AudioCodes Mediant™ Family of Media Gateways & Session Border Controllers

# Connecting AudioCodes' SBC to Microsoft Teams Direct Routing with Local Media Optimization

**Enterprise Model** 

Version 7.2





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# **Abbreviations and Terminology**

Each abbreviation, unless widely used, is spelled out in full when first used.



#### **Related Documentation**

Document Name
Mediant 500 Gateway & E-SBC User's Manual
Mediant 500L Gateway & E-SBC User's Manual
Mediant 800 Gateway & E-SBC User's Manual
Mediant 1000B Gateway & E-SBC User's Manual
Mediant 2600 SBC User's Manual
Mediant 4000 SBC User's Manual
Mediant 9000 SBC User's Manual
Mediant Software SBC User's Manual
Gateway and SBC CLI Reference Guide
SIP Message Manipulation Reference Guide
AudioCodes Configuration Notes

### **Document Revision Record**

LTRT	Description
33450	All information related to Local Media Optimization was removed from document 'Connecting AudioCodes' SBC to Microsoft Teams Direct Routing Enterprise Model' and included in this document. Added Appendix "AudioCodes ARM and SBCs with Teams Direct Local Media Optimization"

#### **Documentation Feedback**

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Configuration Note 1. Introduction

# 1 Introduction

This document describes how to connect AudioCodes' SBC to Teams Direct Routing with Local Media Optimization and refers to the AudioCodes SBC configuration only. For configuring the Office 365 side, please refer to <a href="https://docs.microsoft.com/en-us/microsoftteams/direct-routing-configure">https://docs.microsoft.com/en-us/microsoftteams/direct-routing-configure</a>.

This document is intended for IT or telephony professionals.



**Note:** To zoom in on screenshots of example Web interface configurations, press **Ctrl** and **+**.

# 1.1 About Teams Direct Routing

Teams Direct Routing allows connecting a customer-provided SBC to Microsoft Phone System. The customer-provided SBC can be connected to almost any telephony trunk or connect with third-party PSTN equipment. The connection allows:

- Using virtually any PSTN trunk with Microsoft Phone System
- Configuring interoperability between customer-owned telephony equipment, such as third-party PBXs, analog devices, and Microsoft Phone System

#### 1.2 Validated AudioCodes Version

Microsoft has successfully conducted validation tests with AudioCodes' Mediant SBC Ver. 7.20A.258. Previous firmware versions may run successfully; however, Microsoft did not test such versions. For an updated list, refer to <u>List of Session Border Controllers certified for Direct Routing</u>. Note the following:

- Validate that you have the correct License key. Refer to AudioCodes' device's *User's Manual* for more information on how to view the device's License Key including licensed features and capacity. If you don't have the correct License key, contact your AudioCodes representative to obtain one.
- The main AudioCodes licenses required by the SBC are as follows:
  - SW/TEAMS
  - Number of SBC sessions [Based on requirements]
  - Transcoding sessions [If media transcoding is needed]

# 1.3 About AudioCodes SBC Product Series

AudioCodes' family of SBC devices enables reliable connectivity and security between the enterprise's VoIP network and the service provider's VoIP network.

The 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 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' 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. The SBC can be offered as a Virtualized SBC, supporting the following platforms: Hyper-V, AWS, AZURE, AWP, KVM and VMWare.

# 1.4 Infrastructure Prerequisites

The table below shows the list of infrastructure prerequisites for deploying Direct Routing.

**Table 1-1: Infrastructure Prerequisites** 

Infrastructure Prerequisite	Details			
Certified Session Border Controller (SBC)				
SIP Trunks connected to the SBC				
Office 365 Enterprise tenant				
Domains				
Public IP address for the SBC				
Fully Qualified Domain Name (FQDN) for the SBC	See Microsoft's Plan Direct Pouting			
Public DNS entry for the SBC	See Microsoft's <u>Plan Direct Routing</u> document.			
Public trusted certificate for the SBC				
Firewall ports for Direct Routing signaling				
Firewall IP addresses and ports for Direct Routing media				
Media Transport Profile				
Firewall ports for client media				

# 2 Direct Routing Local Media Optimization

This chapter describes the Direct Routing Local Media Optimization Routing between Microsoft Phone System (Cloud PBX) and SBC devices.

#### Notes:



- The implementation of this feature is only relevant for customers with site topology requiring Local Media Optimization solution
- SIP Signaling is always routed via the Microsoft Phone System Cloud PBX
- For Quick guidelines, see Appendix C "Configuration Quick Guidelines".

#### 2.1 Introduction

The SBC supports the capability to optimize media flow between the Microsoft Phone System (Cloud PBX) and Direct Route SBC devices. It implements network policies for media traffic control flows paths between the Teams clients and the SBC devices for PSTN termination.

Enterprises consider PSTN voice as a business-critical application with high emphasis on voice quality. Media Path Optimization in Media Bypass mode for Direct Routing helps to better manage voice quality by enabling enterprises to do the following:

- Control how the media traffic flows between the Teams clients and customer SBCs;
- Allowing media streams between the Teams clients and SBCs even if SBCs are behind the corporate firewalls with private IPs and not directly visible to Microsoft.

By default, media bypass (referred to as Direct Media by the AudioCodes SBC application) is configured per SIP interface or per SBC device by the parameter Microsoft Teams PowerShell configured parameter MediaBypass (True or False). When enabled, media is routed directly between the Teams user and the SBC, bypassing the Microsoft Phone System Cloud PBX Media Relay or Media Proxy, on the condition that the client and the SBC media interface can establish a routed connection (verified during ICE negotiation).

Affectively this means that traffic does not need to route through an unnecessary loop. For example, the Teams user is in the same building and/or network as the SBC (the Teams client is inside the corporate network and has access to the Internal IP address of the SBC). Alternatively, if the Teams user is outside the corporate network and cannot reach the internal IP address of the SBC, then RTP media needs to pass via the Microsoft Phone System Cloud PBX.

The new functionality of Local Media Optimization uses an additional capability for the location of the Teams user device (for the inbound or the outgoing call). In other words, the SBC offers the correct interface for the media based on the user device location.

The handling is based on supplementary SIP headers supplied by Microsoft Teams HUB:

- **X-MS-UserLocation:** Indicates whether the Teams user is inside or outside the corporate network.
- **X-MS-MediaPath:** Indicates the FQDN of the SBC devices in the network that the call must traverse.
- **X-MS-UserSite**: Indicates the name of the network site



This case applies for the following topologies:

- A distributed mode (central SBC with remote branches with local breakouts)
- A single SBC in a corporate DMZ with two media interfaces (external and internal)

See detailed descriptions of these call scenarios below in Section 2.2.

# 2.2 Typical Call Scenarios

The following describes call scenarios that are implemented using this feature:

- When the destination SBC is the paired SBC (the call breaks out to the SIP Trunk which is connected to this SBC), determines whether the Teams user is calling from inside (internal) or outside the corporate network (external). If the Teams user is calling from inside the corporate network, RTP media flows via the SBC's internal media interface (internal media realm). If the Teams user is calling from outside the corporate network, media flows via the SBC's external media interface (regular media realm).
- When the destination SBC is not the paired SBC (in a distributed topology), determines whether RTP media should traverse it or directly terminate to the remote SBC. For example, when the Teams user is calling from inside the corporate network, located in the same branch as the remote SBC.

The paired SBC serves as a proxy SBC for the downstream (remote) SBCs in the network, which are not directly connected to the Microsoft Phone System; however, are declared in Teams via a new PowerShell command (see below) to the Direct Routing interface. The downstream SBCs are configured on Microsoft Teams with Voice Routes.

#### 2.2.1 Implemented Scenarios

This section describes the implemented scenarios.

#### 2.2.1.1 Central SBC Scenario

In this scenario, all trunks are centralized with media flowing between the central SBC (Site HQ) and the users, based on the user's location. If user is internal, media flows between the internal IP of the central SBC (Site HQ) and the Teams client. If user is external, media flows between the external IP of the SBC and Teams client.

In this example, the administrator is paired to a single SBC (sbc4.contoso.com) to the service, the SBC has a centralized trunk connected to it. When the user is in the internal network, the SBC provides the internal IP of the SBC for media, when the user is outside the corporate network, the SBC provides the external (public) IP of the SBC.

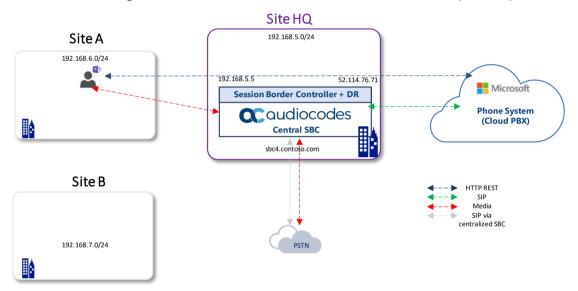
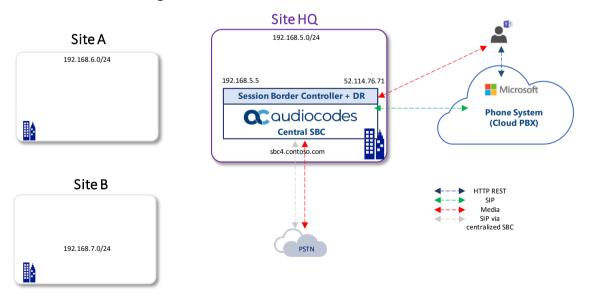


Figure 2-1: Central SBC Traffic Flow - User at "Home" (Internal)







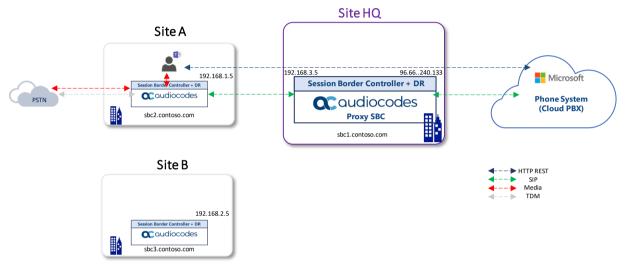
#### 2.2.1.2 Proxy SBC Scenario

For this scenario, the administrator is paired only to a single SBC (sbc1.contoso.com) also referred to as the **Proxy SBC** to the Direct Routing service.

The administrator adds the downstream SBCs using PowerShell command *New-CSOnlinePSTNGateway* (or Via the UMP365 Online Voice Routing), indicating that they can be reached via the proxy SBC. The downstream SBC does not have a WAN interface (it is not configured with a public IP address), however can be assigned to voice routes.

When a user is in an office where the downstream SBC is, the media traffic flows between the user and the SBC directly.

Figure 2-3: Proxy SBC Traffic Flow - user at "home" (Internal)



When a user is outside of the office (on a public internet or in a different office) the media flows from the user to the public IP of the Proxy SBC, which proxies it to the downstream SBC(s).

Site HQ 192.168.3.0/24 Site A 192.168.1.0/24 Microsoft Session Border Controller + DR on Border Controller + DR audiocodes PSTN Phone System Caudiocodes (Cloud PBX) sbc2.contoso.com **Proxy SBC** sbc1.contoso.com Site B 192.168.2.5 LAN Media shr3 contoso com

Figure 2-4: Proxy SBC Traffic Flow - user is external

#### 2.2.1.3 Local Media Optimization Modes

Media Path Optimization technology in the Microsoft Teams network consists of two modes:

- Always Bypass: In the case where the Teams client is internal, the local media candidates of the target SBC will always be offered to the clients.
- OnlyForLocalUsers: The local media candidates of the target SBC is offered only if a user is in the same location as the SBC. For all other cases, either the local or external IP of the proxy SBC is offered.

# 2.3 Online PSTN Gateway Configuration

This section describes the Online PSTN Gateway configuration.

# 2.3.1 Online PSTN Gateway Configuration (Office 365) - Proxy SBC Scenario

Run the following PowerShell commands on the Office 365, in order to configure the Proxy SBC PSTN Gateway on Teams Direct Routing:

**New-CsOnlinePSTNGateway** -Identity **sbc1.contoso.com** -SipSignallingPort **5068** - ForwardCallHistory \$True -ForwardPai \$True -MediaBypass \$True -Enabled \$True

**Set-CsOnlinePSTNGateway** -BypassMode **alwaysbypass** (or **OnlyForLocalUsers**) - ProxySbc \$null

**New-CsTenantTrustedIPAddress** -IPAddress {Public IP (After NAT)} -MaskBits {Subnet Mask Prefix} -Description "Description Text"

Run the following PowerShell commands (O365) for each remote SBC device:

**New-CsOnlinePSTNGateway** -Identity **sbc2.contoso.com** -SipSignallingPort **5068** - ForwardCallHistory \$True -ForwardPai \$True -MediaBypass \$True -Enabled \$True

**Set-CsOnlinePSTNGateway** -BypassMode **alwaysbypass** (or **OnlyForLocalUsers**) - ProxySbc {ProxySBCFQDN} -GatewaySiteId {Location-based routing site-"site address"}

**New-CsTenantTrustedIPAddress** -IPAddress {Public IP (After NAT)} -MaskBits {Subnet Mask Prefix} -Description " Description Text "



**Note:** Enabling Location-based routing policies is not Mandatory for LMO, instead only the assigning of the SBC devices to the sites is required, as shown in the above PowerShell command sets. If you would like to enable Location-based routing, refer to the configuration reference:

https://docs.microsoft.com/en-us/microsoftteams/location-based-routing-enable

Based on the information above the Direct Routing will include three proprietary SIP Headers to SIP Invites and Re-invites.

# 2.3.2 Configure Online PSTN Gateway Configuration via UMP 365 (Optional)

User Management Pack 365 (UMP) is a powerful software application that simplifies user lifecycle and identity management across Skype for Business Server, hosted, cloud, hybrid and Microsoft Teams deployments.

UMP offer Simple to use web-portal user interface, Under System Configuration, the following voice routing components can be configured for use with Microsoft Teams in a direct routing environment:

- Online Dial Plans
- Normalization rule templates for use within Dial Plans



- PSTN Gateways
- PSTN Usage records for use within Voice Routes and Voice Routing Policies
- Voice Routes
- Voice Routing Policies



**Note:** This Chapter is optional, UMP offer simple and easy to use WEB portal user interface that Alleviates need for PowerShell expertise (Chapter 4.3.1)

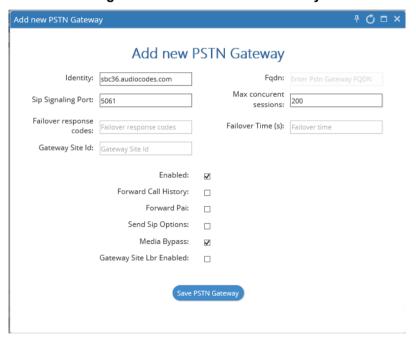
#### 2.3.2.1 Create PSTN Gateway

This section provides an example of how to create a new PSTN Gateway using the AudioCodes User Management Pack 365 install wizard:

#### To create a PSTN Gateway:

Click Add new PSTN Gateway.

