

Oracle SBC as a Local Gateway with Cisco Webex Calling and Contact Center

Technical Application Note



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Revision History

Version	Description of Changes	Date
1.0	 Oracle SBC integration with Cisco Webex Calling as 3rd party Local Gateway (LGW) 	30 th October 2022
1.1	 Added Appendix B section to the document for the new feature which supports Cisco DTMF with OPUS codec 	05 th January 2023
1.2	 Added ACLI and GUI Cert Import Added multitenancy monitoring. Added ACK method to Sip Manip. SBC version changed to 9.x Ice Config for Media Optimization Included GCM ciphers for SRTP. Added support for Cisco Webex Contact Center Modified Caveat to include software fix no available 	08 th November 2024
1.3	SBC v.9.3 Re-Certification Updates	

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1. Intended Audience

This document is intended for use by Oracle Systems Engineers, third party Systems Integrators, Oracle Enterprise customers and partners and end users of the Oracle Enterprise Session Border Controller (SBC). It is assumed that the reader is familiar with basic operations of the Oracle Enterprise Session Border Controller platform along with Cisco Webex Calling and Cisco Webex Contact Center with 3rd Party Local Gateway.

2 Validated Oracle SBC version

Oracle conducted tests with SBC 9.x software – this software with the configuration listed below can run on any of the following products:

- AP 1100
- AP 3900
- AP 3950
- A P4600
- AP 4900
- AP 6350
- AP 6400
- VME
- Oracle SBC on Public Cloud

3 Document Overview

This Oracle technical application note outlines how to configure the Oracle SBC to interwork between PSTN Trunk with Cisco Webex Calling Solution and Cisco Webex Contact Center. The solution contained within this document has been tested using Oracle Communication SBC with software version **OS 9.x version.**

4 Cisco Webex Calling

Cisco Webex Calling is a cloud calling solution that delivers enterprise-grade calling, enabling you to replace your on-premises PBX network with a globally trusted cloud calling solution. Webex Calling easily extends to a complete collaboration experience that includes market-leading calling, meetings, messaging, contact center, and integrated devices for all situations.

Webex Calling Cloud service or in short "Webex Calling" supports "Bring Your Own PSTN" and Enterprise dialing using through what is termed as a Local Gateway that sits at the edge of the Customer's VoIP network. A local gateway is a SIP Session Border Controller that interworks with Webex Calling cloud service in specific ways and this Local gateway MUST operate specified conditions with Webex Calling. Local Gateway feature enables Webex Calling customers to continue using their existing PSTN service provider. *Oracle SBC works with Webex calling as 3rd party Local Gateway in Certificate based Trunking model.*

For additional information on Cisco Webex Calling and certificate-based trunking, please check the below links:

https://www.Webex.com/products/Webex-calling.html

https://help.Webex.com/en-us/article/n0xb944/Configure-Trunks,-Route-Groups,-and-Dial-Plans-for-Webex-Calling#Cisco Reference.dita 20664899-b518-4f5d-bc92-88af4a5c6694

Please note that the IP Addresses, FQDN and configuration names and details given in this document are used for reference purposes only. These same details cannot be used in customer configurations. End users of this document can use the configuration details according to their network requirements. There are some public facing IPs (externally routable IPs) that we use for our testing are masked in this document for security reasons. The customers can configure any publicly routable IPs for these sections as per their network architecture needs.

5 Introduction

5.1 Audience

This is a technical document intended for telecommunications engineers with the purpose of configuring Cisco Webex Calling with 3rd party LGW feature using Oracle Enterprise SBC. There will be steps that require navigating the Oracle SBC GUI interface, understanding the basic concepts of TCP/UDP, IP/Routing, DNS server, SIP/RTP and TLS/SRTP are also necessary to complete the configuration and for troubleshooting, if necessary.

