

Ribbon SBC Core 5K_7K_SWe R9.0 Interop with Zoom BYOC : Interoperability Guide



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Interoperable Vendors



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Document Overview

This document outlines the configuration best practices for the Ribbon SBC Core (SBC 5K, 7K, SWe) when deployed with Zoom Bring Your Own Carrier (BYOC). This means that for all subscribers catering to Zoom customers, the PSTN calls terminating through the local SBC Core are directly connected to the Service Provider of their choice.

A Session Border Controller (SBC) is a network element deployed to protect SIP-based Voice over Internet Protocol (VoIP) networks. Early deployments of SBCs were focused on the borders between two service provider networks in a peering environment. This role has now expanded to include significant deployments between a service provider's access network and a backbone network to provide service to residential and/or enterprise customers. The interoperability compliance testing focuses on verifying inbound and outbound call flows between Ribbon SBC 5K/7K/SWe and Zoom cloud. Ribbon SBC 5K/7K/SWe is deployed on the customer site to resolve any potential numbering format issue between Zoom and the customer's existing carrier dial plan numbering.

This guide contains the following configuration sections:

- [Section A: SBC Core Configuration](#)
 - Captures general SBC Core configurations for deploying with Zoom BYOC.
- [Section B: Configuration for SBC behind NAT](#)
 - Captures additional SBC configuration performed behind NAT.
- [Section C: Zoom Web BYOC configuration](#)
 - Captures the Zoom BYOC configuration.
 - Test all basic calls, along with the supplementary features like call hold, call transfer, and conference with configurations from Section A and Section B.
 - Configure Advanced supplementary features on Zoom as mentioned in [Supplementary Services Configuration on Zoom](#). These include:
 - Auto Receptionist
 - Call Flip
 - Shared Line Appearance (SLA) or Call Delegation
 - Shared Line Group (SLG)



Note

SBC 5x10, 5400, 7000 and SWe are represented as SBC Core in the following sections.

Non-Goals


It is not the goal of this guide to provide detailed configurations that will meet the requirements of every customer. Use this guide as a starting point and build the SBC configurations in consultation with network design and deployment engineers.

Audience

This is a technical document intended for telecommunications engineers with the purpose of configuring both the Ribbon SBCs and the third-party product.

Steps will require navigating the third-party product as well as the Ribbon product using graphical user interface (GUI) or command line interface (CLI).

Understanding of the basic concepts of TCP/UDP/TLS, IP/Routing, and SIP/RTP/SRTP is needed to complete the configuration and any necessary troubleshooting.

**Note**

This configuration guide is offered as a convenience to Ribbon customers. The specifications and information regarding the product in this guide are subject to change without notice. All statements, information, and recommendations in this guide are believed to be accurate but are presented without warranty of any kind, express or implied, and are provided “AS IS”. Users must take full responsibility for the application of the specifications and information in this guide.

Pre-Requisites

The following aspects are required before proceeding with the interop:


- Ribbon SBC 5K /7K/SWe Core
- Ribbon PSX (if using external PSX instead of ERE (Embedded Routing Engine)
- Public IP Addresses
- Zoom BYOC (Bring Your Own Carrier) Trunk
 - Zoom Go account is required.
 - For more details, visit <https://go.zoom.us/signin>
- TLS Certificates for SBC 5K /7K/SWe Core
 - Please refer to [TLS Configuration between Ribbon SBC Core and Zoom](#)

Product and Device Details

The following equipment and software were used for the sample configuration provided:

Table 1: Requirements

	Equipment	Software Version
Ribbon Communications	Ribbon SBC 5K /7K/SWe Core	V09.00.00R0
	Ribbon PSX	V12.02.02R000
Zoom	Zoom Desktop app	5.0.5 (26213.0602)
	Zoom Mobile app	5.0.5 (26211.0602)
Third-party Equipment	Kapanga Softphone	1.00
	Phonerlite	2.77
	Zoiper	5.3.8

**Note**

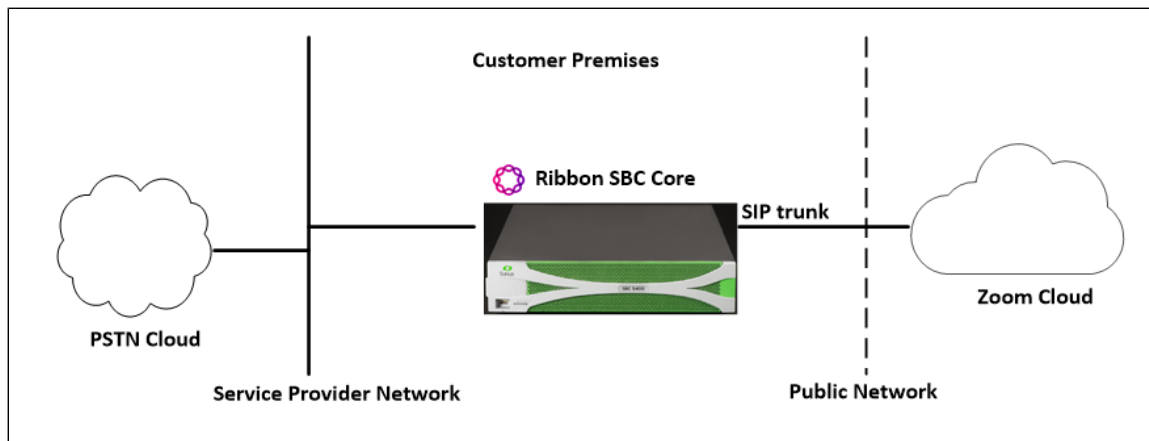
The Ribbon SBC Core portfolio includes SBC 5x10, SBC 5400, SBC 7000 (appliance based), and SBC SWe (virtualized platform). The software version is applicable to Ribbon SBC Core portfolio, and hence, this configuration guide is valid for all these devices.

Network Topology Diagram

This section covers the SBC Core deployment topology and the Interoperability Test Lab Topology.

SBC Core Deployment Topology

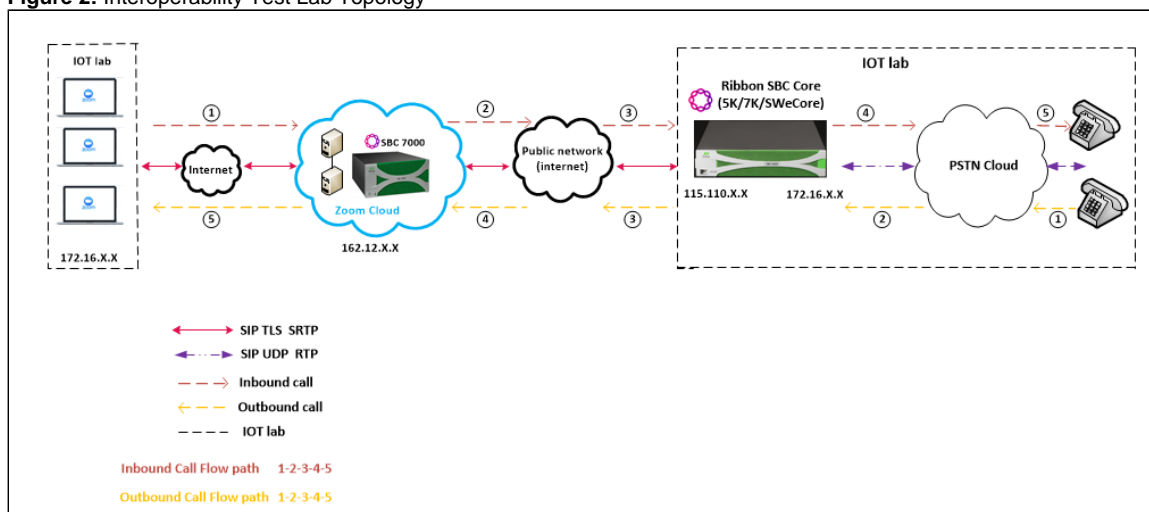
Figure 1: SBC Core Deployment Topology



Interoperability Test Lab Topology

The following lab topology diagram shows connectivity between Zoom and Ribbon SBC Core.

Figure 2: Interoperability Test Lab Topology



Section A: SBC Core Configuration

The following SBC Core configurations are included in this section:

1. [Network and Connectivity](#)
 2. [Static Routes](#)
 3. [TLS Configuration between Ribbon SBC Core and Zoom.](#)
 4. [PSTN Leg Configuration](#)
 5. [Zoom Leg Configuration](#)
- SBC Core can connect to the network as mentioned in [Network and Connectivity](#).
 - Zoom prefers transport as TLS. Establishing a TLS connection between SBC Core and Zoom is covered under [TLS Configuration between Ribbon SBC Core and Zoom](#).
 - SBC Core specific configuration related to PSTN is covered under [PSTN Leg Configuration](#).
 - SBC Core specific configuration related to Zoom is covered under [Zoom Leg Configuration](#).

1. Network and Connectivity

SBC 5400 front and back panel are as shown below:

Figure 3: SBC 5400 Front Panel



Figure 4: SBC 5400 Back Panel



Mgmt - is an RJ45 port and is the management interface of the SBC.

Media 0/1/2/3 depicted as pkt0/pkt1/pkt2/pkt3 are RJ45 ports. Media 0 and Media 1 are used in the current deployment.

2. Static Routes

Static routes are used to create communication to remote networks. In a production environment, static routes are mainly configured for routing from a specific network to a network that can only be accessed through one point or one interface (single path access or default route).



Tip

- For smaller networks with just one or two routes, configuring static routing is preferable. This is often more efficient since a link is not being wasted by exchanging dynamic routing information.
- For networks that have a LAN-side Gateway on Voice VLAN or Multi-Switch Edge Devices (MSEs) with Voice VLAN towards SBC Core, static routing configurations are not required.



Add the static route once PSTN Leg and Zoom Leg configurations are done on the SBC.

Static route towards PSTN

```
set addressContext default staticRoute 0.0.0.0 0 10.54.X.X LIF1 PKT0_V4 preference 100
commit
```

Static route towards Zoom

```
set addressContext default staticRoute 162.12.X.X 24 115.110.X.X LIF2 PKT1_V4 preference 100
commit
```

3. TLS Configuration between Ribbon SBC Core and Zoom

Prerequisites:

- For TLS to work on the public side of the network, a trusted CA (Certificate Authority) is needed. In this scenario, GoDaddy is used as a Trusted CA.
- Enable Zoom BYOC trunk with TLS/SRTP.

Generate a CSR with OpenSSL

To create a Certificate Signing Request (CSR) and key file for a Subject Alternative Name (SAN) certificate with multiple subject alternate names, complete the following procedure:

Create an OpenSSL configuration file (text file) on the local computer by editing the fields to the company requirements.

Note 1: In the example used in this article the configuration file is req.conf.

Note 2: req_extensions will put the subject alternative names in a CSR, whereas x509_extensions would be used when creating an actual certificate file.

```
[req]
    distinguished_name = req_distinguished_name
    req_extensions = v3_req
    prompt = no
    [req_distinguished_name]
    C = US
    ST = VA
    L = SomeCity
    O = MyCompany
    OU = MyDivision
    CN = www.company.com
    [v3_req]
    keyUsage = keyEncipherment, dataEncipherment
    extendedKeyUsage = serverAuth
    subjectAltName = @alt_names
    [alt_names]
    DNS.1 = www.company.com
    DNS.2 = company.com
    DNS.3 = www.company.net
    DNS.4 = company.net
```

Make sure there are no whitespaces at the end of the lines.

#Run the following commands to create the Certificate Signing Request (CSR) and a new Key file:

```
openssl req -new -out company_san.csr -newkey rsa:2048 -nodes -sha256 -keyout company_san.key.temp -config req.conf
```

#Run the following command to verify the Certificate Signing Request:

```
openssl req -text -noout -verify -in company_san.csr
```

After receiving the CSR with above information, provide it to CA (Certificate Authority). You will then receive the proper CA signed certificate in .crt format that is convertible into other formats using openssl.

By default, you should receive two or more certificates from CA (depending upon your CA). One is the SBC certificate, and other is CA's root and intermediate certificate.

Upload the certificates to the SBC at /opt/sonus/external and convert them into SBC-readable format, i.e. SBC certificate is in .pem or .p12 format and root certificate is in .cer or .der.

#Converting .crt to .pem USING OPENSSL for SBC certificate.

```
openssl x509 -in sbc_cert.crt -out sbc_cert.der -outform DER
```

```
openssl x509 -in sbc_cert.der -inform DER -out sbc_cert.pem -outform PEM
```

#After generating sbc_cert.pem file, convert it to .p12 format using below command and the location of the certificate key.

```
openssl pkcs12 -export -out sbc1_cert.p12 -in sbc_cert.pem -inkey /opt/sonus/company_san.key.temp
```

#CONVERTING CRT to CER USING OPENSSL for CA's root and intermediate certificate.

```
openssl x509 -in root_cert.crt -out root_cert.cer -outform DER
```

After converting all these certificates upload them on SBC at /opt/sonus/external location.

Generate Required Certificates

```
#Import Public CA Root Certificate into database.
set system security pki certificate CA_ROOT_CERT type remote fileName root_cert.cer state enabled

#Import Public CA Certified SBC Server Certificate into database.
set system security pki certificate SBC_CERT filename sbcl_cert.pl2 passphrase <Password defined during CSR
generation> state enabled type local
```

TLS Profile

A TLS Profile is required for the TLS handshake between SBC Core and Zoom. This profile defines cipher suites supported by SBC Core. Create the TLS profile as mentioned below:

```
set profiles security tlsProfile TLS_PROF clientCertName SBC_CERT serverCertName SBC_CERT cipherSuite1
tls_ecdhe_rsa_with_aes_256_cbc_sha384 cipherSuite2 tls_ecdhe_rsa_with_aes_128_cbc_sha authClient true allowedRoles
clientandserver acceptableCertValidationErrors invalidPurpose
set profiles security tlsProfile TLS_PROF v1_1 enable
set profiles security tlsProfile TLS_PROF v1_0 disable
set profiles security tlsProfile TLS_PROF v1_2 enable
commit
```

Attach the TLS Profile to the SIP Signaling Port that will be created later in Zoom Leg Configuration.

```
set addressContext default zone ZOOM sipSigPort 7 state disabled mode outOfService
commit
set addressContext default zone ZOOM sipSigPort 7 tlsProfileName TLS_PROF
commit
set addressContext default zone ZOOM sipSigPort 7 state enabled mode inService
commit
```

4. Local Ringback Tone Configuration

This section contains the general SBC configurations.

