



ORACLE

Oracle SBC with Analog Devices and Zoom
Phone

Technical Application Note

ORACLE

COMMUNICATIONS




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1 Related Documentation

1.1 Oracle SBC Documentation

- [Oracle® Enterprise Session Border Controller ACLI Configuration Guide](#)
- [Oracle® Enterprise Session Border Controller Release Notes](#)
- [Oracle® Enterprise Session Border Controller Security Guide](#)

1.2 Zoom Phone Documentation

- <https://zoom.us/docs/doc/Zoom-Bring%20Your%20Own%20Carrier.pdf>
- <https://zoom.us/phonesystem>
- <https://zoom.us/zoom-phone-features>

1.3 Poly OBI302 ATA Documentation

- <https://www.poly.com/in/en/products/phones/obi/obi302>

2 Revision History

Version	Date Revised	Description of Changes
1.0	22/07/2021	Initial publication

3 Intended Audience

This document describes how to connect Analog Devices to Oracle SBC and Zoom Phone. This paper is intended for IT or telephony professionals.

Note: To zoom in on screenshots of Web GUI configuration examples, press Ctrl and +.

3.1 Validated Oracle Versions

We have successfully conducted testing with the Oracle Communications SBC versions:

SCZ840p4a

These software releases with the configuration listed below can run on any of the following products:

- AP 1100
- AP 3900
- AP 4600

- AP 6350
- AP 6300
- VME

3.2 Setup Requirements

Analog Telephony Adapter	See Zoom Documentation for More Details
SIP Trunks connected to the SBC	
Zoom Phone	
Public IP address for the SBC	
Public trusted certificate for the SBC	
Firewall ports for Zoom Voice signaling	
Firewall IP addresses and ports for Zoom Voice media	
Media Transport Profile	
Firewall ports for client media	

4 Zoom Phone Configuration

This Section covers the steps required to configure the Analog Telephony Adapter onto the Zoom Web Portal. For the purpose of Lab Testing we have used **Poly OBI302 Analog Telephone Adapter**. The steps to interwork any other brand ATA Device will remain similar, however there may be changes in the way each device is configured depending upon their make/Model.

4.1 Registration Configuration.

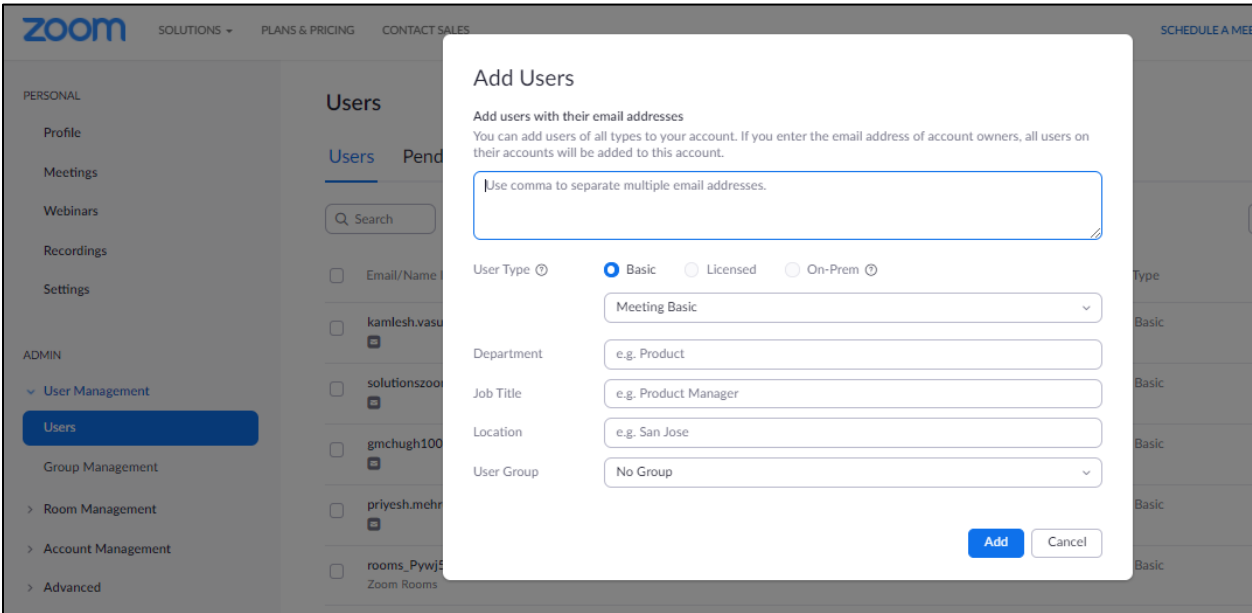
In this Section, we will talk about how to register the Poly OBI302 Device onto Zoom PBX via Oracle SBC.

4.1.1 Create a Zoom User

Navigate to **Admin>User Management > Users**.

Click **Add Users** to create a new Zoom user. Provide the necessary details about the New User and click on Add.

Note : This step can be skipped in case you want use an existing Zoom Phone User.



Once the New User is added it will start reflecting in **Admin >Users** Section on the Web portal.

4.2 Add BYOC Number

Navigate to **Phone Systems Management > Phone Numbers > BYOC**

Select **Add** to add external phone numbers provided by your carrier into the Zoom portal.

Site - Choose the relevant Site on which the Number needs to be added.

For Example, Main Site.

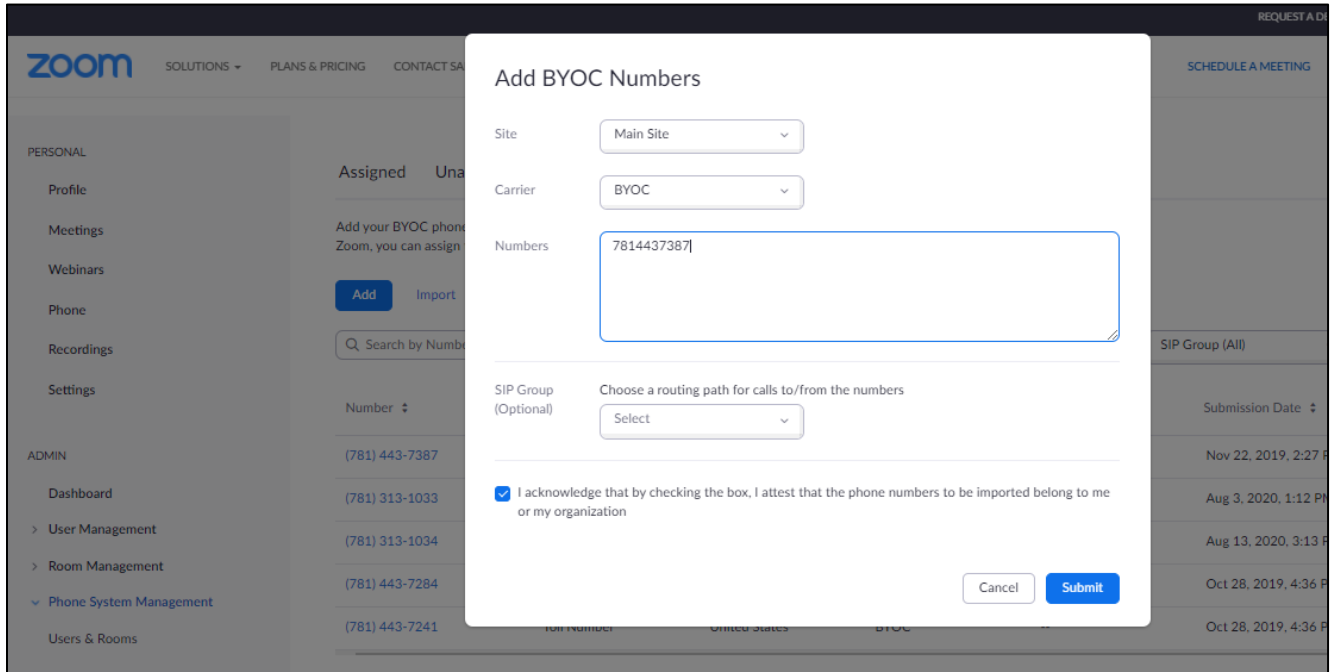
Carrier –Choose BYOC

Numbers- Put the BYOC DID Number provided by your Carrier.

SIP Group – Optional Parameter (Can be Left Blank)

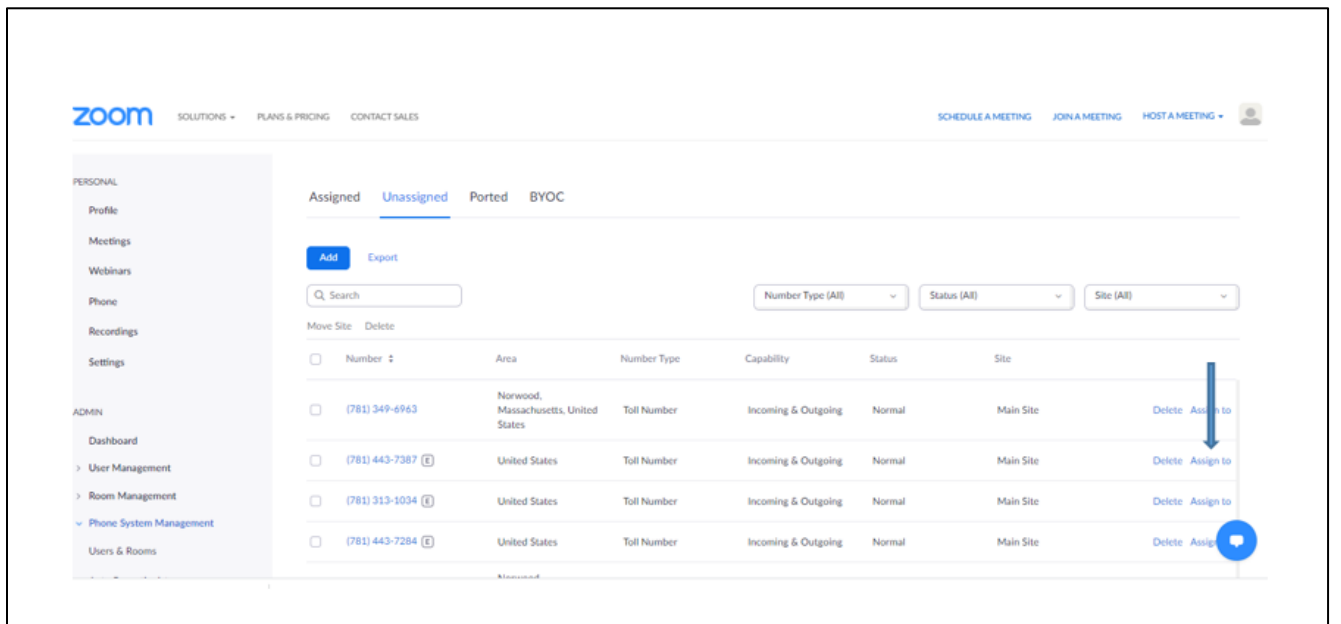
Acknowledge that the Phone Number belongs to your organization.

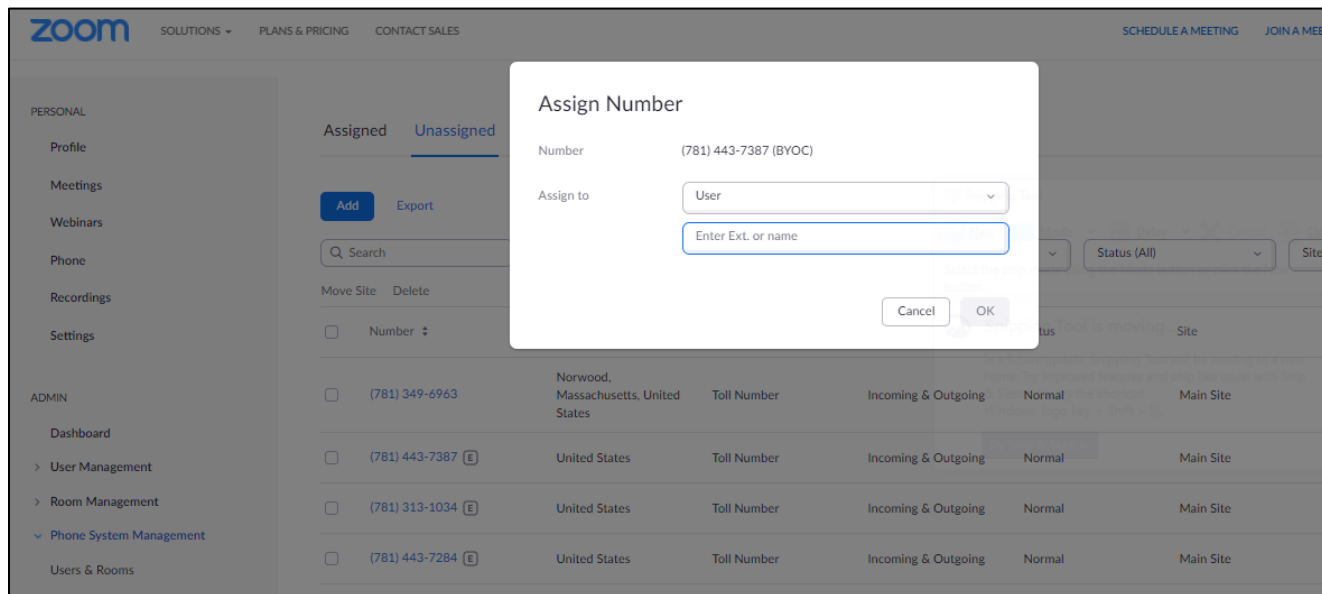
Click **Submit**.