Figure 2-5: Add New PSTN Gateway





**Note:** For detailed description on Adding a PSTN Gateway, refer to LTRT-26685 AudioCodes User Management Pack 365 Release Notes Ver. 7.8.100.382.

# 2.4 Call Scenario Example Topologies

The following call scenario example topologies are shown in this section:

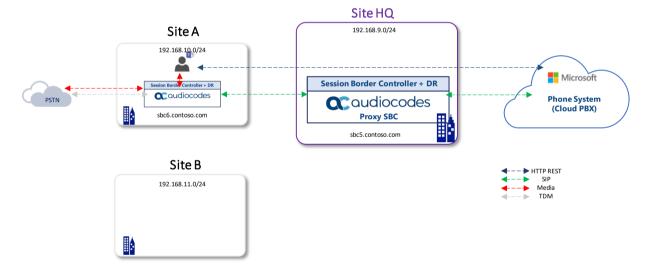
- Always Bypass with Internal Teams User (see Section 2.4.1)
- Always Bypass with External Teams User (see Section 2.4.2)
- Always Bypass with Teams User and SBC in Different Sites (see Section 2.4.3)
- Only for Local Users with Internal Teams User (see Section 2.4.4)
- Only for Local Users with External Teams User (see Section 2.4.5)
- Only for Local Users with Internal Teams User in Different Sites (see Section 2.4.6)

#### 2.4.1 Always Bypass with Internal Teams User

This topology reflects when the Teams user is in the same location as the SBC inside the corporate network and *BypassMode* is configured to *alwaysbypass*:

- The Teams user is located inside the corporate network "Internal" and places an Outbound call from the same location as the SBC – BypassMode is set to alwaysbypass.
- The Teams user is located inside the corporate network "Internal" and places an Inbound call from the same location as the SBC.

Figure 2-6: Always Bypass with Internal Teams User



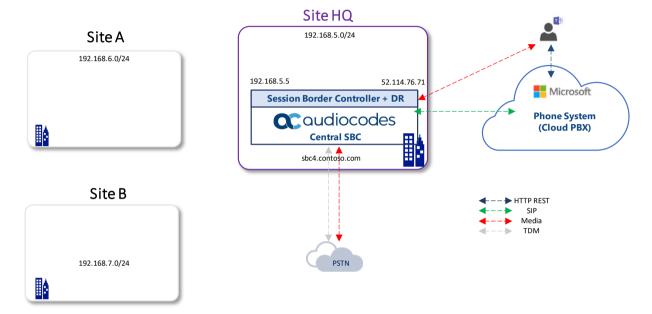


# 2.4.2 Always Bypass with External Teams User

This topology reflects when the Teams user is located outside the corporate network and *BypassMode is configured to alwaysbypass*:

- The Teams user is located outside the corporate network "External" and places an Outbound call *BypassMode* is set to *alwaysbypass*.
- The Teams user is located outside the corporate network "External" and places an Inbound call *BypassMode* is set to *alwaysbypass*.

Figure 2-7: Always Bypass with External Teams User

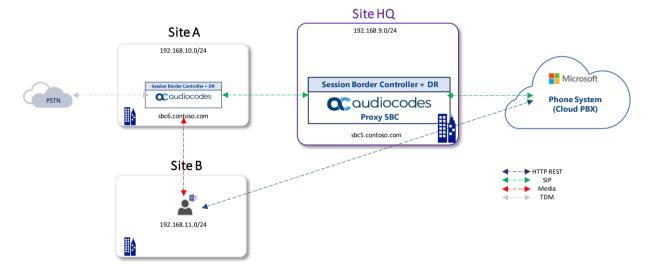


# 2.4.3 Always Bypass with Teams User and SBC in Different Sites

This topology reflects when the Teams user is in a different location to the branch SBC; however located inside the corporate network and *BypassMode* is configured to *alwaysbypass*:

- The Teams user device is inside the corporate network "Internal" and is in a different location to the branch SBC and places an Outbound call *BypassMode* is set to *alwaysbypass*.
- TheTeams user device is inside the corporate network "Internal" and is in a different location to the branch SBC, receiving an Inbound call *BypassMode* is set to *alwaysbypass*.

Figure 2-8: Always Bypass with Teams User and SBC in Different Sites



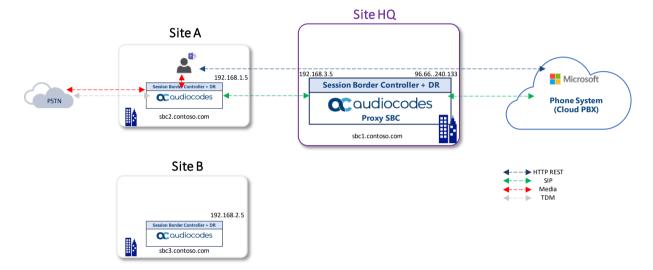


# 2.4.4 Only for Local Users with Internal Teams User

This topology reflects when the Teams user is in the same location as the SBC inside the corporate network and *BypassMode is configured to OnlyForLocalUsers*:

- The Teams user is "Internal" and in the same location as the target SBC, placing an Outbound call (handled the same as above as in Section 2.4.1) *BypassMode* is set to *OnlyForLocalUsers*.
- The Teams user is "Internal" and in the same location as the SBC (as above) making an Inbound call (handled the same as above as in Section 2.4.1) *BypassMode* is set to *OnlyForLocalUsers*.

Figure 2-9: Always Bypass with Internal Teams User



# 2.4.5 Only for Local Users with External Teams User

This topology reflects when the Teams user is located outside the corporate network and *BypassMode is configured to* **OnlyForLocalUsers**:

- (Outbound call) The Teams user is located outside the corporate network "External" and places an Outbound call BypassMode is set to OnlyForLocalUsers.
  In this case, the central SBC ( Proxy SBC) always offers an external interface since the use is outside of the corporate network.
- The Teams user is located outside the corporate network "External" and receives an Inbound call *BypassMode* is set to *OnlyForLocalUsers*.

Site HQ 192.168.3.0/24 Site A 192.168.1.0/24 Microsoft Session Border Controller + DR **a** audiocodes Phone System **a** audiocodes (Cloud PBX) sbc2.contoso.com **Proxy SBC** sbc1.contoso.com Site B ► HTTP REST 192.168.2.5 LAN Media **a** audiocodes sbc3.contoso.com

Figure 2-10: Only for Local Users with External Teams User

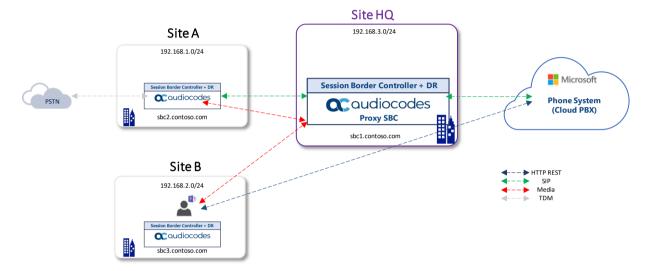


# 2.4.6 Only for Local Users with Internal Teams User in Different Sites

This topology reflects when the Teams User is in a different location to the branch SBC; however located inside the corporate network and *BypassMode* is configured to *OnlyForLocalUsers*:

- The Teams user is inside the corporate network "Internal" and is in a different location to the branch SBC and places an Outbound call BypassMode is set to OnlyForLocalUsers.
- TheTeams user is inside the corporate network "Internal" and is in a different location to the branch SBC, receiving an Inbound call – BypassMode is set to OnlyForLocalUsers.

Figure 2-11: Only for Local Users with Internal Teams User in Different Sites



# 2.5 Configuring SBC for Local Media Optimization (LMO) Proxy SBC

This section describes the configuration required for supporting Local Media Optimization handling on the **Proxy SBC**.

#### Note:



 This document shows how to configure the connection between AudioCodes' SBC and the Teams Direct Routing with a generic SIP Trunk. For detailed configuration of other entities in the deployment such as the SIP Trunk Provider and the local IP-PBX, refer to AudioCodes' SIP Trunk Configuration Notes (in the interoperability suite of documents).

#### 2.5.1 Prerequisites

Before you begin the configuration, make sure you have the following for every SBC you want to pair:

- Public IP address
- FQDN name matching SIP addresses of the Office 365 tenants
- Public certificate, issued by one of the supported CAs

#### 2.5.2 About the SBC Domain Name

The SBC domain name must be from one of the names registered in 'Domains' of the tenant. You cannot use the \*.onmicrosoft.com tenant for the domain name. For example, in Figure 2-2, the administrator registered the following DNS names for the tenant:

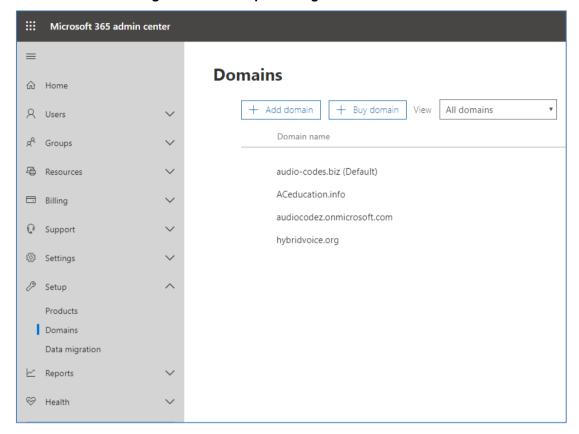
Table 2-1: DNS Names Registered by an Administrator for an Enterprise Office 365 Tenant

DNS name	Can be used for SBC FQDN	Examples of FQDN names
ACeducation.info	Yes	Valid names:
adatumbiz.onmicrosoft.com	No	Using *.onmicrosoft.com domains is not supported for SBC names
hybridvoice.org	Yes	Valid names:



Users can be from any SIP domain registered for the tenant. For example, you can provide users <a href="mailto:user@ACeducation.info">user@ACeducation.info</a> with the SBC FQDN **sbc1.hybridvoice.org** so long as both names are registered for this tenant.

Figure 2-12: Example of Registered DNS Names



The following IP address and FQDN are used as examples in this guide:

Public IP	FQDN Name		
195.189.192.157	sbc.ACeducation.info		

The certificate in the example is from DigiCert.

#### 2.5.3 Validate AudioCodes' License

The following licenses are required on AudioCodes' device:

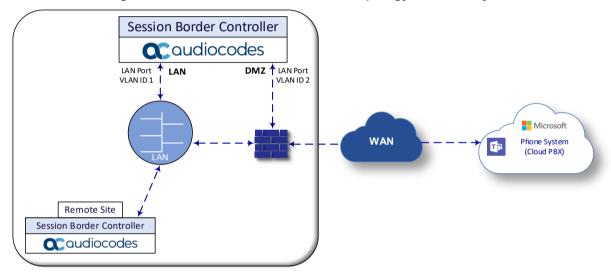
- Enable Microsoft (licensing MSFT) [All AudioCodes media gateways and SBCs are by default shipped with this license. Exceptions: MSBR products and Mediant 500 SBC or Media Gateways]
- Enable TEAMS (licensing SW/TEAMS)
- Number of SBC sessions [based on requirements]
- Transcoding sessions [if media transcoding is needed]
- Coders [based on requirements]

#### 2.5.4 Configure LAN and WAN IP Interfaces

This section describes how to configure the SBC's IP network interfaces. There are several ways to deploy the SBC:

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

Figure 2-13: Network Interfaces in the Topology of the Proxy SBC



#### 2.5.4.1 Validate Configuration of Physical Ports and Ethernet Groups

The physical ports are automatically detected by the SBC. The Ethernet groups are also auto-assigned to the ports. In this step, only parameter validation is necessary.

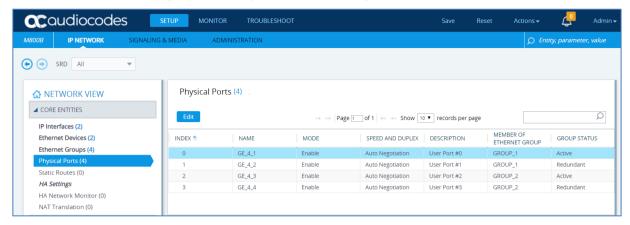
- > To validate physical ports:
- Open the Physical Ports table (Setup menu > IP Network tab > Core Entities folder > Physical Ports).
- 2. Validate that you have at least two physical ports detected by the SBC, one for LAN and the other for WAN. Make sure both ports are in **Enabled** mode.



Note: Based on your hardware configuration, you might have more than two ports.



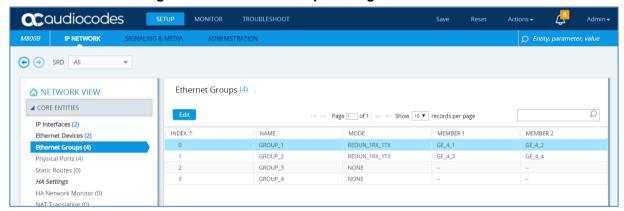
Figure 2-14: Physical Ports Configuration Interface



#### To validate Ethernet Groups:

- Open the Ethernet Groups table (Setup menu > IP Network tab > Core Entities folder > Ethernet Groups).
- Validate that you have at least two Ethernet Groups detected by the SBC, one for LAN and the other for WAN.

Figure 2-15: Ethernet Groups Configuration Interface



#### 2.5.4.2 Configure LAN and WAN VLANs

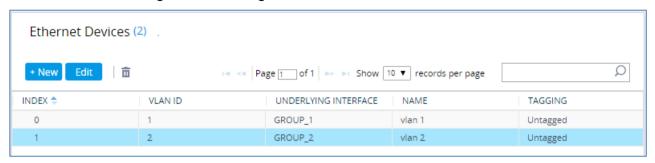
This section describes how to define VLANs for each of the following interfaces:

- LAN (assigned the name "LAN IF")
- WAN (assigned the name "WAN\_IF")

#### To configure the VLANs:

- Open the Ethernet Device table (Setup menu > IP Network tab > Core Entities folder > Ethernet Devices).
- 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

Figure 2-16: Configured VLAN IDs in Ethernet Device



#### 2.5.4.3 Configure Network Interfaces

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

- LAN Interface (assigned the name "LAN IF")
- WAN Interface (assigned the name "WAN IF")
- > To configure network parameters for both LAN and WAN interfaces:
- Open the IP Interfaces table (Setup menu > IP Network tab > Core Entities folder > IP Interfaces).
- 2. Configure the IP interfaces as follows (your network parameters might be different):

Table 2-2: Configuration Example of the Network Interface Table

Index	Name	Application Types	Interface Mode	IP Address	Prefix Length	Gateway	DNS	Ethernet Device
0	LAN_IF	OAMP+ Media + Control	IPv4 Manual	10.15.77.77	16	10.15.0.1	10.15.27.1	vlan 1
1	WAN_IF	Media + Control (as this interface points to the internet, enabling OAMP is not recommended)	IPv4 Manual	195.189.192.157 (DMZ IP address of SBC)	25	195.189.192.129 (router's IP address)	According to your Internet provider's instructions	



The configured IP network interfaces are shown below:

Figure 2-17: Configuration Example of the Network Interface Table



## 2.5.5 Configure TLS Context

The Microsoft Phone System Direct Routing Interface only allows TLS connections from SBCs for SIP traffic with a certificate signed by one of the trusted Certification Authorities. Currently, supported Certification Authorities can be found in the following link:

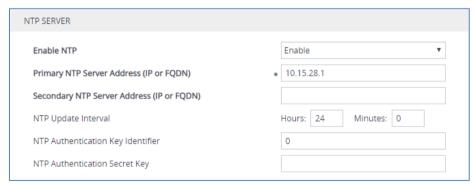
https://docs.microsoft.com/en-us/microsoftteams/direct-routing-plan#public-trusted-certificate-for-the-sbc

#### 2.5.5.1 Configure the NTP Server Address

This section describes how to configure the NTP server's IP address. It is recommended to implement an NTP server (Microsoft NTP server or another global server) to ensure that the SBC receives the current date and time. This is necessary for validating certificates of remote parties. It is important, that NTP Server will locate on the OAMP IP Interface (LAN\_IF in our case) or will be accessible through it.

