**
- ✓ **DSP**
- ✓ **RTP**
- ✓ **SIP**

For more information about the Software License Key, contact your AudioCodes sales representative.

- The scope of this interoperability test and document does **not** cover all security aspects for connecting the SIP Trunk to the IP-PBX environment. Comprehensive security measures should be implemented per your organization's security policies. For security recommendations on AudioCodes' products, refer to the *Recommended Security Guidelines* document.
- Before you begin configuring the E-SBC, ensure that the E-SBC's Web interface Navigation tree is in Advanced-menu display mode. To do this, select the **Advanced** option, as shown below:



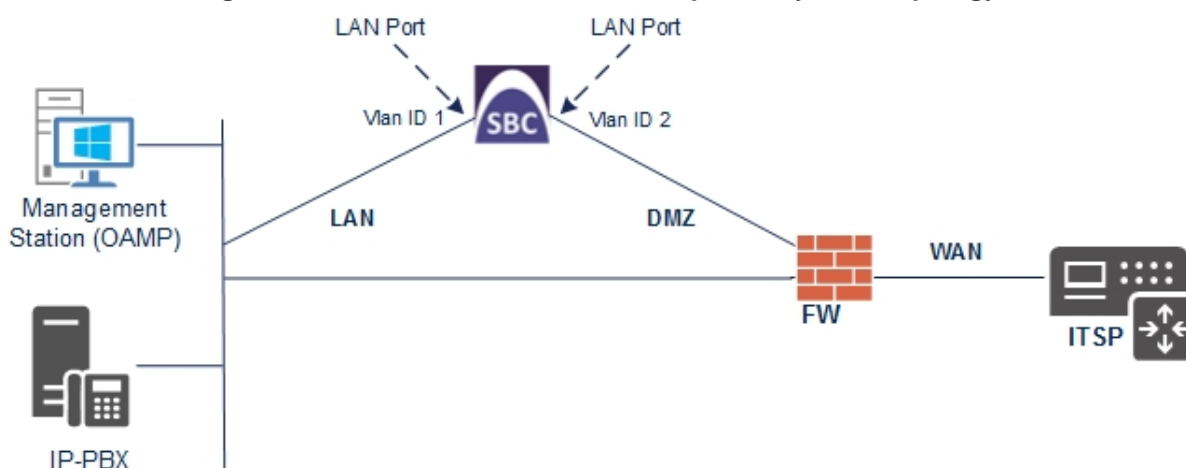
Note that when the E-SBC is reset, the Navigation tree reverts to Basic-menu display.

4.1 Step 1: IP Network Interfaces Configuration

This step describes how to configure the E-SBC's IP network interfaces. There are several ways to deploy the E-SBC; however, this interoperability test topology employs the following deployment method:

- E-SBC interfaces with the following IP entities:
 - IP-PBX, located on the LAN
 - BroadCloud SIP Trunk, located on the WAN
- E-SBC connects to the WAN through a DMZ network
- Physical connection: The type of physical connection to the LAN depends on the method used to connect to the Enterprise's network. In the interoperability test topology, E-SBC connects to the LAN and WAN using dedicated LAN ports (i.e., two ports and two network cables are used).
- E-SBC also uses two logical network interfaces:
 - LAN (VLAN ID 1)
 - WAN (VLAN ID 2)

Figure 4-1: Network Interfaces in Interoperability Test Topology



4.1.1 Step 1a: Configure VLANs

This step describes how to define VLANs for each of the following interfaces:

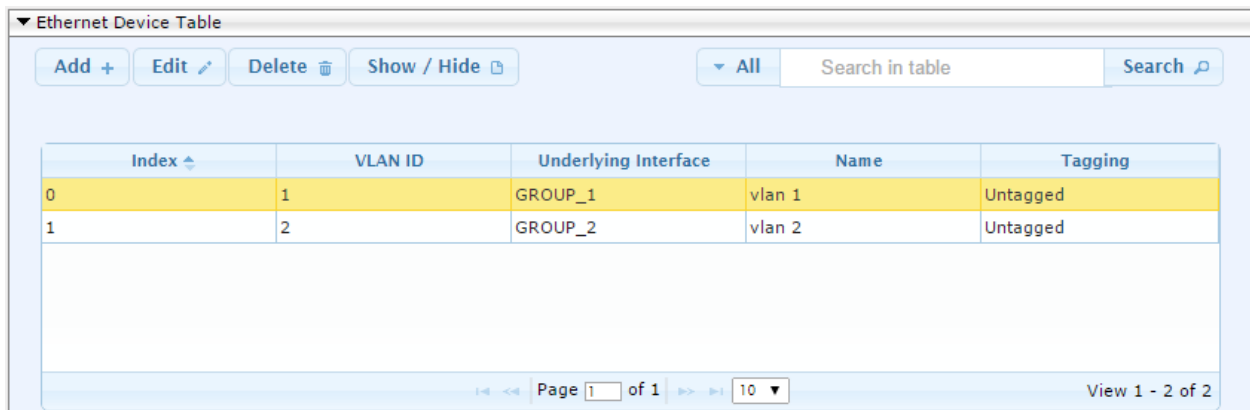
- LAN VoIP (assigned the name "Voice")
- WAN VoIP (assigned the name "WANSP")

➤ **To configure the VLANs:**

1. Open the Ethernet Device Table page (**Configuration** tab > **VoIP** menu > **Network** > **Ethernet Device Table**).
2. There will be one existing row for VLAN ID 1 and underlying interface GROUP_1.
3. Add another VLAN ID 2 for the WAN side as follows:

Parameter	Value
Index	1
VLAN ID	2
Underlying Interface	GROUP_2 (Ethernet port group)
Name	vlan 2
Tagging	Untagged

Figure 4-2: Configured VLAN IDs in Ethernet Device Table



4.1.2 Step 1b: Configure Network Interfaces

This step describes how to configure the IP network interfaces for each of the following interfaces:

- LAN VoIP (assigned the name "Voice")
- WAN VoIP (assigned the name "WANSP")

➤ **To configure the IP network interfaces:**

1. Open the IP Interfaces Table page (**Configuration** tab > **VoIP** menu > **Network** > **IP Interfaces Table**).
2. Modify the existing LAN network interface:
 - a. Select the 'Index' radio button of the **OAMP + Media + Control** table row, and then click **Edit**.
 - b. Configure the interface as follows:

Parameter	Value
IP Address	172.26.100.169 (IP address of E-SBC)
Prefix Length	24 (subnet mask in bits for 255.255.255.0)
Default Gateway	172.26.100.1
VLAN ID	1
Interface Name	Voice (arbitrary descriptive name)
Primary DNS Server IP Address	
Underlying Device	vlan 1

3. Add a network interface for the WAN side:
 - a. Enter **1**, and then click **Add Index**.
 - b. Configure the interface as follows:

Parameter	Value
Application Type	Media + Control
IP Address	65.196.9.185 (WAN IP address)
Prefix Length	28 (for 255.255.255.240)
Default Gateway	65.196.9.177 (router's IP address)
VLAN ID	2
Interface Name	WANSP
Primary DNS Server IP Address	198.6.1.146
Secondary DNS Server IP Address	198.6.1.122
Underlying Device	vlan 2

4. Click **Apply**, and then **Done**.

The configured IP network interfaces are shown below:

Figure 4-3: Configured Network Interfaces in IP Interfaces Table

▼ Interface Table

▼ All

Index ▲	Interface Name	Application Type	Interface Mode	IP Address	Prefix Length	Default Gateway	Primary DNS	Secondary DNS	Underlying Device
0	Voice	OAMP + Medi	IPv4 Manual	172.26.100.1	24	172.26.100.1	0.0.0.0	0.0.0.0	vlan 1
1	WANSP	Media + Cont	IPv4 Manual	65.196.9.185	28	65.196.9.177	198.6.1.146	198.6.1.122	vlan 2

Page 1 of 1

View 1 - 2 of 2

4.2 Step 2: Enable the SBC Application

This step describes how to enable the SBC application.

➤ **To enable the SBC application:**

1. Open the Applications Enabling page (**Configuration** tab > **VoIP** menu > **Applications Enabling** > **Applications Enabling**).

Figure 4-4: Enabling SBC Application



2. From the 'SBC Application' drop-down list, select **Enable**.
3. Click **Submit**.
4. Reset the E-SBC with a burn to flash for this setting to take effect (see Section 4.13 on page 61).

4.3 Step 3: Configure Media Realms

This step describes how to configure Media Realms. The simplest configuration is to create two Media Realms - one for internal (LAN) traffic and one for external (WAN) traffic.

➤ **To configure Media Realms:**

1. Open the Media Realm Table page (**Configuration** tab > **VoIP** menu > **VoIP Network** > **Media Realm Table**).
2. Add a Media Realm for the LAN interface. You can use the default Media Realm (Index 0), but modify it as shown below:

Parameter	Value
Index	0
Media Realm Name	MRLan (descriptive name)
IPv4 Interface Name	Voice
Port Range Start	6000 (as required by IP-PBX)
Number of Media Session Legs	100 (media sessions assigned with port range)

Figure 4-5: Configuring Media Realm for LAN

The screenshot shows a dialog box titled "Edit Row" with a close button (X) in the top right corner. The dialog contains the following fields and values:

- Index: 0
- Name: MRLan
- IPv4 Interface Name: Voice (dropdown menu)
- Port Range Start: 6000
- Number Of Media Session Legs: 100
- Port Range End: 6990
- Default Media Realm: No (dropdown menu)
- QoS Profile: None (dropdown menu)
- BW Profile: None (dropdown menu)

At the bottom of the dialog, there are two buttons: "Save" and "Cancel".

3. Configure a Media Realm for WAN traffic:

Parameter	Value
Index	1
Media Realm Name	MRWan (arbitrary name)
IPv4 Interface Name	WANSP
Port Range Start	7000 (represents lowest UDP port number used for media on WAN)
Number of Media Session Legs	100 (media sessions assigned with port range)

Figure 4-6: Configuring Media Realm for WAN

The configured Media Realms are shown in the figure below:

Figure 4-7: Configured Media Realms in Media Realm Table

Index	Name	IPv4 Interface Name	Port Range Start	Number Of Media Session Legs	Port Range End	Default Media Realm
0	MRLan	Voice	6000	100	6990	No
1	MRWan	WANSP	7000	100	7990	No

4.4 Step 4: Configure SIP Signaling Interfaces

This step describes how to configure SIP Interfaces. For the interoperability test topology, an internal and external SIP Interface must be configured for the E-SBC.