5.2 Requirements

 Fully functioning Cisco Webex Control Hub (Provisioned Webex Control Hub with necessary Webex Calling licenses/Subscription and prepared Webex Calling environment)

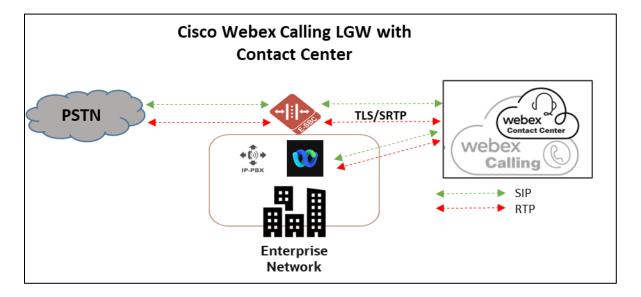
https://help.webex.com/en-us/article/n4cprps/Prepare-Your-Environment-for-Webex-Calling

Oracle Enterprise Session Border Controller (hereafter Oracle SBC) running 9.x version.

The below revision table explains the versions of the software used for each component: This table is Revision 1 as of now:

Software Used	SBC Version
Revision 1	9.x

5.3 Architecture



The configuration, validation and troubleshooting are the focus of this document and will be described in three phases:

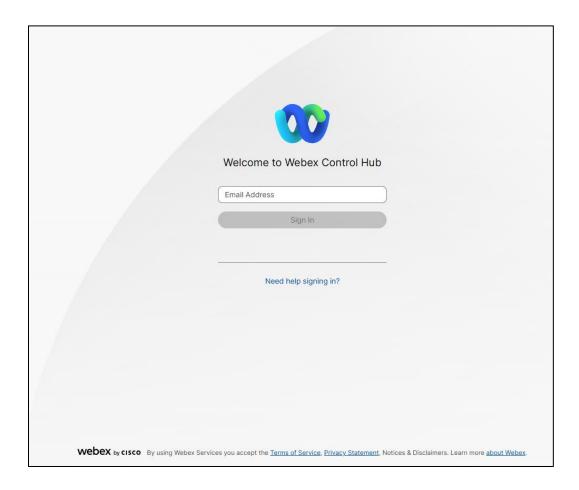
- Phase 1 Configuring Cisco Webex Calling with for the Oracle SBC as a local gateway (LGW)
- Phase 2 Configuration of the Oracle SBC.
- Phase 3 Configuring the Cisco Webex Contact Center.

6 Cisco Webex Configuration

The configuration of Cisco Webex is a mandatory prerequisite before starting the SBC configuration. The Webex admin should <u>Configure Trunks</u>, <u>Route Groups</u>, <u>and Dial Plans</u> for Webex Calling to create a trunk toward Oracle SBC. Once the configuration on Webex Control Hub is complete, the admin will be provided with destination (Webex Edge proxy) Address that need to be configured on the Oracle SBC.

Login to Webex Control Hub with admin credentials.

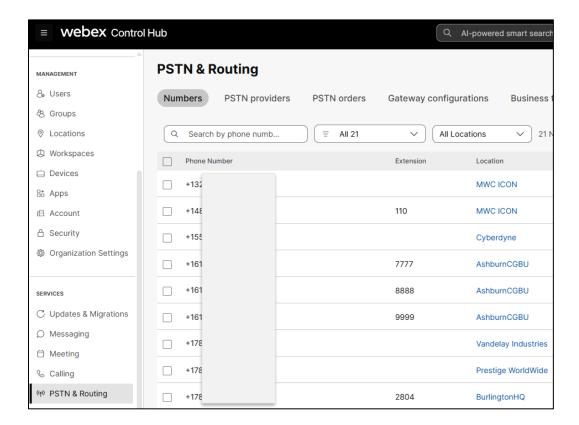
Note: This document assumes that you have already set up your Cisco account, configured your location, and registered the necessary FQDN(s) in the Cisco Control Hub. Please ensure all these prerequisites are completed before proceeding with the trunk configuration to connect to the Oracle SBC



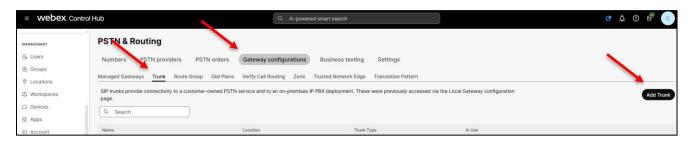
6.1 Trunk Configuration

SIP trunks provide connectivity to a customer-owned PSTN service and to an on-premises IP PBX deployment. These were previously accessed via the Local Gateway configuration page.