4.3 Assign the BYOC Number to a User

The BYOC Number will now be visible in the Unassigned Tab on the portal. Click on Assign to Tab to assign the Number to This User.





4.4 Register the SBC Interface on Zoom Portal

The analog device registers on the Zoom Cloud PBX which is located at the Core Side of Oracle SBC. Oracle SBC is used as a Proxy the register from Poly ATA Device on Zoom Cloud Voice. We will add the SBC Media interface that communicates with the Zoom PBX Registrar, as a device on Zoom Web Portal.

Navigate to **Admin>Phone System Management**

Click Add to Add a New Device

Enter the MAC Address of the SBC Media Interface

Click Assign and then Add to assign this Device to the previously created User.

The SBC MAC Address can be found by running the ACLI command –

➤ **Show Interfaces**

NN4600-139# show interfaces

M10 (media slot 0, port 2)

Flags: UP BROADCAST MULTICAST ARP RUNNING

Type: GIGABIT_ETHERNET

Admin State: enabled

Auto Negotiation: enabled

Internet address: 10.232.50.65 Vlan: 0

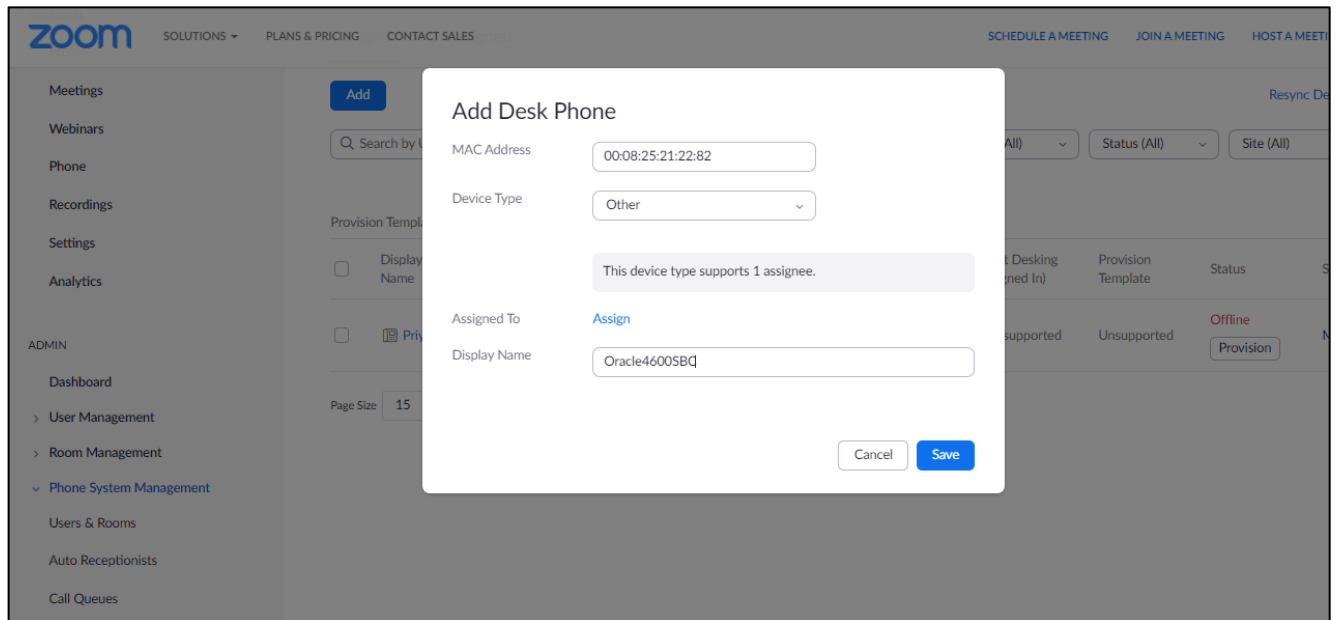
Broadcast Address: 10.232.50.255

Netmask: 255.255.255.0

Gateway: 10.232.50.1

Maximum Transfer Unit size is 1500

Ethernet address is 00:08:25:21:22:82



4.5 Provisioning

Once the Device is successfully added, a **Provision** Button will appear at the bottom of the screen.

Click Provision to discover the Zoom Registrar details as shown in the example below. These details will be required to configure the Poly OBI302 ATA Device.

Note these details as they will be required to configure the device for registration. Download the CA certificates as they will be required for the tls communication of Oracle SBC with Zoom Registrar as mentioned in [Section 6.4 Security configuration](#).

zoom SOLUTIONS ▾ PLANS & PRICING CONTACT SALES SCHEDULE A MEETING

Oracle4600SBC [Rename](#)

Profile

Site	Main Site (Main Site)
Assigned To	Testing Zoom2020 Ext. 12348 ✕
Device Type	Other
MAC Address	00-08-25-21-22-82 Edit
IP Address	--
Provision Template	Unsupported ⓘ
Status	Offline

[Provision](#) [Remove](#)

ADMIN

- Dashboard
- > User Management
- > Room Management
- ▼ Phone System Management
 - Users & Rooms
 - Auto Receptionists
 - Call Queues

zoom SOLUTIONS ▾ PLANS & PRICING SETTING JOIN A MEETING

You will need to enable TLS1.2 for SIP registration and enable SRTP for secure calling on your IP phone. Please refer to your manufacturer's instructions for these processes.

You'll need following information for manual provisioning. For Algo/CyberData Paging/Intercom devices, see [Zoom Phone Supported Devices](#) to view the configuration guide.

SIP Account 1:

1. **SIP Domain:** 110256403.zoom.us
2. **Outbound Proxy:** us01sip0h.ny.zoom.us:5091
3. **User Name:** 54290426471817805175
4. **Authorization ID:** 438070171415
5. **Password:** 0FqS6vOS

Please download CA certificate, [DigiCert Global Root CA](#), [DigiCert Global Root G2](#), [DigiCert Global Root G3](#) and import to your IP phone if they are not in the trust list of the device.

Note: Please note that Zoom support team will not be able to troubleshoot or configure IP phones that are provisioned in this manner. Some Zoom Phone features may not work on manually provisioned phones. It may vary depending on your desk phone model.

ADMIN

- Dashboard
- > User Management
- > Room Management
- ▼ Phone System Management
 - Users & Rooms
 - Auto Receptionists

5 Poly OBI 302 Configuration

Once the ATA Device is connected and successfully powered on, we will configure it to communicate with the Zoom Cloud Voice through the Oracle SBC. The configuration parameters of the Poly OBI 302 ATA Adapter are illustrated in snippets below.

The Registrar Details discovered at the time of provisioning and the SBC configuration will be used to register the Poly OBI 302 ATA Device. Use below example as a reference for your configuration.

Discovered Configuration Element	ATA Configuration Element	Sample Value
Sip Domain	Proxy Server	110256403.zoom.us
Outbound Proxy Port	Proxy Server Port	5091
SBC Access Sip Interface IP Address	Registrar Server	141.146.36.89
SBC Access Sip Interface Port	Registrar Server Port	5065
SBC Access Sip Interface IP Address	Outbound Proxy	141.146.36.89
SBC Access Sip Interface Port	Outbound Proxy Port	5065
Authorization ID	Auth User Name	438070171415
Password	Auth Password	oFqS6vOS
Username@ProxyServer	URI	54290426471817805175@110256403.zoom.us

The screenshot shows the Polycom configuration interface. On the left is a navigation menu with options like 'Setup Wizard', 'Status', 'Router Configuration', 'OBiWiFi Configuration', 'System Management', 'Service Providers', 'ITSP Profile A', 'General', 'SIP', 'RTP', and 'ITSP Profile B'. The main area is titled 'SIP' and 'ITSP Profile A'. It contains a table with the following parameters:

Parameter Name	Value	Default
ProxyServer	110256403.zoom.us	<input type="checkbox"/>
ProxyServerPort	5091	<input type="checkbox"/>
ProxyServerTransport	UDP	<input checked="" type="checkbox"/>
RegistrarServer	141.146.36.89	<input type="checkbox"/>
RegistrarServerPort	5065	<input type="checkbox"/>
UserAgentDomain		<input checked="" type="checkbox"/>
OutboundProxy	141.146.36.89	<input type="checkbox"/>
OutboundProxyPort	5065	<input type="checkbox"/>

Voice Services

- SP1 Service
- SP2 Service
- SP3 Service
- SP4 Service
- OBiTALK Service
- Auto Attendant
- Gateways and Trunk Groups
- OBiBlueTooth 1
- OBiBlueTooth 2
- Page Groups
- Physical Interfaces**
- Codecs**
- Tone Settings**
- Ring Settings**
- Star Codes**
- User Settings**

X_KeepAliveEnable	<input type="checkbox"/>	<input checked="" type="checkbox"/>	?
X_KeepAliveExpires	15	<input checked="" type="checkbox"/>	?
X_KeepAliveServer		<input checked="" type="checkbox"/>	?
X_KeepAliveServerPort	5060	<input checked="" type="checkbox"/>	?
X_KeepAliveMsgType	keep-alive	<input checked="" type="checkbox"/>	?
X_CustomKeepAliveMsg		<input checked="" type="checkbox"/>	?
X_UserAgentPort	5060	<input checked="" type="checkbox"/>	?
X_UserAgentPorts		<input checked="" type="checkbox"/>	?
DirectoryNumber		<input checked="" type="checkbox"/>	?
X_DefaultRing	1	<input checked="" type="checkbox"/>	?
X_CallOnHoldRing	8	<input checked="" type="checkbox"/>	?
X_RepeatDialRing	5	<input checked="" type="checkbox"/>	?
X_BargeInRing	4	<input checked="" type="checkbox"/>	?
X_CallParkedRing	10	<input checked="" type="checkbox"/>	?
X_SipDebugOption	Disable	<input checked="" type="checkbox"/>	?
X_SipDebugExclusion		<input checked="" type="checkbox"/>	?
X_SatelliteMode	<input type="checkbox"/>	<input checked="" type="checkbox"/>	?
X_Proxy	<input type="checkbox"/>	<input checked="" type="checkbox"/>	?
X_ProxyClientConfig		<input checked="" type="checkbox"/>	?
X_AcceptResync	yes without authentication	<input checked="" type="checkbox"/>	?

SIP Credentials

Parameter Name	Value	Default	
AuthUserName	174522534085	<input type="checkbox"/>	?
AuthPassword	*****	<input type="checkbox"/>	?
URI	92302174621264271794@110256403.zoom.us	<input type="checkbox"/>	?

Polycom

Setup Wizard

- Status
- Router Configuration
- OBiWiFi Configuration
- System Management
- Service Providers
- Voice Services
- Physical Interfaces
- PHONE1 Port**

PHONE Port

Parameter Name	Value	Default	
Enable	<input checked="" type="checkbox"/>	<input checked="" type="checkbox"/>	?
DigitMap	{{(1-9)x?(Mpli)[1-9]S9[1-9][0-9]S9[911]**0***# ###	<input checked="" type="checkbox"/>	?
OutboundCallRoute	{{(1-9)x?(Mpli):pp},{{<##>:li},{{<#>:ph2},{{<+7(<input checked="" type="checkbox"/>	?
CallReturnDigitMaps	{pli:(xx)}, {sp1:(<+1>xx)}, {sp2:(<+2>xx)}, {sp3:(<+	<input checked="" type="checkbox"/>	?
PrimaryLine	SP1 Service	<input checked="" type="checkbox"/>	?
ToneOnPrimaryServiceDown	Normal Dial Tone	<input checked="" type="checkbox"/>	?