➤ **To configure SIP Interfaces:**

1. Open the SIP Interface Table page (**Configuration** tab > **VoIP** menu > **VoIP Network** > **SIP Interface Table**).
2. Add a SIP Interface for the LAN interface. You can use the default SIP Interface (Index 0), but modify it as shown below:

Parameter	Value
Index	0
Interface Name	IP-PBX (see Note on page 30)
Network Interface	Voice
Application Type	SBC
UDP Port	5060
TCP and TLS	0
Media Realm	MRLan

3. Configure a SIP Interface for the WAN:

Parameter	Value
Index	1
Interface Name	BroadCloud (see Note on page 30)
Network Interface	WANSP
Application Type	SBC
UDP Port	5060
TCP and TLS	0
Media Realm	MRWan

The configured SIP Interfaces are shown in the figure below:

Figure 4-8: Configured SIP Interfaces in SIP Interface Table

Index	Name	SRD	Network Interface	Application Type	UDP Port	TCP Port	TLS Port	Encapsulating Protocol	Media Realm
0	IP-PBX	DefaultSRD	Voice	SBC	5060	0	0	No encapsulation	MRLan
1	BroadCloud	DefaultSRD	WANSP	SBC	5060	0	0	No encapsulation	MRWan



Note: Unlike in previous software releases where configuration entities (e.g., SIP Interface, Proxy Sets, and IP Groups) were associated with each other using table row indices, Version 7.0 uses the string **names** of the configuration entities. Therefore, it is recommended to configure each configuration entity with meaningful names for easy identification.

4.5 Step 5: Configure Proxy Sets

This step describes how to configure Proxy Sets. The Proxy Set defines the destination address (IP address or FQDN) of the IP entity server. Proxy Sets can also be used to configure load balancing between multiple servers.

For the interoperability test topology, two Proxy Sets need to be configured for the following IP entities:

- IP-PBX
- BroadCloud SIP Trunk

The Proxy Sets will be later applying to the VoIP network by assigning them to IP Groups.

➤ **To configure Proxy Sets:**

1. Open the Proxy Sets Table page (**Configuration** tab > **VoIP** menu > **VoIP Network** > **Proxy Sets Table**).
2. Add a Proxy Set for the IP-PBX. You can use the default Proxy Set (Index 0), however modify it as shown below:

Parameter	Value
Proxy Set ID	0
Proxy Name	IP-PBX (see Note on page 30)
SBC IPv4 SIP Interface	IP-PBX
Proxy Keep Alive	Using Options

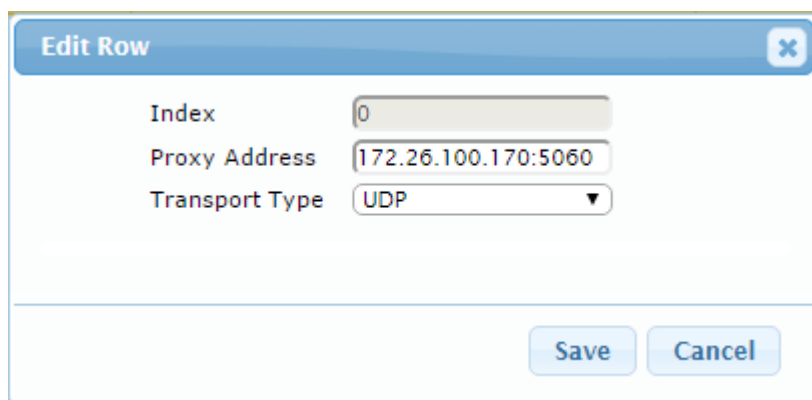
Figure 4-9: Configuring Proxy Set for IP-PBX

Parameter	Value
Index	0
SRD	DefaultSRD
Name	IP-PBX
Gateway IPv4 SIP Interface	None
SBC IPv4 SIP Interface	IP-PBX
Proxy Keep-Alive	Using OPTIONS
Proxy Keep-Alive Time [sec]	60
Redundancy Mode	
Proxy Load Balancing Method	Disable
DNS Resolve Method	
Proxy Hot Swap	Disable
Keep-Alive Failure Responses	
Classification Input	IP Address only
TLS Context Name	None

3. Configure a Proxy Address Table for Proxy Set for IP-PBX:
 - a. Go to **Configuration** tab > **VoIP** menu > **VoIP Network** > **Proxy Sets Table** > **Proxy Address Table**.

Parameter	Value
Index	0
Proxy Address	172.26.100.170:5060 (IP-PBX IP address / FQDN and destination port)
Transport Type	UDP

Figure 4-10: Configuring Proxy Address for IP-PBX



Edit Row
✕

Index

Proxy Address

Transport Type

4. Configure a Proxy Set for the BroadCloud SIP Trunk:

Parameter	Value
Proxy Set ID	1
Proxy Name	BroadCloud (see Note on page 30)
SBC IPv4 SIP Interface	BroadCloud
Proxy Keep Alive	Using Options

Figure 4-11: Configuring Proxy Set for BroadCloud SIP Trunk

Index	1
SRD	DefaultSRD
Name	BroadCloud
Gateway IPv4 SIP Interface	None
SBC IPv4 SIP Interface	BroadCloud
Proxy Keep-Alive	Using OPTIONS
Proxy Keep-Alive Time [sec]	60
Redundancy Mode	
Proxy Load Balancing Method	Disable
DNS Resolve Method	SRV
Proxy Hot Swap	Disable
Keep-Alive Failure Responses	
Classification Input	IP Address only
TLS Context Name	None

- a. Configure a Proxy Address Table for Proxy Set 1:
- b. Go to **Configuration** tab > **VoIP** menu > **VoIP Network** > **Proxy Sets Table** > **Proxy Address Table**.

Parameter	Value
Index	1
Proxy Address	nn6300southsipconnect.adpt-tech.com (IP-PBX IP address / FQDN and destination port)
Transport Type	UDP

Figure 4-12: Configuring Proxy Address for BroadCloud SIP Trunk

Index	1
Proxy Address	nn6300southsipconnec
Transport Type	UDP

The configured Proxy Sets are shown in the figure below:

Figure 4-13: Configured Proxy Sets in Proxy Sets Table

▼ Proxy Sets Table

▼ All

Index ↕	Name	SRD	Gateway IPv4 SIP Interface	SBC IPv4 SIP Interface	Proxy Keep-Alive Time [sec]	Redundancy Mode	Proxy Hot Swap
0	IP-PBX	■ DefaultSRD (#0)	None	IP-PBX	60		Disable
1	BroadCloud	■ DefaultSRD (#0)	None	BroadCloud	60		Disable

Page 1 of 1 10 ▼ View 1 - 2 of 2

4.6 Step 6: Configure IP Profiles

This step describes how to configure IP Profiles. The IP Profile defines a set of call capabilities relating to signaling (e.g., SIP message terminations such as REFER) and media (e.g., coder and transcoding method).

In this interoperability test topology, IP Profiles need to be configured for the following IP entities:

- IP-PBX - to operate in non-secure mode using RTP and UDP
- BroadCloud SIP trunk - to operate in non-secure mode using RTP and UDP

➤ **To configure IP Profile for the IP-PBX:**

1. Open the IP Profile Settings page (**Configuration** tab > **VoIP** > **Coders and Profiles** > **IP Profile Settings**).
2. Click **Add**.
3. Click the **Common** tab, and then configure the parameters as follows:

Parameter	Value
Index	1
Name	IP-PBX (see Note on page 30)

Figure 4-14: Configuring IP Profile for IP-PBX – Common Tab

The screenshot shows a configuration window titled "Edit Row" with a close button (X). Below the title bar, there is a field for "Index" with the value "1". There are four tabs: "Common" (selected), "GW", "SBC Signaling", and "SBC Media". The "Common" tab contains the following parameters and values:

- Name: IP-PBX
- Dynamic Jitter Buffer Minimum Delay [msec]: 10
- Dynamic Jitter Buffer Optimization Factor: 10
- Jitter Buffer Max Delay [msec]: 300
- RTP IP DiffServ: 46
- Signaling DiffServ: 40
- Silence Suppression: Disable
- RTP Redundancy Depth: 0
- Echo Canceler: Line
- Broken Connection Mode: Ignore
- Input Gain (-32 to 31 dB): 0
- Voice Volume (-32 to 31 dB): 0
- Media IP Version: Only IPv4

At the bottom of the window are "Save" and "Cancel" buttons.

4. Click the **SBC Signaling** tab, and then configure the parameters as follows:

Parameter	Value
Remote Update Support	Supported
Remote re-INVITE Support	Supported
Remote REFER Mode	Handle Locally

Figure 4-15: Configuring IP Profile for IP-PBX – SBC Signaling Tab

Edit Row [Close]

Index:

Common | GW | **SBC Signaling** | SBC Media

PRACK Mode:

P-Asserted-Identity Header Mode:

Diversion Header Mode:

History-Info Header Mode:

Session Expires Mode:

Remote Update Support:

Remote re-INVITE:

Remote Delayed Offer Support:

User Registration Time:

NAT UDP Registration Time:

NAT TCP Registration Time:

Remote REFER Mode:

Remote Replaces Mode:

[Save] [Cancel]

- Click the **SBC Media** tab, and then configure the parameters as follows:

Parameter	Value
Media Security Behavior	RTP

Figure 4-16: Configuring IP Profile for IP-PBX – SBC Media Tab

The screenshot shows a configuration window titled "Edit Row" with a close button (X) in the top right corner. Below the title bar, there is an "Index" field containing the number "1". There are four tabs: "Common", "GW", "SBC Signaling", and "SBC Media", with "SBC Media" being the active tab. The configuration parameters and their values are as follows:

- Transcoding Mode: Only If Required
- Extension Coders: None
- Allowed Audio Coders: None
- Allowed Coders Mode: Restriction
- Allowed Video Coders: None
- Allowed Media Types: (empty text box)
- SBC Media Security Mode: RTP
- Media Security Method: SDES
- Enforce MKI Size: Enforce
- SDP Remove Crypto Lifetime: No
- RFC 2833 Mode: As Is
- Alternative DTMF Method: As Is
- RFC 2833 DTMF Payload Type: 0
- Fax Coders: None

At the bottom right of the window, there are "Save" and "Cancel" buttons.