- To configure the NTP server address:
- 1. Open the Time & Date page (Setup menu > Administration tab > Time & Date).
- In the 'Primary NTP Server Address' field, enter the IP address of the NTP server (e.g., 10.15.28.1).

Figure 2-18: Configuring NTP Server Address



3. Click Apply.

#### 2.5.5.2 Create a TLS Context for Teams Direct Routing

The section below shows how to request a certificate for the SBC WAN interface and to configure it based on the example of DigiCert Global Root CA. The certificate is used by the SBC to authenticate the connection with Teams Direct Routing.

The procedure involves the following main steps:

- a. Create a TLS Context for Teams Direct Routing
- Generate a Certificate Signing Request (CSR) and obtain the certificate from a supported Certification Authority
- c. Deploy the SBC and Root/ Intermediate certificates on the SBC
- To create a TLS Context for Teams Direct Routing:
- Open the TLS Contexts page (Setup menu > IP Network tab > Security folder > TLS Contexts).
- 2. Create a new TLS Context by clicking **+New** at the top of the interface, and then configure the parameters using the table below as reference.

Table 2-3: New TLS Context

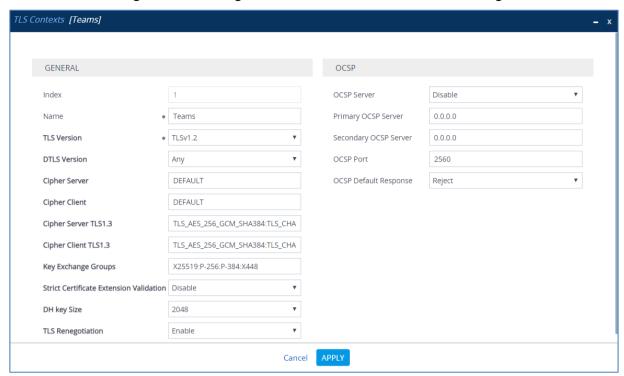
Index	Name	TLS Version			
1	Teams (arbitrary descriptive name)	TLSv1.2			
All other parameters can be left unchanged with their default values.					



**Note:** The table above exemplifies configuration focusing on interconnecting SIP and media. You might want to configure additional parameters according to your company's policies. For example, you might want to configure Online Certificate Status Protocol (OCSP) to check if SBC certificates presented in the online server are still valid or revoked. For more information on the SBC's configuration, see the *User's Manual*, available for download from <a href="https://www.audiocodes.com/library/technical-documents">https://www.audiocodes.com/library/technical-documents</a>.

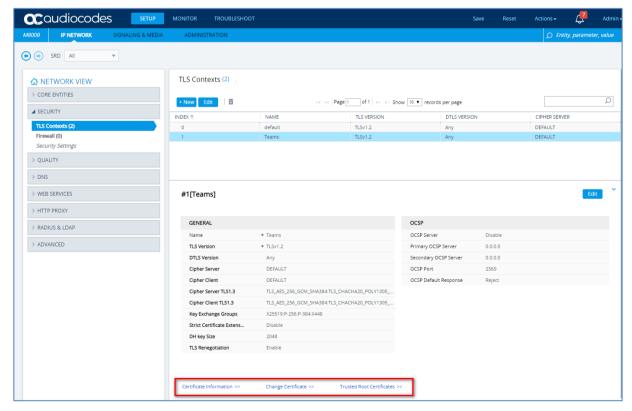


Figure 2-19: Configuration of TLS Context for Direct Routing



3. Click **Apply**; you should see the new TLS Context and option to manage the certificates at the bottom of 'TLS Context' table.

Figure 2-20: Configured TLS Context for Direct Routing and Interface to Manage the Certificates



30

#### 2.5.5.3 Generate a CSR and Obtain the Certificate from a Supported CA

This section shows how to generate a Certificate Signing Request (CSR) and obtain the certificate from a supported Certification Authority.

- > To generate a Certificate Signing Request (CSR) and obtain the certificate from a supported Certification Authority:
- Open the TLS Contexts page (Setup menu > IP Network tab > Security folder > TLS Contexts).
- In the TLS Contexts page, select the Teams TLS Context index row, and then click the Change Certificate link located below the table; the Context Certificates page appears.
- 3. Under the Certificate Signing Request group, do the following:
  - **a.** In the 'Common Name [CN]' field, enter the SBC FQDN name (based on example above, **ACeducation.info**).
  - b. In the '1st Subject Alternative Name [SAN]' field, change the type to 'DNS', and then enter the SBC FQDN name (based on the example above, ACeducation.info).

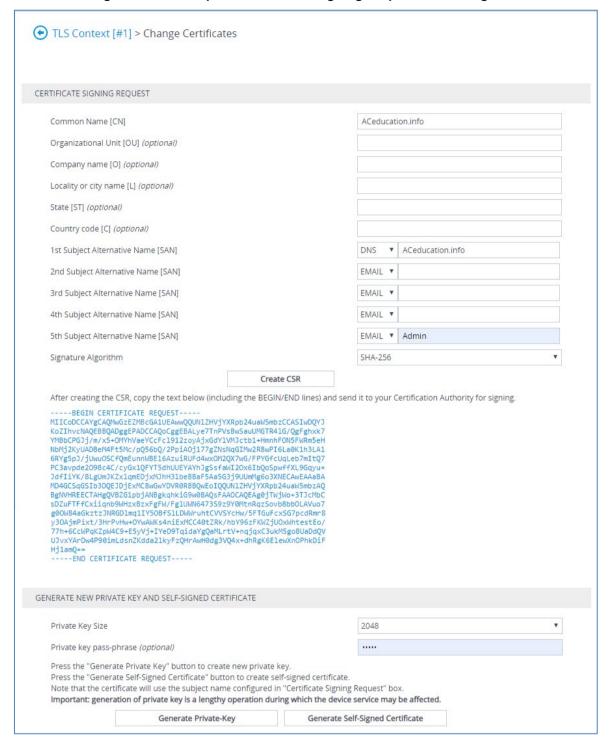


**Note:** The domain portion of the Common Name [CN] and 1<sup>st</sup> Subject Alternative Name [SAN] must match the SIP suffix configured for Office 365 users.

- c. Change the 'Private Key Size' based on the requirements of your Certification Authority. Many CAs do not support private key of size 1024. In this case, you must change the key size to 2048.
  - d. To change the key size on TLS Context, go to: Generate New Private Key and Self-Signed Certificate, change the 'Private Key Size' to 2048 and then click Generate Private-Key. To use 1024 as a Private Key Size value, you can click Generate Private-Key without changing the default key size value.
  - Enter the rest of the request fields according to your security provider's instructions.
  - f.Click the Create CSR button; a textual certificate signing request is displayed in the area below the button:



Figure 2-21: Example of Certificate Signing Request - Creating CSR



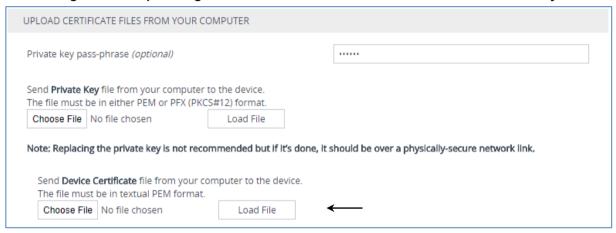
- Copy the CSR from the line "----BEGIN CERTIFICATE" to "END CERTIFICATE
  REQUEST----" to a text file (such as Notepad), and then save it to a folder on your
  computer with the file name, for example certreq.txt.
- 5. Send *certreg.txt* file to the Certified Authority Administrator for signing.

#### 2.5.5.4 Deploy the SBC and Root / Intermediate Certificates on the SBC

After obtaining the SBC signed and Trusted Root/Intermediate Certificate from the CA, install the following:

- SBC certificate
- Root / Intermediate certificates
- To install the SBC certificate:
- 1. In the SBC's Web interface, return to the **TLS Contexts** page and do the following:
  - **a.** In the TLS Contexts page, select the required TLS Context index row, and then click the **Change Certificate** link located below the table; the Context Certificates page appears.
  - b. Scroll down to the Upload certificates files from your computer group, click the Choose File button corresponding to the 'Send Device Certificate...' field, navigate to the certificate file obtained from the CA, and then click Load File to upload the certificate to the SBC.

Figure 2-22: Uploading the Certificate Obtained from the Certification Authority



Validate that the certificate was uploaded correctly: A message indicating that the certificate was uploaded successfully is displayed in blue on the lower part of the page:

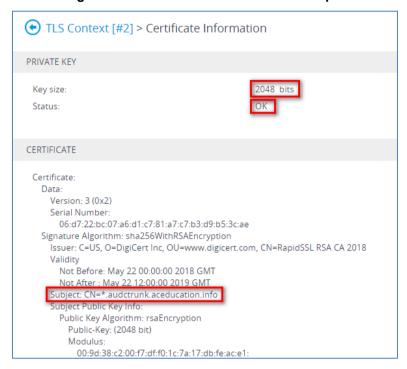
Figure 2-23: Message Indicating Successful Upload of the Certificate



3. In the SBC's Web interface, return to the TLS Contexts page, select the required TLS Context index row, and then click the Certificate Information link, located at the bottom of the TLS. Then validate the Key size, certificate status and Subject Name:

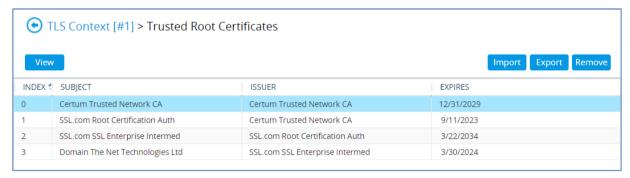


Figure 2-24: Certificate Information Example



- In the SBC's Web interface, return to the TLS Contexts page.
  - c. In the TLS Contexts page, select the required TLS Context index row, and then click the Trusted Root Certificates link, located at the bottom of the TLS Contexts page; the Trusted Certificates page appears.
  - d. Click the **Import** button, and then select all Root/Intermediate Certificates obtained from your Certification Authority to load.
- 5. Click **OK**; the certificate is loaded to the device and listed in the Trusted Certificates store:

Figure 2-25: Example of Configured Trusted Root Certificates



#### 2.5.6 Method of Generating and Installing the Wildcard Certificate

To use the same certificate on multiple devices, you may prefer using 3<sup>rd</sup> party application (e.g. <u>DigiCert Certificate Utility for Windows</u>) to process the certificate request from your Certificate Authority on another machine, with this utility installed.

After you've processed the certificate request and response using the DigiCert utility, test the certificate private key and chain and then export the certificate with private key and assign a password.

#### To install the certificate:

- Open the TLS Contexts page (Setup menu > IP Network tab > Security folder > TLS Contexts).
- In the TLS Contexts page, select the required TLS Context index row, and then click the Change Certificate link located below the table; the Context Certificates page appears.
- Scroll down to the Upload certificates files from your computer group and do the following:
  - **e.** Enter the password assigned during export with the DigiCert utility in the **'Private key pass-phrase'** field.
  - f. Click the **Choose File** button corresponding to the 'Send **Private Key**...' field and then select the SBC certificate file exported from the DigiCert utility.

#### 2.5.7 Deploy Baltimore Trusted Root Certificate



**Note:** Loading Baltimore Trusted Root Certificates to AudioCodes' SBC is mandatory for implementing an MTLS connection with the Microsoft Teams network.

The DNS name of the Teams Direct Routing interface is **sip.pstnhub.microsoft.com**. In this interface, a certificate is presented which is signed by Baltimore Cyber Baltimore CyberTrust Root with Serial Number: 02 00 00 b9 and SHA fingerprint: d4:de:20:d0:5e:66:fc: 53:fe:1a:50:88:2c:78:db:28:52:ca:e4:74.

To trust this certificate, your SBC *must* have the certificate in Trusted Certificates storage. Download the certificate from <a href="https://cacert.omniroot.com/bc2025.pem">https://cacert.omniroot.com/bc2025.pem</a> and follow the steps above to import the certificate to the Trusted Root storage.



**Note:** Before importing the Baltimore root certificate into AudioCodes' SBC, make sure it's in .PEM or .PFX format. If it isn't, you need to convert it to .PEM or .PFX format, otherwise the 'Failed to load new certificate' error message is displayed. To convert to PEM format, use Windows local store on any Windows OS and then export it as 'Base-64 encoded X.509 (.CER) certificate'.



#### 2.5.8 Configure Media Realms

Media Realms allow dividing the UDP port ranges for use on different interfaces. In the example below, two Media Realms are configured:

- One for the LAN interface, with the UDP port starting at 6000 and the number of media session legs 100 (you need to calculate number of media session legs based on your usage)
- One for the WAN interface, with the UDP port range starting at 7000 and the number of media session legs 100

#### To configure Media Realms:

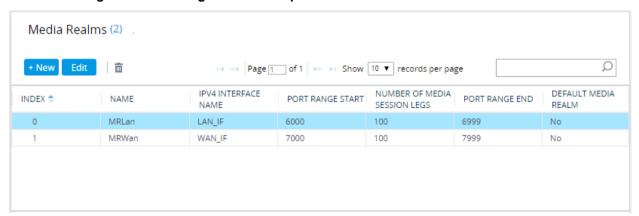
- 1. Open the Media Realms table (**Setup** menu > **Signaling & Media** tab > **Core Entities** folder > **Media Realms**).
- 2. Configure Media Realms as follows (you can use the default Media Realm (Index 0), but modify it):

Table 2-4: Configuration Example Media Realms in Media Realms Table

Index	Name	Topology Location	IPv4 Interface Name	Port Range Start	Number of Media Session Legs
0	MRLan (arbitrary name)		LAN_IF	6000	100 (media sessions assigned with port range)
1	MRWan (arbitrary name)	Up	WAN_IF	7000	100 (media sessions assigned with port range)

The configured Media Realms are shown in the figure below:

Figure 2-26: Configuration Example Media Realms in Media Realm Table



## 2.5.9 Configure SIP Signaling Interfaces

This section shows how to configure a SIP Signaling Interfaces. A SIP Interface defines a listening port and type (UDP, TCP, or TLS) for SIP signaling traffic on a specific logical IP network interface (configured in the Interface Table above) and Media Realm.