6 Oracle SBC Configuration

Below is an outline of the network setup used to conduct all testing between the Oracle SBC, Poly OBI 302 ATA, Zoom Phone and the SIP Trunk.

Note - These instructions cover configuration steps between the Oracle SBC and Zoom Phone. The complete interconnection of other entities, such as connection of the SIP trunk, 3rd Party PBX and/or analog devices are not fully covered in this instruction. The details of such connection are available in other instructions produced by the vendors of retrospective components.

6.1 Prerequisites

In this section we will provide the steps for such interworking of ATA Device with Zoom Phone via Oracle SBC. The ATA Device registers itself on the Zoom PBX through the Oracle SBC. Calls from the ATA are first forwarded to the registrar which are hair pinned back to the SBC from the Zoom BYOC Trunk IP Address. Zoom has an internal routing to send the call to their BYOC IP which connects to Oracle SBC. SBC further breaks out the call to the PSTN Network through the connected SIP Trunk.

Before you begin, make sure that you have the following per every SBC you want to pair to communicate with Zoom BYOC Trunk.

- Public IP address
- Public certificate, issued by one of the supported CAs (refer to [Related Documentation](#) for details about supported Certification Authorities).
- Zoom Public CA certificates to add to trust store of SBC

There are two methods for configuring the Oracle SBC, CLI, or GUI.

For the purposes of this Application note, we'll be using the Oracle SBC GUI for all configuration examples.

This guide assumes the Oracle SBC has been installed, management interface has been configured, product selected and entitlements have been assigned. Also, http-server has been enabled for GUI access. If you require more information on how to install your SBC platform, please refer to the [ACLI configuration guide](#).

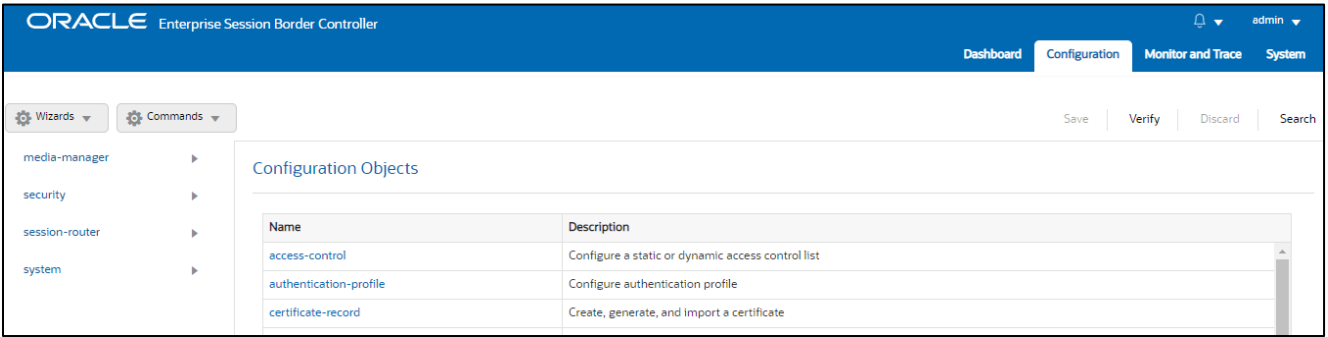
To access the Oracle SBC GUI, enter the management IP address into a web browser. When the login screen appears, enter the username and password to access the ORACLE SBC.

Once you have accessed the Oracle SBC, at the top, click the Configuration Tab. This will bring up the ORACLE SBC Configuration Objects List on the left hand side of the screen.

Any configuration parameter not specifically listed below can remain at the ORACLE SBC default value and does not require a change for connection to Zoom Phone to function properly.

The below configuration example assumes you will be using a secure connection between the Oracle SBC and Zoom Phone Platform for both signalling and media. In this testing, The connection between Poly OBI302 Device and Oracle SBC is UDP in this setup.

Note: All network parameters, ip addresses, hostnames etc..are specific to Oracle Labs, and cannot be used outside of the Oracle Lab environment. They are for example purposes only!!!



6.2 Global Configuration Elements

Before you can configuration more granular parameters on the SBC, there are four global configuration elements that must be enabled (ntp optional) to proceed.

- System-Config
- Media-manager-Config
- SIP-Config
- Ntp-config

6.2.1 System-Config

To configure system level functionality for the ORACLE SBC, you must first enable the system-config

GUI Path: system/system-config

ACL Path: config t→system→system-config

Note: The following parameters are optional but recommended for system config

- Hostname
- Description
- Location
- Default-gateway (*recommend using the management interface gateway for this global setting*)

Modify System Config

system

- host-route
- http-client
- http-server
- network-interface
- ntp-config
- phy-interface
- redundancy-config
- snmp-community
- spl-config
- system-config**

Show All

Hostname: zoom.us

Description: SBC for Zoom Cloud Voice

Location: Burlington MA

Mib System Contact:

Mib System Name:

Mib System Location:

Acp TLS Profile:

OK Delete

network-interface

- ntp-config
- phy-interface
- redundancy-config
- snmp-community
- spl-config
- system-config**

Show All

Page 1 of 1 (1 of 1 items) < 1 >

Options:

Call Trace: enable

Default Gateway: 10.138.194.129

Restart: enable

Telnet Timeout: 0 (Range: 0..65535)

Console Timeout: n (Range: 0..65535)

OK Delete

- Click the OK at the bottom of the screen

6.2.2 Media Manager

To configure media functionality on the SBC, you must first enabled the global media manager

GUI Path: media-manager/media-manager

ACL Path: config t→media-manager→media-manager-config

The following options are recommended for global media manager to help secure the SBC.

- Max-untrusted-signalling
- Min-untrusted-signalling

The values in both these fields are related to the SBC's security configuration. For more detailed security configuration options, please refer to the [SBC's Security Guide](#).

The screenshot shows the 'Modify Media Manager' configuration page. On the left, there is a navigation tree with 'media-manager' selected. The main area contains the following settings:

Parameter	Value	Range
State	<input checked="" type="checkbox"/> enable	
Flow Time Limit	86400	(Range: 0..4294967295)
Initial Guard Timer	300	(Range: 0..4294967295)
Subsq Guard Timer	300	(Range: 0..4294967295)
TCP Flow Time Limit	86400	(Range: 0..4294967295)
TCP Initial Guard Timer	300	(Range: 0..4294967295)
TCP Subsq Guard Timer	300	(Range: 0..4294967295)
Hint Rtcp	<input type="checkbox"/> enable	
Algd Log Level	NOTICE	
Mbcd Log Level	NOTICE	

At the bottom, there are 'OK' and 'Delete' buttons. A 'Show All' toggle is also visible in the bottom left corner.

- Click OK at the bottom

6.2.3 SIP Config

To enable SIP related objects on the ORACLE SBC, you must first configure the global SIP Config element:

GUI Path: session-router/SIP-config

ACL Path: config t→session-router→SIP-config

The following are recommended parameters under the global SIP-config:

- home-realm-id ZoomCore
- registrar-domain *
- registrar-host us01sip0h.ny.zoom.us
- registrar-port 5091
- Options: Click Add, in pop up box, enter the string: **inmanip-before-validate**
- Click Apply/Add another, then enter: **max-udp-length=0**
- Press OK in box

The Values for registrar Host and Port are discovered at the time of Provisioning in [Step 4.4](#).

The home-realm-id is the Core Realm where the Zoom PBX Registrar is located. The values configured here will be used to route the incoming requests from ATA Device towards Zoom.

Configuration View Configuration Q Discard Verify 5

media-manager ▶
 security ▶
 session-router ▼
 access-control
 account-config
 filter-config
 ldap-config
 local-policy
 local-routing-config
 media-profile
 session-agent
 session-group
 session-recording-group
 session-recording-server
 session-translation
 sip-config

Modify SIP Config

State enable

Dialog Transparency enable

Home Realm ID ZoomCore ▼

Egress Realm ID ▼

Nat Mode None ▼

Registrar Domain *

Registrar Host us0sip0h.ny.zoom.us

Registrar Port 5091 (Range: 0.1025..65535)

Init Timer 500 (Range: 0.4294967295)

Max Timer 4000 (Range: 0.4294967295)

Trans Expire 32 (Range: 0.4294967295)

Initial Inv Trans Expire 0 (Range: 0.999999999)

Invite Expire 180 (Range: 0.4294967295)

Session Max Life Limit 0

Enforcement Profile ▼

local-routing-config
 media-profile
 session-agent
 session-recording-group
 session-recording-server
 session-translation
 sip-config
 sip-feature
 sip-interface
 sip-manipulation
 sip-monitoring
 translation-rules

Show All

Red Max Trans 10000 (Range: 0..50000)

Options
 inmanip-before-validate ✕
 max-udp-length=0 ✕

SPL Options

SIP Message Len 4096 (Range: 0..65535)

Enum Sag Match enable

Extra Method Stats enable

Extra Enum Stats enable

Registration Cache Limit 0 (Range: 0.999999999)

Register Use To For Lp enable

Refer Src Routing enable

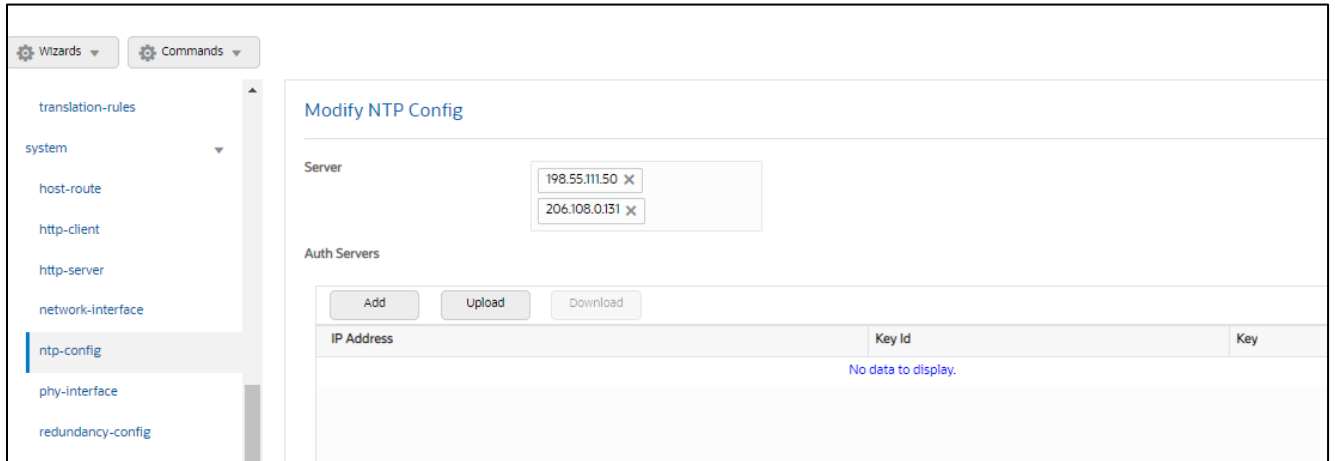
OK Delete

- Click OK at the bottom

6.2.4 NTP Config

GUI Path: system/ntp-config

ACL Path: config t→system→ntp-config



- Click OK at the bottom

6.3 Network Configuration

To connect the SBC to network elements, we must configure both physical and network interfaces. For the purposes of this example, we will configure two physical interfaces, and two network interfaces. One to communicate with ATA Devices and Zoom PBX and BYOC Trunk, the other to connect to PSTN Network.