- To configure an IP Profile for the BroadCloud SIP Trunk:
- 1. Click **Add**.
- 2. Click the **Common** tab, and then configure the parameters as follows:

Parameter	Value
Index	2
Profile Name	BroadCloud (see Note on page 30)

Figure 4-17: Configuring IP Profile for BroadCloud SIP Trunk – Common Tab

Edit Row [X]

Index:

Common
 GW
 SBC Signaling
 SBC Media

Name:

Dynamic Jitter Buffer Minimum Delay [msec]:

Dynamic Jitter Buffer Optimization Factor:

Jitter Buffer Max Delay [msec]:

RTP IP DiffServ:

Signaling DiffServ:

Silence Suppression:

RTP Redundancy Depth:

Echo Canceler:

Broken Connection Mode:

Input Gain (-32 to 31 dB):

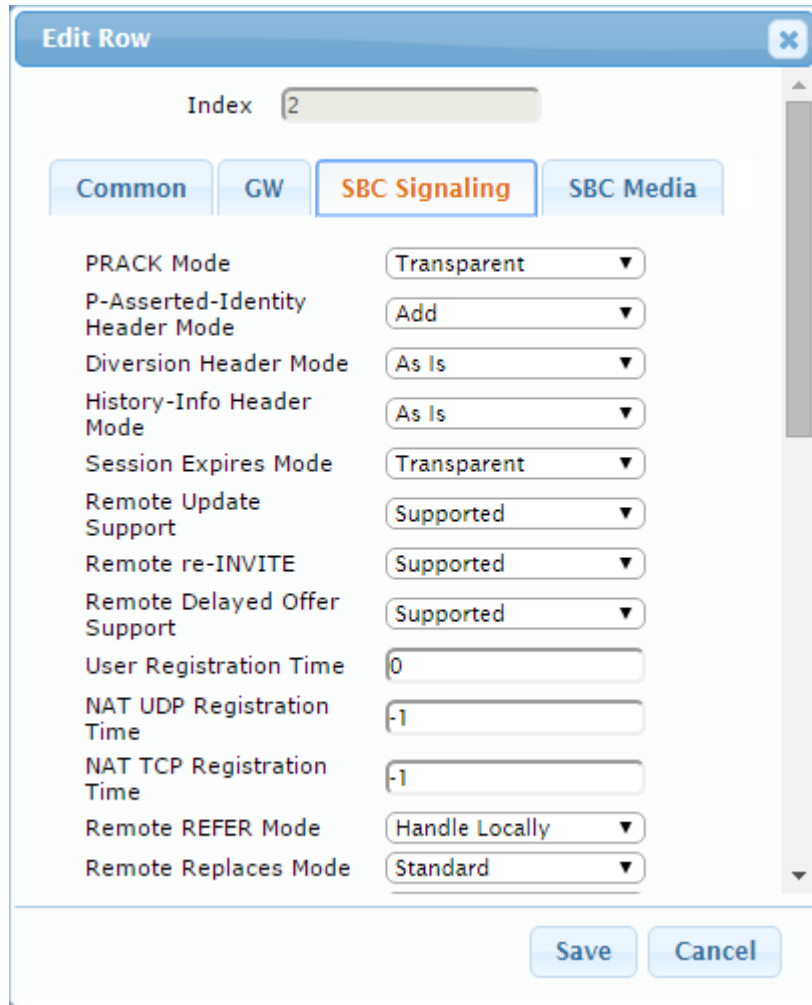
Voice Volume (-32 to 31 dB):

Media IP Version:

- Click the **SBC Signaling** tab, and then configure the parameters as follows:

Parameter	Value
P-Asserted-Identity Header Mode	Add (required for anonymous calls)
Remote REFER Mode	Handle Locally

Figure 4-18: Configuring IP Profile for BroadCloud SIP Trunk – SBC Signaling Tab



4. Click the **SBC Media** tab, and then configure the parameters as follows:

Parameter	Value
Media Security Behavior	RTP

Figure 4-19: Configuring IP Profile for BroadCloud SIP Trunk – SBC Media Tab

Index: 2

Common | GW | SBC Signaling | **SBC Media**

Transcoding Mode: Only If Required ▼

Extension Coders: None ▼

Allowed Audio Coders: None ▼

Allowed Coders Mode: Restriction ▼

Allowed Video Coders: None ▼

Allowed Media Types:

SBC Media Security Mode: RTP ▼

Media Security Method: SDES ▼

Enforce MKI Size: Don't enforce ▼

SDP Remove Crypto Lifetime: No ▼

RFC 2833 Mode: As Is ▼

Alternative DTMF Method: As Is ▼

RFC 2833 DTMF Payload Type: 0

Fax Coders: None ▼

Save Cancel

4.7 Step 7: Configure IP Groups

This step describes how to configure IP Groups. The IP Group represents an IP entity on the network with which the E-SBC communicates. This can be a server (e.g., IP PBX or ITSP) or it can be a group of users (e.g., LAN IP phones). For servers, the IP Group is typically used to define the server's IP address by associating it with a Proxy Set. Once IP Groups are configured, they are used to configure IP-to-IP routing rules for denoting source and destination of the call.

In this interoperability test topology, IP Groups must be configured for the following IP entities:

- IP-PBX located on LAN
- BroadCloud SIP Trunk located on WAN

➤ To configure IP Groups:

1. Open the IP Group Table page (**Configuration** tab > **VoIP** menu > **VoIP Network** > **IP Group Table**).
2. Add an IP Group for the IP-PBX. You can use the default IP Group (Index 0), but modify it as shown below:

Parameter	Value
Index	0
Name	IP-PBX (see Note on page 30)
Type	Server
Proxy Set	IP-PBX
IP Profile	IP-PBX
Media Realm	MRLan
SIP Group Name	172.26.100.170 (according to IP-PBX requirement)

3. Configure an IP Group for the BroadCloud SIP Trunk:

Parameter	Value
Index	1
Name	BroadCloud (see Note on page 30)
Type	Server
Proxy Set	BroadCloud
IP Profile	BroadCloud
Media Realm	MRWan
SIP Group Name	interop.adpt-tech.com (according to ITSP requirement)

The configured IP Groups are shown in the figure below:

Figure 4-20: Configured IP Groups in IP Group Table

Index	Name	SRD	Type	SBC Operation Mode	Proxy Set	IP Profile	Media Realm	SIP Group Name	Classify By Proxy Set	Inbound Message Manipulation Set	Outbound Message Manipulation Set
0	IP-PBX	Default	Server	Not Configured	IP-PBX	IP-PBX	MRLan	172.26.100.170	Enable	-1	-1
1	BroadCloud	Default	Server	Not Configured	BroadCloud	BroadCloud	MRWan	interop.adpt-tech	Enable	-1	4

Page 1 of 1 | 10 | View 1 - 2 of 2

4.8 Step 8: Configure IP-to-IP Call Routing Rules

This step describes how to configure IP-to-IP call routing rules. These rules define the routes for forwarding SIP messages (e.g., INVITE) received from one IP entity to another. The E-SBC selects the rule whose configured input characteristics (e.g., IP Group) match those of the incoming SIP message. If the input characteristics do not match the first rule in the table, they are compared to the second rule, and so on, until a matching rule is located. If no rule is matched, the message is rejected. The routing rules use the configured IP Groups to denote the source and destination of the call. As configured in Section 4.7 on page 34, IP Group 1 represents IP-PBX, and IP Group 2 represents BroadCloud SIP Trunk.

For the interoperability test topology, the following IP-to-IP routing rules need to be configured to route calls between IP-PBX (LAN) and BroadCloud SIP Trunk (WAN):

- Terminate SIP OPTIONS messages on the E-SBC
- Calls from IP-PBX to BroadCloud SIP Trunk
- Calls from BroadCloud SIP Trunk to IP-PBX

➤ **To configure IP-to-IP routing rules:**

1. Open the IP-to-IP Routing Table page (**Configuration** tab > **VoIP** menu > **SBC** > **Routing SBC** > **IP-to-IP Routing Table**).
2. Configure a rule to terminate SIP OPTIONS messages received from the LAN:
 - a. Click **Add**.
 - b. Click the **Rule** tab, and then configure the parameters as follows:

Parameter	Value
Index	0
Name	Terminate OPTIONS (arbitrary descriptive name)
Source IP Group	Any
Request Type	OPTIONS

Figure 4-21: Configuring IP-to-IP Routing Rule for Terminating SIP OPTIONS – Rule Tab

Edit Row
✕

Index

Routing Policy

Rule

Action

Name

Alternative Route Options

Source IP Group

Request Type

Source Username Prefix

Source Host

Destination Username Prefix

Destination Host

Message Condition

Call Trigger

ReRoute IP Group

[Classic View](#)

c. Click the **Action** tab, and then configure the parameters as follows:

Parameter	Value
Destination Type	Dest Address
Destination Address	internal

Figure 4-22: Configuring IP-to-IP Routing Rule for Terminating SIP OPTIONS – Action Tab

3. Configure a rule to route calls from Skype IP-PBX to BroadCloud SIP Trunk:
 - a. Click **Add**.
 - b. Click the **Rule** tab, and then configure the parameters as follows:

Parameter	Value
Index	1
Route Name	IP-PBX to ITSP (arbitrary descriptive name)
Source IP Group	IP-PBX

Figure 4-23: Configuring IP-to-IP Routing Rule for IP-PBX to ITSP – Rule tab

- c. Click the **Action** tab, and then configure the parameters as follows:

Parameter	Value
Destination Type	IP Group
Destination IP Group	BroadCloud
Destination SIP Interface	BroadCloud

Figure 4-24: Configuring IP-to-IP Routing Rule for IP-PBX to ITSP – Action tab

The screenshot shows the 'Edit Row' configuration window for an IP-to-IP Routing Rule. The 'Action' tab is selected. The configuration parameters are as follows:

Parameter	Value
Index	1
Routing Policy	Default_SBCRouting
Destination Type	IP Group
Destination IP Group	BroadCloud
Destination SIP Interface	BroadCloud
Destination Address	
Destination Port	0
Destination Transport Type	
Call Setup Rules Set ID	-1
Group Policy	None
Cost Group	None