DSP Resource Allocation

This configuration only applies if the SBC has been deployed with (hardware) DSP resources. If it has not, executing this configuration step has no negative impact. Do not attempt transcoding, so that the lack of compression resources will not impact the overall SBC configuration in this document.

```
set system mediaProfile compression 75 tone 25
commit
```



This configuration is not required for SBC SWe 7.2 release onwards.

Local Ringback Tone (LRBT) Profile

1. Create a Local Ringback Tone (LRBT) profile that is attached to both PSTN and Zoom leg.
2. Enable Dynamic LRBT.

```
set profiles media toneAndAnnouncementProfile LRBT_PROF
set profiles media toneAndAnnouncementProfile LRBT_PROF localRingBackTone signalingTonePackageState enable
set profiles media toneAndAnnouncementProfile LRBT_PROF localRingBackTone precedence lower
set profiles media toneAndAnnouncementProfile LRBT_PROF localRingBackTone makeInbandToneAvailable enable
set profiles media toneAndAnnouncementProfile LRBT_PROF localRingBackTone flags useThisLrbtForEgress enable
set profiles media toneAndAnnouncementProfile LRBT_PROF localRingBackTone flags useThisLrbtForIngress enable
set profiles media toneAndAnnouncementProfile LRBT_PROF localRingBackTone flags dynamicLRBT enable
commit
```


5. PSTN Leg Configuration

Create profiles with a specific set of characteristics corresponding to PSTN. This includes configuration of the following entities on PSTN leg:

1. [Codec Entry](#)
2. [Packet Service Profile](#)
3. [IP Signaling Profile](#)
4. [IP Interface Group](#)
5. [Zone](#)
6. [SIP Signaling Port](#)
7. [IP Peer](#)
8. [SIP Trunk Group](#)
9. [Routing Label](#)
10. [Call Routing](#)

5.1 Codec Entry

Codec entry allows you to specify the codec used for the call. Create the codec entry for G711Ulaw codec with packet size 20 and rfc2833 method for dtmf.

```
set profiles media codecEntry G711ULAW codec g711
set profiles media codecEntry G711ULAW law ULaw
set profiles media codecEntry G711ULAW packetSize 20
set profiles media codecEntry G711ULAW dtmf relay rfc2833
commit
```

5.2 Packet Service Profile (PSP)

Create a Packet Service Profile (PSP) for the PSTN leg. The PSP is attached to sipTrunkGroup created later in this section.

```
set profiles media packetServiceProfile PSTN_PSP codec codecEntry1 G711ULAW
set profiles media packetServiceProfile PSTN_PSP rtcpOptions rtcp enable
commit
```

5.3 IP Signaling Profile (IPSP)

Create an IP Signaling Profile with appropriate signaling flags towards PSTN.

```
set profiles signaling ipSignalingProfile PSTN_IPSP
set profiles signaling ipSignalingProfile PSTN_IPSP egressIpAttributes flags disable2806Compliance enable
commit
```

5.4 IP Interface Group

Create an IP interface group.



Replace "x.x.x.x" with the SBC's packet interface (pkt) IP address towards PSTN (example pkt0 IP), and "Y" with its prefix length. Provide ceName used during an SBC deployment.

Here the ceName is "ZOOM1".

```
set addressContext default ipInterfaceGroup LIF1 ipInterface PKT0_V4 ceName ZOOM1 portName pkt0
set addressContext default ipInterfaceGroup LIF1 ipInterface PKT0_V4 ipAddress x.x.x.x prefix Y
set addressContext default ipInterfaceGroup LIF1 ipInterface PKT0_V4 mode inService state enabled
commit
```

5.5 Zone

Create Zone towards PSTN and specify the id of the zone.



This Zone groups the set of objects used for the communication towards PSTN.

```
set addressContext default zone PSTN id 2
commit
```

5.6 SIP Signaling Port

Set the SIP Signaling port, which is a logical address used to send and receive SIP call signaling packets and is permanently bound to a specific zone.



Replace "x.x.x.x" with SIP Signaling Port IP of SBC towards PSTN.

```
set addressContext default zone PSTN sipSigPort 3 ipInterfaceGroupName LIF1
set addressContext default zone PSTN sipSigPort 3 ipAddressV4 x.x.x.x
set addressContext default zone PSTN sipSigPort 3 portNumber 5060
set addressContext default zone PSTN sipSigPort 3 transportProtocolsAllowed sip-udp
set addressContext default zone PSTN sipSigPort 3 mode inService
set addressContext default zone PSTN sipSigPort 3 state enabled
commit
```

5.7 IP Peer

Create an IP Peer with the signaling IP address of the PSTN (Service Provider) and assign it to the PSTN Zone.



Replace "x.x.x.x" with the PSTN IP.

```
set addressContext default zone PSTN ipPeer PSTN_IPP ipAddress x.x.x.x
set addressContext default zone PSTN ipPeer PSTN_IPP ipPort 5060
commit
```

5.8 SIP Trunk Group

Create a SIP Trunk Group towards the PSTN and assign corresponding profiles like LRBT, PSP, IPSP created in earlier steps.



You must configure Trunk Group names using capital letters.

```
set addressContext default zone PSTN sipTrunkGroup PSTN_TG media mediaIpInterfaceGroupName LIF1
set addressContext default zone PSTN sipTrunkGroup PSTN_TG mode inService state enabled
commit

set addressContext default zone PSTN sipTrunkGroup PSTN_TG policy signaling ipSignalingProfile PSTN_IPSP
set addressContext default zone PSTN sipTrunkGroup PSTN_TG policy media packetServiceProfile PSTN_PSP
set addressContext default zone PSTN sipTrunkGroup PSTN_TG policy media toneAndAnnouncementProfile LRBT_PROF
set addressContext default zone PSTN sipTrunkGroup PSTN_TG ingressIpPrefix 0.0.0.0 0
commit
```

5.9 Routing Label

Create a Routing Label with a single Routing Label Route to bind the PSTN Trunk Group with the PSTN IP Peer.

```
set global callRouting routingLabel PSTN_RL routingLabelRoute 1 trunkGroup PSTN_TG
set global callRouting routingLabel PSTN_RL routingLabelRoute 1 ipPeer PSTN_IPP
set global callRouting routingLabel PSTN_RL routingLabelRoute 1 inService inService
commit
```

5.10 Call Routing

This entry is used to route all the calls coming from PSTN towards ZOOM endpoints.



Provide ceName used during an SBC deployment. "ZOOM1" is the ceName.

```
set global callRouting route trunkGroup PSTN_TG ZOOM1 standard Sonus_NULL 1 all all ALL none Sonus_NULL
routingLabel ZOOM_RL
commit
```

6. Zoom Leg Configuration

Create profiles with a specific set of characteristics corresponding to Zoom. This includes configuration of the following entities on Zoom leg:

1. [Codec Entry](#)
2. [Packet Service Profile](#)
3. [IP Signaling Profile](#)
4. [IP Interface Group](#)
5. [Zone](#)
6. [SIP Signaling Port](#)
7. [IP Peer](#)
8. [SIP Trunk Group](#)
9. [Routing Label](#)
10. [Call Routing](#)

6.1 Codec Entry

Codec entry allows you to specify the codec used for the call. Create the codec entry for G711Ulaw codec with packet size 20 and rfc2833 method for dtmf.

```
set profiles media codecEntry G711_Zoom codec g711
set profiles media codecEntry G711_Zoom law ULaw
set profiles media codecEntry G711_Zoom packetSize 20
set profiles media codecEntry G711_Zoom dtmf relay rfc2833
commit
```

6.2 Packet Service Profile (PSP)

Create a Packet Service Profile (PSP) for the Zoom leg. The PSP is attached to the sipTrunkGroup that is created later in this section.

Since there is an SRTP between the SBC Core and Zoom, you must create a crypto suite profile.

```
set profiles security cryptoSuiteProfile CRYPT_PROF entry 1 cryptoSuite AES-CM-128-HMAC-SHA1-80
```

The Crypto Suite profile is attached to the ZOOM_PSP.


```

set profiles media packetServiceProfile ZOOM_PSP codec codecEntry1 G711_Zoom
set profiles media packetServiceProfile ZOOM_PSP rtcpOptions rtcp enable
set profiles media packetServiceProfile ZOOM_PSP secureRtpRtcp cryptoSuiteProfile CRYPT_PROF
set profiles media packetServiceProfile ZOOM_PSP secureRtpRtcp flags allowFallback enable
set profiles media packetServiceProfile ZOOM_PSP secureRtpRtcp flags enableSrtp enable
commit

```

6.3 IP Signaling Profile (IPSP)

Create an IP Signaling Profile with appropriate signaling flags towards Zoom.

 The SBC Core to Zoom transport type is TLS and therefore enables the same transport type in ZOOM_IPSP.


```

set profiles signaling ipSignalingProfile ZOOM_IPSP
set profiles signaling ipSignalingProfile ZOOM_IPSP egressIpAttributes flags disable2806Compliance enable
set profiles signaling ipSignalingProfile ZOOM_IPSP egressIpAttributes numberGlobalizationProfile DEFAULT_IP
set profiles signaling ipSignalingProfile ZOOM_IPSP egressIpAttributes transport type1 tlsOverTcp
commit

```

6.4 IP Interface Group

Create an IP interface group.

 Replace "x.x.x.x" with the SBC's packet interface (pkt) IP address towards ZOOM (example pkt1 IP), and "Y" with its prefix length. Provide the ceName used during an SBC deployment.

Here the ceName is "ZOOM1".


```

set addressContext default ipInterfaceGroup LIF2 ipInterface PKT1_V4 ceName ZOOM1 portName pkt1
set addressContext default ipInterfaceGroup LIF2 ipInterface PKT1_V4 ipAddress x.x.x.x prefix Y
set addressContext default ipInterfaceGroup LIF2 ipInterface PKT1_V4 mode inService state enabled
commit

```

6.5 Zone

Create a Zone towards Zoom and specify the id of the zone.

 This Zone groups the set of objects used for communication towards Zoom.


```

set addressContext default zone ZOOM id 6
commit

```

6.6 SIP Signaling Port

Set the SIP Signaling port, which is a logical address used to send and receive SIP call signaling packets and is permanently bound to a specific zone.

 Replace "x.x.x.x" with the SIP Signaling Port IP address of the SBC towards Zoom.

```
set addressContext default zone ZOOM sipSigPort 7 ipInterfaceGroupName LIF2
set addressContext default zone ZOOM sipSigPort 7 ipAddressV4 x.x.x.x
set addressContext default zone ZOOM sipSigPort 7 portNumber 5060
set addressContext default zone ZOOM sipSigPort 7 tlsProfileName TLS_PROF
set addressContext default zone ZOOM sipSigPort 7 transportProtocolsAllowed sip-tls-tcp
set addressContext default zone ZOOM sipSigPort 7 mode inService
set addressContext default zone ZOOM sipSigPort 7 state enabled
commit
```



You created the TLS profile in [TLS Profile](#).



There are a few areas that result in a TLS negotiation issue. One area involves assigning the incorrect port. Ensure the following are accomplished:

- Zoom listens on port number 5061 (default setting).
- Configure port number 5060 on Zoom IP-Peer since Ribbon SBC Core increments the port by 1 when the transport protocol is TLS.

6.7 IP Peer

Create an IP Peer with the signaling IP address of ZOOM and assign it to ZOOM Zone.



Replace "x.x.x.x" with the Zoom SIP signaling IP.

```
set addressContext default zone ZOOM ipPeer ZOOM_IPP ipAddress x.x.x.x
set addressContext default zone ZOOM ipPeer ZOOM_IPP ipPort 5060
commit
```

Path Check Profile

Create a path check profile that attaches to the Zoom side.

```
set profiles services pathCheckProfile ZOOM_OPTIONS protocol sipOptions sendInterval 20 replyTimeoutCount 1
recoveryCount 1
set profiles services pathCheckProfile ZOOM_OPTIONS transportPreference preference1 tls-tcp
commit
```

6.8 SIP Trunk Group

Create a SIP Trunk Group towards ZOOM and assign corresponding profiles like LRBT, PSP, IPSP that were created in earlier steps.



You must configure Trunk Group names using capital letters.

```
set addressContext default zone ZOOM sipTrunkGroup ZOOM_TG media mediaIpInterfaceGroupName LIF2
set addressContext default zone ZOOM sipTrunkGroup ZOOM_TG mode inService state enabled
commit

set addressContext default zone ZOOM sipTrunkGroup ZOOM_TG policy signaling ipSignalingProfile ZOOM_IPSP
set addressContext default zone ZOOM sipTrunkGroup ZOOM_TG policy media packetServiceProfile ZOOM_PSP
set addressContext default zone ZOOM sipTrunkGroup ZOOM_TG policy media toneAndAnnouncementProfile LRBT_PROF
set addressContext default zone ZOOM sipTrunkGroup ZOOM_TG ingressIpPrefix 0.0.0.0 0
commit
```


6.9 Routing Label

Create a Routing Label with a single Routing Label Route to bind the ZOOM Trunk Group with the ZOOM IP Peer.

```
set global callRouting routingLabel ZOOM_RL routingLabelRoute 1 trunkGroup ZOOM_TG
set global callRouting routingLabel ZOOM_RL routingLabelRoute 1 ipPeer ZOOM_IPP
set global callRouting routingLabel ZOOM_RL routingLabelRoute 1 inService inService
commit
```

6.10 Call Routing

This entry is used to route all the calls coming from Zoom towards PSTN endpoints.

 Provide the ceName used during an SBC deployment. "ZOOM1" is the ceName.