Note that the configuration of a SIP interface for the SIP Trunk shows as an example and your configuration might be different. For specific configuration of interfaces pointing to SIP trunks and/or a third-party PSTN environment connected to the SBC, see the trunk / environment vendor documentation.

AudioCodes also offers a comprehensive suite of documents covering the interconnection between different trunks and equipment.

#### To configure SIP Interfaces:

- 1. Open the SIP Interfaces table (**Setup** menu > **Signaling & Media** tab > **Core Entities** folder > **SIP Interfaces**).
- 2. Configure SIP Interfaces. You can use the default SIP Interface (Index 0), but modify it as shown in the table below. The table below shows an example of the configuration. You can change some parameters according to your requirements.



**Note:** The Direct Routing interface can only use TLS for a SIP port. It does not support using TCP due to security reasons. The SIP port might be any port of your choice. When pairing the SBC with Office 365, the chosen port is specified in the pairing command.

Table 2-5: Configuration Example of SIP Signaling Interfaces

Index	Name	Network Interface	Application Type	UDP Port	TCP Port	TLS Port	Enable TCP Keepalive	Classification Failure Response Type	Media Realm	TLS Context Name
0	SitesSIPInterface (arbitrary name)	LAN_IF	SBC	0	0	5061 (according to site requirement)	Disable (leave default value)	500 (leave default value)	MRLan	-
1	Teams (arbitrary name)	WAN_IF	SBC	0 (Phone System does not use UDP or TCP for SIP signaling)	0	5061 (as configured in the Office 365)	Enable	0 (Recommended to prevent DoS attacks)	MRWan	Teams



**Note:** For implementing an MTLS connection with the Microsoft Teams network, configure 'TLS Mutual Authentication' to "Enable" for the Teams SIP Interface.



**Note:** Loading Baltimore Trusted Root Certificates to AudioCodes' SBC is mandatory for implementing an MTLS connection with the Microsoft Teams network. Refer to Section **Error! Reference source not found.** on page **Error! Bookmark not defined.**.



The configured SIP Interfaces are shown in the figure below:

Figure 2-27: Configuration Example of SIP Signaling Interfaces



## 2.5.10 Configure Proxy Sets and Proxy Address

The Proxy Set and Proxy Address defines TLS parameters, IP interfaces, FQDN and the remote entity's port. Proxy Sets can also be used to configure load balancing between multiple servers. The example below covers configuration of a Proxy Sets for Teams Direct Routing and SIP Trunk. Note that the configuration of a Proxy Set for the SIP Trunk (through Site A SBC) shows as an example and your configuration might be different. For specific configuration of interfaces pointing to SIP trunks and/or the third-party PSTN environment connected to the SBC, see the trunk/environment vendor's documentation. AudioCodes also offers a comprehensive suite of documents covering the interconnection between different trunks and the equipment.

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

#### 2.5.10.1 Configure Proxy Sets

To support the Local Media Optimization handling, the Proxy Set that is paired as the Microsoft Teams Direct Routing interface is configured with an FQDN Host Name and each deployed remote (site) SBC is configured with an IP address.

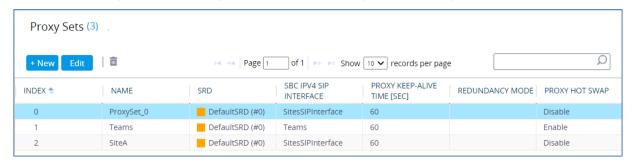
- To configure a Proxy Sets:
- Open the Proxy Sets table (Setup menu > Signaling & Media tab > Core Entities folder > Proxy Sets).
- Configure Proxy Sets as shown in the table below:

Table 2-6: Configuration Example Proxy Sets in Proxy Sets Table

Index	Name	SBC IPv4 SIP Interface	TLS Context Name	Proxy Keep- Alive	Proxy Hot Swap	Proxy Load Balancing Method
1	Teams (arbitrary name)	Teams	Teams	Using Options	Enable	Random Weights
2	SiteA	SitesSIPInterface	Default	Using Options	-	-

The configured Proxy Sets are shown in the figure below:

Figure 2-28: Configuration Example Proxy Sets in Proxy Sets Table



### 2.5.10.2 Configure Proxy Addresses

This section shows how to configure the Proxy Addresses for the Proxy Sets in the Proxy SBC site towards the remote (site) SBC.

#### To configure a Proxy Address for Teams:

- Open the Proxy Sets table (Setup menu > Signaling & Media tab > Core Entities folder > Proxy Sets) and then click the Proxy Set Teams, and then click the Proxy Address link located below the table; the Proxy Address table opens.
- Click +New:
- Configure the address of the Proxy Set according to the parameters described in the table below:

Table 2-7: Configuration Proxy Address for Teams Direct Routing

Index	Proxy Address	Transport Type	Proxy Priority	Proxy Random Weight
0	sip.pstnhub.microsoft.com:5061	TLS	1	1
1	sip2.pstnhub.microsoft.com:5061	TLS	2	1
2	sip3.pstnhub.microsoft.com:5061	TLS	3	1

4. Click **Apply** and then save your settings to flash memory.

#### > To configure a Proxy Address for remote (site) SBCs:

- Open the Proxy Sets table (Setup menu > Signaling & Media tab > Core Entities folder > Proxy Sets) and then click the Proxy Set SIPTrunk, and then click the Proxy Address link located below the table; the Proxy Address table opens.
- Click +New;
- 3. Configure the IP address of the Proxy Set towards Remote SBC in Site A according to the parameters described in the table below:

Table 2-8: Configuration Proxy Address Towards Remote SiteA SBC

Index		Proxy Address	Transport Type
	0	192.168.1.5:5061	TLS

4. Click **Apply** and then save your settings to flash memory.



## 2.5.11 Configure Coder Groups

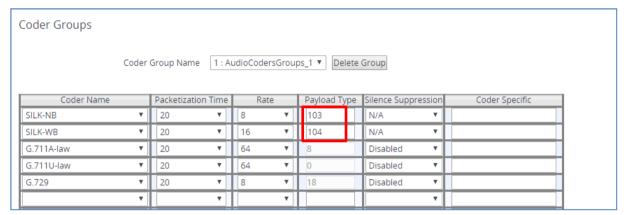
the next section

This section describes how to configure coders (known as *Coder Groups*). Teams Direct Routing supports the SILK and other coders while the network connection to the SIP Trunk may restrict operation with a dedicated coders list. You need to add a Coder Group with the supported coders for each of the following leg, the Teams Direct Routing and the SIP Trunk. Note that the Coder Group ID for this entity will be assigned to its corresponding IP Profile in

#### > To configure a Coder Group:

- Open the Coder Groups table (Setup menu > Signaling & Media tab > Coders & Profiles folder > Coder Groups).
- From the 'Coder Group Name' dropdown, select 1:Does Not Exist and add the required codecs as shown in the figure below.

Figure 2-29: Configuring Coder Group for Teams Direct Routing



3. Click **Apply**, and then confirm the configuration change in the prompt that pops up.

## 2.5.12 Configure IP Profiles

This section describes how to re-configure an IP Profiles in the Proxy SBC site. An IP Profile is a set of parameters with user-defined settings related to signaling (e.g., SIP message terminations such as REFER) and media (e.g., coder type). An IP Profile needs to be assigned to the specific IP Group. See Appendix C for a summary of all IP Profile configurations.

#### > To configure IP Profiles Proxy SBC site:

- Open the Proxy Sets table (Setup menu > Signaling & Media tab > Coders & Profiles folder > IP Profiles).
- Click +New to add the IP Profile for the Teams Direct Routing interface. Configure the parameters using the table below as reference.

Table 2-9: Configuration Example: Teams IP Profile

Parameter	Value
General	
Name	Teams (arbitrary descriptive name)
Media Security	
SBC Media Security Mode	Secured

Parameter	Value			
SBC Early Media				
Remote Early Media RTP Detection Mode	<b>By Media</b> (required, as Teams Direct Routing does not send RTP immediately to remote side when it sends a SIP 18x response)			
SBC Media				
Extension Coders Group	AudioCodersGroups_1			
RTCP Mode	Generate Always (required, as some ITSPs do not send RTCP packets during Hold, but Microsoft expects them)			
ICE Mode	Lite (required only when Media Bypass enabled on Teams)			
SBC Signaling				
SIP UPDATE Support	Not Supported			
Remote re-INVITE Support	Supported Only With SDP			
Remote Delayed Offer Support	Not Supported			
Remote Representation Mode	Add Routing Headers			
SBC Forward and Transfer				
Remote REFER Mode	Regular			
Remote 3xx Mode	Transparent			
SBC Hold				
Remote Hold Format	Inactive (some SIP Trunk may answer with a=inactive and IP=0.0.0.0 in response to the Re-Invite with Hold request from Teams. Microsoft Media Stack doesn't support this format. So, SBC will replace 0.0.0.0 with its IP address)			
All other parameter	All other parameters can be left unchanged at their default values.			

- 3. Click **Apply**, and then save your settings to flash memory.
- 4. Click **+New** to add the IP Profile for the SIP Trunk (through Site A SBC). Configure the parameters using the table below as a reference.

Table 2-10: Configuration Example: SIP Trunk IP Profile (toward remote Site A SBC)

Parameter	Value		
General			
Name	SiteA (arbitrary name)		
SBC Forward and Transfer	SBC Forward and Transfer		
Remote REFER Mode	Regular		
Remote Replaces Mode	Standard		
Remote 3xx Mode	Transparent		
All other parameters can be left unchanged with their default values.			

5. Click **Apply**, and then save your settings to flash memory.



## 2.5.13 Configure IP Groups

This section describes how to configure IP Groups. The IP Group represents an IP entity on the network with which the SBC communicates. This can be a server (e.g., IP-PBX or SIP Trunk) 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.

- To configure IP Group for Microsoft Teams Direct Routing for Media optimization:
- Open the IP Groups table (Setup menu > Signaling & Media tab > Core Entities folder > IP Groups).
- Click Edit to re-configure the IP Group for the Microsoft Teams Direct Routing paired SBC (Proxy SBC):

Table 2-11: Configuration Example: IP Group for Microsoft Teams Direct Routing

Parameter	Value
Name	Teams (arbitrary name)
Topology Location	Up
Туре	Server
Proxy Set	Teams
IP Profile	Teams
Media Realm	MRWan
Internal Media Realm	MRLan  This parameter is relevant when the 'Teams Local Media Optimization Handling' parameter (see below) is configured to any value other than "None" and the X-MS-UserLocation header in the incoming SIP message is set to 'Internal'. In this case, the Internal Media Realm determines the UDP port range and maximum sessions for Media traffic on this IP interface.  If X-MS-UserLocation=Internal response is received from Teams, a new IP address/port is allocated using the Internal Media Realm only if the call is non-direct media i.e. media traverses the paired SBC to the remote SBCs.
Classify by Proxy Set	Disable
Local Host Name	<b><fqdn< b=""> name of the SBC in the enterprise tenant&gt; (For example, sbc.ACeducation.info defines the host name (string) that the device uses in the SIP message's Via and Contact headers. This defines the FQDN as the host name that is recognized by Microsoft Teams. The device uses this string for Via and Contact headers in outgoing INVITE messages sent to a specific IP Group, and the Contact header in SIP 18x and 200 OK responses for incoming INVITE messages received from the other configured IP Groups (SiteA and SiteB).</fqdn<></b>
Always Use Src Address	Yes
Teams Local Media Optimization Handling	<b>Teams Decides</b> (The routing decision is made according to the Microsoft Teams headers for the primary route)

Parameter	Value		
	This parameter is relevant for inbound calls to Teams when "Teams Local Media Optimization Handling" is set to "Teams Decides" or "SBC Decides":		
	Direct Media (default) – Perform direct media call towards Teams.		
Teams Local Media Optimization Initial Behavior	• Internal - Perform non-direct media call (media traverses the paired SBC from the remote SBC) towards Teams using Internal Media Realm.		
	External – Perform non-direct media call (media traverses the paired SBC from the remote SBC) towards Teams using external (regular) Media Realm.		
	<b>Note:</b> The value of this parameter can be variable depending on particular setup		
Proxy Keep-Alive using IP Group settings	Enable		
Inbound Message Manipulation Set	0		
Outbound Message Manipulation Set	1		
Call Setup Rules Set ID	0		
All other paramet	All other parameters can be left unchanged with their default values.		

3. Click +New to add the IP Group for the SBC located at Site A and connected to the SIP Trunk. Configure the parameters using the table below as reference:

Table 2-12: Configuration Example: IP Group for Site A SBC

Parameter	Value
Name	SiteA (arbitrary name)
Topology Location	Down
Туре	Server
Proxy Set	SiteA
IP Profile	SiteA
Media Realm	MRLan
Tags	Site={RemotePSTNGateWayFQDN}
All other parameters can be left unchanged with their default values.	

The Site Tag should be defined as the **remote** site SBC's FQDN and should be discoverable by DNS from the Proxy SBC.

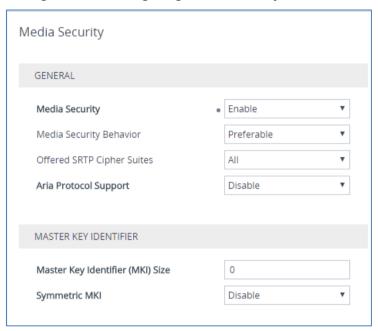


## 2.5.14 Configure SRTP

This section describes how to configure media security. The Direct Routing Interface requires the use of SRTP only, so you need to configure the SBC to operate in the same manner.

- To configure media security:
- Open the Media Security page (Setup menu > Signaling & Media tab > Media folder > Media Security).
- 2. From the 'Media Security' drop-down list, select **Enable** to enable SRTP.

Figure 2-30: Configuring Media Security Parameter



3. Click Apply.

## 2.5.15 Configure Message Condition Rules

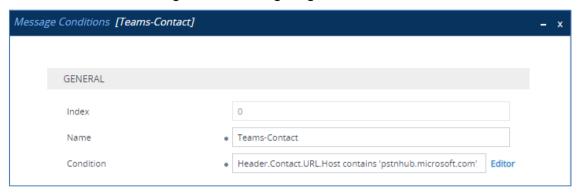
This section describes how to configure the Message Condition Rules. A Message Condition defines special conditions (requisites) for incoming SIP messages. These rules can be used as additional matching criteria for the IP-to-IP routing rules in the IP-to-IP Routing table.

The following condition verifies that the Contact header contains Teams FQDN.

- > To configure a Message Condition rule:
- Open the Message Conditions table (Setup menu > Signaling & Media tab > Message Manipulation folder > Message Conditions).
- 2. Click **New**, and then configure the parameters as follows:

Parameter	Value
Index	0
Name	Teams-Contact (arbitrary descriptive name)
Condition	header.contact.url.host contains 'pstnhub.microsoft.com'

Figure 2-31: Configuring Condition Table



Click Apply.