• In the left menu, select **PSTN & Routing** from the menu:

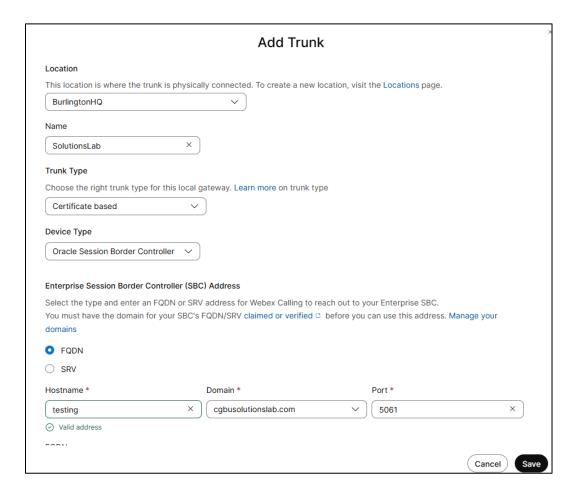


Next, at the top, click on Gateway configurations/Trunk, top right, click Add Trunk:



In the Add Trunk Configuration window, configure the following:

- Choose a Location from the drop down menu
- Name
- Trunk Type (Certificate Based)
- Select Oracle Session Border Controller for Device Type
- Select the radio button for FQDN
- Select your domain from the drop down menu
- Add the hostname
- Add Port
- Enter the Maximum number of concurrent Calls (must be in range <250, 6500>)



Click Save at the bottom

6.2 Route Group

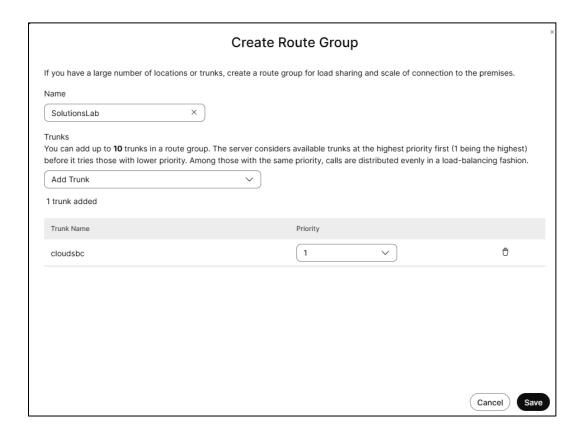
A group of trunks that allows further scale and redundancy with the connection to the premises.

Under PSTN & Routing, select **Route Group** from the top menu, then select **Create Route Group**:



In the Create Route Group window, configure the following:

- Name
- Under Trunks, select the trunk we just added from the drop down menu
- Select a priority from the drop down menu (Priority 1 is highest)



Click Save at the bottom.

6.3 Dial Plans

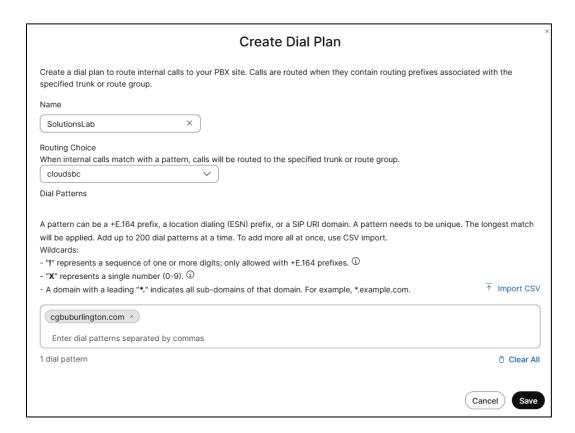
Dial plans allow you to route calls to on-premises extensions via your trunk or route group.