Buttons: Save, Cancel. Link: Classic View

4. To configure rule to route calls from BroadCloud SIP Trunk to IP-PBX:
 - a. Click **Add**.
 - b. Click the **Rule** tab, and then configure the parameters as follows:

Parameter	Value
Index	2
Route Name	ITSP to IP-PBX (arbitrary descriptive name)
Source IP Group	BroadCloud

Figure 4-25: Configuring IP-to-IP Routing Rule for ITSP to IP-PBX – Rule tab

Edit Row
✕

Index

Routing Policy

Rule

Action

Name

Alternative Route Options

Source IP Group

Request Type

Source Username Prefix

Source Host

Destination Username Prefix

Destination Host

Message Condition

Call Trigger

ReRoute IP Group

[Classic View](#)

- c. Click the **Action** tab, and then configure the parameters as follows:

Parameter	Value
Destination Type	IP Group
Destination IP Group	IP-PBX
Destination SIP Interface	IP-PBX

Figure 4-26: Configuring IP-to-IP Routing Rule for ITSP to IP-PBX – Action tab

The configured routing rules are shown in the figure below:

Figure 4-27: Configured IP-to-IP Routing Rules in IP-to-IP Routing Table

Index	Name	Routing Policy	Alternative Route Options	Source IP Group	Request Type	Source Username Prefix	Destination Username Prefix	Destination Type	Destination IP Group	Destination SIP Interface	Destination Address
0	Terminate OPTI	Default_SBC	Route Row	Any	OPTIONS	*	*	Dest Address	None	None	internal
1	IP-PBX to ITSP	Default_SBC	Route Row	IP-PBX	All	*	*	IP Group	BroadCloud	BroadCloud	
2	ITSP to IP-PBX	Default_SBC	Route Row	BroadCloud	All	*	*	IP Group	IP-PBX	IP-PBX	



Note: The routing configuration may change according to your specific deployment topology.

4.9 Step 9: Configure IP-to-IP Manipulation Rules

This step describes how to configure IP-to-IP manipulation rules. These rules manipulate the source and / or destination number. The manipulation rules use the configured IP Groups to denote the source and destination of the call. As configured in Section 4.7 on page 34, IP Group 0 represents IP-PBX, and IP Group 1 represents BroadCloud SIP Trunk.



Note: Adapt the manipulation table according to you environment dial plan.

For example, for this interoperability test topology, a manipulation was configured to add the prefix to the destination number for calls from the IP-PBX IP Group to the BroadCloud SIP Trunk IP Group for specific destination username prefix.

➤ **To configure a number manipulation rule:**

1. Open the IP-to-IP Outbound Manipulation page (**Configuration** tab > **VoIP** menu > **SBC > Manipulations SBC > IP-to-IP Outbound**).
2. Click **Add**.
3. Click the **Rule** tab, and then configure the parameters as follows:

Parameter	Value
Index	0
Name	Call to desk
Source IP Group	IP-PBX
Destination IP Group	BroadCloud
Destination Username Prefix	4347

Figure 4-28: Configuring IP-to-IP Outbound Manipulation Rule – Rule Tab

4. Click the **Action** tab, and then configure the parameters as follows:

Parameter	Value
Manipulated Item	Destination URI
Prefix to Add	0119723976

Figure 4-29: Configuring IP-to-IP Outbound Manipulation Rule - Action Tab

5. Click **Submit**.

The figure below shows an example of configured IP-to-IP outbound manipulation rules for calls between IP-PBX IP Group and BroadCloud SIP Trunk IP Group:

Figure 4-30: Example of Configured IP-to-IP Outbound Manipulation Rules

Inde:	Name	Routing Policy	Additio Manipi	Source IP Group	Destinatio IP Group	Source Usernam Prefix	Destinatio Usernam Prefix	Manipul: Item	Remove From Left	Remove From Right	Leave From Right	Prefix to Add	Suffix to Add
0	Call to desk	Default_SI	No	IP-PBX	BroadCloud	*	4347	Destinatio	0	0	255	01197239	
1	Call to mobile	Default_SI	No	IP-PBX	BroadCloud	*	4774	Destinatio	1	0	255	01197254	
2	For Anonymo	Default_SI	No	IP-PBX	BroadCloud	*	*	Source UR	0	0	255		

4.10 Step 10: Configure Message Manipulation Rules

This step describes how to configure SIP message manipulation rules. SIP message manipulation rules can include insertion, removal, and/or modification of SIP headers. Manipulation rules are grouped into Manipulation Sets, enabling you to apply multiple rules to the same SIP message (IP entity).

Once you have configured the SIP message manipulation rules, you need to assign them to the relevant IP Group (in the IP Group table) and determine whether they must be applied to inbound or outbound messages.

➤ **To configure SIP message manipulation rule:**

1. Open the Message Manipulations page (**Configuration** tab > **VoIP** menu > **SIP Definitions** > **Msg Policy & Manipulation** > **Message Manipulations**).
2. Configure a new manipulation rule (Manipulation Set 4) for BroadCloud SIP Trunk. This rule applies to messages sent to the BroadCloud SIP Trunk IP Group. This replaces the host part of the SIP From Header with the value from the SIP To Header.

Parameter	Value
Index	0
Name	Change From host
Manipulation Set ID	4
Message Type	any.request
Action Subject	header.from.url.host
Action Type	Modify
Action Value	header.to.url.host

Figure 4-31: Configuring SIP Message Manipulation Rule 0 (for BroadCloud SIP Trunk)

The screenshot shows a web-based 'Edit Row' dialog box with the following configuration:

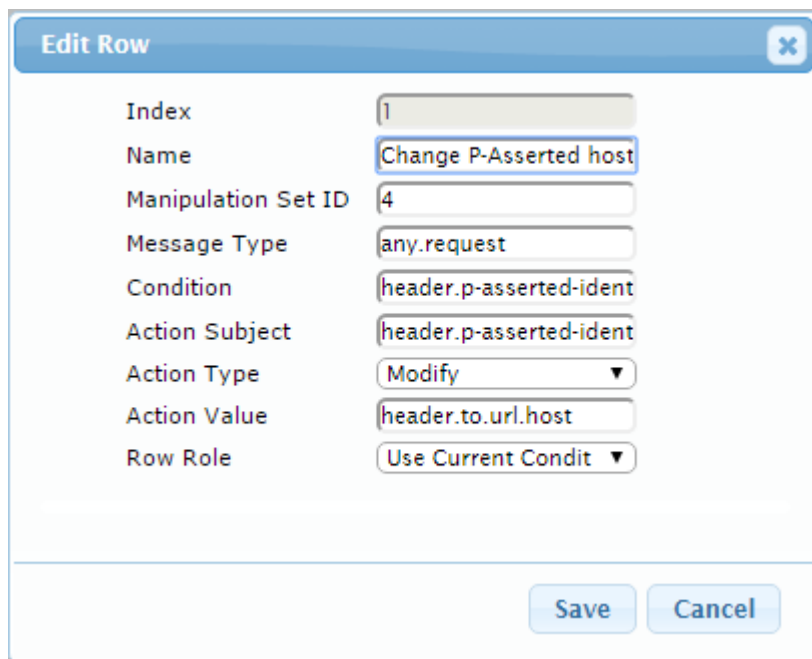
- Index:** 0
- Name:** Change From host
- Manipulation Set ID:** 4
- Message Type:** any.request
- Condition:** (empty field)
- Action Subject:** header.from.url.host
- Action Type:** Modify
- Action Value:** header.to.url.host
- Row Role:** Use Current Condit

Buttons for 'Save' and 'Cancel' are located at the bottom right of the dialog.

- Configure another manipulation rule (Manipulation Set 4) for the BroadCloud SIP Trunk. This rule applies to messages sent to the BroadCloud SIP Trunk IP Group. This replaces the host part of the SIP P-Asserted-Identity Header with the value from the SIP To Header.

Parameter	Value
Index	1
Manipulation Name	Change P-Asserted host
Manipulation Set ID	4
Message Type	any.request
Condition	header.p-asserted-identity exists
Action Subject	header.p-asserted-identity
Action Type	Modify
Action Value	header.to.url.host

Figure 4-32: Configuring SIP Message Manipulation Rule 1 (for BroadCloud SIP Trunk)



Parameter	Value
Index	1
Name	Change P-Asserted host
Manipulation Set ID	4
Message Type	any.request
Condition	header.p-asserted-ident
Action Subject	header.p-asserted-ident
Action Type	Modify
Action Value	header.to.url.host
Row Role	Use Current Condit

- Configure another manipulation rule (Manipulation Set 4) for the BroadCloud SIP Trunk. This rule applies to messages sent to the BroadCloud SIP Trunk IP Group for a Call Forward initiated by the IP-PBX. This replaces the user part of the SIP From Header with the value from the Diversion Header.

Parameter	Value
Index	2
Name	Call Forward
Manipulation Set ID	4
Message Type	any.request
Condition	header.diversion exists

Action Subject	header.from.url.user
Action Type	Modify
Action Value	header.diversion.url.user

Figure 4-33: Configuring SIP Message Manipulation Rule 2 (for BroadCloud SIP Trunk)

5. Configure another manipulation rule (Manipulation Set 4) for the BroadCloud SIP Trunk. This rule applies to messages sent to the BroadCloud SIP Trunk IP Group for a Call Forward initiated by the IP-PBX based on previous rule condition. This replaces the host part of the SIP Diversion Header with the value from the From Header.