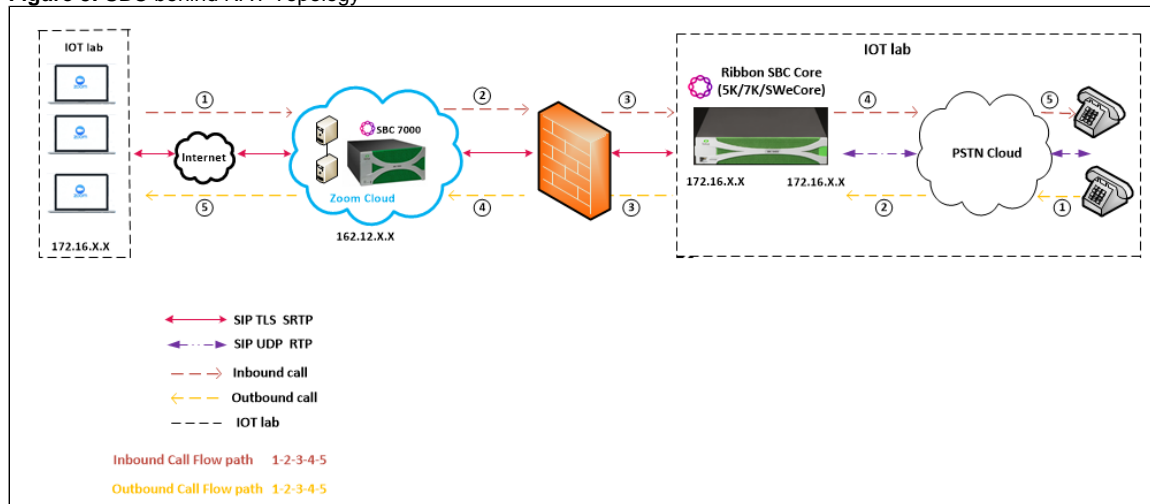
```
set global callRouting route trunkGroup ZOOM_TG ZOOM1 standard Sonus_NULL 1 all all ALL none Sonus_NULL
routingLabel PSTN_RL
commit
```

Section B: Configuration for SBC behind NAT

Telecom operators do not expose the WAN side of the SBC directly to the public network. The SBC is deployed in the DMZ behind a NAT'd device having WAN interface configured with a private IP. To achieve this, certain SIP Message Manipulation (SMM) rules are applied in the SBC for converting Private IP to Public IP.

SBC behind NAT Topology

Figure 5: SBC behind NAT Topology



Additional configuration for SBC behind NAT

SIP Trunk Group towards Zoom

Add the following additional configuration to the SIP Trunk Group towards the Zoom leg.

```
set addressContext default zone ZOOM sipTrunkGroup ZOOM_TG services natTraversal signalingNat enabled
set addressContext default zone ZOOM sipTrunkGroup ZOOM_TG services natTraversal mediaNat enabled
commit
```

Outbound Profile

An SMM rule "HeaderModification" is used to replace the Private IP with the Public IP.



Replace the <Private_IP> and <Public_IP> with actual IP's.

```
set profiles signaling sipAdaptorProfile HeaderModifications state enabled
set profiles signaling sipAdaptorProfile HeaderModifications profileType messageManipulation
set profiles signaling sipAdaptorProfile HeaderModifications rule 1 applyMatchHeader one
set profiles signaling sipAdaptorProfile HeaderModifications rule 1 criterion 1 type message
set profiles signaling sipAdaptorProfile HeaderModifications rule 1 criterion 1 message
set profiles signaling sipAdaptorProfile HeaderModifications rule 1 criterion 1 message messageTypes all
set profiles signaling sipAdaptorProfile HeaderModifications rule 1 action 1 type messageBody
set profiles signaling sipAdaptorProfile HeaderModifications rule 1 action 1 operation regsub
set profiles signaling sipAdaptorProfile HeaderModifications rule 1 action 1 from
set profiles signaling sipAdaptorProfile HeaderModifications rule 1 action 1 from type value
set profiles signaling sipAdaptorProfile HeaderModifications rule 1 action 1 from value <Public_IP>
set profiles signaling sipAdaptorProfile HeaderModifications rule 1 action 1 to
set profiles signaling sipAdaptorProfile HeaderModifications rule 1 action 1 to type messageBody
set profiles signaling sipAdaptorProfile HeaderModifications rule 1 action 1 to messageBodyValue all
set profiles signaling sipAdaptorProfile HeaderModifications rule 1 action 1 regexp
set profiles signaling sipAdaptorProfile HeaderModifications rule 1 action 1 regexp string <Private_IP>
set profiles signaling sipAdaptorProfile HeaderModifications rule 1 action 1 regexp matchInstance all
set profiles signaling sipAdaptorProfile HeaderModifications rule 2 applyMatchHeader one
set profiles signaling sipAdaptorProfile HeaderModifications rule 2 criterion 1 type message
set profiles signaling sipAdaptorProfile HeaderModifications rule 2 criterion 1 message
set profiles signaling sipAdaptorProfile HeaderModifications rule 2 criterion 1 message messageTypes all
set profiles signaling sipAdaptorProfile HeaderModifications rule 2 criterion 2 type header
set profiles signaling sipAdaptorProfile HeaderModifications rule 2 criterion 2 header
set profiles signaling sipAdaptorProfile HeaderModifications rule 2 criterion 2 header name Contact
set profiles signaling sipAdaptorProfile HeaderModifications rule 2 criterion 2 header condition exist
set profiles signaling sipAdaptorProfile HeaderModifications rule 2 criterion 2 header hdrInstance all
set profiles signaling sipAdaptorProfile HeaderModifications rule 2 action 1 type header
set profiles signaling sipAdaptorProfile HeaderModifications rule 2 action 1 operation regsub
set profiles signaling sipAdaptorProfile HeaderModifications rule 2 action 1 headerInfo fieldValue
set profiles signaling sipAdaptorProfile HeaderModifications rule 2 action 1 from
set profiles signaling sipAdaptorProfile HeaderModifications rule 2 action 1 from type value
set profiles signaling sipAdaptorProfile HeaderModifications rule 2 action 1 from value <Public_IP>
set profiles signaling sipAdaptorProfile HeaderModifications rule 2 action 1 to
set profiles signaling sipAdaptorProfile HeaderModifications rule 2 action 1 to type header
set profiles signaling sipAdaptorProfile HeaderModifications rule 2 action 1 to value Contact
set profiles signaling sipAdaptorProfile HeaderModifications rule 2 action 1 regexp
set profiles signaling sipAdaptorProfile HeaderModifications rule 2 action 1 regexp string <Private_IP>
set profiles signaling sipAdaptorProfile HeaderModifications rule 2 action 1 regexp matchInstance all
```

Attach the SMM rule to the OutputAdapter Profile of ZOOM_TG

```
set addressContext default zone ZOOM sipTrunkGroup ZOOM_TG signaling messageManipulation outputAdapterProfile
HeaderModifications
commit
```

Section C: SBC Core Configuration with External PSX

SBC Core configuration

The following SBC configurations remain the same as mentioned in [Section A: SBC Core Configuration](#).

1. [Network and Connectivity](#)
2. [Static Routes](#)
3. [TLS Configuration between Ribbon SBC Core and Zoom](#)
4. [Local Ringback Tone Configuration](#) – Configure only the [DSP Resource Allocation](#) on SBC Core. [Local Ringback Tone \(LRBT\) Profile](#) is configured in the external PSX, as shown later.
5. [PSTN Leg Configuration](#) – Configure the [IP Interface Group](#), [Zone](#), [SIP Signaling Port](#) and [SIP Trunk Group](#) as mentioned in this section.
6. [Zoom Leg Configuration](#) – Configure the [IP Interface Group](#), [Zone](#), [SIP Signaling Port](#) and [SIP Trunk Group](#) as mentioned in this section.

Enable Remote PSX Server

By default, the SBC Core is enabled with a local PSX server. You must disable the local PSX to enable the remote PSX.

Command to disable the local PSX server.

```
set system policyServer localServer PSX_LOCAL_SERVER state disabled mode outOfService
```

Command to enable the remote PSX server.

```
set system policyServer remoteServer RBBNPSX ipAddress 172.16.X.X state enabled mode active
```



"RBBNPSX" is the name of the remote PSX server. Provide the valid IP of the remote PSX server.

Configuration on the PSX

This section provides the configuration aspects required on the PSX side.

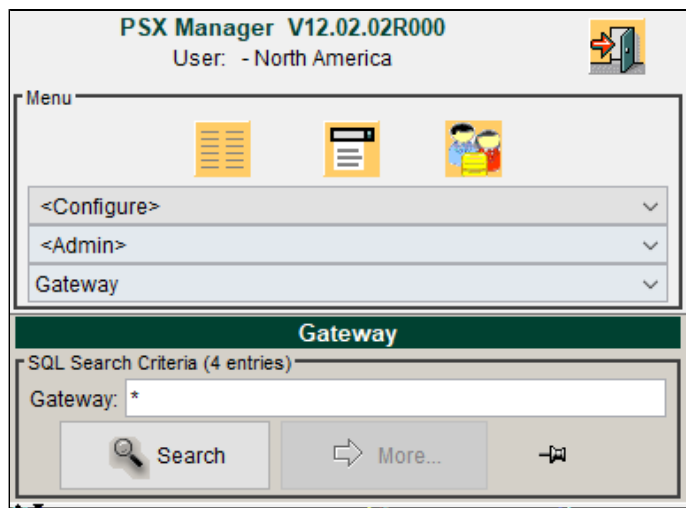


The sequence of steps covering various snapshots & information provided in the snapshots for PSX configuration must be meticulously followed.

Gateway

Configure a gateway with SBC name and it's management IP address.

Figure 6: Gateway Creation



Host: 172.16.100.216 @ 4330
Master (SWe) - V12.02.02R000

View: Gateway Close All Perspective: Full View

GATEWAY: ZOOM LRNS

Switch: ZOOM

Gateway Group: DEFAULT

Cluster Profile: <None>

Default Trunk Group: SIP

Charge Band Profile: <None>

Traffic Control Escape Profile: <None>

Mobile Switch ID: 1 ☒ None

Signaling Gateway Group: <None>

Enum Authority Profile: <None>

Address Reachability Service Profile: <None>

SMM Profile Group: <None>

Peer Throttling Profile: <None>

P-Origination-ID:

Flags

☐ CAMEL Services Supported ☐ Route CAMEL Subscription Calls

☐ CDP Gateway ☒ Traffic Management

☐ MTRR Supported ☐ Logical SBC

Display

☐ Allow Mixed Characters in Gateway Name

Flags

☐ CAMEL Services Supported ☐ Route CAMEL Subscription Calls

☐ CDP Gateway ☒ Traffic Management

☐ MTRR Supported ☐ Logical SBC

Display

☐ Allow Mixed Characters in Gateway Name

H.323 Control

☐ Prune Routes

Network

IPv4 Address: 10 . 54 . . Port Number: 2569

IPv6 Address: 0 : 0 : 0 : 0 : 0 : 0 : 0 : 0

☒ Prefer IPv4 ☐ Prefer IPv6

☐ H.323 IPv4 Address: 0 . 0 . 0 . 0 H.323 Port Number: 1720

☐ H.323 IPv6 Address: 0 : 0 : 0 : 0 : 0 : 0 : 0 : 0

☐ Set As Default H.323 Gateway For This IP Address

☒ Prefer IPv4 ☐ Prefer IPv6

☐ SIP IPv4 Address: 0 . 0 . 0 . 0 SIP Port Number: 5060

☐ SIP IPv6 Address: 0 : 0 : 0 : 0 : 0 : 0 : 0 : 0

☐ Set As Default SIP Gateway For This IP Address

Save Cancel Delete

Tone and Announcement Profile

Configure the Tone and Announcement Profile as shown below:

Figure 7: Tone And Announcement Profile

Tone And Announcement Profile: ZOOM_LRBT

Local Ring Back Tone

☒ **Signaling Tone Package:** 1 - DEFAULT

Precedence: ☒ Lower ☐ Higher

☒ **Make Inband Tone Available**

Flags

☐ Force Local Ring Back Tone

☒ Dynamic Local Ring Back Tone

☒ Use This Local Ring Back Tone For Egress

☒ Use This Local Ring Back Tone For Ingress

☐ Announcement Based Tones

☐ 180WithOrWithOutSdp

EarlyMediaMethod: None

Tone And Announcement Profile

☐ **Announcement Package:** 1 - DEFAULT

Signaling Tone Package: 1 - DEFAULT

☐ Override

Tone Generation Criteria

Tone Generation Criteria: <None>

Crypto Suite Profile

Select the Crypto Suite as "AES CM 128 HMAC SHA1 80".

Figure 8: Crypto Suite Profile

Crypto Suite Profile: ZOOM_CRYPT

Description: Secure Crypto Suite Profile for Zoom

Crypto Suites

Sequence: 1

Crypto Suite: AES CM 128 HMAC SHA1 80

Session Parameter Flags

☐ Unauthenticated SRTP

☐ Unencrypted SRTP

☐ Unencrypted SRTCP

Add/Update

Sequence	Crypto Suite
1	AES CM 128 HMAC SHA1 80

Element Routing Priority

Assign the highest priority to the Entity Type Trunk Group for all the required Call Types.