6.3.1 Physical Interfaces

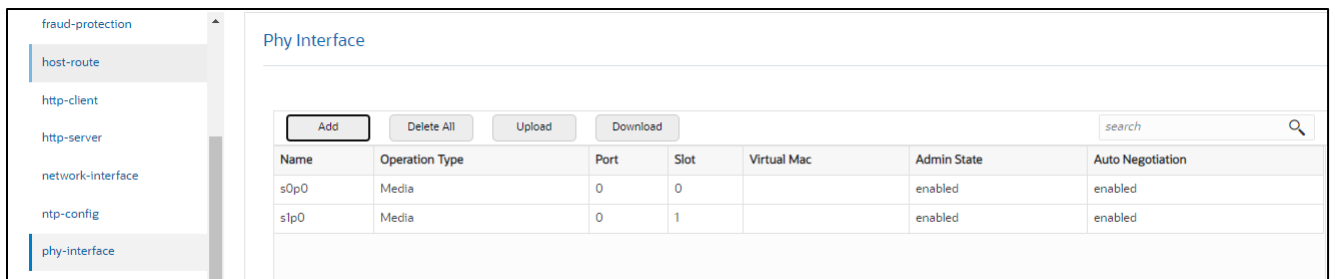
GUI Path: system → phy-interface

ACL Path: config t→system→phy-interface

- Click Add, use the following table as a configuration example:

Config Parameter	Zoom	PSTN
Name	s0p0	s1p0
Operation Type	Media	Media
Slot	0	1
Port	0	0

Note: Physical interface names, slot and port may vary depending on environment



- Click OK at the bottom of each after entering config information

6.3.2 Network Interfaces

GUI Path: system/network-interface

ACL Path: config t→system→network-interface

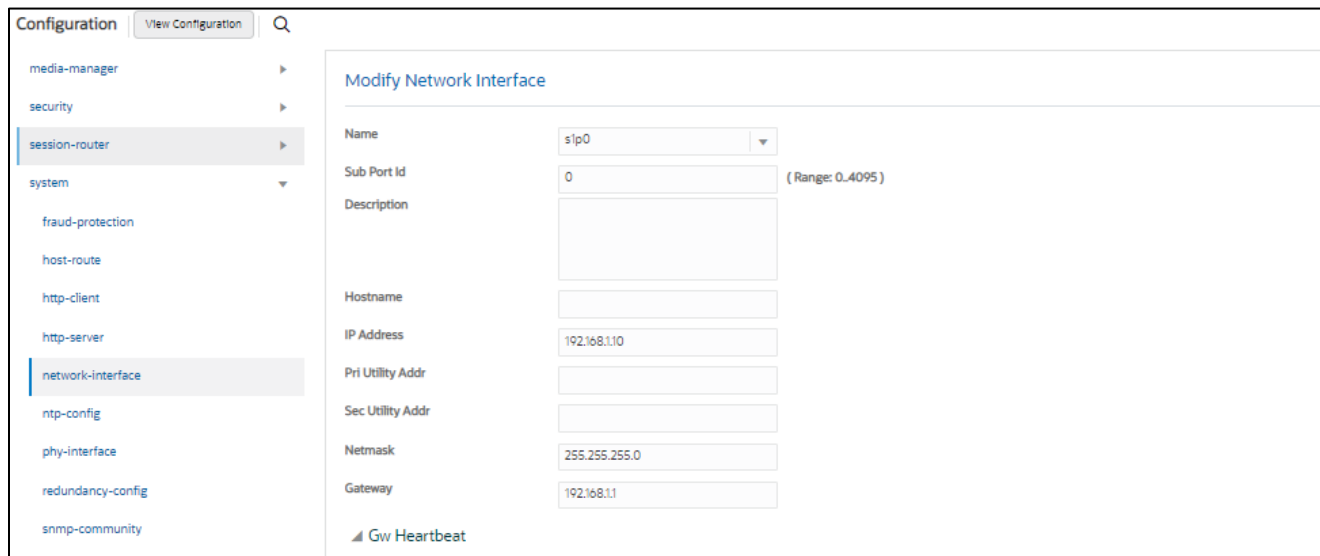
- Click Add, use the following table as a configuration example:

Configuration Parameter	Zoom	PSTN
Name	s0p0	s1p0
Hostname	Domain (if applicable)	Domain (if applicable)
IP Address	141.146.36.101	192.168.1.10
Netmask	255.255.255.192	255.255.255.0
Gateway	141.146.36.65	192.168.1.1
DNS Primary IP	8.8.8.8	
DNS Domain	Domain(if applicable)	

The screenshot shows a web-based configuration interface for a network device. On the left is a navigation menu with categories like 'media-manager', 'security', 'session-router', 'system', 'fraud-protection', 'host-route', 'http-client', 'http-server', 'network-interface' (highlighted), 'ntp-config', 'phy-interface', 'redundancy-config', 'snmp-community', 'spl-config', 'system-config', and 'trap-receiver'. The main area is titled 'Modify Network Interface' and contains the following fields:

- Name: s0p0 (dropdown menu)
- Sub Port Id: 0 (text input, range: 0-4095)
- Description: (empty text area)
- Hostname: (empty text input)
- IP Address: 141.146.36.101 (text input)
- Pri Utility Addr: (empty text input)
- Sec Utility Addr: (empty text input)
- Netmask: 255.255.255.192 (text input)
- Gateway: 141.146.36.65 (text input)
- GW Heartbeat:
 - State: enable
 - Heartbeat: 10 (text input, range: 0-65535)
 - Retry Count: 3 (text input, range: 0-65535)
 - Retry Timeout: 3 (text input, range: 1-65535)
 - Health Score: 0 (text input, range: 0-100)
- DNS IP Primary: 8.8.8.8 (text input)
- DNS Domain: (empty text input)

At the bottom, there are 'OK' and 'Back' buttons, and a 'Show All' toggle switch.



- Click OK at the bottom of each after entering config information.

6.4 Security Configuration

This section describes how to configure the SBC for both TLS and SRTP communication with Zoom Phone BYOC Platform. The connection between Zoom Phone allows TCP or TLS connections from SBC's for SIP traffic, and RTP or SRTP for media traffic. For our testing, the connection between the Oracle SBC and Zoom Phone platform was secured via TLS/SRTP. This setup requires a certificate signed by one of the trusted Certificate Authorities. The connection between ATA Device and Sip Trunk with the SBC is UDP so this section does not apply to ATA configuration on the SBC.

6.4.1 Certificate Records

“Certificate-records” are configuration elements on Oracle SBC which captures information for a TLS certificate such as common-name, key-size, key-usage etc.

This section walks you through how to configure certificate records, create a certificate signing request, and import the necessary certificates into the SBC's configuration.

GUI Path: security/certificate-record

ACL Path: config t→security→certificate-record

For the purposes of this application note, we'll create below certificate records.

SBC Certificates

- ✓ SBCEnterpriseCert.pem
- ✓ DigiCertGlobalRootCA.crt.pem
- ✓ DigiCertInter.pem

Zoom PBX and Zoom BYOC Trunk Certificates

- ✓ sbc_ca.pem
- ✓ DigiCertGlobalRootCA.crt.pem
- ✓ DigiCertGlobalRootG2.crt.pem
- ✓ DigiCertGlobalRootG3.crt.pem

6.4.2 SBC End Entity Certificate

The SBC's end entity certificate **SBCEnterpriseCert.pem** is what is presented to Zoom Phone signed by your CA authority. In this example we are using DigiCert as our signing authority.

The certification must include a **common name**.

For this, we are using an FQDN as the common name.

- Common name: (**telechat.o-test06161977.com**)

To Configure the certificate record:

- Click Add, and configure the SBC certificate as shown below:

The screenshot displays the 'Modify Certificate Record' configuration page. On the left, a navigation sidebar lists various system components, with 'certificate-record' highlighted. The main configuration area includes the following fields and options:

- Name:** SBCEnterpriseCert
- Country:** US
- State:** California
- Locality:** Redwood City
- Organization:** Oracle Corporation
- Unit:** (empty)
- Common Name:** telechat.o-test06161977.com
- Key Size:** 2048
- Alternate Name:** (empty)
- Trusted:** enable
- Key Usage List:** digitalSignature, keyEncipherment
- Extended Key Usage List:** serverAuth, ClientAuth

At the bottom of the configuration area, there are 'OK' and 'Back' buttons. A 'Show All' toggle is located at the bottom of the sidebar.

- Click OK at the bottom

6.5 Root CA and Intermediate Certificates

Using this same procedure, configure certificate records for Root CA and Intermediate Certificates for SBC and Zoom.

6.5.1 Oracle SBC and Zoom Certificate

Oracle SBC certificate are signed by DigiCert. Zoom also provides DigiCert certificate for the purpose of TLS connection with SBC. In this Setup the root CA for both Oracle SBC and Zoom is same so only one certificate record entry is created which covers both SBC and Zoom. DigiCertInter is the intermediate CA certificate for the SBC. We will create certificate-record entry for each certificate discovered at the time of registration configuration in [Step 4.5](#) and the SBC CA certificate. The same certificates are required for Zoom BYOC Trunk so can be reused.

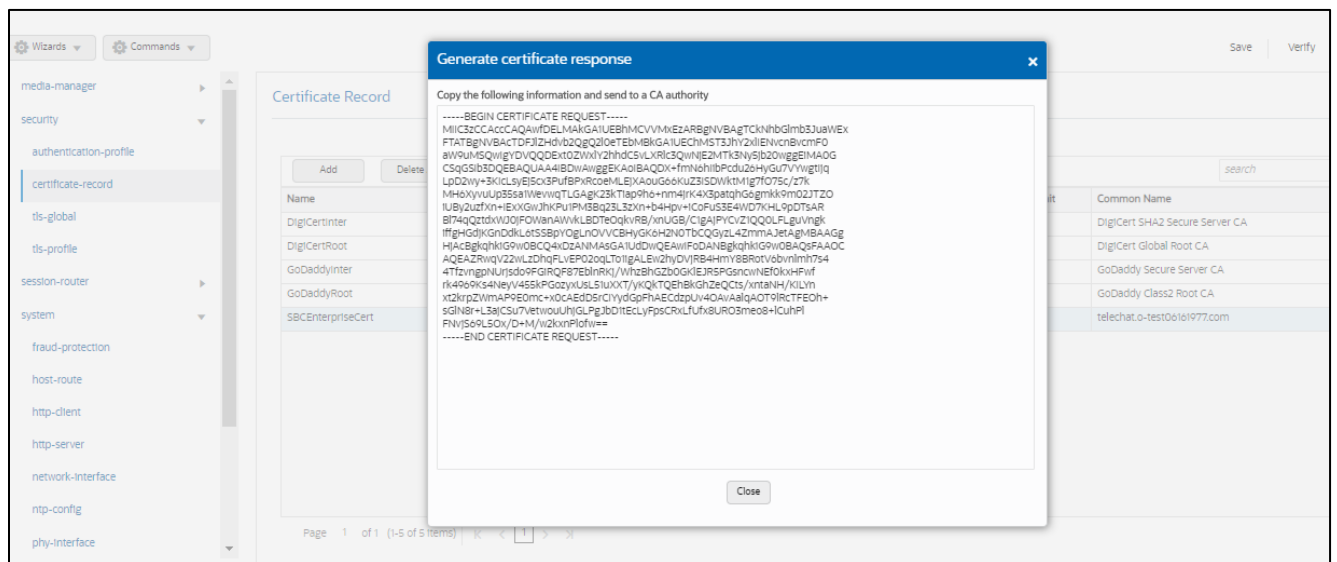
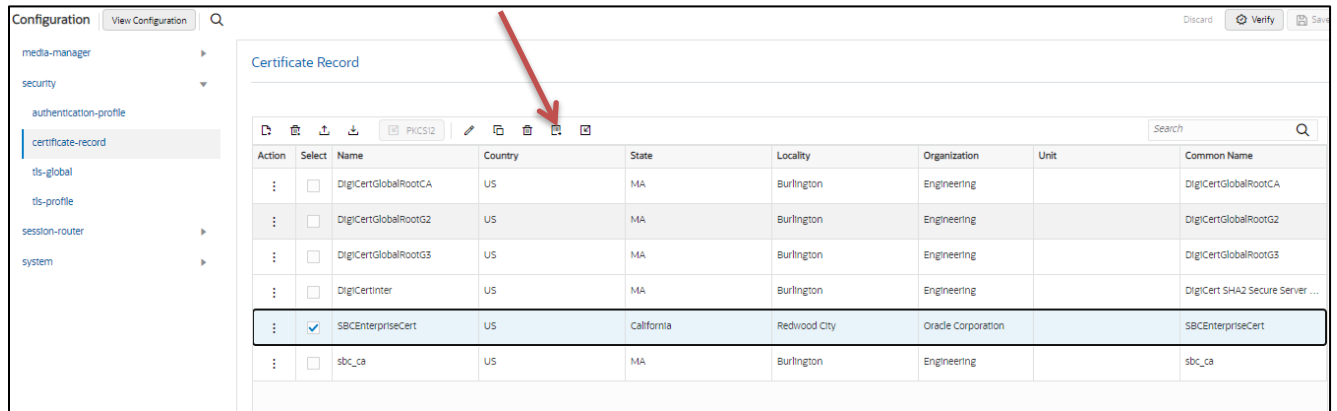
<https://support.zoom.us/hc/en-us/articles/360054176992-BYOC-BYOP-Public-CA-Certificate-and-SIP-proxy-address-change>