Parameter	Value
Index	3
Name	Call Forward
Manipulation Set ID	4
Message Type	
Condition	
Action Subject	header.diversion.url.host
Action Type	Modify
Action Value	header.from.url.host
Row Role	Use Previous Condition

Figure 4-34: Configuring SIP Message Manipulation Rule 3 (for BroadCloud SIP Trunk)

Edit Row
✕

Index	<input type="text" value="3"/>
Name	<input type="text" value="Call Forward"/>
Manipulation Set ID	<input type="text" value="4"/>
Message Type	<input type="text"/>
Condition	<input type="text"/>
Action Subject	<input type="text" value="header.diversion.url.ho:"/>
Action Type	<input type="text" value="Modify"/>
Action Value	<input type="text" value="header.from.url.host"/>
Row Role	<input type="text" value="Use Previous Condi"/>

Figure 4-35: Example of Configured SIP Message Manipulation Rules

Index	Name	Manipulatic Set ID	Message Type	Condition	Action Subject	Action Type	Action Value	Row Role
0	Change From host	4	any.request		header.from.url.ho:	Modify	header.to.url.host	Use Current Cor
1	Change P-Asserted	4	any.request	header.p-asserted-i	header.p-asserted-	Modify	header.to.url.host	Use Current Cor
2	Call Forward	4	any.request	header.diversion ex	header.from.url.us	Modify	header.diversion.ur	Use Current Cor
3	Call Forward	4			header.diversion.ur	Modify	header.from.url.ho:	Use Previous Cc

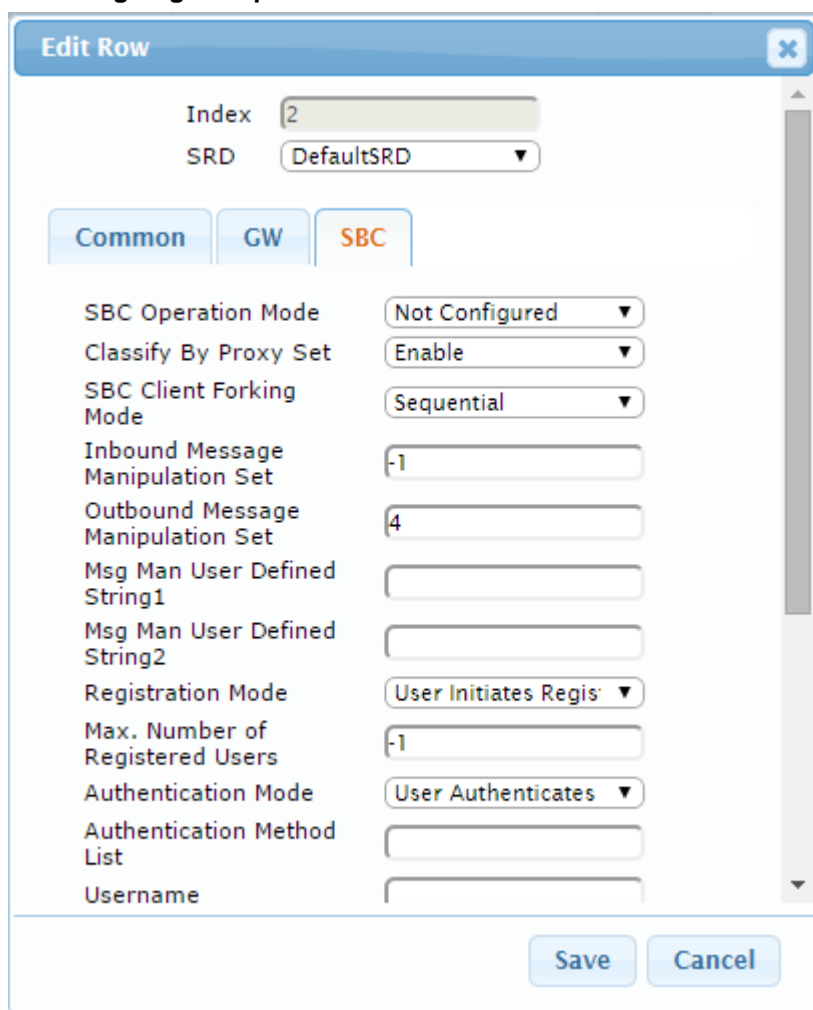
The table displayed below includes SIP message manipulation rules which are bound together by commonality via the Manipulation Set ID 4, which are executed for messages sent to the BroadCloud SIP Trunk IP Group. These rules are specifically required to enable proper interworking between BroadCloud SIP Trunk and IP-PBX. Refer to the *User's Manual* for further details concerning the full capabilities of header manipulation.

Rule Index	Rule Description	Reason for Rule
0	This rule applies to messages sent to the BroadCloud SIP Trunk IP Group. This replaces the host part of the SIP From Header with the value from the SIP To Header.	BroadCloud SIP Trunk requires that all messages should be from known hosts and BroadCloud users.

Rule Index	Rule Description	Reason for Rule
1	This rule applies to messages sent to the BroadCloud SIP Trunk IP Group. This replaces the host part of the SIP P-Asserted-Identity Header with the value from the SIP To Header.	
2	This rule applies to messages sent to the BroadCloud SIP Trunk IP Group for a Call Forward initiated by the IP-PBX. This replaces the user part of the SIP From Header with the value from the Diversion Header.	
3	This rule applies to messages sent to the BroadCloud SIP Trunk IP Group for a Call Forward initiated by the IP-PBX based on a previous rule condition. This replaces the host part of the SIP Diversion Header with the value from the From Header.	

6. Assign Manipulation Set ID 4 to the BroadCloud SIP trunk IP Group:
 - a. Open the IP Group Table page (**Configuration** tab > **VoIP** menu > **VoIP Network** > **IP Group Table**).
 - b. Select the row of the BroadCloud SIP trunk IP Group, and then click **Edit**.
 - c. Click the **SBC** tab.
 - d. Set the 'Outbound Message Manipulation Set' field to 4.

Figure 4-36: Assigning Manipulation Set 4 to the BroadCloud SIP Trunk IP Group



Edit Row

Index:

SRD:

Common | **GW** | **SBC**

SBC Operation Mode:

Classify By Proxy Set:

SBC Client Forking Mode:

Inbound Message Manipulation Set:

Outbound Message Manipulation Set:

Msg Man User Defined String1:

Msg Man User Defined String2:

Registration Mode:

Max. Number of Registered Users:

Authentication Mode:

Authentication Method List:

Username:

- e. Click **Submit**.

4.11 Step 11: Configure Registration Accounts

This step describes how to configure SIP registration accounts. This is required so that the E-SBC can register with the BroadCloud SIP Trunk on behalf of IP-PBX. The BroadCloud SIP Trunk requires registration and authentication to provide service.

In the interoperability test topology, the Served IP Group is IP-PBX IP Group and the Serving IP Group is BroadCloud SIP Trunk IP Group.

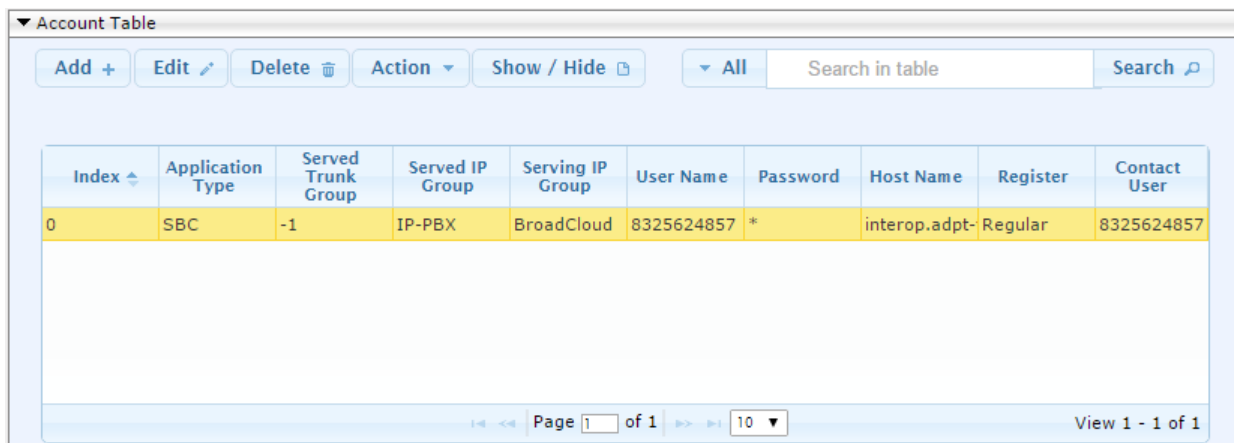
➤ **To configure a registration account:**

1. Open the Account Table page (**Configuration** tab > **VoIP** menu > **SIP Definitions** > **Account Table**).
2. Enter an index number (e.g., "0"), and then click **Add**.
3. Configure the account according to the provided information from , for example:

Parameter	Value
Application Type	SBC
Served IP Group	IP-PBX
Serving IP Group	BroadCloud
Username	As provided by BroadCloud
Password	As provided by BroadCloud
Host Name	interop.adpt-tech.com
Register	Regular
Contact User	8325624857 (pilot number)

4. Click **Apply**.

Figure 4-37: Configuring SIP Registration Account



4.12 Step 12: Miscellaneous Configuration

This section describes miscellaneous E-SBC configuration.

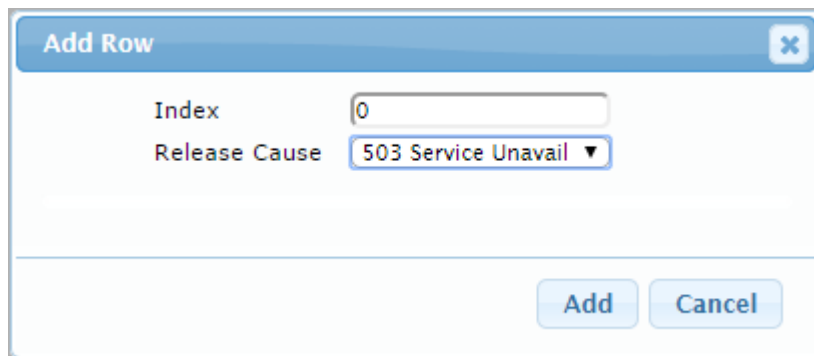
4.12.1 Step 12a: Configure SBC Alternative Routing Reasons

This step describes how to configure the E-SBC's handling of SIP 503 responses received for outgoing SIP dialog-initiating methods, e.g., INVITE, OPTIONS, and SUBSCRIBE messages. In this case E-SBC attempts to locate an alternative route for the call.

➤ **To configure SIP reason codes for alternative IP routing:**

1. Open the SBC Alternative Routing Reasons page (**Configuration** tab > **VoIP** menu > **SBC** > **Routing SBC** > **SBC Alternative Routing Reasons**).
2. Click **Add**; the following dialog box appears:

Figure 4-38: SBC Alternative Routing Reasons Table - Add Record



Add Row	
Index	<input type="text" value="0"/>
Release Cause	<input type="text" value="503 Service Unavail"/>
<input type="button" value="Add"/> <input type="button" value="Cancel"/>	

3. Click **Submit**.

4.13 Step 13: Reset the E-SBC

After you have completed the configuration of the E-SBC described in this chapter, save ("burn") the configuration to the E-SBC's flash memory with a reset for the settings to take effect.