Figure 9: Element RP

Element Routing Priority: ZOOM_ERP

Call Property

Call Type: 1+
Priority: 1
Network: All
Toll Indication: <All>

Entity Type: Trunk Group
Priority: 1

Add Update

Call Type	Call Priority	Network	Toll Indication	Entity Type	Priority
Private	1	All	<All>	<None>	1
0+	1	All	<All>	<None>	1
0-	1	All	<All>	<None>	1
1+	1	All	<All>	Trunk Group	1
1+	2	All	<All>	<None>	2
IDDD	1	All	<All>	<None>	1
0+ IDDD	1	All	<All>	<None>	1
00	1	All	<All>	<None>	1
IP VPN Service	1	All	<All>	<None>	1
Test	1	All	<All>	<None>	1
Transit	1	All	<All>	<None>	1
Other Carrier Chosen	1	All	<All>	<None>	1
Carrier Cut Through	1	All	<All>	<None>	1
User Name	1	All	<All>	<None>	1
Mobile	1	All	<All>	<None>	1

Delete

Back To Softlink Save Cancel Delete

Routing Criteria

Use the Routing criteria "DEFAULT_IP" as shown below. The configuration is as follows:

Figure 10: Routing C

Routing Criteria: DEFAULT_IP

Call Property

Call Type: 1+
Priority: 1
Network: All
Toll Indication: <All>

☒ Use Destination ☒ Use Partition

Add Update

Call Type	Call Priority	Network	Toll Indication	Use Destination	Use Partition
Private	1	All	<All>	Enabled	Enabled
0+	1	All	<All>	Enabled	Enabled
0-	1	All	<All>	Enabled	Enabled
1+	1	All	<All>	Enabled	Enabled
IDDD	1	All	<All>	Enabled	Enabled
0+ IDDD	1	All	<All>	Enabled	Enabled
00	1	All	<All>	Enabled	Enabled
IP VPN Service	1	All	<All>	Enabled	Enabled
Test	1	All	<All>	Enabled	Enabled
Transit	1	All	<All>	Enabled	Enabled
Other Carrier Chosen	1	All	<All>	Enabled	Enabled
Carrier Cut Through	1	All	<All>	Enabled	Enabled
User Name	1	All	<All>	Enabled	Enabled
Mobile	1	All	<All>	Enabled	Enabled

Delete

Back To Softlink Save Cancel Delete

PSTN Leg Configuration

IP Signaling Profile (IPSP)

Create an IP Signaling Profile with the appropriate signaling flags towards PSTN.

Figure 11: IP Signaling Profile

IP SIGNALING PROFILE: PSTN_IPSP

Common IP Attributes - Communicating With The Peer Regardless Of Call Direction

☐ Accept Alert Info

☐ Add P-Charging Function Addr

☐ Add Path/Service Route Per TG

☐ Audio Codec Change through Empty TCS

☐ Call Hold Interworking

☐ Calling Party Type Number If Present

☐ Clearmode For Data Calls

☐ Create P-Charging-Vector

☐ Create P-Visited-Network Id

☐ Create Path Header

☐ Create Service-Route Header

☐ Customized Session Timer Behavior

☐ Disable Also Header

☐ Disable Constrained Capacities

☐ Disable Host Translation

☐ Disable Media Lock Down

☐ Disable Refer-To URI Parameters

☐ Discard Received Reason Header

☐ Do Not Include SS Attribute In Re-INVITE

☐ No Content Disposition

☐ No Port Number 5060

☐ No Userinfo In Contact Header

☐ Only Selected Codec In Session Refresh

☐ Override Relay For Non SIP Egress Leg

☐ P-Called-Party-Id-Support

☐ P-ChgMsg-Info

☐ Relay Data Path Mode Changes To The Other Leg

☐ Reject REFER

☐ Replace Host On Via Header

☐ Reject REFER With IP

☐ Reject REFER With TN

☐ ReQuery PSX on REGISTER Refresh

☐ Restrict History Info Header

☐ Route Using Received FQDN

☐ SDP O-line Only Compares

☐ Send All Allowed Codecs For Late Media Invite Or Re-Invite

☐ Send Direct Media Info In SDP Attribute

☐ Send Empty TCS

☐ Don't Send REFER With TN

☐ End To End BYE

☐ End To End RE-INVITE

☐ End To End UPDATE

☐ Suppress End To End Session Refresh

☐ End To End PRACK

☐ Enable Default PUI Procedures

☐ Enable Dial String Handling

☐ Include G729 with G729A when offer PSP has G729A

☐ Include IP Ports In FROM And TO Headers

☐ Include Reason Header (Q.850)

☐ Include SS Attribute In Initial Invite

☐ Include Transport Type In Contact Header

☐ Insert Peer Address As Top Route Header

☐ Lockdown Preferred Codec

☐ Map Cause Location

☐ Map SGD In P-Sig-Info Header

☐ Map Suspend/Resume Event In P-Svc-Info Header

☐ Map UUI In P-Sig-Info Header

☐ MIME Cause Precede Reason Header Cause

☐ Send PTIME In SDP

☐ Send RTCP Port In SDP

☐ Session Timer Refresh Update

☐ Set Accept Header To Application SDP Only

☐ Set Oline Dash

☐ Set Session Version Zero

☐ Set Sline Dash

☐ Store P-Charging Function Addr

☒ Store P-Charging Vector

☐ Store Path Header

☐ Store Service-Route Header

☐ Suppress Min-SE if not received

☐ Terminal Portability Interworking

☐ Send RTCP BandWidth Info

☐ Validate Access Nw Info Header

☐ Use Psx Route for Registered Invite

☐ From Header Anonymisation

☐ Create ISUP Message Body

☐ Disable Transparently Passing ISUP Message Body

☐ aiToPemInterworking

Relay Flags	
<input type="checkbox"/> Conference Event Package	<input type="checkbox"/> PUBLISH
<input type="checkbox"/> Dialog Event Package	<input type="checkbox"/> REFER
<input type="checkbox"/> DTMF Body	<input type="checkbox"/> Reg Event Package
<input type="checkbox"/> Force 503 To 500 Relay	<input type="checkbox"/> Ribbon Media Body
<input type="checkbox"/> Info	<input type="checkbox"/> Status Code 3XX
<input type="checkbox"/> Message	<input type="checkbox"/> Status Code 4XX-6XX
<input type="checkbox"/> Notify	<input type="checkbox"/> Third Party Bodies
<input type="checkbox"/> Options	<input type="checkbox"/> Update without SDP
Refer To Header Relay <input checked="" type="radio"/> Reject the REFER request if no match is found <input type="radio"/> relay the REFER request if no match is found <input type="radio"/> relay the REFER request without matching	

Transparency Flags	
<input type="checkbox"/> Accept-Contact Header	<input type="checkbox"/> Reason Header
<input type="checkbox"/> Accept-Language Header	<input type="checkbox"/> Referred-By Header
<input type="checkbox"/> Accept Header	<input type="checkbox"/> Resource Priority Option Tag
<input type="checkbox"/> Alert Information Header	<input type="checkbox"/> Request-URI
<input type="checkbox"/> Allow Header	<input type="checkbox"/> Resource-Lists Body
<input type="checkbox"/> Authcode Headers	<input type="checkbox"/> RLMI Body
<input type="checkbox"/> Call-Info Header	<input type="checkbox"/> Route Header
<input type="checkbox"/> Contact Header*	<input type="checkbox"/> Server Header

Egress IP Attributes - Sending A Call In The Forward Direction To The Peer	
IP Protocol Type:	<input checked="" type="radio"/> SIP Only <input type="radio"/> SIP-I <input type="radio"/> H.323 <input type="radio"/> Wireless
IP Signaling MIME Content Type:	ISUP
IP Signaling Treatment:	Interwork
MIME Content Type Version:	1 - ansi88
Globalize Number Profile:	<None>
Phone-Context Parameter Length:	0
Media Qos Kpi Profile:	<None>
Signaling Qos Kpi Profile:	<None>
Flags	
<input type="checkbox"/> Accept 3XX With RN	<input type="checkbox"/> Qos Based Routing
<input type="checkbox"/> BGCF Target Scheme Transparency	<input type="checkbox"/> Prefix RN to Dialed Digits
<input type="checkbox"/> Convert Inactive To Sendrecv	<input type="checkbox"/> Reject 3XX With IP
<input type="checkbox"/> Delay Cut Through	<input type="checkbox"/> Reject 3XX With TN
<input checked="" type="checkbox"/> Disable 2806 Compliance	<input type="checkbox"/> Same CallId For Required Authorization
<input type="checkbox"/> Disable Optional Register Parameters	<input type="checkbox"/> Transit PAI From Unregistered Peer
<input checked="" type="checkbox"/> Disposition Handling Required	<input type="checkbox"/> Suppress UNREGISTER
<input type="checkbox"/> Don't Send Fast Start Proposal	<input type="checkbox"/> TTC-ISUP Mapping
<input type="checkbox"/> Enable 3261 Cancel Handling	<input type="checkbox"/> Use Called Party In Request URI
<input type="checkbox"/> Include ENUM Parameters	<input type="checkbox"/> Use Colon In SDP Media Type Parameter

Privacy

☐ Transparency
 ☐ AnonymizeHostIpAddress

Privacy Information:
 ☒ P-Preferred-ID
 ☐ P-Asserted-ID
 ☐ Remote-Party-ID

Flags

☒ Include Privacy
 ☐ Privacy Required by Proxy
 ☐ MS Lync Privacy Support
 ☐ Include Embedded PAI Header in Redirected INVITE
 ☐ Do Not Include Tel URI In PAI Header

Redirect

Mode: Accept Redirection

Contact Handling:
 ☒ Merge Received Contacts with Existing Contacts
 ☐ Purge Existing Contacts

Flags

☐ Skip Crankback Profile And Always Crankback
 ☐ Honor Embedded Headers in 3xx
 ☐ Force Re-query for Redirection
 ☐ Enhanced Local Redirection

SIP Cause Mapping

Internal To SIP Cause Mapping: 1 - DEFAULT

SIP To Internal Cause Mapping: 1 - DEFAULT

Internal to SIP Cause Mapping Profile Name

SIP to Internal Cause Mapping Profile Name

Call Forwarding

☐ Diversion-History Info Interworking (RFC 6044 compliance)

Redirection Information

☒ Diversion
 ☐ Diversion With Transparency
 ☐ PK Header

History Information

☐ Include History-Info
 ☐ Cause Parameter In RFC 4458
 ☐ Reason With Cause Value As Per RFC 4244

CPC Mapping Flags

☐ Map CPC when Presentation Indicator is Restricted
 ☒ Any CPC
 ☐ CPC=Priority

Send CPC Param In

☒ Default
 ☐ PAI
 ☐ From
 ☐ Both (PAI and From)

P Charge Info

☐ Transparency

P-Charge-Info Information:
 ☒ URI Parameter
 ☐ User Parameter
 ☐ Header Parameter

Flags

☐ Include NPI
 ☐ Include NOA

Transport Type
Transport Type 1: <None>
Transport Type 2: <None>
Transport Type 3: <None>
Transport Type 4: <None>
☐ Use configured transport for egress leg

Ingress IP Attributes - Signaling Back A Message To The Peer That We Receive A Call From
Flags

☐ 181 Supported
☐ 182 Supported
☐ Convert Progress To Alert
☐ Don't Send Facility Message
☐ Don't Send 3XX With IP
☐ Don't Send 3XX With TN
☐ Map Called Party Category In P-Sig-Info Header
☐ No SDP In 180 Supported
☐ Refuse Fast Start Proposal
☐ Registration Expires in Expires Header
☐ Map Subsequent 180 to 183
☐ Early Media Authorization

☐ Registration Support 3xx
☐ Send 183 On Initiating Disconnect Treatment
☐ Send Fast Start Response In CP
☐ Send SDP In 200 OK If 18x Reliable
☐ Send Updated SDP In 200 OK
☐ Send SDP In Subsequent 18x
☐ Send TLS Connection Failure Response
☐ Suppress 183 For 3xx Redirect Response
☐ Suppress 183 Without SDP
☐ Override 3xx Relay
☐ Send BIT-H Of BCI In Outgoing Invite
☐ Convert Alert To Progress

Save

Cancel

Delete

Packet Service Profile (PSP)

Create a Packet Service Profile (PSP) for the PSTN leg. The PSP is attached to the TrunkGroup that is created later in this section.

Figure 12: Packet Service Profile

Packet Service Profile: PSTN_PSP
Silence Factor: 40
Voice Initial Playout Buffer Delay (ms): 10
Type Of Service: 0
AAL1 Payload Size: 47
Preferred RTP Payload Type For DTMF Relay: <None>
Media Packet COS: 0
Monitoring Profile: <None>

Codec Entry

Codec Entry: <None>

Add
Update

Codec Entry	Value
1	G711Ulaw_T38_2833

Delete

Media Control: IPv4 Only

Packet-To-Packet Control	
Transcode	
<input type="radio"/> Only <input checked="" type="radio"/> Conditional <input type="radio"/> Determined By PSP For Other Leg <input type="radio"/> Transcode	
Conditions In Addition To "No Common Codec"	
<input type="checkbox"/> Apply Fax Tone Treatment <input type="checkbox"/> Different DTMF Relay <input type="checkbox"/> Different Packet Size	<input type="checkbox"/> Different Silence Suppression <input type="checkbox"/> Honor Answer Preference <input type="checkbox"/> Honor Offer Preference <input type="checkbox"/> Different 2833 Payload Type
Codecs Allowed For Transcoding	
This Leg: <input type="checkbox"/> G.711 A <input type="checkbox"/> G.711 U <input type="checkbox"/> G.722 <input type="checkbox"/> G.722.2 <input type="checkbox"/> G.723.1 <input type="checkbox"/> G.726 <input type="checkbox"/> G.729 <input type="checkbox"/> OPUS <input type="checkbox"/> EVS <input type="checkbox"/> SILK <input type="checkbox"/> T.38 <input type="checkbox"/> iLBC <input type="checkbox"/> AMR Other Leg: <input type="checkbox"/> G.711 A <input type="checkbox"/> G.711 U <input type="checkbox"/> G.722 <input type="checkbox"/> G.722.2 <input type="checkbox"/> G.723.1 <input type="checkbox"/> G.726 <input type="checkbox"/> G.729 <input type="checkbox"/> OPUS <input type="checkbox"/> EVS <input type="checkbox"/> SILK <input type="checkbox"/> T.38 <input type="checkbox"/> iLBC <input type="checkbox"/> AMR	
RTCP	
<input checked="" type="checkbox"/> RTCP Packet Loss Threshold (Packets Lost/100,000 Packets): <input type="text" value="0"/>	
RR Bandwidth: <input type="text" value="250"/>	
RS Bandwidth: <input type="text" value="250"/>	
Packet Loss Action	
<input checked="" type="radio"/> None <input type="radio"/> Trap <input type="radio"/> Trap And Disconnect	
<input type="checkbox"/> Enable RTCP Only For HELD Calls <input type="checkbox"/> Termination For Pass-Through Calls	
<input type="checkbox"/> RTCP-MUX <input type="checkbox"/> Generate RTCP for T140 if not received from other leg	
RTCP-XR	
<input type="checkbox"/> Relay <input type="checkbox"/> Relay Or Terminate	