Config Parameter	sbc_ca	DigiCertGlobalRootCA	DigiCertGlobalRootG2	DigiCertGlobalRootG3	DigiCertInter
Common Name	sbc_ca	DigiCertGlobalRootCA	DigiCertGlobalRootG2	DigiCertGlobalRootG3	DigiCertInter
Key Size	2048	2048	2048	2048	2048
Key-Usage-List	digitalSignature keyEncipherment	digitalSignature keyEncipherment	digitalSignature keyEncipherment	digitalSignature keyEncipherment	digitalSignature keyEncipherment
Extended Key Usage List	serverAuth	serverAuth	serverAuth	serverAuth	serverAuth
cKey algor	rsa	rsa	rsa	rsa	rsa
Digest-algor	Sha256	Sha256	Sha256	Sha256	Sha256

6.5.2 Generate Certificate Signing Request

Now that the SBC's certificate has been configured, create a certificate signing request for the SBC's end entity only. **This is not required for any of the Root CA or intermediate certificates that have been created.**

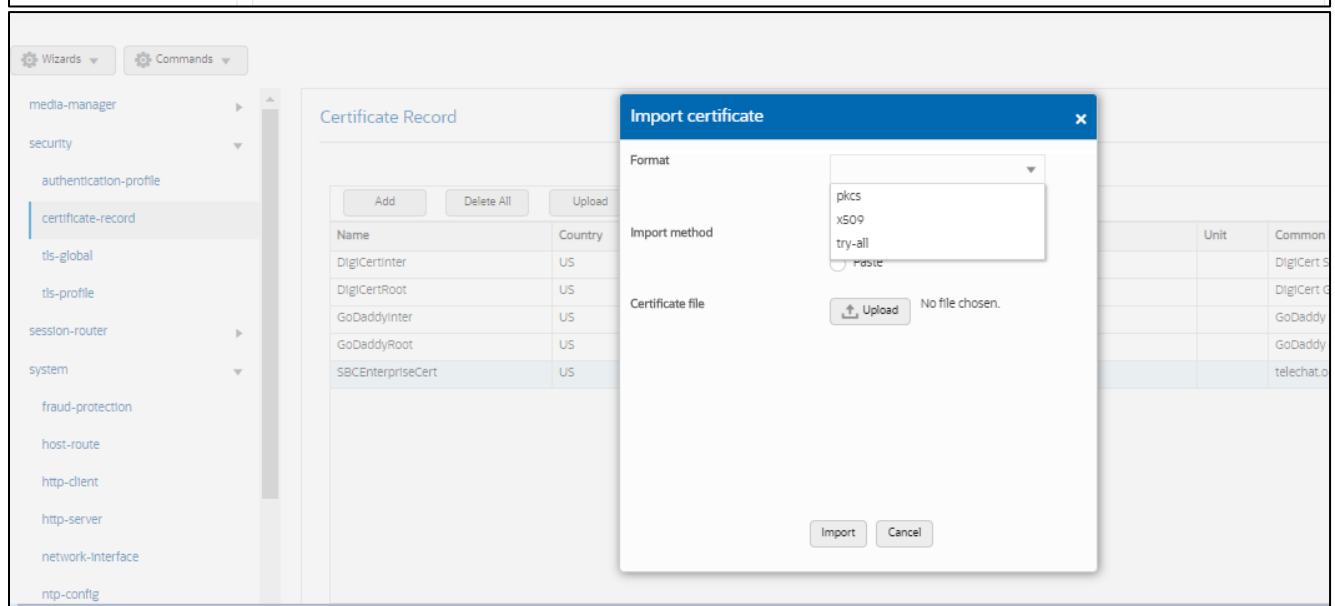
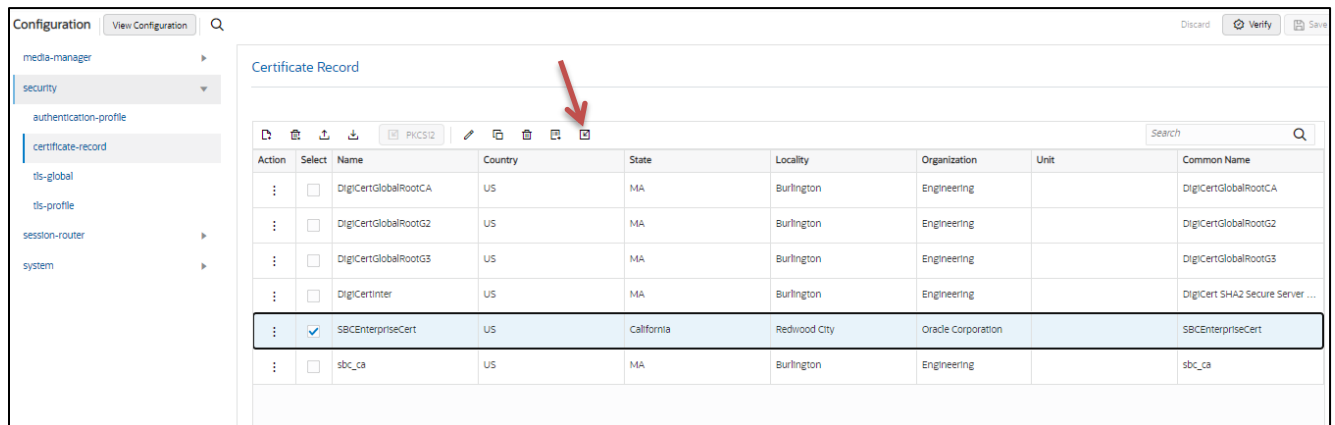
On the certificate record page in the Oracle SBC GUI, select the SBC's end entity certificate that was created above, and click the "generate" tab



- copy/paste the text that gets printed on the screen as shown above and upload to your CA server for signature.
- Also note, at this point, **a save and activate is required** before you can import the certificates to each certificate record created above.

6.5.3 Import Certificates to SBC

Once certificate signing request has been completed – import the signed certificate to the SBC. Please note – all certificates including root and intermediate certificates are required to be imported to the SBC. Once all certificates have been imported, issue **save/activate** from the WebGUI



Repeat these steps to import all the root and intermediate CA certificates into the SBC:

- ✓ sbc_ca
- ✓ DigiCertGlobalRootCA
- ✓ DigiCertGlobalRootG2
- ✓ DigiCertGlobalRootG3
- ✓ DigiCertInter

At this stage, all required certificates have been imported.

6.5.4 TLS Profile

TLS profile configuration on the SBC allows for specific certificates to be assigned.

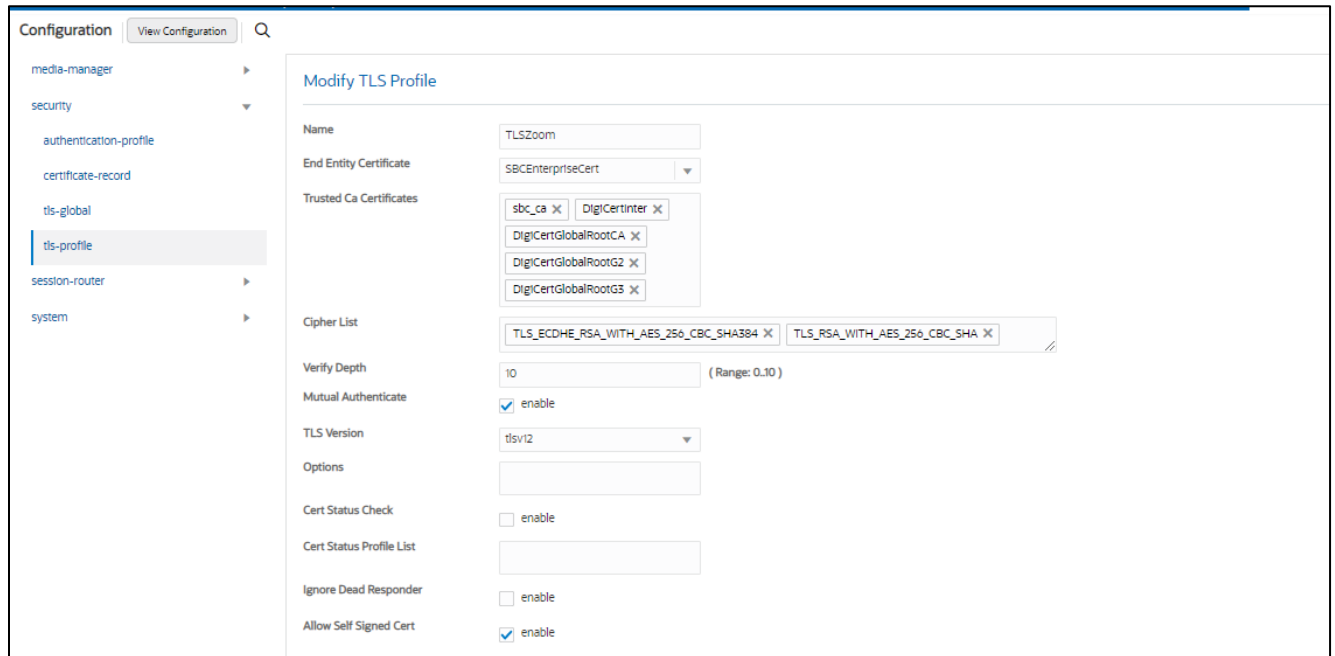
GUI Path: security/tls-profile

ACL Path: config t→security→tls-profile

- Click Add, use the example below to configure

Zoom supports the following signalling ciphers that need to be added to the TLS profile:

- TLS-ECDHE-RSA-WITH-AES-256-CBC-SHA-384
- RSA-WITH-AES-256-CBC-SHA-256



Note: Only the DigiCert Certificates need to be added to the tls-profile to authenticate the certificate presented to the SBC from Zoom Phone.

- Click OK at the bottom

6.6 Media Security Configuration

This section outlines how to configure support for media security between the ORACLE SBC and Zoom Cloud Voice.

6.6.1 Sdes-profile

This is the first element to be configured for media security, where the algorithm and the crypto's to be used are configured.

GUI Path: security/media-security/sdes-profile

ACL Path: config t→security→media-security→sdes-profile

Oracle SBC and Zoom Cloud Voice Support the following media ciphers for SRTP:

- AES-CM-128-HMAC-SHA1-80
- AES-CM-128-HMAC- SHA1-32

Click Add, and use the example below to configure

- Click OK at the bottom

6.6.2 Media Security Policy

Media-sec-policy instructs the SBC how to handle the SDP received/sent under a realm (RTP, SRTP or any of them) and, if SRTP needs to be used, the sdes-profile that needs to be used

In this example, we are configuring two media security policies. One to secure and decrypt media toward Zoom, the other for non-secure media facing PSTN and the ATA Device.

GUI Path: security/media-security/media-sec-policy

ACL Path: config t→security→media-security→media-sec-policy

- Click Add, use the examples below to configure

- media-manager ▶
- security ▼
 - admin-security ▶
 - auth-params
 - authentication
 - authentication-profile
 - cert-status-profile
 - certificate-record
 - ike ▶
 - ipsec ▶
 - media-security ▼
 - dtls-srtp-profile
 - media-sec-policy
 - sdes-profile
 - sipura-profile
 - password-policy

Show All

Modify Media Sec Policy

Name

Pass Through enable

Options

▲ Inbound

Profile

Mode

Protocol

Hide Egress Media Update enable

▲ Outbound

Profile

Mode

Protocol

- media-manager ▶
- security ▼
 - admin-security ▶
 - auth-params
 - authentication
 - authentication-profile
 - cert-status-profile
 - certificate-record
 - ike ▶
 - ipsec ▶
 - media-security ▼
 - dtls-srtp-profile
 - media-sec-policy
 - sdes-profile
 - sipura-profile
 - password-policy

Show All

Modify Media Sec Policy

Name

Pass Through enable

Options

▲ Inbound

Profile

Mode

Protocol

Hide Egress Media Update enable

▲ Outbound

Profile

Mode

Protocol

6.7 Media Configuration

This section will guide you through the configuration of realms and steering pools, both of which are required for the SBC to handle signaling and media flows toward Zoom and PSTN.