➤ **To save the configuration to flash memory:**

1. Open the Maintenance Actions page (**Maintenance** tab > **Maintenance** menu > **Maintenance Actions**).

Figure 4-39: Resetting the E-SBC

The screenshot displays a web-based configuration interface for the E-SBC. It is organized into three main sections, each with a dropdown arrow on the left:

- Reset Configuration:** This section contains three rows. The first row is 'Reset Board' with a 'Reset' button. The second row is 'Burn To FLASH' with a dropdown menu set to 'Yes'. The third row is 'Graceful Option' with a dropdown menu set to 'No'.
- LOCK / UNLOCK:** This section contains two rows. The first row is 'Lock' with a 'LOCK' button. The second row is 'Graceful Option' with a dropdown menu set to 'No'. Below this section, the text 'Gateway Operational State' is followed by 'UNLOCKED'.
- Save Configuration:** This section contains one row: 'Burn To FLASH' with a 'BURN' button.

2. Ensure that the 'Burn to FLASH' field is set to **Yes** (default).
3. Click the **Reset** button.

This page is intentionally left blank.

A AudioCodes INI File

The *ini* configuration file of the E-SBC, corresponding to the Web-based configuration as described in Section 4 on page 21, is shown below:



Note: To load and save an ini file, use the Configuration File page (**Maintenance** tab > **Software Update** menu > **Configuration File**).

```

;*****
;** Ini File **
;*****

;Board: Mediant 800 E-SBC
;HW Board Type: 69  FK Board Type: 72
;Serial Number: 5916116
;Slot Number: 1
;Software Version: 7.00A.035.012
;DSP Software Version: 5014AE3_R => 700.40
;Board IP Address: 172.26.100.169
;Board Subnet Mask: 255.255.255.0
;Board Default Gateway: 172.26.100.1
;Ram size: 496M  Flash size: 64M  Core speed: 500Mhz
;Num of DSP Cores: 3  Num DSP Channels: 90
;Num of physical LAN ports: 4
;Profile: NONE
;;;Key features:;Board Type: 72 ;Coders: G723 G729 GSM-FR ;Channel Type:
RTP DspCh=90 ;IP Media: VXML ;DATA features: ;DSP Voice features:
IpmDetector V150=50 ;Security: IPSEC MediaEncryption StrongEncryption
EncryptControlProtocol ;E1Trunks=2 ;T1Trunks=2 ;E&M Ports=6 ;BRITrunks=6
;Control Protocols: SIP SASurvivability SBC=100 FEU=600 ;Default
features:;Coders: G711 G726;

;----- HW components-----
;
; Slot # : Module type : # of ports
;-----
;      1 : FALC56      : 1
;      2 : Empty
;      3 : Empty
;-----

[SYSTEM Params]

SyslogServerIP = 172.26.100.116
EnableSyslog = 1
NTPServerUTCOffset = 7200
;VpFileLastUpdateTime is hidden but has non-default value
NTPServerIP = '0.0.0.0'
;LastConfigChangeTime is hidden but has non-default value
;PM_gwINVITEDialogs is hidden but has non-default value
;PM_gwSUBSCRIBEDialogs is hidden but has non-default value
;PM_gwSBCRegisteredUsers is hidden but has non-default value
;PM_gwSBCMediaLegs is hidden but has non-default value

```

```
;PM_gwSBCTranscodingSessions is hidden but has non-default value
```

```
[BSP Params]
```

```
PCMLawSelect = 3  
UdpPortSpacing = 10  
EnterCpuOverloadPercent = 99  
ExitCpuOverloadPercent = 95
```

```
[Analog Params]
```

```
[ControlProtocols Params]
```

```
AdminStateLockControl = 0
```

```
[MGCP Params]
```

```
[MEGACO Params]
```

```
EP_Num_0 = 0  
EP_Num_1 = 1  
EP_Num_2 = 1  
EP_Num_3 = 0  
EP_Num_4 = 0
```

```
[PSTN Params]
```

```
[SS7 Params]
```

```
[Voice Engine Params]
```

```
ENABLEMEDIASECURITY = 1  
CallProgressTonesFilename = 'usa_tones_13.dat'
```

```
[WEB Params]
```

```
UserProductName = 'Mediant 800 E-SBC'  
WebLogoText = 'BroadCloud'  
UseWeblogo = 1  
;UseLogoInWeb is hidden but has non-default value  
UseProductName = 1  
HTTPSCipherString = 'RC4:EXP'  
;HTTPSPkeyFileName is hidden but has non-default value
```

```
[SIP Params]
```

```
MEDIACHANNELS = 30  
GWDEBUGLEVEL = 5  
;ISPRACKREQUIRED is hidden but has non-default value  
ENABLESBCAPPLICATION = 1  
MSLDAPPRIMARYKEY = 'telephoneNumber'  
MEDIACDRREPORTLEVEL = 1
```



```
SBCFORKINGHANDLINGMODE = 1
ENERGYDETECTORCMD = 587202560
ANSWERDETECTORCMD = 10486144
;GWAPPCONFIGURATIONVERSION is hidden but has non-default value

[SCTP Params]

[IPsec Params]

[Audio Staging Params]

[SNMP Params]

[ PhysicalPortsTable ]

FORMAT PhysicalPortsTable_Index = PhysicalPortsTable_Port,
PhysicalPortsTable_Mode, PhysicalPortsTable_SpeedDuplex,
PhysicalPortsTable_PortDescription, PhysicalPortsTable_GroupMember,
PhysicalPortsTable_GroupStatus;
PhysicalPortsTable 0 = "GE_4_1", 1, 4, "User Port #0", "GROUP_1",
"Active";
PhysicalPortsTable 1 = "GE_4_2", 1, 4, "User Port #1", "GROUP_1",
"Redundant";
PhysicalPortsTable 2 = "GE_4_3", 1, 4, "User Port #2", "GROUP_2",
"Active";
PhysicalPortsTable 3 = "GE_4_4", 1, 4, "User Port #3", "GROUP_2",
"Redundant";

[ \PhysicalPortsTable ]

[ EtherGroupTable ]

FORMAT EtherGroupTable_Index = EtherGroupTable_Group,
EtherGroupTable_Mode, EtherGroupTable_Member1, EtherGroupTable_Member2;
EtherGroupTable 0 = "GROUP_1", 2, "GE_4_1", "GE_4_2";
EtherGroupTable 1 = "GROUP_2", 2, "GE_4_3", "GE_4_4";
EtherGroupTable 2 = "GROUP_3", 0, "", "";
EtherGroupTable 3 = "GROUP_4", 0, "", "";

[ \EtherGroupTable ]

[ DeviceTable ]

FORMAT DeviceTable_Index = DeviceTable_VlanID,
DeviceTable_UnderlyingInterface, DeviceTable_DeviceName,
DeviceTable_Tagging;
DeviceTable 0 = 1, "GROUP_1", "vlan 1", 0;
DeviceTable 1 = 2, "GROUP_2", "vlan 2", 0;

[ \DeviceTable ]
```

```

[ InterfaceTable ]

FORMAT InterfaceTable_Index = InterfaceTable_ApplicationTypes,
InterfaceTable_InterfaceMode, InterfaceTable_IPAddress,
InterfaceTable_PrefixLength, InterfaceTable_Gateway,
InterfaceTable_InterfaceName, InterfaceTable_PrimaryDNSServerIPAddress,
InterfaceTable_SecondaryDNSServerIPAddress,
InterfaceTable_UnderlyingDevice;
InterfaceTable 0 = 6, 10, 172.26.100.169, 24, 172.26.100.1, "Voice",
0.0.0.0, 0.0.0.0, "vlan 1";
InterfaceTable 1 = 5, 10, 65.196.9.185, 28, 65.196.9.177, "WANSP",
198.6.1.146, 198.6.1.122, "vlan 2";

[ \InterfaceTable ]

[ DspTemplates ]

;
; *** TABLE DspTemplates ***
; This table contains hidden elements and will not be exposed.
; This table exists on board and will be saved during restarts.
;

[ \DspTemplates ]

[ WebUsers ]

FORMAT WebUsers_Index = WebUsers_Username, WebUsers_Password,
WebUsers_Status, WebUsers_PwAgeInterval, WebUsers_SessionLimit,
WebUsers_SessionTimeout, WebUsers_BlockTime, WebUsers_UserLevel,
WebUsers_PwNonce;
WebUsers 0 = "Admin",
"$1$z/3i5+fh5+Hn5rvq4+vruby+1NDS14XdhYPQ3onZjoYiZPDw8HAxpTCnJvLw8rIxppmZ
WczZ2c+P20xODluOzc=", 1, 0, 2, 15, 60, 200,
"a4e40b4a1ef60fad38601e9bf6d0c1ce";
WebUsers 1 = "User",
"$1$EiUhIXBycnohfit/L3otExUbFkYcFBJMERNJGUwYGVIGV1UFB1VSD18MA1hbDA5ydHdx
CR/Jn15Ln1le38qMWg=", 1, 0, 2, 15, 60, 50,
"a5bdea28146076a2e00cabbb04f2139f";

[ \WebUsers ]

[ TLSContexts ]

FORMAT TLSContexts_Index = TLSContexts_Name, TLSContexts_TLSVersion,
TLSContexts_ServerCipherString, TLSContexts_ClientCipherString,
TLSContexts_OcspEnable, TLSContexts_OcspServerPrimary,
TLSContexts_OcspServerSecondary, TLSContexts_OcspServerPort,
TLSContexts_OcspDefaultResponse;
TLSContexts 0 = "default", 0, "RC4:EXP", "ALL:!ADH", 0, , , 2560, 0;

[ \TLSContexts ]

[ IpProfile ]
    
```

```

FORMAT IpProfile_Index = IpProfile_ProfileName, IpProfile_IpPreference,
IpProfile_CodersGroupID, IpProfile_IsFaxUsed,
IpProfile_JitterBufMinDelay, IpProfile_JitterBufOptFactor,
IpProfile_IPDiffServ, IpProfile_SigIPDiffServ, IpProfile_SCE,
IpProfile_RTPRedundancyDepth, IpProfile_RemoteBaseUDPport,
IpProfile_CNGmode, IpProfile_VxxTransportType, IpProfile_NSEMode,
IpProfile_IsDTMFUsed, IpProfile_PlayRBTone2IP,
IpProfile_EnableEarlyMedia, IpProfile_ProgressIndicator2IP,
IpProfile_EnableEchoCanceller, IpProfile_CopyDest2RedirectNumber,
IpProfile_MediaSecurityBehaviour, IpProfile_CallLimit,
IpProfile_DisconnectOnBrokenConnection, IpProfile_FirstTxDtmfOption,
IpProfile_SecondTxDtmfOption, IpProfile_RxDTMFOption,
IpProfile_EnableHold, IpProfile_InputGain, IpProfile_VoiceVolume,
IpProfile_AddIEInSetup, IpProfile_SBCExtensionCodersGroupID,
IpProfile_MediaIPVersionPreference, IpProfile_TranscodingMode,
IpProfile_SBCAllowedMediaTypes, IpProfile_SBCAllowedCodersGroupID,
IpProfile_SBCAllowedVideoCodersGroupID, IpProfile_SBCAllowedCodersMode,
IpProfile_SBCMediaSecurityBehaviour, IpProfile_SBCRFC2833Behavior,
IpProfile_SBCAlternativeDTMFMethod, IpProfile_SBCAssertIdentity,
IpProfile_AMDSensitivityParameterSuit, IpProfile_AMDSensitivityLevel,
IpProfile_AMDMaxGreetingTime, IpProfile_AMDMaxPostSilenceGreetingTime,
IpProfile_SBCDiversionsMode, IpProfile_SBCHistoryInfoMode,
IpProfile_EnableQSIGTunneling, IpProfile_SBCFaxCodersGroupID,
IpProfile_SBCFaxBehavior, IpProfile_SBCFaxOfferMode,
IpProfile_SBCFaxAnswerMode, IpProfile_SbcPrackMode,
IpProfile_SBCSessionExpiresMode, IpProfile_SBCRemoteUpdateSupport,
IpProfile_SBCRemoteReinviteSupport,
IpProfile_SBCRemoteDelayedOfferSupport, IpProfile_SBCRemoteReferBehavior,
IpProfile_SBCRemote3xxBehavior, IpProfile_SBCRemoteMultiple18xSupport,
IpProfile_SBCRemoteEarlyMediaResponseType,
IpProfile_SBCRemoteEarlyMediaSupport, IpProfile_EnableSymmetricMKI,
IpProfile_MKISize, IpProfile_SBCEnforceMKISize,
IpProfile_SBCRemoteEarlyMediaRTP, IpProfile_SBCRemoteSupportsRFC3960,
IpProfile_SBCRemoteCanPlayRingback, IpProfile_EnableEarlyl183,
IpProfile_EarlyAnswerTimeout, IpProfile_SBC2833DTMFPayloadType,
IpProfile_SBCUserRegistrationTime, IpProfile_ResetSRTPStateUponRekey,
IpProfile_AmdMode, IpProfile_SBCReliableHeldToneSource,
IpProfile_GenerateSRTPKeys, IpProfile_SBCPlayHeldTone,
IpProfile_SBCRemoteHoldFormat, IpProfile_SBCRemoteReplacesBehavior,
IpProfile_SBCSDPptimeAnswer, IpProfile_SBCPreferredPTime,
IpProfile_SBCUseSilenceSupp, IpProfile_SBCRTPRedundancyBehavior,
IpProfile_SBCPlayRBTToTransferee, IpProfile_SBCRTPMode,
IpProfile_SBCJitterCompensation,
IpProfile_SBCRemoteRenegotiateOnFaxDetection,
IpProfile_JitterBufMaxDelay,
IpProfile_SBCUserBehindUdpNATRegistrationTime,
IpProfile_SBCUserBehindTcpNATRegistrationTime,
IpProfile_SBCSDPHandlerTCPAttribute,
IpProfile_SBCRemoveCryptoLifetimeInSDP, IpProfile_SBCIceMode,
IpProfile_SBCRTPMux, IpProfile_SBCMediaSecurityMethod,
IpProfile_SBCHandleXDetect, IpProfile_SBCRTPFeedback,
IpProfile_SBCRemoteRepresentationMode, IpProfile_SBCKeepVIAHeaders,
IpProfile_SBCKeepRoutingHeaders, IpProfile_SBCKeepUserAgentHeader,
IpProfile_SBCRemoteMultipleEarlyDialogs,
IpProfile_SBCRemoteMultipleAnswersMode, IpProfile_SBCDirectMediaTag,
IpProfile_SBCAdaptRFC2833BWTtoVoiceCoderBW;
IpProfile 1 = "IP-PBX", 1, 0, 0, 10, 10, 46, 40, 0, 0, 0, 0, 2, 0, 0, 0,
0, -1, 1, 0, 0, -1, 0, 4, -1, 1, 1, 0, 0, "", -1, 0, 0, "", -1, -1, 0, 2,
0, 0, 0, 0, 8, 300, 400, 0, 0, 0, -1, 0, 0, 1, 3, 0, 2, 2, 1, 3, 0, 1, 0,
1, 0, 0, 0, 0, 0, 1, 0, 0, 0, 0, 0, 0, 1, 0, 0, 0, 0, 0, 0, 0, 0, 0,
0, 0, 300, -1, -1, 0, 0, 0, 0, 0, 0, -1, -1, -1, -1, -1, 0, "", 0;
IpProfile 2 = "BroadCloud", 1, 0, 0, 10, 10, 46, 40, 0, 0, 0, 0, 2, 0, 0,
0, 0, -1, 1, 0, 0, -1, 0, 4, -1, 1, 1, 0, 0, "", -1, 0, 0, "", -1, -1, 0,
2, 0, 0, 1, 0, 8, 300, 400, 0, 0, 0, -1, 0, 0, 1, 3, 0, 2, 2, 1, 3, 0, 1,
0, 1, 0, 0, 0, 0, 0, 1, 0, 0, 0, 0, 0, 0, 1, 0, 0, 0, 0, 0, 0, 0, 0,
0, 0, 0, 300, -1, -1, 0, 0, 0, 0, 0, 0, 0, -1, -1, -1, -1, -1, 0, "", 0;