Peer Absence Action	
<input checked="" type="radio"/> None <input type="radio"/> Trap <input type="radio"/> Trap And Disconnect	
Silence Insertion Descriptor	
G.711 Silence Insertion Descriptor RTP Payload Type: <input type="text" value="13"/>	
<input checked="" type="checkbox"/> Silence Insertion Descriptor Heartbeat	
Data Calls	
Initial Playout Buffer Delay (ms): <input type="text" value="50"/>	
Packet Size: <input type="text" value="20"/>	
Preferred RTP Payload Type: <input type="text" value="56"/>	
Video Calls	
Maximum Video Bandwidth (kbps): <input type="text" value="0"/>	
Video Bandwidth Reduction Factor (%): <input type="text" value="0"/>	
<input checked="" type="checkbox"/> Audio Only If Video Is Prevented	
IPv4 TOS: <input type="text" value="0"/>	
IPv6 Traffic Class: <input type="text" value="0"/>	
IEEE 802.1Q VLAN COS: <input type="text" value="0"/>	
Codec List Profile: <input type="text" value="<None>"/>	
Qos Values	
MSRP DSCP: <input type="text" value="0"/>	
DTLS SCTP DSCP: <input type="text" value="0"/>	
T140 DSCP: <input type="text" value="0"/>	
Application DSCP: <input type="text" value="0"/>	

Secure RTP/RTCP

Crypto Suite Profile: <None>

Flags

☐ Allow Fallback
 ☐ Enable SRTP
 ☐ Reset ROC On Session Key Change
 ☐ Reset Enc/Dec/ROC on Decryption Key Change
 ☐ Update Crypto On Modify
 ☐ Allow Pass Through

DTLS/SRTP

Crypto Suite Profile: <None>

Flags

☐ Allow Fallback
 ☐ Enable DTLS
 ☐ Relay DTLS SRTP
 ☐ Relay DTLS SCTP

Flags

☐ DSCP Passthrough
 ☐ Interwork DTMF OOB-2833 Without Transcoding
 ☐ Digit Detect Send Enabled
 ☐ Use Direct Media
 ☐ Disallow Data Calls
 ☐ Validate Peer Support for DTMF Events
 ☐ SSRC Randomize
 ☐ HD Codec Preferred
 ☐ Reserve BW for Preferred Audio Common Codec
 ☐ Prefer NB PassThru Over HDTranscode
 ☐ Police on Heaviest Audio Codec
 ☐ Match Offered Codec Group if Nb Only
 ☐ t140 Call
 ☐ Force Route PSP Order
 ☐ Allow Audio Transcode For MultiStream Call
 ☐ Allow Mid Call SSRC Modification
 ☐ Generate and Signal SSRC and CName

Packet Service Profile ID Group

Create the Packet Service Profile ID Group and attach the Packet Service Profile created earlier.

Figure 13: Packet Service Profile ID Group

Packet Service Profile ID Group: PSTN

Packet Service Profile: PSTN_PSP

HPC Packet Service Profile: <None>

IP Signaling Peer Group

Add the PSTN IP Address as shown below:

Figure 14: IP Signaling Peer Group

IP Signaling Peer Group: PSTN_IPP

Description:

Flags

☐ Send All Peer IP Addresses/FQDNs

Peer Group Data

Sequence Number: 0

☒ IPv4 Address:
 172.16. . 16. . . Port Number: 5060

☐ IPv6 Address:
 0 : 0 : 0 : 0 : 0 : 0 : 0 : 0 Port Number: 0

☐ Server FQDN: Port Number: 0

☒ In Service

Add/Update

Sequence Number	IP Address	Port Number	Server FQDN	Port Number	Send	Service Status
0	172.16. . 16. .	5060		0	IP Address	In Service

Trunk Group

Create a SIP Trunk Group towards the PSTN and assign corresponding profiles like LRBT, PSP, IPSP created in earlier steps.

Figure 15: Trunk Group

Trunk Group:	PSTN_TG	<input type="checkbox"/> Unrestricted
Gateway:	ZOOM	
Description:		
Auto Recall Profile:	<None>	
Call Processing Localization Variant:	North America	
Calling Area:	<None>	
Carrier:	0000	
Carrier Selection Priority:	<None>	
Country:	1 - USA, Canada and Caribbean	
DDI Range Profile:	<None>	
Destination Switch Type:	Access	
Direction:	Two Way	
Element Routing Priority Profile:	ZOOM_ERP	
Feature Control Profile:	DEFAULT_FC_ACCESS	
IP Signaling Profile:	PSTN_IPSP	
LATA:	<None>	
Local Recursion Profile:	<None>	
Maximum Satellite Hops:	Three or More Satellite Hops	
Network Data Partition:	0	
Network Data Net:	0	
Next Hop Domain:	<None>	
Number Analysis Profile:	<None>	
Number Length Enforcement:	<None>	
Originating Carrier:	<None>	
PPR Profile:	<None>	
Pseudo Carrier:	<None>	
Remote Sip Peer Type:	None	
Region:	<None>	
Routing Criteria Profile:	DEFAULT_IP	
SCP Business Service Group:	0	
Signaling Profile:	DEFAULT_IP_PROFILE	
Signaling Flag:	SIP	
SIP Domain:	<None>	
SIP Response Code Profile:	<None>	
TDM Type:	Other	
Tone And Announcement Profile:	ZOOM_LRBT	
Trunk Group COS:		
Trunk Group COS Profile:	<None>	
Trunk Group Domain:	<None>	
Trunk Number:		
Zone Index Profile:	DEFAULT	
ZZ Profile:	<None>	
Charge Band Profile:	<None>	

Ingress

Charge Indicator: None

Default CPC: <None>

Default OLIP: <None>

Dial Plan Profile: <None>

Forced OLIP Value: <None>

In DM/PM Rule: <None>

Info Transfer Capability Profile: <None>

IP Version Preference: IPv4 Only

ONI:

JIP:

NPA:

Numbering Plan: NANP_ACCESS

In Policy Profile Group: <None>

Flags

<input type="checkbox"/> Allow Hex Digits In Cdpn	<input type="checkbox"/> Non-Zero Video Bandwidth Based Routing for H.323
<input type="checkbox"/> Discard NPDI	<input type="checkbox"/> Non-Zero Video Bandwidth Based Routing for SIP
<input type="checkbox"/> Discard RN	<input type="checkbox"/> Overlap Dialing
<input type="checkbox"/> HD Preferred Routing	<input type="checkbox"/> TNS Circuit Code Based Routing
<input type="checkbox"/> HD Supported Routing	<input checked="" type="checkbox"/> Use IPTG Routing (Hop By Hop Routing) For Ingress

Egress

Out DM/PM Rule: <None>

Out Policy Profile Group: <None>

Trunk Context:

R-URI Host: R-URI Host Port:

Flags

<input type="checkbox"/> Disable Crankback
<input type="checkbox"/> Enable JIP Interwork
<input type="checkbox"/> Use Preferred Identity

Billing

Billing Plan: <None>

Billing Information: <None>

Default Billing Number:

Nature Of Address: <None>

Numbering Plan Indicator: <None>

Calling Party Number

Calling Party:

Nature Of Address: <None>

Numbering Plan Indicator: <None>

Presentation: <None>

Flags

☒ Do Not Use For Fallback Bearer Capability ☐ Out Of Service

☐ Escaped ☐ Satellite Trunk

☐ Use Sac NonSac Call Types For ZZ Profile

IP TG

IP Signaling Peer Group: PSTN_IPP

☐ IP Peer Supported

Packet Service Profile ID Group: PSTN

☒ Egress IP Signaling Profile: PSTN_IPSP

Packet Service Profile

Preferred Packet Service Profile ID Group: <None>

☐ Destination Override

Traffic Management Options

Trunk Group Reservation Level 1: 10

Trunk Group Reservation Level 2: 5

VPN Information

Business Group: <None>

Business Location: <None>

☐ Business Group From CLI

Services

☒ Not Screened ☐ Screened - Normal ☐ Screened - Fraud

Class Of Service: DEFAULT_IP

Service Exception Profile: <None>

Save Cancel Delete

Routing Label

Configure the Routing Label as follows:

Routing Label: PSTN_RL

Destination

☒ Ignore ☐ Do not Use ☐ Use

Route Prioritization Type

☒ Sequence ☐ Proportion ☐ Round Robin ☐ All Proportion ☐ Least Cost Routing

Route Prioritization Type For Equal Cost Routes: Sequence

☐ Use TAR Routes

TAR Route Prioritization Type

☒ Sequence ☐ Proportion ☐ Round Robin ☐ All Proportion ☐ Least Cost Routing

Route Prioritization Type For Equal Cost Routes: Sequence

Local Routes

☐ Pass Only Local Routes ☐ Prioritize Local Routes ☒ Do Nothing

Flags

☐ Continue Number Translation ☐ Continue CNAM Translation ☐ No Connect Signal To Be Sent


Routes

Type	Endpoint 1	Endpoint 2	IP Peer	Sequence	Proportion	Status	TAR Action	TAR Loc...	DM/PM R...	Apply Later	Testing	Cost	Skip LR	STI Type
GSX Gate...	PSTN_TG	ZOOM		1	0	In Service	Normal	0		Do Not A...	Normal	1000000	Disabled	0

New Open Delete

Save Cancel Delete

Create a new Route and attach the Gateway and Trunk Group as created earlier.


Route
✕

Type: GSX Gateway

Gateway: ZOOM

Trunk Group: PSTN_TG

IP Peer: <None>

Sequence: 1

Proportion: 0

Cost: 1000000

TAR Action: Normal

TAR Location: 0

DM/PM Rule: <None>
☐ Apply Later

Testing:
☒ Normal
☐ Test
☐ Non-Test

☒ In Service
☐ Skip Local Recursion

☐ Signing
☐ Local Tagging
☐ Verification

OK
Cancel

Zoom Leg Configuration

IP Signaling Profile (IPSP)

Create an IP Signaling Profile with appropriate signaling flags towards Zoom.

Figure 16: IP Signaling Profile

Host: 172.16.100.216 @ 4330
Master (SWe) - V12.02.02R000

View: IP Signaling Profile Close All

Perspective: Full View

IP SIGNALING PROFILE: ZOOM_IPSP

Common IP Attributes - Communicating With The Peer Regardless Of Call Direction

<input type="checkbox"/> Accept Alert Info	<input type="checkbox"/> No Content Disposition
<input type="checkbox"/> Add P-Charging Function Addr	<input type="checkbox"/> No Port Number 5060
<input type="checkbox"/> Add Path/Service Route Per TG	<input type="checkbox"/> No Userinfo In Contact Header
<input type="checkbox"/> Audio Codec Change through Empty TCS	<input type="checkbox"/> Only Selected Codec In Session Refresh
<input type="checkbox"/> Call Hold Interworking	<input type="checkbox"/> Override Relay For Non SIP Egress Leg
<input type="checkbox"/> Calling Party Type Number If Present	<input checked="" type="checkbox"/> P-Called-Party-Id-Support
<input type="checkbox"/> Clearmode For Data Calls	<input type="checkbox"/> P-ChgMsg-Info
<input type="checkbox"/> Create P-Charging-Vector	<input type="checkbox"/> Relay Data Path Mode Changes To The Other Leg
<input type="checkbox"/> Create P-Visited-Network Id	<input type="checkbox"/> Reject REFER
<input type="checkbox"/> Create Path Header	<input type="checkbox"/> Replace Host On Via Header
<input type="checkbox"/> Create Service-Route Header	<input type="checkbox"/> Reject REFER With IP
<input type="checkbox"/> Customized Session Timer Behavior	<input type="checkbox"/> Reject REFER With TN
<input type="checkbox"/> Disable Also Header	<input type="checkbox"/> ReQuery PSX on REGISTER Refresh
<input type="checkbox"/> Disable Constrained Capacities	<input type="checkbox"/> Restrict History Info Header
<input type="checkbox"/> Disable Host Translation	<input type="checkbox"/> Route Using Received FQDN
<input type="checkbox"/> Disable Media Lock Down	<input type="checkbox"/> SDP O-line Only Compares
<input type="checkbox"/> Disable Refer-To URI Parameters	<input type="checkbox"/> Send All Allowed Codecs For Late Media Invite Or Re-Invite
<input type="checkbox"/> Discard Received Reason Header	<input type="checkbox"/> Send Direct Media Info In SDP Attribute
<input type="checkbox"/> Do Not Include SS Attribute In Re-INVITE	<input type="checkbox"/> Send Empty TCS

<input type="checkbox"/> Don't Send REFER With IP	<input type="checkbox"/> Send Only Preferred Codec
<input type="checkbox"/> Don't Send REFER With TN	<input type="checkbox"/> Send PTIME In SDP
<input type="checkbox"/> End To End BYE	<input type="checkbox"/> Send RTCP Port In SDP
<input type="checkbox"/> End To End RE-INVITE	<input type="checkbox"/> Session Timer Refresh Update
<input type="checkbox"/> End To End UPDATE	<input type="checkbox"/> Set Accept Header To Application SDP Only
<input type="checkbox"/> Suppress End To End Session Refresh	<input type="checkbox"/> Set Oline Dash
<input type="checkbox"/> End To End PRACK	<input type="checkbox"/> Set Session Version Zero
<input type="checkbox"/> Enable Default PUI Procedures	<input type="checkbox"/> Set Sline Dash
<input type="checkbox"/> Enable Dial String Handling	<input type="checkbox"/> Store P-Charging Function Addr
<input type="checkbox"/> Include G729 with G729A when offer PSP has G729A	<input checked="" type="checkbox"/> Store P-Charging Vector
<input type="checkbox"/> Include IP Ports In FROM And TO Headers	<input type="checkbox"/> Store Path Header
<input type="checkbox"/> Include Reason Header (Q.850)	<input type="checkbox"/> Store Service-Route Header
<input type="checkbox"/> Include SS Attribute In Initial Invite	<input type="checkbox"/> Suppress Min-SE if not received
<input type="checkbox"/> Include Transport Type In Contact Header	<input type="checkbox"/> Terminal Portability Interworking
<input type="checkbox"/> Insert Peer Address As Top Route Header	<input type="checkbox"/> Send RTCP BandWidth Info
<input type="checkbox"/> Lockdown Preferred Codec	<input type="checkbox"/> Validate Access Nw Info Header
<input type="checkbox"/> Map Cause Location	<input type="checkbox"/> Use Psx Route for Registered Invite
<input type="checkbox"/> Map SGD In P-Sig-Info Header	<input type="checkbox"/> From Header Anonymisation
<input type="checkbox"/> Map Suspend/Resume Event In P-Svc-Info Header	<input type="checkbox"/> Create ISUP Message Body