## 2.5.16 Configure Classification Rules

This section describes how to configure Classification rules. A Classification rule classifies incoming SIP dialog-initiating requests (e.g., INVITE messages) to a "source" IP Group. The source IP Group is the SIP entity that sends the SIP dialog request. Once classified, the device uses the IP Group to process the call (manipulation and routing).

You can also use the Classification table for employing SIP-level access control for successfully classified calls, by configuring Classification rules with whitelist and blacklist settings. If a Classification rule is configured as a whitelist ("Allow"), the device accepts the SIP dialog and processes the call. If the Classification rule is configured as a blacklist ("Deny"), the device rejects the SIP dialog.

#### > To configure a Classification rule:

- Open the Classification table (Setup menu > Signaling & Media tab > SBC folder > Classification Table).
- 2. Click **New**, and then configure the parameters as follows:

Parameter	Value
Index	0
Name	Teams
Source SIP Interface	Teams
Source IP Address	52.114.*.*
Destination Host	sbc.ACeducation.info (example)
Message Condition	Teams-Contact
Action Type	Allow
Source IP Group	Teams



lassification [Teams] SRD #0 [DefaultSRD] MATCH ACTION Destination Routing Policy Source SIP Interface #1 [Teams] Source IP Group IP Group Selection Source IP Address 52.114.\*.\* Source IP Group #2 [Teams] ▼ View Source Transport Type IP Group Tag Name default ▼ View IP Profile Source Username Pattern Source Host Destination Username Pattern Destination Host sbc.ACeducation.info Message Condition #0 [Teams-Contact] ▼ View

Cancel

Figure 2-32: Configuring Classification Rule

Click Apply.

## 2.5.17 Configure Call Setup Rules

This section describes how to configure Call Setup Rules based on the site hostname, extracted from the Request-URI header. Call Setup rules define various sequences (site destination in this case) that are run upon receipt of an incoming call (dialog) at call setup, before the device routes the call to its destination.

Configured Call Setup Rules need be assigned to specific IP Group.

- To configure a Call Setup Rules based on Site FQDN:
- Open the Call Setup Rules table (Setup menu > Signaling & Media tab > SIP Definitions folder > Call Setup Rules).
- 2. Click New
- 3. Configure a Call Setup rule according to the parameters described in the table below.

**Table 2-13: Call Setup Rules Table** 

Index	Rules Set ID	Condition	Action Subject	Action Type	Action Value
0	0	Var.Session.0 == "	Var.Session.0	Modify	Header.Request-URI.URL.Host.Name
1	0	Var.Session.0 != "	DstTags.Site	Modify	Var.Session.0

4. Click **Apply** and then save your settings to flash memory.

## 2.5.18 Configure Message Manipulation Rules

This section 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 rules:
- Open the Message Manipulations page (Setup menu > Signaling & Media tab > Message Manipulation folder > Message Manipulations).
- Configure a new manipulation rule (Manipulation Set 0) for Teams IP Group. This rule applies to messages received from the Teams IP Group. This remove the privacy header to enable CLI identity.

Parameter	Value
Index	0
Name	Privacy Header
Manipulation Set ID	0
Condition	Header.Privacy contains 'id'
Action Subject	Header.Privacy
Action Type	Remove

3. Configure another manipulation rule (Manipulation Set 1) for Teams IP Group. This rule applies to messages sent to the Teams IP Group. This replace the host part of the Contact Header with the value from the To Header.

Parameter	Value
Index	1
Name	Replace Host in Contact
Manipulation Set ID	1
Message Type	Invite.Request
Action Subject	Header.Contact.URL.Host
Action Type	Modify
Action Value	Header.To.URL.Host



## 2.5.19 Configure IP-to-IP Call Routing Rules

This section 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 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 example shown below only covers IP-to-IP routing, though you can route the calls from SIP Trunk (through Site A SBC) to Teams and vice versa. See AudioCodes' SBC documentation for more information on how to route in other scenarios.

The following IP-to-IP Routing Rules will be defined:

- Calls from Teams Direct Routing to SIP Trunk (through Site A SBC)
- Calls from SIP Trunk (through Site A SBC) to Teams Direct Routing
- > To re-configure IP-to-IP routing rules:
- Open the IP-to-IP Routing table (Setup menu > Signaling & Media tab > SBC folder > Routing > IP-to-IP Routing).
- 2. Configure routing rules as shown in the table below:

Table 2-14: IP-to-IP Call Routing Rules

Index	Name	Source IP Group	Request Type	Dest Type	Dest IP Group	Routing Tag Name	Internal Action
0	Terminate OPTIONS	Any	OPTIONS	Internal			Reply (Response ='200')
1	Teams to SIP Trunk (arbitrary name)	Teams		Destination Tag		Site	
2	SIP Trunk to Teams (arbitrary name)	Any		IP Group	Teams		



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

## 2.5.20 Configure Firewall Settings



**Note:** AudioCodes highly advised to configure firewall with network traffic filtering rules **in front of** WAN interface of the SBC. For detailed list of ports, which needed to be open please refer to: <a href="https://docs.microsoft.com/en-us/microsoftteams/direct-routing-plan#sip-signaling-fqdns-and-firewall-ports">https://docs.microsoft.com/en-us/microsoftteams/direct-routing-plan#sip-signaling-fqdns-and-firewall-ports</a>.

As an extra security to the above note, there is option to configure traffic filtering rules (access list) for incoming traffic on AudioCodes SBC. For each packet received on the configured network interface, the SBC searches the table from top to bottom until the first matching rule is found. The matched rule can permit (allow) or deny (block) the packet. Once a rule in the table is located, subsequent rules further down the table are ignored. If the end of the table is reached without a match, the packet is accepted. Please note that the firewall is stateless. The blocking rules will apply to all incoming packets, including UDP or TCP responses.

- > To configure a firewall rule:
- 1. Open the Firewall table (Setup menu > IP Network tab > Security folder> Firewall).
- 2. Configure the following Access list rules for Teams Direct Rout IP Interface:

Table 2-15: Firewall Table Rules

Index	Source IP	Subnet Prefix	Start Port	End Port	Protocol	Use Specific Interface	Interface ID	Allow Type
0	<public dns="" ip="" server=""> (e.g. 8.8.8.8)</public>	32	0	65535	Any	Enable	WAN_IF	Allow
1	52.114.148.0	32	0	65535	TCP	Enable	WAN_IF	Allow
2	52.114.132.46	32	0	65535	TCP	Enable	WAN_IF	Allow
3	52.114.75.24	32	0	65535	TCP	Enable	WAN_IF	Allow
4	52.114.76.76	32	0	65535	TCP	Enable	WAN_IF	Allow
5	52.114.7.24	32	0	65535	TCP	Enable	WAN_IF	Allow
6	52.114.14.70	32	0	65535	TCP	Enable	WAN_IF	Allow
49	0.0.0.0	0	0	65535	Any	Enable	WAN_IF	Block



**Note:** Be aware, that if in your configuration, connectivity to other entities (except Teams) is performed through the same IP Interface as Teams (WAN\_IF in our example), you must add rules to allow traffic from these entities.



# 2.6 Configuring SBC for Local Media Optimization (LMO) Remote Site SBCs

This section describes the configuration required for supporting Local Media Optimization handling on the remote site SBCs.

## 2.6.1 Configure LAN and WAN IP Interfaces

Configuration of the SBC's IP network interfaces done in the same way as in Proxy SBC. Please refer to Section 2.5.4 on page 25 above.

## 2.6.2 Configure Media Realms

Configuration of the SBC's IP network interfaces done in the same way as in Proxy SBC. Please refer to Section 2.5.8 on page 36 above.

## 2.6.3 Configure SIP Interfaces

This section shows how to configure a SIP Signaling Interfaces on the remote site SBC.

#### To configure SIP interfaces:

- 1. Open the SIP Interfaces table (**Setup** menu > **Signaling & Media** tab > **Core Entities** folder > **SIP Interfaces**).
- 2. Enable TLS port for SIP signaling. You can use the default SIP Interface (Index 0), however modify it as shown in the table below. The table below shows an example of the configuration. You can change some parameters according to your requirements.

Table 2-16: Configuration Example: Site SBC SIP Interfaces

Index	Name	Network Interface	Application Type	UDP Port	TCP Port	TLS Port	Enable TCP Keepalive	Classification Failure Response Type	Media Realm	TLS Context Name
0	SIPTrunk (arbitrary name)	WAN_IF	SBC	5060 (according to Service Provider requirement)	0	0	Disable (leave default value)	0 (Recommended to prevent DoS attacks)	MRWan	-
1	ProxySBC (arbitrary name)	LAN_IF	SBC	0	0	5061	Enable	500 (leave default value)	MRLan	-

## 2.6.4 Configure Proxy Sets and Proxy Address

This section describes how to configure Proxy Sets and Proxy address for remote SBCs.

#### To configure a Proxy Sets:

- 1. Open the Proxy Sets table (**Setup** menu > **Signaling & Media** tab > **Core Entities** folder > **Proxy Sets**).
- 2. Configure Proxy Sets as shown in the table below

**Table 2-17: Configuration Example: Site Proxy Sets** 

Index	Name	SBC IPv4 SIP Interface	TLS Context Name	Proxy Keep-Alive	
1	SIPTrunk (arbitrary name)	SIPTrunk	Default	Using Options	
2	ProxySBC (arbitrary name)	ProxySBC	Default	Using Options	
	All other CID configuration can be left unchanged with their default values				

All other SIP configuration can be left unchanged with their default values.

#### > To configure a Proxy Address for SIP Trunk:

- Open the Proxy Sets table (Setup menu > Signaling & Media tab > Core Entities folder > Proxy Sets) and then click the Proxy Set SIPTrunk, and then click the Proxy Address link located below the table; the Proxy Address table opens.
- Click +New:
- Configure the address of the SIP Trunk according to the parameters described in the table below:

Table 2-18: Configuration Proxy Address for SIP Trunk

Index	Proxy Address	Transport Type	
0	SIPTrunk.com:5060 (SIP Trunk IP / FQDN and port)	UDP	
А	All other Proxy Addresses can be left unchanged with their default values.		

Click Apply.

#### ➤ To configure a Proxy Address for Proxy SBCs:

- Open the Proxy Sets table (Setup menu > Signaling & Media tab > Core Entities folder > Proxy Sets) and then click the Proxy Set ProxySBC, and then click the Proxy Address link located below the table; the Proxy Address table opens.
- Configure the address of the Proxy SBC according to the parameters described in the table below:

Table 2-19: Configuration Example: Proxy SBC Address

Index	Proxy Address	Transport Type	
0	{ProxySBC IP}:5061	TLS	
А	All other Proxy Addresses can be left unchanged with their default values.		

Click Apply.



## 2.6.5 Configure an IP Profiles

This section describes how to configure the IP Profiles for the SBC at the remote site. See Appendix C for a summary of all IP Profile configurations.

- To configure IP Profile to each remote site SBC:
- 1. Open the Proxy Sets table (**Setup** menu > **Signaling & Media** tab > **Coders & Profiles** folder > **IP Profiles**).
- 2. Click **+New** to add the IP Profile for the Microsoft Teams (through Proxy SBC). Configure the parameters using the table below as a reference.

Table 2-20: Configuration Example: Teams IP Profile (through the Proxy SBC)

Parameter	Value
General	
Name	ProxySBC (arbitrary name)
Media Security	
SBC Media Security Mode	Secured
SBC Media	
Extension Coders Group	AudioCodersGroups_1
ICE Mode	Lite
SBC Signaling	
Remote Update Support	Not Supported
Remote re-INVITE Support	Supported Only With SDP
Remote Delayed Offer Support	Not Supported
Remote Representation Mode	Replace Contact
SBC Forward and Transfer	
Remote REFER Mode	Handle Locally
Remote Replaces Mode	Handle Locally
Remote 3xx Mode	Handle Locally
SBC Hold	
Remote Hold Format	Inactive
All other paramete	ers can be left unchanged at their default values.

- 3. Click Apply.
- 4. Click **+New** to add the IP Profile for the SIP Trunk. Configure the parameters using the table below as reference.

Table 2-21: Configuration Example: SIP Trunk IP Profile (toward SIP Provider/ Media Gateway)

Parameter	Value
General	
Name	SIPTrunk (arbitrary name)
SBC Signaling	
P-Asserted-Identity Header Mode	Add (required for anonymous calls)

SBC Forward and Transfer		
Remote REFER Mode Handle Locally		
Remote Replaces Mode	Handle Locally	
Remote 3xx Mode Handle Locally		
All other parameters can be left unchanged at their default values.		

5. Click Apply.

## 2.6.6 Configure IP Groups

This section describes how to configure the IP Groups for the SBC in each remote site.

- > To configure an IP Groups:
- 1. Open the IP Groups table (**Setup** menu > **Signaling & Media** tab > **Core Entities** folder > **IP Groups**).
- 2. Click **+New** to add the IP Group for the SIP Trunk:

Table 2-22: Configuration Example: Site SBC IP Group towards SIP Trunk

Parameter	Value						
Name	SIPTrunk						
Туре	Server						
Proxy Set	SIPTrunk						
IP Profile	SIPTrunk						
Media Realm	MRLan or MRWan (according to your network environment)						
SIP Group Name (according to ITSP requirement)							
All other para	All other parameters can be left unchanged with their default values.						

- 3. Click Apply.
- 4. Click +New to add the IP Group towards Teams (through Proxy SBC) in the remote site SBC:

Table 2-23: Configuration Example: Site SBC IP Group towards Teams (through Proxy SBC)

Parameter	Value						
Name	ProxySBC (arbitrary name)						
Туре	Server						
Proxy Set	ProxySBC						
IP Profile	ProxySBC						
Media Realm	MRLan						
SIP Group Name {MSFT - CSOnlinePSTNGateway}							
All other parameters can be left unchanged with their default values.							

5. Click Apply.



## 2.6.7 Configure SRTP

Configuration of the SRTP done in the same way as in Proxy SBC. Please refer to Section 2.5.14 on page 44 above.

## 2.6.8 Configure IP-to-IP Call Routing Rules

This section describes how to configure IP-to-IP call routing rules on the remote site SBC.

- > To configure IP-to-IP routing rule:
- Open the IP-to-IP Routing table (Setup menu > Signaling & Media tab > SBC folder > Routing > IP-to-IP Routing).
- 2. Configure routing rules as shown in the table below:

Table 2-24: Site IP-to-IP Call Routing Rule

Index	Name	Source IP Group	Request Type	Call Triger	ReRoute IP Group	Dest Type	Dest IP Group	Internal Action
0	Terminate OPTIONS	Any	OPTIONS			Internal		Reply (Response='200')
1	Terminate Refer (arbitrary name)	Any	Any	REFER	ProxySBC	IP Group	ProxySBC	
2	Teams to SIP Trunk (arbitrary name)	ProxySBC				IP Group	SIPTrunk	
3	SIP Trunk to Teams (arbitrary name)	SIPTrunk				IP Group	ProxySBC	

# 2.7 Adapt Gateway to Work with Local Media Optimization

This section provides guidelines for configuring PSTN Gateway Application at the remote sites for supporting Local Media Optimization handling. In order to do this, SBC entities needed to be configured on the device.