6.7.1 Realm Config

Realms are a logical distinction representing routes (or groups of routes) reachable by the Oracle Session Border Controller and what kinds of resources and special functions apply to those routes. Realms are used as a basis for determining ingress and egress associations to network interfaces.

GUI Path; media-manager/realm-config

ACL Path: config t→media-manager→realm-config

- Click Add, and use the following table as a configuration example for the four realms used in this configuration example
- **Access-** Realm Facing the ATA Device
- **Core -** Realm facing the Zoom Registrar
- **Zoom Realm-** Realm for the Zoom BYOC Trunk
- **PSTN Realm-** Breakout Realm for calls towards PSTN

Config Parameter	Access	Core	Zoom Trunk	PSTN
Identifier	Access	ZoomCore	Zoom	SIPTrunk
Network Interface	s0p0:0	s0p0:0	s0p0:0	s1p0:0
Mm in realm	<input checked="" type="checkbox"/>	<input checked="" type="checkbox"/>	<input checked="" type="checkbox"/>	<input checked="" type="checkbox"/>
Access-control-trust-level	Low	High	High	High
Media Sec policy	RTP	sdespolicy	sdesPolicy	RTP

Action	Select	Identifier	Description	Addr Prefix	Network Interfaces	Media Realm List	Mm In Realm	Mm In Network
:	<input type="checkbox"/>	Access		0.0.0.0	M00:0,4		enabled	enabled
:	<input type="checkbox"/>	SIPTrunk		0.0.0.0	M00:0		enabled	enabled
:	<input type="checkbox"/>	Zoom	Realm for Zoom Cloud Voice	0.0.0.0	M00:0		enabled	enabled
:	<input type="checkbox"/>	ZoomCore		0.0.0.0	M00:0,4		enabled	enabled

6.7.2 Steering Pools

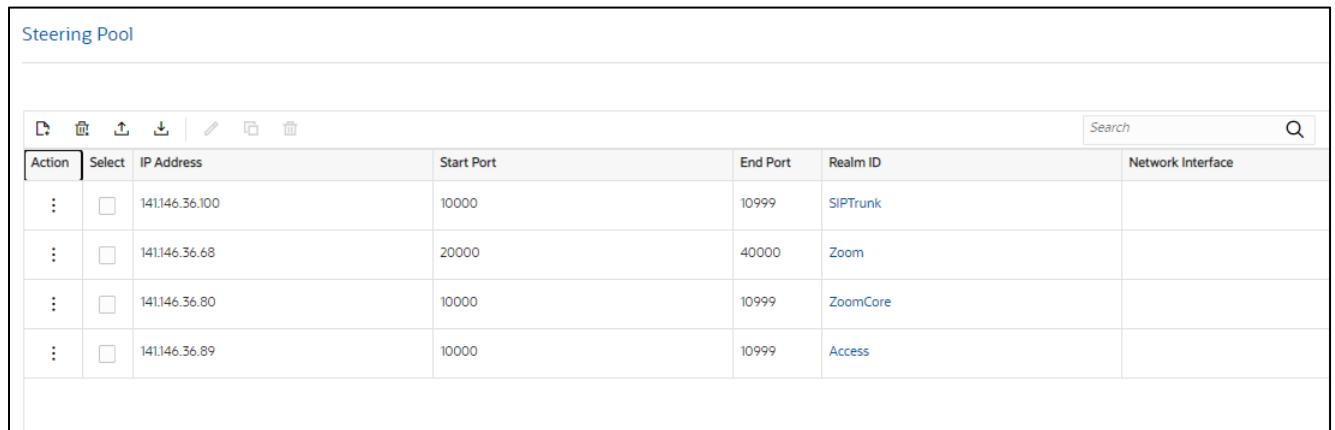
Steering pools define sets of ports that are used for steering media flows through the Oracle SBC. These selected ports are used to modify the SDP to cause receiving session agents to direct their media toward this system.

We will configure steering pool for PSTN each realm configured.

GUI Path: media-manager/steering-pool

ACL Path: config t→media-manager→steering-pool

- Click Add, and use the below examples to configure



The screenshot shows the 'Steering Pool' configuration page in a GUI. It features a toolbar with icons for refresh, delete, add, download, edit, and save. A search bar is located on the right. Below the toolbar is a table with the following data:

Action	Select	IP Address	Start Port	End Port	Realm ID	Network Interface
:	<input type="checkbox"/>	141.146.36.100	10000	10999	SIPTrunk	
:	<input type="checkbox"/>	141.146.36.68	20000	40000	Zoom	
:	<input type="checkbox"/>	141.146.36.80	10000	10999	ZoomCore	
:	<input type="checkbox"/>	141.146.36.89	10000	10999	Access	

6.8 SIP Configuration

This section outlines the configuration parameters required for processing, modifying and securing SIP signaling traffic.

6.8.1 SIP Manipulations

In order to comply with the signaling message requirements of Carrier and Zoom we have applied following sip-manipulations.

Note: Applying these manipulations are not compulsory is dependent upon the requirement of your Carrier. The requirement may vary from carrier to carrier so the HMRs are subjected to change.

6.8.1.1 Manipulation towards Zoom Side

For calls to be presented to Zoom Phone from the Oracle SBC, the Oracle SBC requires alterations to the SIP signaling natively created. To do this, we should we can use the prebuilt HMR ACME_NAT_TO_FROM_IP

The following SIP manipulation is applied as the out-manipulationId to the sip-interface created for Zoom Trunk and modifies packets generated by the Oracle SBC to Zoom Phone:

The manipulation performs the following modifications to SIP packets

1. Changes the host portion of From address with the SBC sip-interface IP Address.
2. Changes the host portion of To Header with Zoom IP Address.

6.8.1.2 Manipulation towards Carrier Side.

The following SIP manipulation is applied as the out-manipulationId on the Session-Agent created for the Carrier Trunk. This manipulation modifies packets generated by the Oracle SBC to Carrier Side as stated below:

1. Removes the unwanted headers inserted by Zoom in the signaling when forwarding the message to Carrier.
2. Changes the Host portion of From Header with the Local SBC IP Address.
3. Changes the Host portion of To Header with Carrier side IP Address
4. Changes the Host portion of P-Asserted Identity with Carrier side IP Address.

Name	Element Type
XTraceID	header-rule
XInstanceID	header-rule

Header-Rules

Below is an example to remove the X-TraceID header towards Carrier. In similar fashion other header-rules can be created to remove other headers such as XInstanceID, XDInfo etc.

The screenshot shows the configuration interface for a SIP manipulation rule. The sidebar on the left lists various configuration categories, with 'sip-manipulation' selected. The main panel is titled 'Modify Sip manipulation / header rule' and contains the following fields:

- Name: XTraceID
- Header Name: X-Trace-ID[^]
- Action: delete
- Comparison Type: case-sensitive
- Msg Type: request
- Methods: INVITE
- Match Value: (empty)
- New Value: (empty)
- CfgRules: (empty)

At the bottom of the main panel, there is an 'Add' button and 'OK' and 'Back' buttons.

Similar Header-rules are created to remove the other X headers which are inserted by Zoom on the Sip Signaling.

The screenshot shows the 'Modify SIP Manipulation' configuration page. The sidebar on the left lists various configuration categories, with 'sip-manipulation' selected. The main panel is titled 'Modify SIP Manipulation' and contains a table of existing header rules:

Name	Element Type
XTraceID	header-rule
XInstanceID	header-rule
XDInfo	header-rule
XCapability	header-rule
xpublicip	header-rule
xorigcontact	header-rule
xorigcallid	header-rule
xtocarrier	header-rule
xFSSupport	header-rule
changeFromIP	header-rule
changeToIP	header-rule
changeAssertedIP	header-rule

On the same Sip-manipulation we have called the ACME_NAT_TO_FROM_IP Manipulation which performs the topology hiding as below -

1. Changes the host portion of From Header with the Local SBC IP Address.
2. Changes the host portion of To Header with Carrier side IP Address

3. Changes the host portion of P Asserted Identity with Carrier side IP Address.
Header-rule

Wizards Commands

- local-policy
- local-routing-config
- media-profile
- session-agent
- session-group
- session-recording-group
- session-recording-server
- session-translation
- sip-config
- sip-feature
- sip-interface
- sip-manipulation
- sip-monitoring
- translation-rules
- system

Show All

Modify Sip manipulation / header rule

Name: callAcme

Header Name: From

Action: sip-manip

Comparison Type: case-sensitive

Msg Type: request

Methods:

Match Value:

New Value: ACME_NAT_TO_FROM_IP

CfgRules

Name	Element Type
No data to display	

OK Back

Below Portion of the HMR Changes the Host portion of P-Asserted Identity with Carrier side IP Address.

Header-rule

local-policy

- local-routing-config
- media-profile
- session-agent
- session-group
- session-recording-group
- session-recording-server
- session-translation
- sip-config
- sip-feature
- sip-interface
- sip-manipulation

Modify Sip manipulation / header rule

Name: changeAssertedIP

Header Name: P-Asserted-Identity

Action: manipulate

Comparison Type: pattern-rule

Msg Type: request

Methods: INVITE

Match Value:

New Value:

CfgRules

Element Rule

Modify Sip manipulation / header rule / element rule

Name:

Parameter Name:

Type:

Action:

Match Val Type:

Comparison Type:

Match Value:

New Value:

OK Back

6.8.1.3 Manipulation for OPTIONS Ping.

The following SIP manipulation can be applied as the in-manipulationId to be applied to Options Requests generated by Zoom to the SBC. This will allow the SBC to respond locally to Options Requests.

Modify SIP Manipulation

Name:

Description:

Split Headers:

Join Headers:

CfgRules

Add

Name	Element Type
Respond2OPTIONS	header-rule

Header Rule:

- local-policy
- local-routing-config
- media-profile
- session-agent
- session-group
- session-recording-group
- session-recording-server
- session-translation
- sip-config
- sip-feature
- sip-interface
- sip-manipulation
- sip-monitoring
- translation-rules

Modify Sip manipulation / header rule

Name	<input type="text" value="Respond2OPTIONS"/>
Header Name	<input type="text" value="from"/>
Action	<input type="text" value="reject"/>
Comparison Type	<input type="text" value="case-sensitive"/>
Msg Type	<input type="text" value="any"/>
Methods	<input type="text" value="OPTIONS"/>
Match Value	<input type="text"/>
New Value	<input type="text" value="'200 OK'"/>
CfgRules	<input type="text"/>

Name	Element Type
No data to display	

Please note, If running release SCZ830m1p7 or later, there is a new configuration parameters on the Session Agent Config element, called [ping-response](#). When enabled on each agent, it will take that place of the following SIP-Manipulation.

- local-policy
- local-routing-config
- media-profile
- session-agent
- session-group
- session-recording-group
- session-recording-server
- session-translation
- sip-config
- sip-feature
- sip-interface
- sip-manipulation
- sip-monitoring
- translation-rules

Modify Session Agent

SPL Options	<input type="text"/>
Media Profiles	<input type="text"/>
In Translationid	<input type="text"/>
Out Translationid	<input type="text" value="addPlus"/>
Trust Me	<input type="checkbox"/> enable
Local Response Map	<input type="text"/>
Ping Response	<input checked="" type="checkbox"/> enable
In Manipulationid	<input type="text" value="RespondOPTIONS"/>
Out Manipulationid	<input type="text" value="ZoomManipulation"/>
Manipulation String	<input type="text"/>
Manipulation Pattern	<input type="text"/>

6.9 Session-Translation

The following session-translation is created and applied as out-translationid on the Session-Agent towards Zoom. This session-translation is created to add a +1 and can be used towards as Zoom requires calls to be presented in E.164 format.