```

```

[ \IpProfile ]

[ CpMediaRealm ]

FORMAT CpMediaRealm_Index = CpMediaRealm_MediaRealmName,
CpMediaRealm_IPv4IF, CpMediaRealm_IPv6IF, CpMediaRealm_PortRangeStart,
CpMediaRealm_MediaSessionLeg, CpMediaRealm_PortRangeEnd,
CpMediaRealm_IsDefault, CpMediaRealm_QoeProfile, CpMediaRealm_BWProfile;
CpMediaRealm 0 = "MRLan", "Voice", "", 6000, 100, 6990, 1, "", "";
CpMediaRealm 1 = "MRWan", "WANSP", "", 7000, 100, 7990, 0, "", "";

[ \CpMediaRealm ]

[ SBCRoutingPolicy ]

FORMAT SBCRoutingPolicy_Index = SBCRoutingPolicy_Name,
SBCRoutingPolicy_LCREnable, SBCRoutingPolicy_LCRAverageCallLength,
SBCRoutingPolicy_LCRDefaultCost, SBCRoutingPolicy_LdapServerGroupName;
SBCRoutingPolicy 0 = "Default_SBCRoutingPolicy", 0, 1, 1, "";

[ \SBCRoutingPolicy ]

[ SRD ]

FORMAT SRD_Index = SRD_Name, SRD_BlockUnRegUsers, SRD_MaxNumOfRegUsers,
SRD_EnableUnAuthenticatedRegistrations, SRD_SharingPolicy,
SRD_UsedByRoutingServer, SRD_SBCOperationMode,
SRD_SBCRegisteredUsersClassificationMethod, SRD_SBCRoutingPolicyName;
SRD 0 = "DefaultSRD", 0, -1, 1, 0, 0, 0, -1, "Default_SBCRoutingPolicy";

[ \SRD ]

[ SIPInterface ]

FORMAT SIPInterface_Index = SIPInterface_InterfaceName,
SIPInterface_NetworkInterface, SIPInterface_ApplicationType,
SIPInterface_UDPPort, SIPInterface_TCPPort, SIPInterface_TLSPort,
SIPInterface_SRDName, SIPInterface_MessagePolicyName,
SIPInterface_TLSContext, SIPInterface_TLSMutualAuthentication,
SIPInterface_TCPKeepaliveEnable,
SIPInterface_ClassificationFailureResponseType,
SIPInterface_PreClassificationManSet, SIPInterface_EncapsulatingProtocol,
SIPInterface_MediaRealm, SIPInterface_SBCDirectMedia,
SIPInterface_BlockUnRegUsers, SIPInterface_MaxNumOfRegUsers,
SIPInterface_EnableUnAuthenticatedRegistrations,
SIPInterface_UsedByRoutingServer;
SIPInterface 0 = "IP-PBX", "Voice", 2, 5060, 5060, 0, "DefaultSRD", "",
"", -1, 0, 500, -1, 0, "MRLan", 0, -1, -1, -1, 0;
SIPInterface 1 = "BroadCloud", "WANSP", 2, 5060, 0, 0, "DefaultSRD", "",
"", -1, 0, 500, -1, 0, "MRWan", 0, -1, -1, -1, 0;

[ \SIPInterface ]

[ ProxySet ]
    
```

```

FORMAT ProxySet_Index = ProxySet_ProxyName,
ProxySet_EnableProxyKeepAlive, ProxySet_ProxyKeepAliveTime,
ProxySet_ProxyLoadBalancingMethod, ProxySet_IsProxyHotSwap,
ProxySet_SRDName, ProxySet_ClassificationInput, ProxySet_TLSContextName,
ProxySet_ProxyRedundancyMode, ProxySet_DNSResolveMethod,
ProxySet_KeepAliveFailureResp, ProxySet_GWIPv4SIPInterfaceName,
ProxySet_SBCIPv4SIPInterfaceName, ProxySet_SASIPv4SIPInterfaceName,
ProxySet_GWIPv6SIPInterfaceName, ProxySet_SBCIPv6SIPInterfaceName,
ProxySet_SASIPv6SIPInterfaceName;
ProxySet 0 = "IP-PBX", 1, 60, 0, 0, "DefaultSRD", 0, "", -1, -1, "", "",
"IP-PBX", "", "", "", "", "";
ProxySet 1 = "BroadCloud", 1, 60, 0, 0, "DefaultSRD", 0, "", -1, 1, "",
"", "BroadCloud", "", "", "", "", "";

[ \ProxySet ]

[ IPGroup ]

FORMAT IPGroup_Index = IPGroup_Type, IPGroup_Name, IPGroup_ProxySetName,
IPGroup_SIPGroupName, IPGroup_ContactUser, IPGroup_SipReRoutingMode,
IPGroup_AlwaysUseRouteTable, IPGroup_SRDName, IPGroup_MediaRealm,
IPGroup_ClassifyByProxySet, IPGroup_ProfileName,
IPGroup_MaxNumOfRegUsers, IPGroup_InboundManSet, IPGroup_OutboundManSet,
IPGroup_RegistrationMode, IPGroup_AuthenticationMode, IPGroup_MethodList,
IPGroup_EnableSBCClientForking, IPGroup_SourceUriInput,
IPGroup_DestUriInput, IPGroup_ContactName, IPGroup_Username,
IPGroup_Password, IPGroup_UUIFormat, IPGroup_QOEProfile,
IPGroup_BWProfile, IPGroup_MediaEnhancementProfile,
IPGroup_AlwaysUseSourceAddr, IPGroup_MsgManUserDef1,
IPGroup_MsgManUserDef2, IPGroup_SIPConnect, IPGroup_SBCPSAPMode,
IPGroup_DTLSContext, IPGroup_CreatedByRoutingServer,
IPGroup_UsedByRoutingServer, IPGroup_SBCOperationMode,
IPGroup_SBCRouteUsingRequestURIPort;
IPGroup 0 = 0, "IP-PBX", "IP-PBX", "172.26.100.170", "", -1, 0,
"DefaultSRD", "MRLan", 1, "IP-PBX", -1, -1, -1, 0, 0, "", 0, -1, -1, "",
"", "$1$gQ==", 0, "", "", "", 0, "", "", 0, 0, "", 0, 0, -1, 0;
IPGroup 1 = 0, "BroadCloud", "BroadCloud", "interop.adpt-tech.com", "", -
1, 0, "DefaultSRD", "MRWan", 1, "BroadCloud", -1, -1, 4, 0, 0, "", 0, -1,
-1, "", "", "$1$gQ==", 0, "", "", "", 0, "", "", 0, 0, "", 0, 0, -1, 0;

[ \IPGroup ]

[ SBCAlternativeRoutingReasons ]

FORMAT SBCAlternativeRoutingReasons_Index =
SBCAlternativeRoutingReasons_ReleaseCause;
SBCAlternativeRoutingReasons 0 = 503;

[ \SBCAlternativeRoutingReasons ]

[ ProxyIp ]

FORMAT ProxyIp_Index = ProxyIp_ProxySetId, ProxyIp_ProxyIpIndex,
ProxyIp_IpAddress, ProxyIp_TransportType;
ProxyIp 0 = "0", 0, "172.26.100.170:5060", 0;
ProxyIp 1 = "1", 0, "nn6300southsipconnect.adpt-tech.com", 0;

[ \ProxyIp ]