#### Notes:



- This section is only relevant for implementation, where the remote site is populated with PSTN connectivity (through Gateway Application).
- The Gateway configuration can vary from customer to customer, therefore in this document, we only provide the configuration changes that are necessary to adopt the Gateway to work with Local Media Optimization.
- Device should be populated with the appropriate (SBC session and IP security) licenses.

## 2.7.1 Configure SBC SIP Signaling Interface

This section shows how to configure SBC SIP Signaling Interface. To configure SBC SIP interface:

- 1. Open the SIP Interfaces table (**Setup** menu > **Signaling & Media** tab > **Core Entities** folder > **SIP Interfaces**).
- Click +New to add SBC SIP Interface (if there is already a configured SIP Interface with Application Type 'SBC', this interface can be used). You can change some parameters according to your requirements.

Table 2-25: Configuration Example: Site SIP Interface

Index	Name	Name Application Type							
1	ProxySBC (arbitrary name)	SBC	5061 (arbitrary port)						
	All other SIP configuration can be left unchanged with their default values.								

## 2.7.2 Configure SBC Proxy Set

This section describes how to configure SBC Proxy Set towards Proxy SBC.

- 1. Open the Proxy Sets table (**Setup** menu > **Signaling & Media** tab > **Core Entities** folder > **Proxy Sets**).
- 2. Click +New to add the ProxySBC Proxy Set as shown in the table:

Table 2-26: Configuration Example: Site Proxy Set

Index	Name	SBC IPv4 SIP Interface	TLS Context Name	Proxy Keep-Alive			
1	ProxySBC	ProxySBC	Default	Using Options			
All other Proxy Sets can be left unchanged with their default values.							



## 2.7.3 Configure SBC Proxy Address

This section describes how to configure a Proxy address of the Proxy SBC.

- > To configure a Proxy Address for remote SBC:
- Open the Proxy Sets table (Setup menu > Signaling & Media tab > Core Entities folder > Proxy Sets) and then click the Proxy Set ProxySBC, and then click the Proxy Address link located below the table; the Proxy Address table opens.
- 2. Configure Proxy Set Address as shown in the table:

Table 2-27: Configuration Example: Site Proxy Address

Index	Proxy Address	Transport Type					
0	{ProxySBC IP}:5061	TLS					
Α	All other Proxy Addresses can be left unchanged with their default values.						

## 2.7.4 Configure SBC IP Profile

This section describes how to configure the IP Profile for the Proxy SBC.

- > To configure IP Profile to the Proxy SBC:
- Open the Proxy Sets table (Setup menu > Signaling & Media tab > Coders & Profiles folder > IP Profiles).
- 2. Click **+New** to add the IP Profile for the Teams Direct Routing interface through the Proxy SBC. Configure the parameters using the table below as reference.

Table 2-28: Configuration Example: Teams IP Profile (through the Proxy SBC)

Parameter	Value
General	
Name	ProxySBC (arbitrary name)
Media Security	
SBC Media Security Mode	Secured
SBC Media	
Extension Coders Group	AudioCodersGroups_1
ICE Mode	Lite
SBC Signaling	
Remote Update Support	Not Supported
Remote re-INVITE Support	Supported Only With SDP
Remote Delayed Offer Support	Not Supported
Remote Representation Mode	Add Routing Headers
SBC Forward and Transfer	
Remote REFER Mode	Handle Locally
Remote Replaces Mode	Handle Locally
Remote 3xx Mode	Handle Locally

SBC Hold					
Remote Hold Format	Inactive				
All other parameters can be left unchanged at their default values.					

## 2.7.5 Configure SBC IP Group

This section describes how to configure the IP group towards Proxy SBC in the remote site.

- > To configure an IP Group:
- 1. Open the IP Groups table (**Setup** menu > **Signaling & Media** tab > **Core Entities** folder > **IP Groups**).
- 2. Click +New to add the IP Group towards Proxy SBC:

Table 2-29: Configuration Example: Site IP Group

Parameter	Value					
Name	ProxySBC (arbitrary name)					
Туре	Server					
Proxy Set	ProxySBC					
IP Profile	ProxySBC					
Media Realm	(according to your network environment)					
SIP Group Name {MSFT - CSOnlinePSTNGateway}						
All other parameters can be left unchanged with their default values.						

## 2.7.6 Configure SBC IP-to-IP Routing Rule

This section describes how to configure IP-to-IP routing rule for calls from Teams (through Proxy SBC) to the Gateway Application.

The following IP-to-IP Routing Rules are defined:

- Terminate SIP OPTIONS messages on the SBC
- Terminate REFER messages to Teams Direct Routing
- Calls from Teams Direct Routing to the Gateway
- ➤ To configure SBC IP-to-IP routing rules:
- Open the IP-to-IP Routing table (Setup menu > Signaling & Media tab > SBC folder > Routing > IP-to-IP Routing).
- 2. Configure routing rules as shown in the table below:

Table 2-30: SBC IP-to-IP Routing Rules

Index	Name	Source IP Group	Request Type	Call Triger	ReRoute IP Group	Dest Type	Dest IP Group	Dest SIP Interface	Dest Address	Dest Port
0	Terminate OPTIONS	Any	OPTIONS			Dest Address			internal	
1	Refer from Teams (arbitrary name)	Any		REFER	ProxySBC	Request URI	ProxySBC			



Index	Name	Source IP Group	Request Type	Call Triger	ReRoute IP Group	Dest Type	Dest IP Group	Dest SIP Interface	Dest Address	Dest Port
2	Teams to GW (arbitrary name)	ProxySBC				Dest Address		{GW SIP Interface}	{GW IP Interface}	{GW SIP Interface port}

## 2.7.7 Configure Gateway Tel-to-IP Routing Rule

This section describes how to configure Gateway Tel-to-IP routing rule for routing calls from PSTN to Teams through Proxy SBC.

- To configure Tel-to-IP routing rules:
- 1. Open the Tel-to-IP Routing table (**Setup** menu > **Signaling & Media** tab > **Gateway** folder > **Routing** > **Tel** > **IP Routing**).
- 2. Click **New** and configure routing rule as shown in the table below:

Table 2-31: Gateway Tel-to-IP Routing Rule

Route Name	IP Profile	Dest IP Group Name						
GW to Teams (arbitrary name)	Teams	ProxySBC						
All other para	All other parameters can be left unchanged with their default values.							

# 3 Verify the Pairing Between the SBC and Direct Routing

After you have paired the SBC with Direct Routing using the *New-CsOnlinePSTNGateway* PowerShell command, validate that the SBC can successfully exchange OPTIONS with Direct Routing.

- To validate the pairing using SIP OPTIONS:
- 1. Open the Proxy Set Status page (Monitor menu > VolP Status tab> Proxy Set Status).
- 2. Find the Direct SIP connection and verify that 'Status' is online. If you see a failure, you need to troubleshoot the connection first, before configuring voice routing.

Figure 3-1: Proxy Set Status

				This page refreshes every 60 seconds					
PROXY SET ID	NAME	MODE	KEEP ALIVE	ADDRESS	PRIORITY	WEIGHT	SUCCESS		STATUS
0	ProxySet_0	Parking	Disabled						NOT RESOLVED
1	SIPTrunk	Parking	Enabled						ONLINE
				10.15.40.35(*)	-	-	1023	37	ONLINE
2	Teams	Load Balancing	Enabled						ONLINE
				sip.pstnhub.microsoft.com(52.114.75.24:5061)(*)	1	1.00	1	1	ONLINE
				sip2.pstnhub.microsoft.com(52.114.132.46:5061) (*)	2	1.00	1	0	ONLINE
				sip3.pstnhub.microsoft.com(52.114.7.24:5061)(*)	3	1.00	1	0	ONLINE



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## A Syntax Requirements for SIP Messages 'INVITE' and 'OPTIONS'

The syntax of SIP messages must conform with Direct Routing requirements.

This section covers the high-level requirements for the SIP syntax used in 'INVITE' and 'OPTIONS' messages. You can use the information presented here as a first step when troubleshooting unsuccessful calls. AudioCodes has found that most errors are related to incorrect syntax in SIP messages.

## A.1 Terminology

Must	Strictly required. The deployment does not function correctly without the correct configuration of these parameters.
------	--

## A.2 Syntax Requirements for 'INVITE' Messages

Figure A-1: Example of an 'INVITE' Message

```
INVITE sip:+97249888108@10.15.40.55;user=phone SIP/2.0
Via: SIP/2.0/TLS sbc.ACeducation.info:5068;alias;branch=z9hG4bKac496289557
Max-Forwards: 69
From: <sip:+97239762000@10.15.77.12>;tag=lc1642854452
To: <sip:+97249888108@10.15.40.55;user=phone>
Call-ID: 1167963076285201992217@ACeducation.info
CSeq: 1 INVITE
Contact: <sip:+97239762000@sbc.ACeducation.info:5068;transport=tls>
Supported: em,100rel,timer,replaces,path,resource-priority,sdp-anat
Allow: REGISTER,OPTIONS,INVITE,ACK,CANCEL,BYE,NOTIFY,PRACK,REFER,INFO,SUBSCRIBE,UPDATE
User-Agent: 10.15.40.55/v.7.20A.250.273
Content-Type: application/sdp
Content-Length: 1114
```

#### Contact header

- MUST: When placing calls to the Direct Routing interface, the 'CONTACT' header must have the SBC FQDN in the URI hostname
- Syntax: Contact: <phone number>@<FQDN of the SBC>:<SBC Port>;<transport type>
- If the parameter is not configured correctly, calls are rejected with a '403 Forbidden' message.



# A.3 Syntax Requirements for 'INVITE' Messages in Media Optimization

#### Figure A-2: Example of an 'INVITE' Message (External user)

```
(N 3129751) ---- Incoming SIP Message from 52.114.132.46:4736 to SIPInterface #0 (SIPInterface_0) TLS TO(#3107) SocketID(93) --
INVITE sip:+122225888@mosbc71.audctrunk.aceducation.info:5061;user=phone;transport=tls SIP/2.0
FROM: MO2<sip:+15551002@sip.psthhub.microsoft.com:5061;user=phone>;tag=98ecalf975c9499c95c2c9e66c317524
TO: <sip:+122225888@mosbc71.audctrunk.aceducation.info:5061;user=phone>
CSEQ: 1 INVITE
CALL-ID: 2245c7b314c152cdb897d5ce79cec2de
MAX-FORWARDS: 70
VIA: SIP/2.0/TLS 52.114.132.46:5061;branch=z9hG4bK12f46b6
RECORD-ROUTE: <sip:sip-du-a-us.psthhub.microsoft.com:5061;transport=tls;lr>
CONTACT: <sip:api-du-c-euwe.psthhub.microsoft.com:443;transport=tls;x-i=71634df0-a8ef-40c8-865b-85039cc5e0df;x-c=/v1/ngc/call/22
CONTENT-LENGTH: 1781
USER-AGENT: Microsoft.PSTNHub.SIPProxy v.2019.11.28.2 i.USEA.4
CONTENT-TYPE: application/sdp
ALLOW: INVITE,ACK,OPTIONS,CANCEL,BYE,NOTIFY
P.ACCEDITED.TDENTITY: <tel:+15551002>,<sip:mo2@mo.audctrunk.aceducation.info>
PRIVACY: id

X-MS-MediaPath: mosbc.audctrunk.aceducation.info,mosbc71.audctrunk.aceducation.info
```

#### Figure A-3: Example of an 'INVITE' Message (Internal User)

- Privacy header removed
- X-MS- headers receive by Teams

# A.4 Syntax Requirements for 'INVITE' Messages in site for Media Optimization

Figure A-4: Example of an 'INVITE' Message From Site to Teams

```
(N 2672058) ---- Outgoing SIP Message to 52.114.75.24:5061 from SIPInterface #0 (SIPInterface INVITE sip:+15551002@mosbc71.audctrunk.aceducation.info;user=phone SIP/2.0
Via: SIP/2.0/TLS mosbc.audctrunk.aceducation.info:5061;alias;branch=z9hG4bKac275253736
Max-Forwards: 68
From: <sip:+122225888@10.15.40.29>;tag=1c1490933092
To: <sip:+15551002@mosbc71.audctrunk.aceducation.info;user=phone>
Call-ID: 11184345432212201913191@mosbc.audctrunk.aceducation.info
CSeq: 1 INVITE
Contact: <sip:+122225888@mosbc71.audctrunk.aceducation.info;5061;transport=tls>
Supported: em,100rel,timer,reptaces,patn,resource-priority,sdp-anat
Allow: REGISTER,OPTIONS,INVITE,ACK,CANCEL,BYE,NOTIFY,PRACK,REFER,INFO,SUBSCRIBE,UPDATE
User-Agent: Mediant VE SBC/v.7.20A.254.975
Content-Type: application/sdp
Content-Length: 1109
```

Contact header with Source Site FQDN

## A.5 Requirements for 'OPTIONS' Messages Syntax

Figure A-5: Example of 'OPTIONS' message

```
OPTIONS sip:195.189.192.171 SIP/2.0

Via: SIP/2.0/TLS sbc.ACeducation.info:5068;alias;branch=z9hG4bKac1385438539

Max-Forwards: 70

From: <sip:195.189.192.171>;tag=1c1890841146

To: <sip:195.189.192.171>
Call-ID: 59585523229520193103@ACeducation.info
CSeq: 1 OPTIONS
Contact: <sip sbc.ACeducation.info:5068;transport=tls>
Allow: REGISTER,OPTIONS,INVITE,ACK,CANCEL,BYE,NOTIFY,PRACK,REFER,INFO,SUBSCRIBE,UPDATE
User-Agent: 10.15.40.55/v.7.20A.250.273
Accept: application/sdp, application/simple-message-summary, message/sipfrag
Content-Length: 0
```

#### Contact header

- MUST: When sending OPTIONS to the Direct Routing interface, the 'CONTACT' header must have the SBC FQDN in the URI hostname
- Syntax: Contact: <phone number>@<FQDN of the SBC>:<SBC Port>;<transport type>
- If the parameter is not configured correctly, the calls are rejected with a '403 Forbidden' message

The table below shows where in the Web interface the parameters are configured and where in this document you can find the configuration instructions.

Table A-1: Syntax Requirements for an 'OPTIONS' Message

Parameter	Where Configured	How to Configure
Contact	Setup > Signaling and Media > Core Entities > IP Groups> <group name=""> &gt; Local Host Name</group>	See Section 2.12.
	In IP Group, 'Contact' must be configured. In this field ('Local Host Name'), define the local host name of the SBC as a string, for example, sbc.ACeducation.info. The name changes the host name in the call received from the IP Group.	



## A.6 Connectivity Interface Characteristics

The table below shows the technical characteristics of the Direct Routing interface.

In most cases, Microsoft uses RFC standards as a guide during development, but does not guarantee interoperability with SBCs - even if they support all the parameters in the table below - due to the specifics of the implementation of the standards by SBC vendors.