<input type="checkbox"/> Map UII In P-Sig-Info Header	<input type="checkbox"/> Disable Transparently Passing ISUP Message Body
<input type="checkbox"/> MIME Cause Precede Reason Header Cause	<input type="checkbox"/> aiToPemInterworking
<input type="checkbox"/> Minimize Relaying Of Media Changes From Other Call Leg	<input type="checkbox"/> Send SBC Supported Codecs For Late Media Re-Invite
<input type="checkbox"/> No Service Route Hdr For Emergency Registration	
<input type="checkbox"/> Publish IP In Hold SDP	
<input type="checkbox"/> Insert PAccess Network Info	
<input type="checkbox"/> Contact Transparency For Isfocus Media Tag	
<input type="checkbox"/> Support S-CSCF Restoration Procedures	
<input type="checkbox"/> Insert UE Flow Info	
<input type="checkbox"/> Include SIP Reason Header	
Call Preservation Flags	
<input type="checkbox"/> Call Preservation	
Call Preservation Time Out: 5	
Call Transfer Flags	
Handle IP Addresses Not Present In Network Selector Table (NST): Route Via Transferring IPTG	
<input type="checkbox"/> Force Re-Route Via PSX Query	
<input type="checkbox"/> Skip Re-Route Via PSX Query	
Local Media Control Flags	
<input type="checkbox"/> Enable HOLD on REFER	
Option Tag In Require Header	
<input type="checkbox"/> Suppress Replace Tag	

PreConditions Profile	
<input type="checkbox"/> State	
<input type="checkbox"/> Support If Egress IPTG	<input type="checkbox"/> Strength Optional Policy
<input type="checkbox"/> Strength Mandatory Policy	<input type="checkbox"/> UPDATE Preconditions Policy
Strength Mandatory Priority:	1
Strength Optional Priority:	1
UPDATE Preconditions Priority:	1
Relay Flags	
<input type="checkbox"/> Conference Event Package	<input type="checkbox"/> PUBLISH
<input type="checkbox"/> Dialog Event Package	<input type="checkbox"/> REFER
<input type="checkbox"/> DTMF Body	<input type="checkbox"/> Reg Event Package
<input type="checkbox"/> Force 503 To 500 Relay	<input type="checkbox"/> Ribbon Media Body
<input type="checkbox"/> Info	<input type="checkbox"/> Status Code 3XX
<input type="checkbox"/> Message	<input type="checkbox"/> Status Code 4XX-6XX
<input type="checkbox"/> Notify	<input type="checkbox"/> Third Party Bodies
<input type="checkbox"/> Options	<input type="checkbox"/> Update without SDP
Refer To Header Relay	
<input checked="" type="radio"/> Reject the REFER request if no match is found <input type="radio"/> relay the REFER request if no match is found <input type="radio"/> relay the REFER request without matching	
Transparency Flags	
<input type="checkbox"/> Accept-Contact Header	<input type="checkbox"/> Reason Header
<input type="checkbox"/> Accept-Language Header	<input type="checkbox"/> Referred-By Header

<input type="checkbox"/> Accept Header	<input type="checkbox"/> Resource Priority Option Tag
<input type="checkbox"/> Alert Information Header	<input type="checkbox"/> Request-URI
<input type="checkbox"/> Allow Header	<input type="checkbox"/> Resource-Lists Body
<input type="checkbox"/> Authcode Headers	<input type="checkbox"/> RLMI Body
<input type="checkbox"/> Call-Info Header	<input type="checkbox"/> Route Header
<input type="checkbox"/> Contact Header*	<input type="checkbox"/> Server Header
<input type="checkbox"/> Error Info	<input type="checkbox"/> Service-Route Header
<input type="checkbox"/> Event Header	<input type="checkbox"/> Simple-Filter Body
<input type="checkbox"/> External Body	<input type="checkbox"/> SIP Body
<input type="checkbox"/> From Header	<input type="checkbox"/> SIPFRAG Body
<input type="checkbox"/> Geo Location Error	<input type="checkbox"/> Target-Dialog Header
<input type="checkbox"/> Geo Location Header	<input type="checkbox"/> To Header
<input type="checkbox"/> Geo Location Route	<input type="checkbox"/> Tone Body
<input type="checkbox"/> History Info	<input type="checkbox"/> Unknown Body
<input type="checkbox"/> Image Body	<input type="checkbox"/> Unknown Header
<input type="checkbox"/> Max_forwards Header	<input type="checkbox"/> User-Agent Header
<input type="checkbox"/> MWI Body	<input type="checkbox"/> User-To-User Header
<input type="checkbox"/> Pass Complete Contact Header	<input type="checkbox"/> Via Header
<input type="checkbox"/> P-Access-Network-Info Header	<input type="checkbox"/> Warning Header
<input type="checkbox"/> P-Called-Party-Id	<input type="checkbox"/> Watcherinfo Body

<input type="checkbox"/> P-Charging-Vector Header	<input type="checkbox"/> X-ATP
<input type="checkbox"/> P-Early-Media	
<input type="checkbox"/> P-Visited-Network ID Header	
<input type="checkbox"/> Path Header	
<input type="checkbox"/> Pdf Body	
<input type="checkbox"/> Pdf-Diff Body	
<input type="checkbox"/> QSIG Body	
<input type="checkbox"/> PDCS-Billing Info Header	
<input type="checkbox"/> Transparency	
<input type="checkbox"/> Include Privacy	
<input type="checkbox"/> Sip In Core	
<input type="checkbox"/> Use SIP In Core	
<input type="checkbox"/> Header Encryption Flags	
<input type="checkbox"/> Path Header	<input type="checkbox"/> Service Route Header
<input type="checkbox"/> Subscription Package Support	
<input type="checkbox"/> Support Reg Event	<input type="checkbox"/> Use PSX Route For SBC Initiated Subscribe
<input type="checkbox"/> Registrar Recovery	
<input type="checkbox"/> Register to Alternate on Primary Down	<input type="checkbox"/> Override Internal Expires Timer
<input type="checkbox"/> Revert to Primary On Recovery	<input type="checkbox"/> Deregister Alternate on Primary Recovery
Egress IP Attributes - Sending A Call In The Forward Direction To The Peer	

Egress IP Attributes - Sending A Call In The Forward Direction To The Peer

IP Protocol Type: ☒ SIP Only ☐ SIP-I ☐ H.323 ☐ Wireless

IP Signaling MIME Content Type: ISUP

IP Signaling Treatment: Interwork

MIME Content Type Version: 1 - ansi88

Globalize Number Profile: DEFAULT_IP

Phone-Context Parameter Length: 0

Media Qos Kpi Profile: <None>

Signaling Qos Kpi Profile: <None>

Flags

<input type="checkbox"/> Accept 3XX With RN	<input type="checkbox"/> Qos Based Routing
<input type="checkbox"/> BGCF Target Scheme Transparency	<input type="checkbox"/> Prefix RN to Dialed Digits
<input type="checkbox"/> Convert Inactive To Sendrecv	<input type="checkbox"/> Reject 3XX With IP
<input type="checkbox"/> Delay Cut Through	<input type="checkbox"/> Reject 3XX With TN
<input checked="" type="checkbox"/> Disable 2806 Compliance	<input type="checkbox"/> Same CallId For Required Authorization
<input type="checkbox"/> Disable Optional Register Parameters	<input type="checkbox"/> Transit PAI From Unregistered Peer
<input checked="" type="checkbox"/> Disposition Handling Required	<input type="checkbox"/> Suppress UNREGISTER
<input type="checkbox"/> Don't Send Fast Start Proposal	<input type="checkbox"/> TTC-ISUP Mapping
<input type="checkbox"/> Enable 3261 Cancel Handling	<input type="checkbox"/> Use Called Party In Request URI
<input type="checkbox"/> Include ENUM Parameters	<input type="checkbox"/> Use Colon In SDP Media Type Parameter
<input type="checkbox"/> Insert In Band Indication	<input type="checkbox"/> Use JIP from 3XX Response in PDCS-Billing-Info-Header

<input type="checkbox"/> Add Loop Back Route Header	<input type="checkbox"/> Validate ISUB Address
<input type="checkbox"/> Map 181 Or 182 Message To 183	<input type="checkbox"/> Wait Till Connect Before Abandon FastStart
<input type="checkbox"/> Map 3xx Contact URL To Route Header	<input type="checkbox"/> Restrict User Equal To Phone
<input type="checkbox"/> Map Contractor Number In P-Sig-Info Header	<input type="checkbox"/> Ignore SDP After Offer Answer Completed
<input type="checkbox"/> Use Network Provided Screening Indicator For Calling Number	<input type="checkbox"/> Map Diversion Header To Charge Number
<input type="checkbox"/> MonitorRtpOnEgressUpdate	<input type="checkbox"/> Map RN, OCN, RDI To Diversion Header
<input type="checkbox"/> Honor Subsequent SDP Answer	

BCI

<input type="checkbox"/> BCI Interwork Encountered	<input type="checkbox"/> BCI ISDN Access
----------------------------------------------------	------------------------------------------

Carrier Information

<input type="checkbox"/> Disconnect If Neither Terminating CA Nor CIC Received	<input type="checkbox"/> Use Terminating CIC From SIP
<input type="checkbox"/> Use Terminating CA From SIP	

Domain Name

<input type="checkbox"/> Preserve Ingress FROM Domain Name	<input type="checkbox"/> Use Lower Case Domain Names
<input type="checkbox"/> Preserve Ingress R-URI Domain Name	<input type="checkbox"/> Use SIP Domain Name In FROM Field
<input type="checkbox"/> Use IP Signaling Peer Domain In R-URI	<input type="checkbox"/> Use Zone Level Domain Name In Contact
<input type="checkbox"/> Use DM/PM Manipulated Host Name In R-URI	<input type="checkbox"/> Use SIP Domain Name In Request URI
<input type="checkbox"/> Use Zone Level Domain Name In Path Header	<input type="checkbox"/> Use Called URI As R-URI
<input type="checkbox"/> Use SIP Domain Name In PAI Header	

ISUB

IP SIGNALING PROFILE: ZOOM_IPSP

- ☐ Allow NSAP ISUB ☐ Include Called Party ISUB
☐ Allow User Specified ISUB ☐ Include Calling Party ISUB

Number Portability Attributes

NPDI Options: ☒ Include npdi ☐ Include npdi=yes ☐ Do Not Include npdi

Flags

- ☐ Disable rn

Privacy

- ☐ Transparency
☐ AnonymizeHostIpAddress

Privacy Information: ☒ P-Preferred-ID ☐ P-Asserted-ID ☐ Remote-Party-ID

Flags

- ☒ Include Privacy ☐ Privacy Required by Proxy
☐ MS Lync Privacy Support ☐ Include Embedded PAI Header in Redirected INVITE
☐ Do Not Include Tel URI In PAI Header

Redirect

Mode: Accept Redirection

Contact Handling: ☒ Merge Received Contacts with Existing Contacts ☐ Purge Existing Contacts

Flags

- ☐ Skip Crankback Profile And Always Crankback ☐ Honor Embedded Headers in 3xx
☐ Force Re-query for Redirection ☐ Enhanced Local Redirection

SIP Cause Mapping

Internal To SIP Cause Mapping: 1 - DEFAULT

SIP To Internal Cause Mapping: 1 - DEFAULT

Internal to SIP Cause Mapping Profile Name

SIP to Internal Cause Mapping Profile Name

SIP Headers And Parameters

Include Charge Information: ☒ Include None ☐ Include P-Charge-Info

Session-Expires Refresher: ☒ Not Send ☐ UAC ☐ UAS

SIP TO Header Mapping: ☒ None ☐ Original Called Number (OCN) ☐ Called Number ☐ GAP Dialed Number

☐ PI Allowed Send CPC In: ☒ DEFAULT ☐ FROM ☐ PAI ☐ BOTH

Destination Trunk Group Options: Include None

Originating Trunk Group Options: Include None

Generate Call-ID Using: Do not Use Ingress Call-Id

Flags

- ☐ Include CIC ☐ Include PSTN Parameters
☐ Include CPC Information ☐ Include Qvalue
☐ Include NPI ☐ Skip CSeq Check In Early Dialog
☐ Include OLIP ☒ Transparency For Destination Trunk Group Parameter
☐ Include P-K-Adn ☐ End To End Ack
☐ No CDR Change In End To End Ack

Call Forwarding

☐ Diversion-History Info Interworking (RFC 6044 compliance)

Redirection Information

☒ Diversion
☐ Diversion With Transparency

☐ PK Header

History Information

☐ Include History-Info
☐ Cause Parameter In RFC 4458
☐ Reason With Cause Value As Per RFC 4244

CPC Mapping Flags

☐ Map CPC when Presentation Indicator is Restricted

☒ Any CPC
☐ CPC=Priority

Send CPC Param In

☒ Default
☐ PAI
☐ From
☐ Both (PAI and From)

P Charge Info

☐ Transparency

P-Charge-Info Information:
☒ URI Parameter
☐ User Parameter
☐ Header Parameter

Flags

☐ Include NPI
☐ Include NOA

SIP RPH ETS

Action For ETS 400 Response With 417 Reason Code:

Retry Without ETS

ETS Default Priority Value:

0

Flags

☐ Add/Modify ETS Resource Priority Header
☐ Use Incoming ETS Resource Value

☐ Do Not Include Require RPH

Transport Type

Transport Type 1:

TLS Over TCP

Transport Type 2:

<None>

Transport Type 3:

<None>

Transport Type 4:

<None>

☐ Use configured transport for egress leg

Ingress IP Attributes - Signaling Back A Message To The Peer That We Receive A Call From

Flags

☐ 181 Supported
☐ 182 Supported
☐ Convert Progress To Alert
☐ Don't Send Facility Message
☐ Don't Send 3XX With IP
☐ Don't Send 3XX With TN
☐ Map Called Party Category In P-Sig-Info Header
☐ No SDP In 180 Supported
☐ Refuse Fast Start Proposal

☐ Registration Support 3xx
☐ Send 183 On Initiating Disconnect Treatment
☐ Send Fast Start Response In CP
☐ Send SDP In 200 OK If 18x Reliable
☐ Send Updated SDP In 200 OK
☐ Send SDP In Subsequent 18x
☐ Send TLS Connection Failure Response
☐ Suppress 183 For 3xx Redirect Response
☐ Suppress 183 Without SDP

<input type="checkbox"/> Registration Expires in Expires Header <input type="checkbox"/> Map Subsequent 180 to 183 <input type="checkbox"/> Early Media Authorization <input type="checkbox"/> Report Early Media Auth	<input type="checkbox"/> Override 3xx Relay <input type="checkbox"/> Send BIT-H Of BCI In Outgoing Invite <input type="checkbox"/> Convert Alert To Progress
Carrier Information <input type="checkbox"/> Generate Terminating CA	
History Information <input type="checkbox"/> Include History-Info <input type="checkbox"/> Cause Parameter In RFC 4458 <input type="checkbox"/> Reason With Cause Value As Per RFC 4244	
Access Transfer Profile: <None>	
Trf Parameters Preferred Trf Uri: <input style="width: 80%;" type="text"/> Preferred Mrb Uri: <input style="width: 80%;" type="text"/>	
Enum Parameters TTL: <input style="width: 80%;" type="text" value="0"/>	
<input type="button" value="Save"/> <input type="button" value="Cancel"/> <input type="button" value="Delete"/>	

Packet Service Profile (PSP)

Create a Packet Service Profile (PSP) for the Zoom leg. The PSP is attached to the TrunkGroup that is created later in this section. Since there is an SRTP between the SBC Core and Zoom, you must create a crypto suite profile as shown below:

Figure 17: Packet Service Profile

Packet Service Profile: ZOOM_PSP	
Silence Factor:	40
Voice Initial Playout Buffer Delay (ms):	10
Type Of Service:	0
AAL1 Payload Size:	47
Preferred RTP Payload Type For DTMF Relay:	<None>
Media Packet COS:	0
Monitoring Profile:	<None>
Codec Entry	
Codec Entry:	<None>
<input type="button" value="Add"/> <input type="button" value="Update"/>	
Codec Entry	Value
1	G711Ulaw_T38_2833

Media Control: IPv4 Only

T.38

Number of Redundant Packets ☐ 0 ☒ 1

Low Speed Number of Redundant Packets ☐ 0 ☒ 1

T.38v0 Maximum Bit Rate ☐ 2.4 kbits/s ☐ 4.8 kbits/s ☐ 9.6 kbits/s

Data Rate Management Type ☐ Type 1 - Local Generation of TCF ☒ Type 2 - Transfer of TCF

Use Max Bit Rate Only ☒ Disabled ☐ Enabled

ECM ☐ ECM Preferred

T38FaxMaxDatagram Size without Redundancy ☒ Disabled ☐ Enabled

T.38FaxProtocolVersion: T.38(v0)

Honor Remote Precedence ☒ Disabled ☐ Enabled

Send Route PSP Precedence ☐ Disabled ☒ Enabled

Packet-To-Packet Control

Transcode ☐ Only ☒ Conditional ☐ Determined By PSP For Other Leg ☐ Transcode

Conditions In Addition To "No Common Codec"

☐ Apply Fax Tone Treatment ☐ Different Silence Suppression

☐ Different DTMF Relay ☐ Honor Answer Preference

☐ Different Packet Size ☐ Honor Offer Preference

☐ Different 2833 Payload Type

Codecs Allowed For Transcoding

This Leg: ☐ G.711 A ☐ G.711 U ☐ G.722 ☐ G.722.2 ☐ G.723.1 ☐ G.726 ☐ G.729 ☐ OPUS ☐ EVS ☐ SILK ☐ T.38 ☐ iLBC ☐ AMR

Other Leg: ☐ G.711 A ☐ G.711 U ☐ G.722 ☐ G.722.2 ☐ G.723.1 ☐ G.726 ☐ G.729 ☐ OPUS ☐ EVS ☐ SILK ☐ T.38 ☐ iLBC ☐ AMR

RTCP

☒ RTCP Packet Loss Threshold (Packets Lost/100,000 Packets): 0

RR Bandwidth: 250

RS Bandwidth: 250

Packet Loss Action ☒ None ☐ Trap ☐ Trap And Disconnect

☐ Enable RTCP Only For HELD Calls ☐ Termination For Pass-Through Calls

☐ RTCP-MUX ☐ Generate RTCP for T140 if not received from other leg

RTCP-XR

☐ Relay ☐ Relay Or Terminate

Secure RTP/RTCP

Crypto Suite Profile: ZOOM_CRYPT

Flags

☒ Allow Fallback ☒ Enable SRTP

☐ Reset ROC On Session Key Change ☐ Reset Enc/Dec/ROC on Decryption Key Change

☐ Update Crypto On Modify ☐ Allow Pass Through

DTLS/SRTP

Crypto Suite Profile: <None>

Flags

☐ Allow Fallback ☐ Enable DTLS

☐ Relay DTLS SRTP ☐ Relay DTLS SCTP

Flags

☐ DSCP Passthrough ☐ Interwork DTMF OOB-2833 Without Transcoding

☐ Digit Detect Send Enabled ☐ Use Direct Media

☐ Disallow Data Calls ☐ Validate Peer Support for DTMF Events

☐ SSRC Randomize ☐ HD Codec Preferred

☐ Reserve BW for Preferred Audio Common Codec ☐ Prefer NB PassThru Over HDTranscode

☐ Police on Heaviest Audio Codec ☐ Match Offered Codec Group If Nb Only

☐ t140 Call ☐ Force Route PSP Order

☐ Allow Audio Transcode For MultiStream Call ☐ Allow Mid Call SSRC Modification

☐ Generate and Signal SSRC and CName

< >

Save Cancel Delete

Packet Service Profile ID Group

Create a Packet Service Profile ID Group and attach the Packet Service Profile created earlier.

Figure 18: Packet Service Profile ID Group

Packet Service Profile ID Group: ZOOM

Packet Service Profile: ZOOM_PSP

HPC Packet Service Profile: <None>

IP Signaling Peer Group

Add the Zoom IP Address as shown below:

Figure 19: IP Signaling Peer Group

IP Signaling Peer Group: ZOOM_IPP

Description:

Flags

☐ Send All Peer IP Addresses/FQDNs

Peer Group Data

Sequence Number: 0

☒ IPv4 Address: 162 . 12 . . Port Number: 5060

☐ IPv6 Address: 0 : 0 : 0 : 0 : 0 : 0 : 0 : 0 Port Number: 0

☐ Server FQDN: Port Number: 0

☒ In Service

Add/Update

Sequence Number	IP Address	Port Number	Server FQDN	Port Number	Send	Service Status
0	162.12 . .	5060		0	IP Address	In Service

Trunk Group

Create a SIP Trunk Group towards Zoom and assign the corresponding profiles like LRBT, PSP, IPSP created in earlier steps.

Figure 20: Trunk Group

Trunk Group:	ZOOM_TG	<input type="checkbox"/> Unrestricted
Gateway:	ZOOM	
Description:		
Auto Recall Profile:	<None>	
Call Processing Localization Variant:	North America	
Calling Area:	<None>	
Carrier:	0000	
Carrier Selection Priority:	<None>	
Country:	1 - USA, Canada and Caribbean	
DDI Range Profile:	<None>	
Destination Switch Type:	Access	
Direction:	Two Way	
Element Routing Priority Profile:	ZOOM_ERP	
Feature Control Profile:	DEFAULT_SIP_FC_ACCESS	
IP Signaling Profile:	ZOOM_IPSP	
LATA:	<None>	
Local Recursion Profile:	<None>	
Maximum Satellite Hops:	Three or More Satellite Hops	
Network Data Partition:	0	
Network Data Net:	0	
Next Hop Domain:	<None>	
Number Analysis Profile:	<None>	
Number Length Enforcement:	<None>	
Originating Carrier:	<None>	
PPR Profile:	<None>	
Pseudo Carrier:	<None>	
Remote Sip Peer Type:	None	
Region:	<None>	
Routing Criteria Profile:	DEFAULT_IP	
SCP Business Service Group:	0	
Signaling Profile:	DEFAULT_IP_PROFILE	
Signaling Flag:	SIP	
SIP Domain:	<None>	
SIP Response Code Profile:	<None>	
TDM Type:	Other	
Tone And Announcement Profile:	ZOOM_LRBT	
Trunk Group COS:		
Trunk Group COS Profile:	<None>	
Trunk Group Domain:	<None>	
Trunk Number:		
Zone Index Profile:	DEFAULT	
ZZ Profile:	<None>	
Charge Band Profile:	<None>	

Ingress

Charge Indicator: None

Default CPC: <None>

Default OLIP: <None>

Dial Plan Profile: <None>

Forced OLIP Value: <None>

In DM/PM Rule: <None>

Info Transfer Capability Profile: <None>

IP Version Preference: IPv4 Only

ONI:

JIP:

NPA:

Numbering Plan: NANP_ACCESS

In Policy Profile Group: <None>

Flags

<input type="checkbox"/> Allow Hex Digits In Cdpn	<input type="checkbox"/> Non-Zero Video Bandwidth Based Routing for H.323
<input type="checkbox"/> Discard NPDI	<input type="checkbox"/> Non-Zero Video Bandwidth Based Routing for SIP
<input type="checkbox"/> Discard RN	<input type="checkbox"/> Overlap Dialing
<input type="checkbox"/> HD Preferred Routing	<input type="checkbox"/> TNS Circuit Code Based Routing
<input type="checkbox"/> HD Supported Routing	<input checked="" type="checkbox"/> Use IPTG Routing (Hop By Hop Routing) For Ingress

Egress

Charge Indicator: None

Out DM/PM Rule: <None>

Out Policy Profile Group: <None>

Trunk Context:

R-URI Host: R-URI Host Port: 0

Flags

<input type="checkbox"/> Disable Crankback
<input type="checkbox"/> Enable JIP Interwork
<input type="checkbox"/> Use Preferred Identity

Billing

Billing Plan: <None>

Billing Information: <None>

Default Billing Number:

Nature Of Address: <None>

Numbering Plan Indicator: <None>

Calling Party Number

Calling Party:

Nature Of Address: <None>

Numbering Plan Indicator: <None>

IPSG

IP Signaling Peer Group: ZOOM_IPP

☐ IP Peer Supported

Packet Service Profile ID Group: ZOOM

☒ Egress IP Signaling Profile: ZOOM_IPSP

Packet Service Profile

Preferred Packet Service Profile ID Group: <None>

☐ Destination Override

Traffic Management Options

Trunk Group Reservation Level 1: 10

Trunk Group Reservation Level 2: 5

VPN Information

Business Group: <None>

Business Location: <None>

☐ Business Group From CLI

Services

☒ Not Screened ☐ Screened - Normal ☐ Screened - Fraud

Class Of Service: DEFAULT_IP

Service Exception Profile: <None>

Save Cancel Delete

Routing Label

Configure the Routing Label as follows:

Figure 21: Routing Label

Routing Label: ZOOM_RL

Action

☒ Routes ☐ Script ☐ Route Hopping ☐ LCR

Number Of Routes Requested: 10 ☒ All

Number Of Routes Per Call: 10

Script: <None> [Runtime Variables](#)

Partition: <None>

DM/PM Rule: <None> ☐ Apply Later

CPC Screening: <None>

Overflow Number:

Overflow Nature Of Address: <None>

Overflow Numbering Plan Indicator: <None>

Call Parameter Filter Group: <None>

Call Parameter Filter Profile Script: <None>

Routing Criteria

☐ Use Entity Type: <None>

Partition

☒ Ignore ☐ Do not Use ☐ Use

Destination

☒ Ignore ☐ Do not Use ☐ Use

Route Prioritization Type

☒ Sequence ☐ Proportion ☐ Round Robin ☐ All Proportion ☐ Least Cost Routing

Route Prioritization Type For Equal Cost Routes: Sequence

☐ Use TAR Routes

TAR Route Prioritization Type

☒ Sequence ☐ Proportion ☐ Round Robin ☐ All Proportion ☐ Least Cost Routing

Route Prioritization Type For Equal Cost Routes: Sequence

Local Routes

☐ Pass Only Local Routes ☐ Prioritize Local Routes ☒ Do Nothing

Flags


☐ Continue Number Translation ☐ Continue CNAM Translation ☐ No Connect Signal To Be Sent

Routes

Type	Endpoint 1	Endpoint 2	IP Peer	Sequence	Proportion	Status	TAR Action	TAR Loc...	DM/PM R...	Apply Later	Testing	Cost	Skip LR	STI Type
GSX Gate...	ZOOM_TG	ZOOM		1	0	In Service	Normal	0		Do Not A...	Normal	1000000	Disabled	0

New Open Delete

Create a new Route and attach the Gateway and Trunk Group as created earlier.