The screenshot shows a web-based configuration interface for session translation. On the left is a navigation menu with a search bar and a 'Show All' toggle. The menu items include: local-policy, local-routing-config, media-profile, session-agent, session-group, session-recording-group, session-recording-server, session-translation (highlighted), sip-config, sip-feature, sip-Interface, sip-manipulation, sip-monitoring, translation-rules, and system. At the top of the main area are 'Wizards' and 'Commands' tabs. The main area is titled 'Modify Session Translation' and contains the following fields:

Id	<input type="text" value="addPlus"/>
Rules Calling	<input type="text" value="addPlus x"/>
Rules Called	<input type="text" value="addPlus x"/>
Rules Asserted Id	<input type="text"/>
Rules Redirect	<input type="text"/>
Rules Isup Cdpn	<input type="text"/>
Rules Isup Cgpn	<input type="text"/>
Rules Isup Gn	<input type="text"/>
Rules Isup Rdn	<input type="text"/>
Rules Isup Ocn	<input type="text"/>

At the bottom right of the form are 'OK' and 'Back' buttons.

local-policy
local-routing-config
media-profile
session-agent
session-group
session-recording-group
session-recording-server
session-translation
sip-config
sip-feature
sip-interface
sip-manipulation
sip-monitoring
translation-rules

Modify Translation Rules

Id: addPlus
Type: add
Add String: +1
Add Index: 0
Delete String:
Delete Index: 0 (Range: 0.999999999)

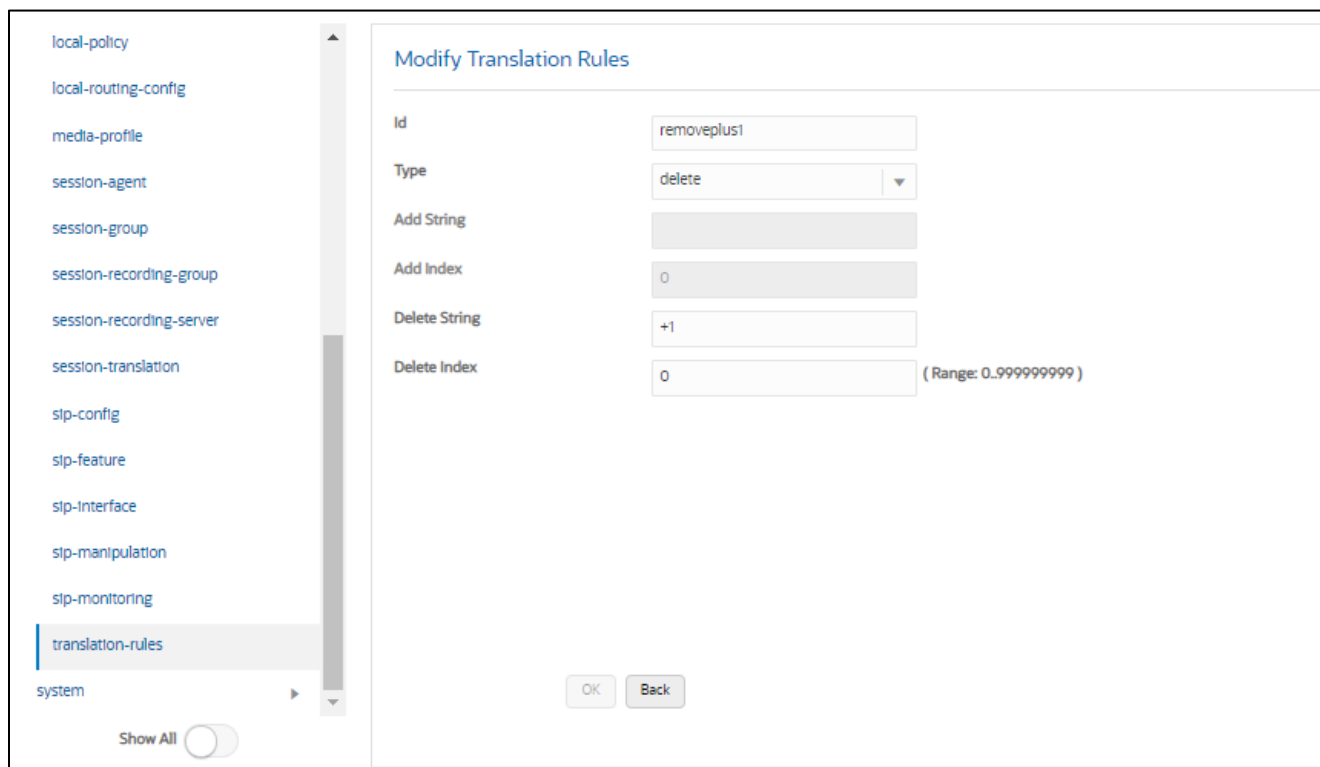
The following session-translation is created and applied as out-translationid on the Session-Agent towards Carrier. This session-translation is created to add remove +1 when call is sent towards Carrier as Carrier in this case requires calls to be presented in 10-digit dial format.

local-policy
local-routing-config
media-profile
session-agent
session-group
session-recording-group
session-recording-server
session-translation
sip-config
sip-feature
sip-interface
sip-manipulation
sip-monitoring
translation-rules
system

Modify Session Translation

Id: removeE164
Rules Calling: removeplus1
Rules Called: removeplus1
Rules Asserted Id: removeplus1
Rules Redirect:
Rules Isup Cdpn:
Rules Isup Cgpn:
Rules Isup Gn:
Rules Isup Rdn:
Rules Isup Ocn:
OK Back

Show All



6.10 Session Timer Profile (Optional)

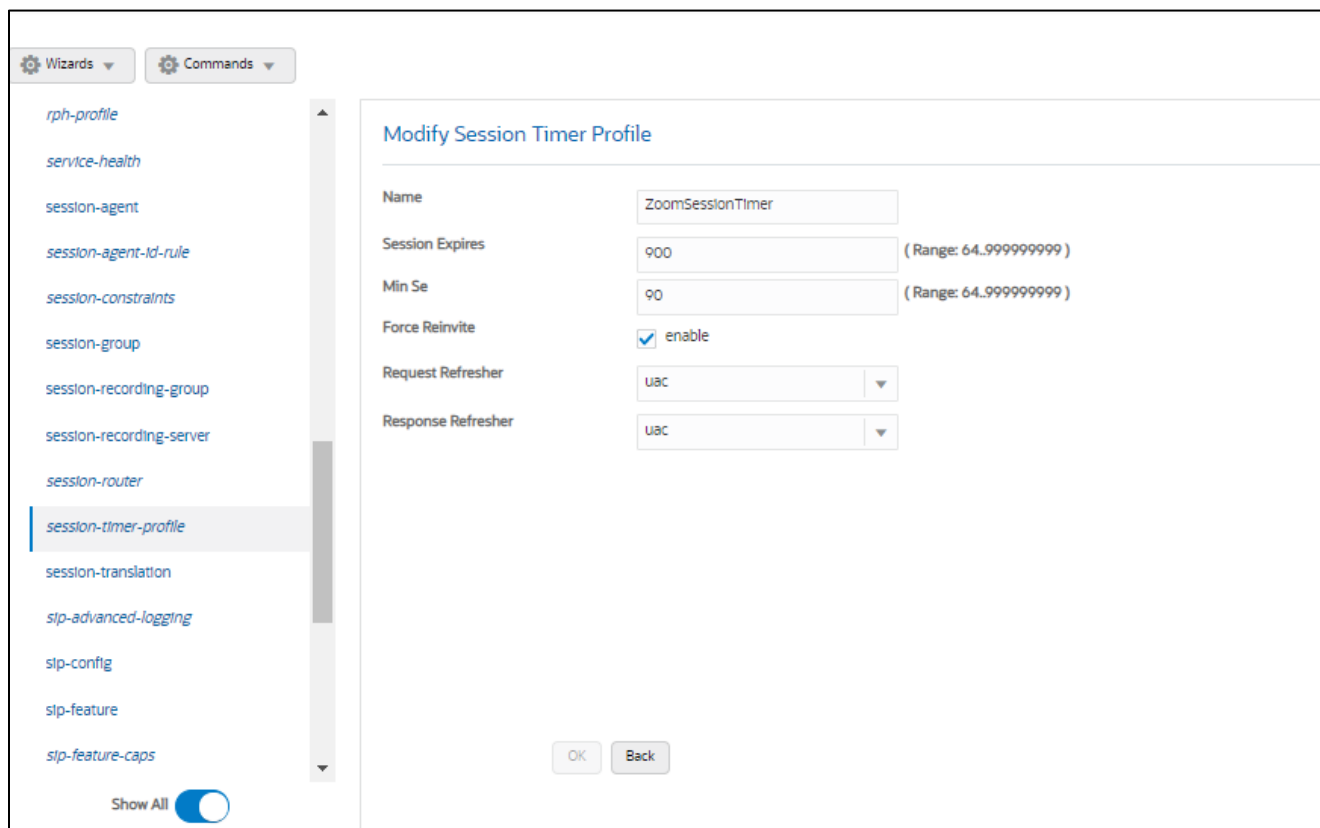
Zoom Phone does support RFC 4028 Session Timers in SIP. In many cases, RFC 4028 is not supported by carriers providing SIP trunking services to their customers. In order to accommodate this, the SBC will interwork between PSTN carrier and Zoom Phone in order to provide support for Session Timers in SIP.

For more information about the Oracle SBC's support for RFC4028, please see the [840 Configuration Guide, page 4-300](#)

GUI Path: session-router/session-timer-profile

ACL Path: config t→session-router→session-timer-profile

Use the following as an example to configure session timer profile on your Oracle SBC. Some parameters may vary to fit your specific environment.



6.11 SIP Interface

The SIP interface defines the transport addresses (IP address and port) upon which the Oracle SBC receives and sends SIP messages

GUI Path: session-router/SIP-interface

ACL Path: config t→session-router→SIP-interface

We will configure the sip-interface for each configured Realm object in this setup –

Click Add, and use the table below as an example to Configure:

Config Parameter	Access	Core	Zoom Trunk	PSTN
Realm ID	Access	ZoomCore	Zoom	SIPTrunk
Out manipulation id	ACME_NAT_TO_FRO M_IP	ACME_NAT_TO_FRO M_IP	ACME_NAT_TO_FRO M_IP	SIPTrunkManipulation
SIP Port Config Parameter				
Address	141.146.36.89	141.146.36.80	141.146.36.68	192.168.1.10
Port	5065	5061	5061	5060
Transport protocol	UDP	TLS	TLS	UDP

TLS profile		TLSZoom	TLSZoom	
Allow anonymous	registered	agents-only	agents-only	agents-only
Session Timer Profile			ZoomSessionTimer	
nat-traversal	always			
registration-caching	enabled			
route-to-registrar	enabled			

Please note, here we will assign some of the configuration elements configured earlier in this document, i.e.....

- TLS Profile
- Session-timer-profile
- SIP-Manipulations

Since the Access realm is configured to handle registrations the following parameter must be enabled on this realm to allow SBC to cache the registrations on this sip-interface.

nat-traversal - always
 registration-caching – enabled
 route-to-registrar – enabled

To forwards call requests from ATA Device towards the registrar IP Address and Port configured in the sip-config Section of the document.

Alternatively, a local-policy configuration can also be used in case route-to-registrar is not configured.

Action	Select	State	Realm ID	Description	Carriers	Trans Expire	Initial Inv Trans Expire
:	<input type="checkbox"/>	enabled	Access				0
:	<input type="checkbox"/>	enabled	SIPTrunk				0
:	<input type="checkbox"/>	enabled	Zoom				0
:	<input type="checkbox"/>	enabled	ZoomCore				0

6.12 Session Agents

Session Agents are configuration elements which are trusted agents that can both send and receive traffic from the ORACLE SBC with direct access to the trusted data path.

GUI Path: session-router/session-agent

ACLI Path: config t→session-router→session-agent

You will need to configure three session agents for Zoom Trunk, Zoom Registrar and one for SIPTrunk.

- Click Add, and use the table below to configure:

Config parameter	Zoom Trunk	Zoom Registrar	SIPTrunk
Hostname	162.12.232.59	us01sip0h.ny.zoom.us	192.168.1.10
IP Address	162.12.232.59		192.168.1.10
Port	5061	5091	5060
Transport method	StaticTLS	StaticTLS	UDP+TCP
Realm ID	Zoom	ZoomCore	Peer_SIPTrunk
Ping Method	OPTIONS	OPTIONS	OPTIONS
Ping Interval	30	30	30
Ping Response	Enabled	Enabled	Enabled
out-manipulationid	ZoomE164		

Note: Ping Response enabled takes the place of the [Respond Options Sip Manipulation Rule](#)

Hostname	IP Address	Port	State	App Protocol	Realm ID	Description
162.12.232.59	162.12.232.59	5061	enabled	SIP	Core_Zoom	SA to Zoom TLS
162.12.233.59	162.12.233.59	5061	enabled	SIP	Core_Zoom	SA to Zoom TLS
68.68.117.67	68.68.117.67	5060	enabled	SIP	Peer_SIPTrunk	

- Hit the OK tab at the bottom of each when applicable

Click OK at the bottom

6.13 Local Policy Configuration

Local Policy config allows for the SBC to route calls from one end of the network to the other based on routing criteria.