Microsoft has a partnership with some SBC vendors and guarantees their devices' interoperability with the interface. All validated devices are listed on Microsoft's website. Microsoft only supports devices that are validated in order to connect to the Direct Routing interface.

AudioCodes is one of the vendors who are in partnership with Microsoft.

AudioCodes' SBCs are validated by Microsoft to connect to the Direct Routing interface.

Table A-2: Teams Direct Routing Interface - Technical Characteristics

Category	Parameter	Value	Comments
Ports and IP ranges	SIP Interface FQDN Name	See Microsoft's document Deploying Direct Routing Guide.	-
	IP Addresses range for SIP interfaces	See Microsoft's document Deploying Direct Routing Guide.	-
	SIP Port	5061	-
	IP Address range for Media	See Microsoft's document Deploying Direct Routing Guide.	-
	Media port range on Media Processors	See Microsoft's document Deploying Direct Routing Guide.	-
	Media Port range on the client	See Microsoft's document Deploying Direct Routing Guide.	-
Transport	SIP transport	TLS	-
and Security	Media Transport	SRTP	-
•	SRTP Security Context	DTLS, SIPS  Note: Support for DTLS is pending. Currently, SIPS must be configured. When support for DTLS will be announced, it will be the recommended context.	https://tools.ietf.org/html/rfc57
	Crypto Suite	AES_CM_128_HMAC_ SHA1_80, non-MKI	-
	Control protocol for media transport	SRTCP (SRTCP-Mux recommended)	Using RTCP MUX helps reduce the number of required ports

Category	Parameter	Value	Comments
	Supported Certification Authorities	See the Deployment Guide	-
	Transport for Media Bypass (of configured)	<ul> <li>ICE-lite (RFC5245)         <ul> <li>recommended</li> </ul> </li> <li>Client also has         <ul> <li>Transport Relays</li> </ul> </li> </ul>	-
	Audio codecs	<ul> <li>G711</li> <li>Silk (Teams clients)</li> <li>Opus (WebRTC clients) - only if Media Bypass is used</li> <li>G729</li> </ul>	-
Codecs	Other codecs	<ul> <li>CN</li> <li>Required narrowband and wideband</li> <li>RED - Not required</li> <li>DTMF - Required</li> <li>Events 0-16</li> <li>Silence Suppression - Not required</li> </ul>	-



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# **B** SIP Proxy Direct Routing Requirements

Teams Direct Routing has three FQDNs:

- sip.pstnhub.microsoft.com [Global FQDN. The SBC attempts to use it as the first priority region. When the SBC sends a request to resolve this name, the Microsoft Azure DNS server returns an IP address pointing to the primary Azure datacenter assigned to the SBC. The assignment is based on performance metrics of the datacenters and geographical proximity to the SBC. The IP address returned corresponds to the primary FQDN.]
- **sip2.pstnhub.microsoft.com** [Secondary FQDN. Geographically maps to the second priority region.]
- sip3.pstnhub.microsoft.com [Tertiary FQDN. Geographically maps to the third priority region.]

These three FQDNs must be placed in the order shown above to provide optimal quality of experience (less loaded and closest to the SBC datacenter assigned by querying the first FQDN).

The three FQDNs provide a failover if a connection is established from an SBC to a datacenter that is experiencing a temporary issue.

### **B.1** Failover Mechanism

The SBC queries the DNS server to resolve **sip.pstnhub.microsoft.com**. The primary datacenter is selected based on geographical proximity and datacenters performance metrics.

If during the connection the primary datacenter experiences an issue, the SBC will attempt **sip2.pstnhub.microsoft.com** which resolves to the second assigned datacenter, and in rare cases if datacenters in two regions are unavailable, the SBC retries the last FQDN (**sip3.pstnhub.microsoft.com**) which provides the tertiary datacenter IP address.

The SBC must send SIP OPTIONS to all IP addresses that are resolved from the three FQDNs, that is, sip.pstnhub.microsoft.com, sip2.pstnhub.microsoft.com and sip3.pstnhub.microsoft.com.



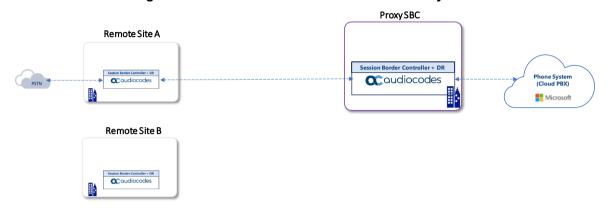
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# **C** Configuration Quick Guidelines

This appendix provides quick guidelines for configuring of the SBC's (Proxy SBC and the remote sites SBC's) to support Local Media Optimization.

## C.1 Proxy SBC Scenario Topology

Figure C-1: IP Profile for Remote Sites and Proxy SBC



## C.2 SIP Interface

Table C-1: SIP Interface Proxy SBC Configuration Summary

Index	Name	Network Interface	Application Type	UDP Port	TCP Port	TLS Port	Enable TCP Keepalive	Classification Failure Response Type	Media Realm	TLS Context Name
0	SiteSIPInt erface (arbitrary name)	LAN_IF	SBC	0	0	5061 (accordi ng to site requirem ent)	Disable (leave default value)	500 (leave default value)	MRLan	-
1	Teams (arbitrary name)	WAN_IF	SBC	0	0	5061 (as configur ed in the Office 365)	Enable	0 (Recommend ed to prevent DoS attacks)	MRWan	Teams



Table C-2: SIP Interface Remote SBC Configuration Summary

Index	Name	Network Interface	Application Type	UDP Port	TCP Port	TLS Port	Enable TCP Keepalive	Classification Failure Response Type	Media Realm	TLS Context Name
0	SIPTrunk (arbitrary name)	WAN_IF	SBC	5060 (according to Service Provider requirement)	0	0	Disable (leave default value)	0 (Recommended to prevent DoS attacks)	MRWan	-
1	ProxySBC (arbitrary name)	LAN_IF	SBC	0	0	5061	Enable	500 (leave default value)	MRLan	-

## C.3 Proxy Set

Table C-3: Proxy Set Proxy SBC Configuration Summary

Index	Name	SBC IPv4 SIP Interface	TLS Context Name	Proxy Keep- Alive	Proxy Hot Swap	Proxy Load Balancing Method	Proxy Address	Transport Type	Proxy Priority
1	Teams (arbitrary name)	Teams	Teams	Using Options	Enable	Random Weights	sip.pstnhub.micros oft.com:5061 sip2.pstnhub.micr osoft.com:5061 sip3.pstnhub.micr osoft.com:5061	TLS TLS TLS	1 2 3
2	SiteA (arbitrary name)	SiteSIPInterface (arbitrary name)	Default	Using Options	-	-	192.168.1.5:5061 (IP address of the SiteA SBC)	TLS	
3	SiteB (arbitrary name)	SiteSIPInterface (arbitrary name)	Default	Using Options	-	-	192.168.2.5:5061 (IP address of the SiteB SBC)	TLS	

### Table C-4: Proxy SET Remote SBC Configuration Summary

Index	Name	SBC IPv4 SIP Interface	TLS Context Name	Proxy Keep- Alive	Proxy Hot Swap	Proxy Load Balancing Method	Proxy Address	Transport Type	Proxy Priority
1	SIPTrunk (arbitrary name)	SIPTrunk	Default	Using Options	-	-	SIPTrunk.com:5060	UDP	-
2	ProxySBC (arbitrary name)	ProxySBC	Default	Using Options	-	-	{ProxySBC IP}:5061	TLS	-

## C.4 IP Profile

**Table C-5: IP Profile Configuration Summary** 

	Remo	te SBC	Proxy	y SBC			
Parameter	Value	Value	Value	Value			
General							
Name	SIPTrunk (toward SIP Provider/ MGW appl)	ProxySBC	SiteA (toward site A Remote SBC)	Teams			
SBC Media S	ecurity						
SBC Media Security Mode	Secured (should be synchronized with SIP provider)	Not Secured (according to customer needs)	Not Secured (according to customer needs)	Secured			
SBC Early M	edia						
Remote Early Media RTP Detection Mode  By Signaling (Default)		By Signaling (Default)	By Signaling (Default)	By Media (required, as Teams Direct Routing does not send RTP immediately to remote side when it sends a SIP 18x response)			
SBC Media							
Extension Coders Group	AudioCodersGro ups_1	AudioCodersGro ups_1	AudioCodersGro ups_1	AudioCodersGro ups_1			
RTCP Mode	Transparent (Default)	Transparent (Default)	<b>Transparent</b> (Default)	Generate Always (required, as some ITSPs do not send RTCP packets during Hold, but Microsoft expects them)			
ICE Mode	<b>Disable</b> (Default)	Lite	Disable (Default)	Lite			
SBC Signalir	SBC Signaling						
P-Asserted- Identity Header Mode	Add (required for anonymous calls)	As Is (Default)	As Is (Default)	As Is (Default)			
SIP UPDATE Support	Supported (Default)	Not Supported	Supported (Default)	Not Supported			



	Remot	te SBC	Proxy	SBC		
Remote re- INVITE Support	Supported (Default)	Supported Only With SDP	Supported (Default)	Supported Only With SDP		
Remote Delayed Offer Support	Supported (Default)	Not Supported	Supported (Default)	Not Supported		
Remote Representa tion Mode	According to Operation Mode (Default)  Replace Contact		According to Operation Mode (Default)	Add Routing Headers		
SBC Forward	SBC Forward and Transfer					
Remote REFER Mode	Handle Locally	Handle Locally	Regular (Default)	Regular (Default)		
Remote Replaces Mode	Handle Locally	Handle Locally	Standard (Default)	Standard (Default)		
Remote 3xx Mode	Handle Locally	Handle Locally	Transparent (Default)	Transparent (Default)		
SBC Hold						
Remote Hold Format	Transparent (Default)	Inactive	Transparent (Default)	Inactive		
	All other parameters can be left unchanged at their default values.					

# C.5 IP Group

Table C-6: IP Group Proxy SBC toward Teams Configuration Summary

Parameter	Value
Name	Teams
Туре	Server
Proxy Set	Teams
IP Profile	Teams
Media Realm	MRWan
	MRLan
Internal Media Realm	This parameter is relevant when the 'Teams Local Media Optimization Handling' parameter (see below) is configured to any value other than "None" and the X-MS-UserLocation header in the incoming SIP message is set to 'Internal'. In this case, the Internal Media Realm determines the UDP port range and maximum sessions for Media traffic on this IP interface.

Parameter	Value	
	If X-MS-UserLocation=Internal response is received from Teams, a new IP address/port is allocated using the Internal Media Realm only if the call is non-direct media i.e. media traverses the paired SBC to the remote SBCs.	
Classify by Proxy Set	Disable	
Local Host Name	<fqdn enterprise="" in="" name="" of="" sbc="" tenant="" the=""> (For example, sbc.ACeducation.info defines the host name (string) that the device uses in the SIP message's Via and Contact headers. This defines the FQDN as the host name that is recognized by Microsoft Teams. The device uses this string for Via and Contact headers in outgoing INVITE messages sent to a specific IP Group, and the Contact header in SIP 18x and 200 OK responses for incoming INVITE messages received from the other configured IP Groups (SiteA and SiteB).</fqdn>	
Always Use Src Address	Yes	
Teams Local Media Optimization Handling	<b>Teams Decides</b> (The routing decision is made according to the Microsoft Teams headers for the primary route)	
Teams Local Media Optimization Initial Behavior	<ul> <li>This parameter is relevant for inbound calls to Teams when "Teams Local Media Optimization Handling" is set to "Teams Decides" or "SBC Decides":</li> <li>Direct Media (default) – Perform direct media call towards Teams.</li> <li>Internal - Perform non-direct media call (media traverses the paired SBC from the remote SBC) towards Teams using Internal Media Realm.</li> <li>External – Perform non-direct media call (media traverses the paired SBC from the remote SBC) towards Teams using external (regular) Media Realm.</li> <li>Note: The value of this parameter can be variable depending on particular setup</li> </ul>	
Proxy Keep-Alive using IP Group settings	Enable	
Inbound Message Manipulation Set	0	
Outbound Message Manipulation Set	1	
Call Setup Rules Set ID 0		



Table C-7: IP Group Proxy SBC toward Remote SBC's Configuration Summary

Parameter	Value		
Name	SiteA/SiteB		
Туре	Server		
Proxy Set	SiteA/SiteB		
IP Profile	SiteA/SiteB		
Media Realm	MRLan		
Tags  Site={RemotePSTNGateWayFQDN} The Site Tag should be defined as the remote site SBC's FQDN and should be discoverable by DNS from the Proxy SBC.			
All other parameters can be left unchanged with their default values.			

Table C-8: IP Group Remote SBC toward Proxy SBC Configuration Summary

Parameter	Value			
Name	ProxySBC (arbitrary name)			
Topology Location	Down			
Туре	Server			
Proxy Set	ProxySBC			
IP Profile ProxySBC				
Media Realm	MRLan			
SIP Group Name {MSFT - CSOnlinePSTNGateway}				
All other parameters can be left unchanged with their default values.				

Table C-9: IP Group Remote SBC toward SIP Trunk (PSTN) Configuration Summary

Parameter	Value			
Name	SIPTrunk (arbitrary name)			
Topology Location	Down			
Туре	Server			
Proxy Set	SIPTrunk			
IP Profile	SIPTrunk			
Media Realm	MRWan			
Classify by Proxy Set	Enable			
SIP Group Name	(according to ITSP requirement)			
All other parameters can be left unchanged with their default values.				

# C.6 IP-To-IP Routing

Table C-10: IP-To-IP Routing in the Proxy SBC

Index	Name	Source IP Group	Request Type	Dest Type	Dest IP Group	Routing Tag Name	Internal Action
0	Terminate OPTIONS	Any	OPTIONS	Internal			Reply (Response='200')
1	Teams to SIP Trunk (arbitrary name)	Teams		Destination Tag		Site	
2	SIP Trunk to Teams (arbitrary name)	Any		IP Group	Teams		

### Table C-11: IP-To-IP Routing in the Remote Site SBC

Index	Name	Source IP Group	Request Type	Call Triger	ReRoute IP Group	Dest Type	Dest IP Group	Dest Address
0	Terminate OPTIONS	Any	OPTIONS			Internal		Reply (Response='200')
1	Terminate Refer (arbitrary name)	Any	Any	REFER	ProxySBC	IP Group	ProxySBC	
2	Teams to SIP Trunk (arbitrary name)	ProxySBC				IP Group	SIPTrunk	
3	SIP Trunk to Teams (arbitrary name)	SIPTrunk				IP Group	ProxySBC	



# **C.7** Message Manipulations

Table C-12: Proxy SBC Message Manipulation Index 0

Parameter	Value
Index	0
Name	Privacy Header
Manipulation Set ID	0
Condition	Header.Privacy contains 'id'
Action Subject	Header.Privacy
Action Type	Remove

Table C-13: Proxy SBC Message Manipulation Index 1

Parameter	Value
Index	1
Name	Replace Host in Contact
Manipulation Set ID	1
Message Type	Invite.Request
Action Subject	Header.Contact.URL.Host
Action Type	Modify
Action Value	Header.To.URL.Host

# D AudioCodes ARM and SBCs with Teams Direct Local Media Optimization

This appendix describes how to provision all the system components involved in the ARM and SBC solution for Teams Direct Routing Local Media Optimization.