 **Route** ✕

Type: GSX Gateway

Gateway: ZOOM

Trunk Group: ZOOM_TG

IP Peer: <None>

Sequence: 1

Proportion: 0

Cost: 1000000

TAR Action: Normal

TAR Location: 0

DM/PM Rule: <None> ☐ Apply Later

Testing: ☒ Normal ☐ Test ☐ Non-Test

☒ In Service ☐ Skip Local Recursion

☐ Signing ☐ Local Tagging ☐ Verification

OK Cancel

Standard Route

PSX uses this entry to route all the calls coming from the PSTN towards ZOOM End Points.

Figure 22: Standard Route 1

Entity Type:	Trunk Group		
Trunk Group:	PSTN_TG		<input type="checkbox"/> All
GATEWAY:	ZOOM		
Not Applicable			
<input checked="" type="radio"/> Call Parameter Filter Profile:	<None>		
<input type="radio"/> Call Parameter Filter Profile Group:	<None>		
Destination National:			
Destination Country:	1 - USA, Canada and Caribbean		
Domain Name:	<None>		
<input checked="" type="radio"/>			
<input type="radio"/> IP Address:	0	0	0
Partition:	DEFAULT		
Routing Label:	ZOOM_RL		
Call Type	<div> <div></div> <div> <input checked="" type="checkbox"/> All Call Type Bits </div> </div>		
Transmission Medium	<div> Speech 3.1 KHz Audio 7.0 KHz Audio 56 kbps 64 kbps Packet Multirate 384 kbps 1536 kbps </div>		
Time Range:	ALL		
<input type="checkbox"/> Always Confirm Changes <div> <div>Create New</div> <div>Save</div> <div>Cancel</div> <div>Delete</div> </div>			

PSX uses this entry to route all the calls coming from ZOOM towards PSTN End Points.

Figure 23: Standard Route 2

Entity Type:	Trunk Group		
Trunk Group:	ZOOM_TG		<input type="checkbox"/> All
GATEWAY:	ZOOM		
Not Applicable			
<input checked="" type="radio"/> Call Parameter Filter Profile:	<None>		
<input type="radio"/> Call Parameter Filter Profile Group:	<None>		
Destination National:			
Destination Country:	1 - USA, Canada and Caribbean		
Domain Name:	<None>		
<input checked="" type="radio"/>			
<input type="radio"/> IP Address:	0	0	0
Partition:	DEFAULT		
Routing Label:	PSTN_RL		
Call Type	<div> <div></div> <div> <input checked="" type="checkbox"/> All Call Type Bits </div> </div>		
Transmission Medium	<div> Speech 3.1 KHz Audio 7.0 KHz Audio 56 kbps 64 kbps Packet Multirate 384 kbps 1536 kbps </div>		
Time Range:	ALL		
<input type="checkbox"/> Always Confirm Changes <div> <div>Create New</div> <div>Save</div> <div>Cancel</div> <div>Delete</div> </div>			

Section D: Zoom Web BYOC Configuration

Prerequisites :

- **Zoom Go BYOC account:** This is a special type of Zoom account that has an outbound/inbound SIP trunk that peers between the Zoom Phone Cloud and the customer's PSTN carrier connection.
- **Customer's existing carrier/carrier equipment:** Any carrier offering PSTN services and the carrier equipment can be router/gateway or another SBC that supports SIP trunk connectivity. The carrier provides several DID's to use as external BYOC numbers.
- **Trunk Registration:** BYOC is a "static" trunk between two static IP endpoints, therefore no trunk registration is done here.



Note

Ensure a Zoom BYOC SIP trunk is built between the Zoom SBC and the Ribbon SBC Core deployed on a customer site.

Once the Zoom Go account is available, login to the Zoom Web BYOC portal at <https://go.zoom.us/>.

The following Zoom BYOC configurations are included in this section:

1. [Add External Number](#)
2. [Create Zoom Users](#)
3. [Supplementary Services Configuration on Zoom](#)

Add External Number

1. Navigate to **Phone Systems Management > Phone Numbers > External**.

Figure 24: Add External Number

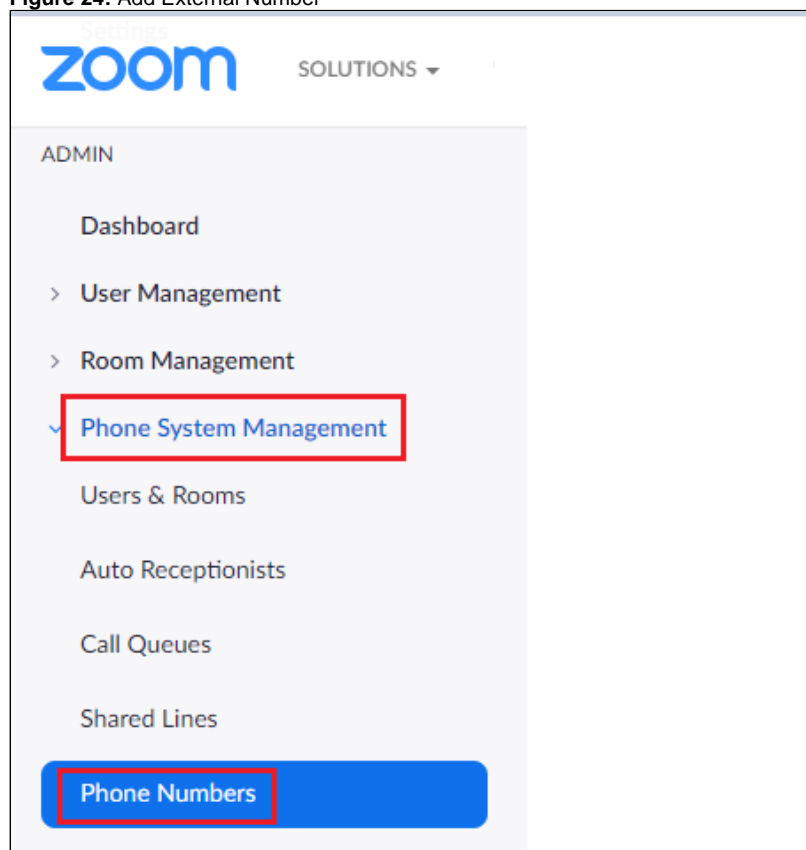


Figure 25: External

Assigned Unassigned Ported **External**

Add your external phone numbers running by your own carriers into Zoom from here. Like other numbers running by Zoom, you can assign these numbers to any extensions.

Add Import Export

Search by Numbers

2. Select **Add** to add external phone numbers provided by your carrier into the Zoom portal. These numbers are the DID numbers provided by your carrier.
3. Select **BYOC** as the carrier.

Figure 26: Add External Number

Add External Numbers

Carrier **BYOC**

Numbers

Example: +19991234567, +19991234568

Enter the existing customer phone numbers (from carrier) separated by commas.

4. Click **Submit**.
5. Verify the external numbers have been created successfully as shown below.

Figure 27: External Number created successfully

Assigned Unassigned Ported **External**

Add your external phone numbers running by your own carriers into Zoom from here. Like other numbers running by Zoom, you can assign these numbers to any extensions.

Add Import Export

Search by Numbers

Number Type (All)

Number	Number Type	Carrier	Country	Submission Date
(512) 567-1233	Toll Number	BYOC	United States	May 8, 2020, 12:05 AM

Create Zoom Users

Zoom Users are created in order to login to Zoom clients on desktop or mobile. The steps for creating a user are as follows:

1. Navigate to **User Management > Users** .
2. Click **Add** to create new Zoom users.
3. Navigate to **Phone System Management > Users & Rooms**.
4. Check that the User status is " **Active** ".
5. Navigate to **Assign Calling Plan > Assign BYOC Calling Plan** .
6. Click " **Confirm and Assign Numbers** ".

Figure 28: Create Zoom User

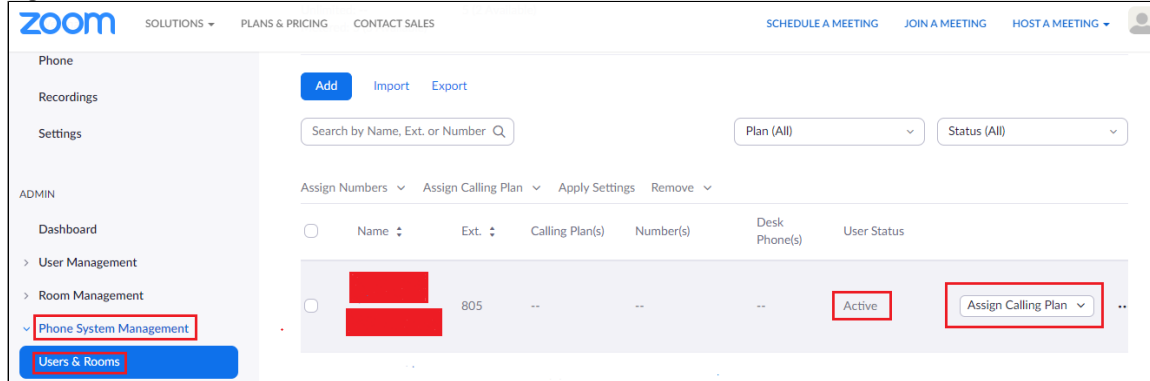
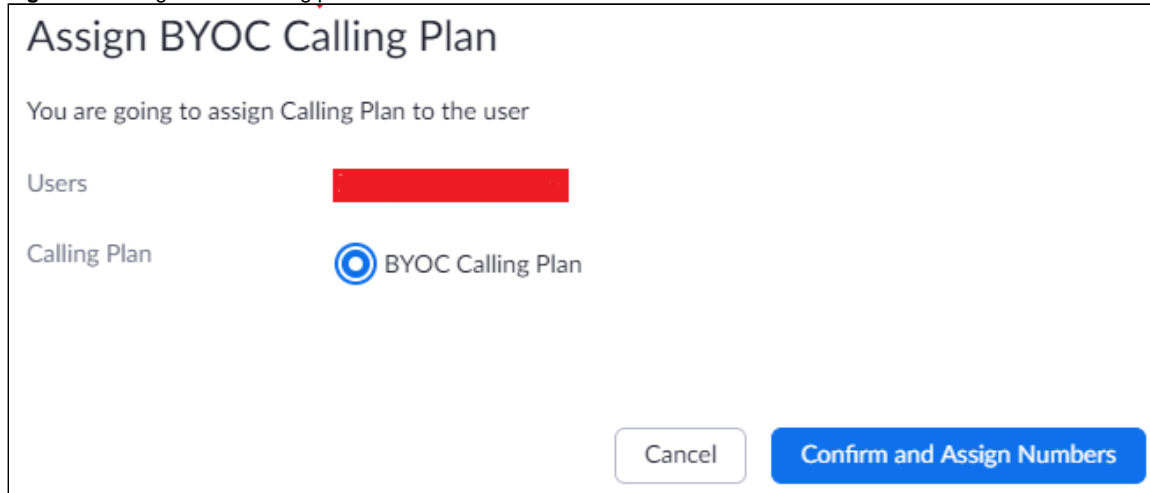


Figure 29: Assign BYOC calling plan



7. Assign the External Numbers created previously in [Add External Number](#).

Figure 30: Choose from Unassigned Numbers

Choose from Unassigned External Numbers

Number Type (All)

<input checked="" type="checkbox"/>	Number	Location	Number Type
<input checked="" type="checkbox"/>	(512) 567-1233 E	United States	Toll Number

Page Size
10
Total 1

8. Click **Confirm** to finish. Once the User is assigned with a Calling Plan and Number, it should look like the following example:

Figure 31: Configured User

Plan (All)

Assign Numbers
Assign Calling Plan
Apply Settings
Remove

<input type="checkbox"/>	Name	Ext.	Calling Plan(s)	Number(s)	Desk Phone(s)	User Status
<input type="checkbox"/>	[REDACTED]	805	BYOC	(512) 567-1233 E	--	Active

Supplementary Services Configuration on Zoom

Zoom supports multiple supplementary services. To configure different supplementary services in Zoom, refer to the following links:

1. Auto Receptionist: https://support.zoom.us/hc/en-us/articles/360001297663-Getting-started-with-Zoom-Phone-admin-#h_a625f531-94c6-4291-909e-3d68ad685b68
2. Call Flip: <https://support.zoom.us/hc/en-us/articles/360034613311-Using-Call-Flip>
3. Shared Line Appearance (SLA) or Call Delegation: <https://support.zoom.us/hc/en-us/articles/360032881731>
4. Shared Line Group/SLG: <https://support.zoom.us/hc/en-us/articles/360038850792/>

Supplementary Services & Features Coverage

The following checklist depicts the set of services/features covered through the configuration defined in this Interop Guide.

Sr.No.	Supplementary Features/Services	Coverage
1	Basic Registration over UDP,TCP & TLS	N/A

2	Basic Call Setup	✓
3	Basic Call Termination	✓
4	Auto Receptionist (Auto Attendant)	✓
5	Call Waiting	✓
6	Call Hold/Resume	✓
7	Call Transfer - Blind (Cold Transfer)	✓
8	Call Transfer - Consult (Warm Transfer)	✓
9	Call Queue	✓
10	Conference	✓
11	Shared Line Group (SLG)	✓
12	Shared Line Appearance (SLA) or Call Delegation	✓
13	Call Recording	✓
14	Call Flip	✓

Legend

✓	Supported
✗	Not Supported
N/A	Not Applicable

Caveats

Note the following items in relation to this Interop:

- Potential issue has been observed on SBC 5400 for long duration calls (HOLD/UNHOLD from PSTN after 30mins) where SBC management tends to go down. The fix for this issue is available in SBC release 9.2 and beyond. We recommend to use SBC 9.2 release.

Support

For any support related queries about this guide, please contact your local Ribbon representative, or use the details below:

- Sales and Support: 1-833-742-2661
- Other Queries: 1-877-412-8867
- Website: <https://ribboncommunications.com/about-us>

References

For detailed information about Ribbon products & solutions, please visit:

<https://ribboncommunications.com/products>

For detailed information about Zoom products & solutions, please visit:

<https://zoom.us/>

Conclusion

This Interoperability Guide describes a successful configuration covering Zoom interop involving the SBC Core. All the necessary features and serviceability aspects stand covered as per the details provided in this interoperability document.