```

```

[ Account ]

FORMAT Account_Index = Account_ServedTrunkGroup,
Account_ServedIPGroupName, Account_ServingIPGroupName, Account_Username,
Account_Password, Account_HostName, Account_Register,
Account_ContactUser, Account_ApplicationType;
Account 0 = -1, "IP-PBX", "BroadCloud", "8325624857",
"$1$SSg/LyUiDSA0NCFhZGRj", "interop.adpt-tech.com", 1, "8325624857", 2;

[ \Account ]

[ IP2IPRouting ]

FORMAT IP2IPRouting_Index = IP2IPRouting_RouteName,
IP2IPRouting_RoutingPolicyName, IP2IPRouting_SrcIPGroupName,
IP2IPRouting_SrcUsernamePrefix, IP2IPRouting_SrcHost,
IP2IPRouting_DestUsernamePrefix, IP2IPRouting_DestHost,
IP2IPRouting_RequestType, IP2IPRouting_MessageConditionName,
IP2IPRouting_ReRouteIPGroupName, IP2IPRouting_Trigger,
IP2IPRouting_CallSetupRulesSetId, IP2IPRouting_DestType,
IP2IPRouting_DestIPGroupName, IP2IPRouting_DestSIPInterfaceName,
IP2IPRouting_DestAddress, IP2IPRouting_DestPort,
IP2IPRouting_DestTransportType, IP2IPRouting_AltRouteOptions,
IP2IPRouting_GroupPolicy, IP2IPRouting_CostGroup;
IP2IPRouting 0 = "Terminate OPTIONS", "Default_SBCRoutingPolicy", "Any",
"*, ", " ", " ", " ", " ", 6, " ", "Any", 0, -1, 1, " ", " ", "internal", 0, -1, 0,
0, " ";
IP2IPRouting 1 = "IP-PBX to ITSP", "Default_SBCRoutingPolicy", "IP-PBX",
"*, ", " ", " ", " ", " ", 0, " ", "Any", 0, -1, 0, "BroadCloud", "BroadCloud",
" ", 0, -1, 0, 0, " ";
IP2IPRouting 2 = "ITSP to IP-PBX", "Default_SBCRoutingPolicy",
"BroadCloud", " ", " ", " ", " ", " ", 0, " ", "Any", 0, -1, 0, "IP-PBX", "IP-
PBX", " ", 0, -1, 0, 0, " ";

[ \IP2IPRouting ]

[ IPOutboundManipulation ]

FORMAT IPOutboundManipulation_Index =
IPOutboundManipulation_ManipulationName,
IPOutboundManipulation_RoutingPolicyName,
IPOutboundManipulation_IsAdditionalManipulation,
IPOutboundManipulation_SrcIPGroupName,
IPOutboundManipulation_DestIPGroupName,
IPOutboundManipulation_SrcUsernamePrefix, IPOutboundManipulation_SrcHost,
IPOutboundManipulation_DestUsernamePrefix,
IPOutboundManipulation_DestHost,
IPOutboundManipulation_CallingNamePrefix,
IPOutboundManipulation_MessageConditionName,
IPOutboundManipulation_RequestType,
IPOutboundManipulation_ReRouteIPGroupName,
IPOutboundManipulation_Trigger, IPOutboundManipulation_ManipulatedURI,
IPOutboundManipulation_RemoveFromLeft,
IPOutboundManipulation_RemoveFromRight,
IPOutboundManipulation_LeaveFromRight, IPOutboundManipulation_Prefix2Add,
IPOutboundManipulation_Suffix2Add,
IPOutboundManipulation_PrivacyRestrictionMode;
    
```

```

IPOutboundManipulation 0 = "Call to desk", "Default_SBCRoutingPolicy", 0,
"IP-PBX", "BroadCloud", "*", "*", "4347", "*", "*", "", 0, "Any", 0, 1,
0, 0, 255, "0119723976", "", 0;
IPOutboundManipulation 1 = "Call to mobile", "Default_SBCRoutingPolicy",
0, "IP-PBX", "BroadCloud", "*", "*", "4774", "*", "*", "", 0, "Any", 0,
1, 1, 0, 255, "011972546262", "", 0;
IPOutboundManipulation 2 = "For Anonymous", "Default_SBCRoutingPolicy",
0, "IP-PBX", "BroadCloud", "*", "*", "*", "*", "*", "", 0, "Any", 0, 0,
0, 0, 255, "", "", 0;

[ \IPOutboundManipulation ]

[ CodersGroup0 ]

FORMAT CodersGroup0_Index = CodersGroup0_Name, CodersGroup0_pTime,
CodersGroup0_rate, CodersGroup0_PayloadType, CodersGroup0_Sce,
CodersGroup0_CoderSpecific;
CodersGroup0 0 = "g711Alaw64k", 20, 255, -1, 0, "";

[ \CodersGroup0 ]

[ MessageManipulations ]

FORMAT MessageManipulations_Index =
MessageManipulations_ManipulationName, MessageManipulations_ManSetID,
MessageManipulations_MessageType, MessageManipulations_Condition,
MessageManipulations_ActionSubject, MessageManipulations_ActionType,
MessageManipulations_ActionValue, MessageManipulations_RowRole;
MessageManipulations 0 = "Change From host", 4, "any.request", "",
"header.from.url.host", 2, "header.to.url.host", 0;
MessageManipulations 1 = "Change P-Asserted host", 4, "any.request",
"header.p-asserted-identity exists", "header.p-asserted-
identity.url.host", 2, "header.to.url.host", 0;
MessageManipulations 2 = "Call Forward", 4, "any.request",
"header.diversion exists", "header.from.url.user", 2,
"header.diversion.url.user", 0;
MessageManipulations 3 = "Call Forward", 4, "", "",
"header.diversion.url.host", 2, "header.from.url.host", 1;

[ \MessageManipulations ]

[ RoutingRuleGroups ]

;
; *** TABLE RoutingRuleGroups ***
; This table contains hidden elements and will not be exposed.
; This table exists on board and will be saved during restarts.
;

[ \RoutingRuleGroups ]

[ GwRoutingPolicy ]

FORMAT GwRoutingPolicy_Index = GwRoutingPolicy_Name,
GwRoutingPolicy_LCREnable, GwRoutingPolicy_LCRAverageCallLength,
GwRoutingPolicy_LCRDefaultCost, GwRoutingPolicy_LdapServerGroupName;

```

```
GwRoutingPolicy 0 = "GwRoutingPolicy", 0, 1, 1, "";  
  
[ \GwRoutingPolicy ]  
  
[ ResourcePriorityNetworkDomains ]  
  
FORMAT ResourcePriorityNetworkDomains_Index =  
ResourcePriorityNetworkDomains_Name,  
ResourcePriorityNetworkDomains_Ip2TelInterworking;  
ResourcePriorityNetworkDomains 1 = "dsn", 1;  
ResourcePriorityNetworkDomains 2 = "dod", 1;  
ResourcePriorityNetworkDomains 3 = "drsn", 1;  
ResourcePriorityNetworkDomains 5 = "uc", 1;  
ResourcePriorityNetworkDomains 7 = "cuc", 1;  
  
[ \ResourcePriorityNetworkDomains ]
```


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