## D.1 About AudioCodes Routing Manager (ARM)

The ARM is a LINUX-based, software-only, telephony management product which expedites and streamlines IP telephony routing for enterprises with multiple globally distributed branches. The ARM determines the quickest, least expensive, and best call quality routes in packet networks. Routing data, previously located on the SBC, Unified Communications (UC) application (e.g., Microsoft's Skype for Business), or Media Gateway, is now located on the ARM server. If an enterprise has an SBC in every branch, a single ARM, deployed in HQ, can route all calls in the globally distributed corporate network to PSTN, the local provider, enterprise headquarters, or to the IP network. Routing rules, configured by the IT manager in the ARM's Routing Table, perform the routing.

If an enterprise has only one or two branches, its IT manager can easily independently implement maintenance changes. In globally distributed enterprises, IT managers until now had to laboriously implement changes, multiple times, per branch. With the ARM, IT managers implement changes only once, saving significant labor and time resources and costs.

### **D.2** Solution Overview

In Teams Direct Local Media Optimization, ARM handles call signaling for all solution SBCs (Proxy SBC and Remote SBCs). Remote SBCs handle the Local Media Optimization business logic. ARM significantly simplifies the provisioning of connectivity between the Proxy SBC and Regional SBC and visualizes the topology.

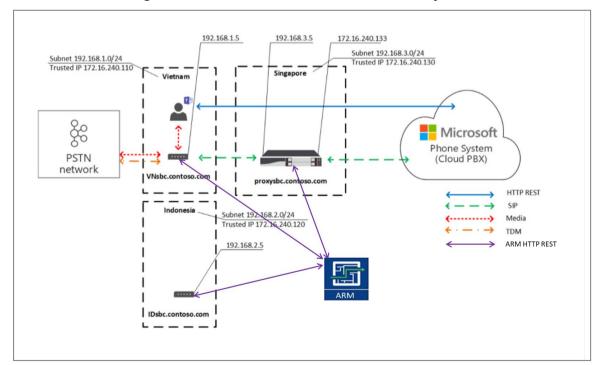


Figure D-1: IP Profile for Remote Sites and Proxy SBC



The following sections describe the exact steps for configuring:

- Proxy and Remote SBCs
- ARM Nodes, Connections, Routing Rules

## D.3 Configuration of the SBCs

#### Notes:



- Validate that your AudioCodes' Mediant SBCs are loaded with the correct firmware version (7.20A.258.354 or later).
- The following sections assumes that an SBC configuration is deployed in production working with Local Media Optimization and you wish to add support for working with ARM.

### D.3.1 Configuring Proxy SBC for Local Media Optimization (LMO)

This section describes the configuration required for supporting Local Media Optimization handling on the **Proxy SBC**.

- To configure LMO on Proxy SBC:
- 1. Configure the IP Interfaces as described in Section 2.5.4.
- 2. Configure the TLS Context as described in Section 2.5.5.
- 3. Generate and install a certificate as described in Section 2.5.6.
- Load the Baltimore Trusted Root Certificates as described in Section 2.5.7.
- Open the SRDs table (Setup menu > Signaling & Media tab > Core Entities folder > SRDs) and set the 'Used by Routing Server' parameter to 'Used' for all SRDs.



**Note:** If the SBC is already provisioned to work with Local Media Optimization, and the solution is being extended to work with ARM, set the 'Used by Routing Server' parameter to 'Used' for all SRDs.

Configure the Media Realms as described in Section 2.5.8 and set the 'Used by Routing Server' parameter to 'Used' for both Media Realms.



**Note:** If the SBC is already provisioned to work with Local Media Optimization and the solution is being extended to work with ARM, set the 'Used by Routing Server' parameter to 'Used' for both Media Realms.

7. Configure SIP Interfaces as described in Section 2.5.9 and set the 'Used by Routing Server' parameter to 'Used' for both interfaces.



**Note:** If SBC is already provisioned to work with Local Media Optimization, and the solution is being extended to work with ARM, set the 'Used by Routing Server' parameter to 'Used' for both interfaces.

- 8. Add a Proxy Set towards Teams as described in Section 2.5.10.
  - a. Configure the Proxy Set according to Table 2-6 (Teams).
  - b. Configure the Proxy Address for this Proxy Set as described in Table 2-7.

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**Configuration Note** 

**Note:** If SBC is already provisioned to work with Local Media Optimization, and the solution is being extended to work with ARM, delete all Proxy Sets towards sites.

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**9.** Configure the Coder Groups as described in Section 2.5.11.

- 10. Add an IP Profiles for Teams as described in Section 2.5.12 according to Table 2-9.
- **11.** Add an IP Group toward Teams as described in Section 2.5.13 according to Table 2-11 and set the 'Used by Routing Server' parameter to 'Used'.



**Note:** If SBC is already provisioned to work with Local Media Optimization, and the solution is being extended to work with ARM, set the 'Used by Routing Server' parameter to 'Used' and delete all IP Groups towards sites.

- **12.** Configure the SRTP as described in Section 2.5.14.
- 13. Add a Message Condition as described in Section 2.5.15.
- 14. Add a Classification Rules as described in Section 2.5.16.
- 15. Add Message Manipulations as described in Chapter 2.5.18.
- 16. Open the IP-to-IP Routing table (Setup menu > Signaling & Media tab > SBC folder > Routing > IP-to-IP Routing).
- 17. Insert a new row to work with the ARM (should be before the other INVITE rules):

Index	Name	Source IP Group	Request Type	Destination Type	
1	ARM	Any	INVITE	Routing Server	

# D.3.2 Configuring Remote Site SBCs for Local Media Optimization (LMO)

This section describes the configuration required for supporting Local Media Optimization handling on the **remote site** SBCs.

- To configure remote site SBCs for LMO:
- 1. Configure the IP Interfaces as described in Section 2.6.1.
- 2. Open the SRDs table (**Setup** menu > **Signaling & Media** tab > **Core Entities** folder > **SRDs**) and set the 'Used by Routing Server' parameter to 'Used' for all SRDs.



**Note:** If the SBC is already provisioned to work with Local Media Optimization, and the solution is now being extended to work with ARM, set the 'Used by Routing Server' parameter to 'Used' for all SRDs.

3. Configure the Media Realms as described in Section 2.6.2 and set the 'Used by Routing Server' parameter to 'Used' for both Media Realms.



**Note:** If the SBC is already provisioned to work with Local Media Optimization, and the solution is now being extended to work with ARM, set the 'Used by Routing Server' parameter to 'Used' for both Media Realms.

**4.** Configure SIP Interfaces as described in Section 2.6.3 and set 'Used by Routing Server' parameter to 'Used' for both interfaces.



**Note:** If the SBC is already provisioned to work with Local Media Optimization, and the solution is now being extended to work with ARM, set the 'Used by Routing Server' parameter to 'Used' for both interfaces.

- 5. Add a Proxy Set toward SIP Trunk as described in Section 2.6.4.
  - a. Configure the Proxy Set according to Table 2-17.
  - **b.** Configure the Proxy Address for this Proxy Set as described in Table 2-18.



**Note:** If SBC is already provisioned to work with Local Media Optimization, and now solution is extended with ARM, you have to delete Proxy Set towards Proxy SBC.



- 6. Add an IP Profiles for SIP Trunk as described in Section 2.6.5 according to Table 2-21.
- 7. Add an IP Group toward SIP Trunk as described in Section 2.6.6 according to Table 2-22 and set the 'Used by Routing Server' parameter to 'Used'.



**Note:** If the SBC is already provisioned to work with Local Media Optimization, and the solution is being extended to work with ARM, set the 'Used by Routing Server' parameter to 'Used' and delete IP Group towards Proxy SBC.

- 8. Configure the SRTP as described in Section 2.6.7.
- Open the IP-to-IP Routing table (Setup menu > Signaling & Media tab > SBC folder > Routing > IP-to-IP Routing).
- 10. Insert a new row to work with the ARM (should be before the other INVITE rules):

Index	Name	Source IP Group	Request Type	Destination Type
1	ARM	Any	INVITE	Routing Server

# **D.4** ARM Configuration

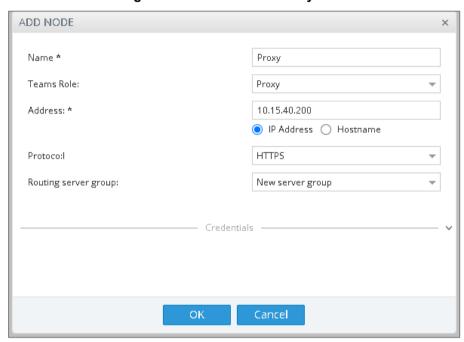
This section describes how to configure the ARM Web interface.

## **D.4.1 Defining SBC Nodes**

This section describes how to define SBC nodes.

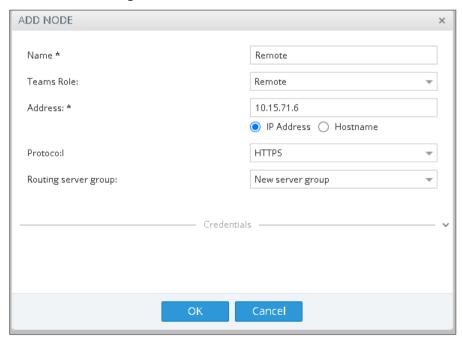
- > To define SBC nodes:
- 1. In ARM GUI Interface, add an AC Node for the Proxy\_sbc.

Figure D-2: AC Node for Proxy SBC



2. Add an AC Node for the Remote\_SBC.

Figure D-3: AC Node for Remote SBC





3. Unlock the Peer Connections. Wait for Sync. Nodes will be enabled.

Figure D-4: Enable Nodes

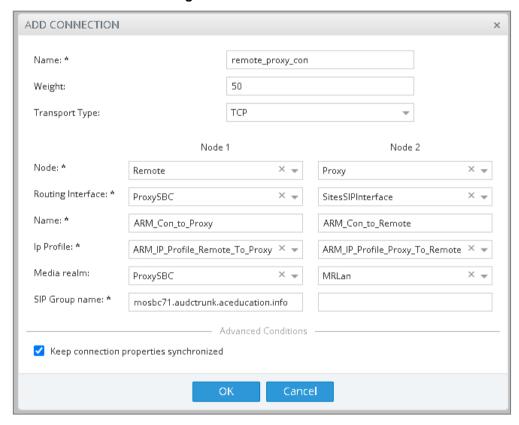


### **D.4.2 Defining Connection**

This section describes how to define the connection between the remote SBC and the proxy SBC.

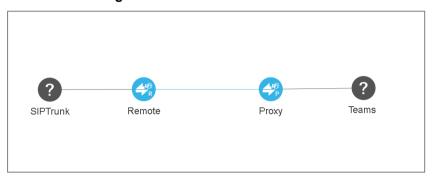
- ➤ To define connection between the remote SBC and the proxy SBC:
- 1. Drag a Connection from the Remote\_sbc to the Proxy\_sbc.
- 2. Select the protocol type, Routing Interface, Name, Ip Profile, Media Realm for both Nodes.
- 3. Configure Sip Group Name for the Remote Node.

Figure D-5: Add Connection



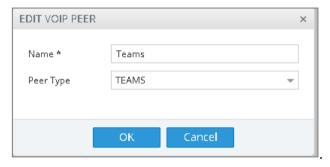
4. Wait for sync. The Connection will be enabled.

Figure D-6: Established Connection



5. Edit the Teams Voip-Peer and select TEAMS.

Figure D-7: Teams Voip-Peer

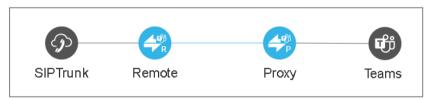


6. Edit the SIPTrunk Voip-Peer and select SIP\_TRUNK.

Figure D-8: SIPTrunk VoIP-Peer



Figure D-9: Established Connection





## **D.4.3 Defining Routing Rules**

This section describes how to define routing rules.

### D.4.3.1 Calls from Teams

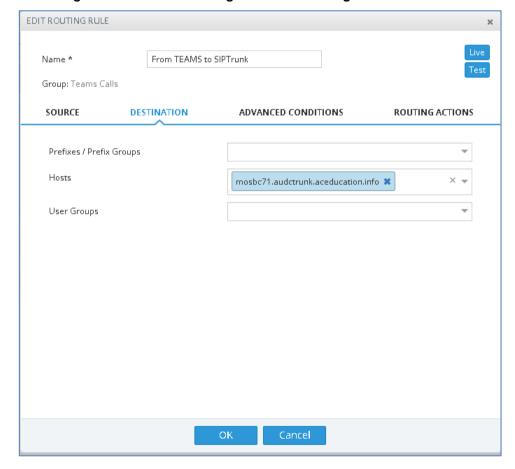
This section describes how to define routing rules for calls from Teams.

- To define a routing rule for calls from Teams.
- 1. Add a Routing Rule for incoming call from Teams.

```
Name = From TEAMS to SIPTrunk
Source Peer Connection = The Peer Connection of Proxy_sbc toward
Teams
Destination Host = {MSFT - CSOnlinePSTNGateway} of Remote_sbc
Routing Action = The Peer Connection of Remote_sbc toward
SIPTrunk
```

2. Click **Live** to activate the routing rule; the rule is now activated in the ARM.

Figure D-10: Add a Routing Rule for Incoming Call from Teams



### D.4.3.2 Calls to Teams

This section describes how to define routing rules for calls to Teams.

- To define a routing rule for calls to Teams:
- 1. Add a Routing Rule for incoming call from SIPTrunk.

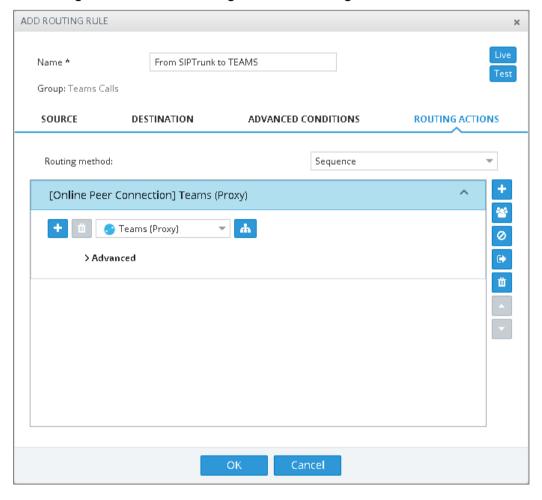
```
Name = From SIPTrunk to TEAMS

Source Peer Connection = The SIPTrunk Peer Connection of Remote_sbc

Routing Action = The Peer Connection of Proxy sbc toward TEAMS
```

- 2. Define the characteristics of the route request, e.g., the User Group and phone prefix of the originator/destination.
- 3. Click **Live** to activate the routing rule; the rule is now activated in the ARM.

Figure D-11: Add a Routing Rule for Incoming Call from SIP Trunk



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