GUI Path: session-router/local-policy

ACL Path: config t→session-router→local-policy

In order to route SIP traffic to and from Zoom Phone Platform, the following local-policies will need to be configured.

- Click Add and use the following and an example to configure:

6.13.1 Route Calls from Zoom Trunk To PSTN:

The screenshot shows the 'Modify Local Policy' configuration page. The left sidebar lists various configuration sections, with 'local-policy' selected. The main area contains the following fields:

- From Address: * X
- To Address: * X
- Source Realm: Zoom X
- Description: (empty)
- State: enable
- Policy Priority: none

Below these fields is a 'Policy Attributes' table:

Action	Select	Next Hop	Realm	Action	Terminate Recursion	Cost	State	App Protocol	Lookup	Next Key
:	<input type="checkbox"/>	68.68.117.67	SIPTrunk	none	disabled	0	enabled		single	

Policy Attribute:

The screenshot shows the 'Modify Local policy / policy attribute' configuration page. The left sidebar lists various configuration sections, with 'local-policy' selected. The main area contains the following fields:

- Next Hop: 68.68.117.67
- Realm: SIPTrunk
- Action: none
- Terminate Recursion: enable
- Cost: 0 (Range: 0.999999999)
- State: enable
- App Protocol: (empty)
- Lookup: single
- Next Key: (empty)

6.13.2 Route Calls from PSTN To Zoom:

Action	Select	Next Hop	Realm	Action	Terminate Recursion	Cost	State	App Protocol	Lookup	Next Key
:	<input type="checkbox"/>	162.12.232.59	Zoom	none	disabled	0	enabled		single	

Policy Attribute:

Next Hop	Realm	Action	Terminate Recursion	Cost	State	App Protocol	Lookup	Next Key
162.12.232.59	Zoom	none	<input type="checkbox"/> enable	0 (Range: 0.999999999)	<input checked="" type="checkbox"/> enable		single	

- Click OK at the bottom of each when applicable:
-

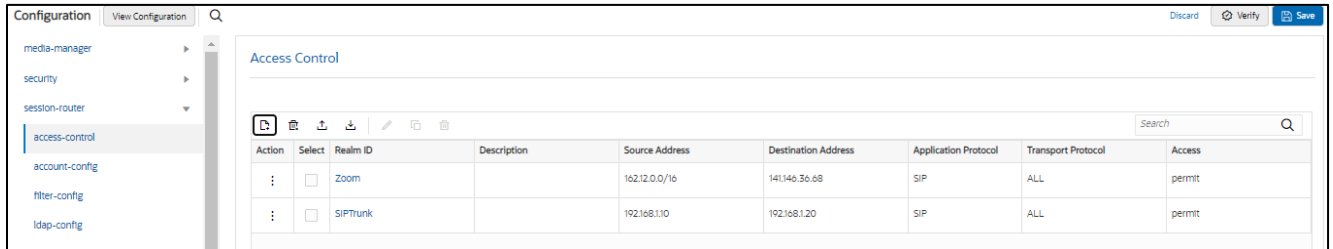
6.14 Access Controls

To enhance the security of your Oracle Session Border Controller, we recommend configuration access controls to limit traffic to only trusted IP addresses on all public facing interfaces

GUI Path: session-router/access-control

ACLI Path: config t→session-router→access-control

Please use the example below to configure access controls in your environment for both Zoom IP's, as well as SIPTrunk IP's (if applicable)



Notice the trust level on this ACL is set to high. When the trust level on an ACL is set to the same value of as the access control trust level of its associated realm, this create an implicit deny, so only traffic from IP addresses configured as ACL's with the same trust level will be allowed to send traffic to the SBC. For more information about trust level on ACL's and Realms, please see the [SBC Security Guide, Page 3-10](#).

- Click OK at the bottom

Save and activate your configuration!

The SBC configuration is now complete. Move to verify the connection with Zoom.

7 Verify Connectivity

7.1 ORACLE SBC Options Ping

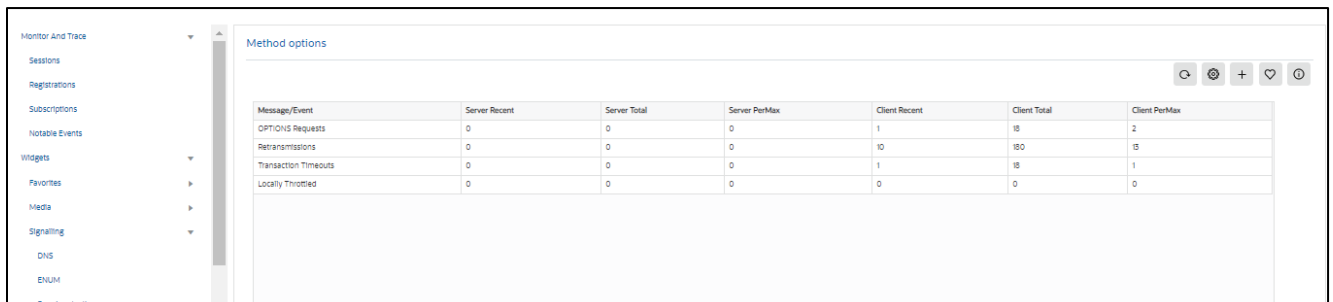
After you've paired the ORACLE SBC with Zoom, validate that the SBC can successfully exchange SIP Options with Zoom Cloud Voice.

While in the ORACLE SBC GUI, Utilize the "Widgets" to check for OPTIONS to and from the SBC.

- At the top, click "Widgets"

This brings up the Widgets menu on the left hand side of the screen

GUI Path: Monitor and Trace/Signaling/SIP/Methods/OPTIONS



- Looking at both the **Server Recent** and **Client Recent**, verify the counters are showing OPTIONS Requests and 200OK responses.

8 SBC Behind NAT SPL configuration

This configuration is needed when your SBC is behind a NAT device. This is configured to avoid loss in voice path and SIP signaling.

The Support for SBC Behind NAT SPL plug-in changes information in SIP messages to hide the end point located inside the private network. The specific information that the Support for SBC Behind NAT SPL plug-in changes depends on the direction of the call.

For example, from the NAT device to the SBC or from the SBC to the NAT device.

Configure the Support for SBC Behind NAT SPL plug-in for each SIP interface that is connected to a NAT device. One public-private address pair is required for each SIP interface that uses the SPL plug-in, as follows.

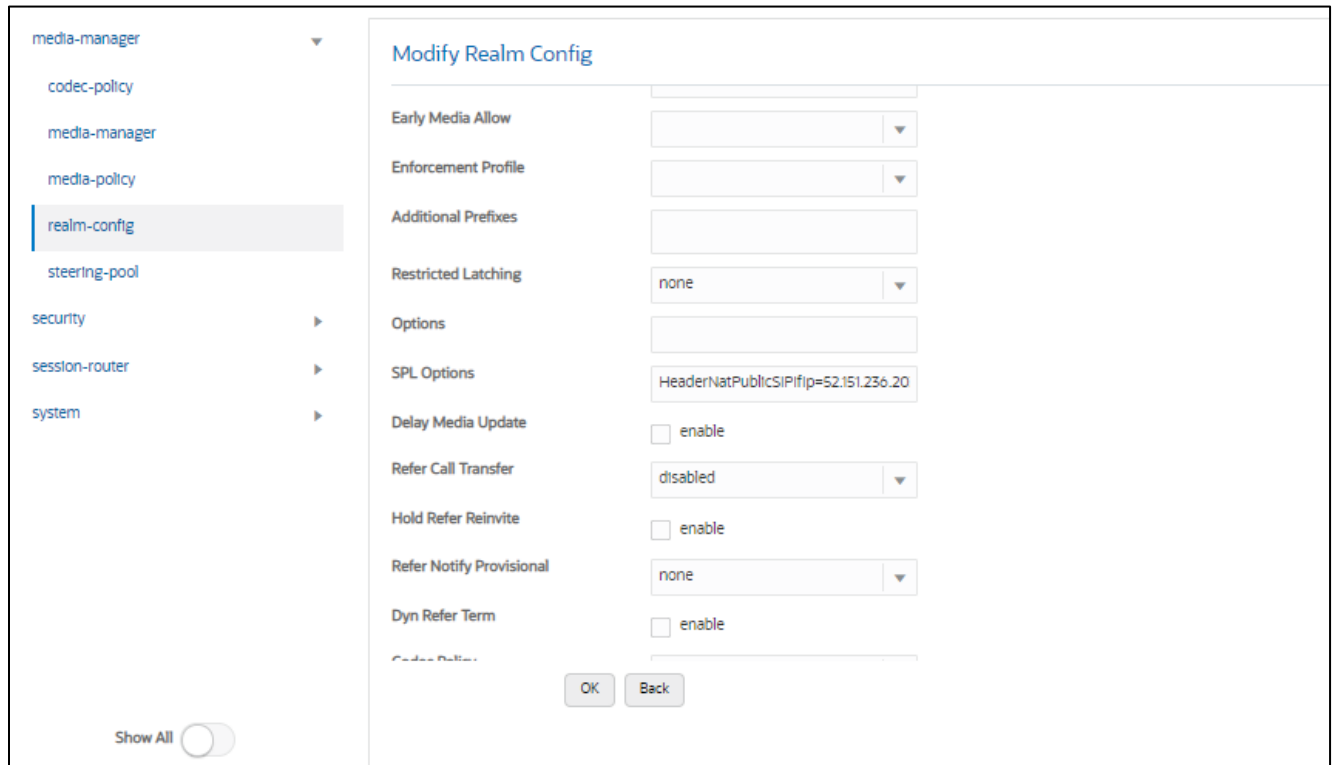
- The private IP address must be the same as the SIP Interface IP address.
- The public IP address must be the public IP address of the NAT device

Here is an example configuration with SBC Behind NAT SPL config. The SPL is applied to the Zoom side SIP interface.

To configure SBC Behind NAT SPL Plug in, go to session-router->SIP-interface->spl-options and input the following value, save and activate.

```
HeaderNatPublicSIPIfIp=52.151.236.203, HeaderNatPrivateSIPIfIp=10.0.4.4
```

Here HeaderNatPublicSIPIfIp is the public interface ip and HeaderNatPrivateSIPIfIp is the private ip.



This configuration would be applied to each SIP Interface in the ORACLE SBC configuration that was deployed behind a Nat Device.


9 Caveat

9.1 Transcoding Opus Codec

Opus is an audio codec developed by the IETF that supports constant and variable bitrate encoding from 6 kbit/s to 510 kbit/s and sampling rates from 8 kHz (with 4 kHz bandwidth) to 48 kHz (with 20 kHz bandwidth, where the entire hearing range of the human auditory system can be reproduced). It incorporates technology from both Skype's speech-oriented SILK codec and Xiph.Org's low-latency CELT codec. This feature adds the Opus codec as well as support for transrating, transcoding, and pooled transcoding. Opus can be adjusted seamlessly between high and low bit rates, and transitions internally between linear predictive coding at lower bit rates and transform coding at higher bit rates (as well as a hybrid for a short overlap). Opus has a very low algorithmic delay (26.5 ms by default), which is a necessity for use as part of a low audio latency communication link, which can permit natural conversation, networked music performances, or lip sync at live events. Opus permits trading-off quality or bit rate to achieve an even smaller algorithmic delay, down to 5 ms. Its delay is very low compared to well over 100 ms for popular music formats such as MP3, Ogg Vorbis, and HE-AAC; yet Opus performs very competitively with these formats in terms of quality across bit rates.

Zoom Phone fully support the use of OPUS, but advertises a static value of 40000 for max average bit rate

Although the range for maxaveragebitrate is 6000 to 51000, only bit rates of 6000 to 30000 bps are transcodable by the DSP's on the Oracle SBC. A media profile configured with a value for maxaveragebitrate greater than 30000 is not transcodable and cannot be added on egress in the codec-policy element.



The Oracle SBC will however support the entire range of of maxaveragebitrate if negotiated between the parties of each call flow.



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Integrated Cloud Applications & Platform Services

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