Under PSTN & Routing, select **Dial Plans** from the top menu, then select **Create Dial Plan:**



In the Create Dial Plan window, configure the following:

- Name
- Routing Choice (you can select either Trunk or Route Group. In this example, we'll select the trunk)
- Enter a Dial pattern or import CSV (optional)

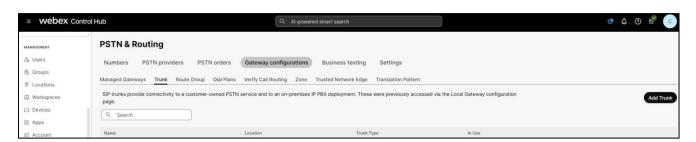


Click Save at the bottom.

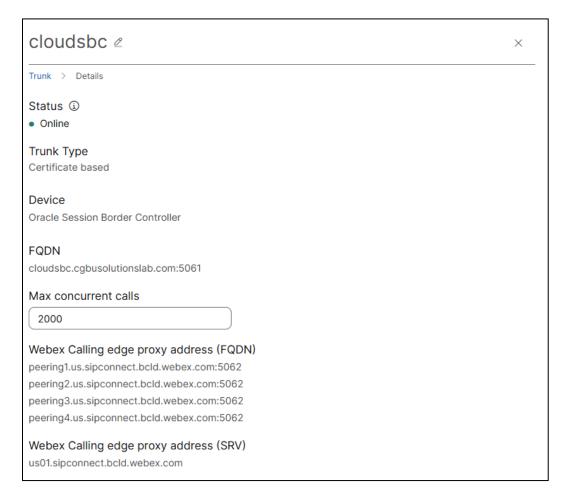
6.4 Webex Edge Proxy Address

To view the destination (Webex Edge proxy) address, which will be used as the Session Agent in the Oracle SBC to connect to Cisco Webex:

Under PSTN & Routing, select Trunk from the top menu:



Next, click on the trunk created in section 6.1



Cisco recommends using SRV based Webex Calling edge address for connection your Oracle SBC as a local gateway.

Please note that Webex Calling Proxy Addresses given below are example addresses which are used for testing and these values will vary from region to region. For more information about the Webex Calling Proxy Addresses, please contact your Cisco team.

This concludes the minimum configuration required for the gateway. We'll now move on to configuring your Oracle SBC as a local gateway.

7 Configuring the SBC

This chapter provides step-by-step instructions for configuring the Oracle SBC for Cisco Webex Calling LGW. In this configuration, both signaling and media between the Oracle SBC and Cisco Webex are secured using TLS and SRTP. A generic example of configuring a non-secure PSTN service on the Oracle SBC is also included for illustration purposes only; however, it is not the primary focus of this chapter.

There are two methods for configuring the Oracle SBC: ACLI or GUI. For the purposes of this note, we'll be using the Oracle SBC GUI for all configuration examples. We will however provide the ACLI path to each element.

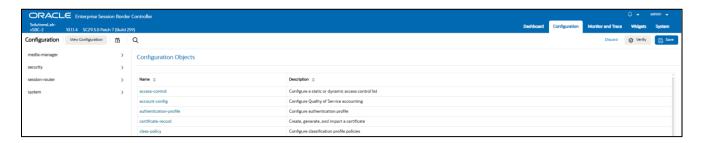
This guide assumes the Oracle SBC has been installed, the management interface has been configured, the product has been selected, and entitlements have been assigned. Additionally, web-server-config should be enabled for GUI access. If you need more information on installing your SBC platform, please refer to the <u>ACLI configuration guide</u>.

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

Once you have access to the Oracle SBC GUI, click the **Configuration** tab at the top. This will display the Oracle SBC Configuration Objects List on the left-hand side of the screen.

Any configuration parameter not specifically listed below can remain at its Oracle SBC default value and does not require a change for the connection to Cisco Webex Calling to function properly.

Note: The configuration examples below were captured from a system running the latest GA software, version 9.3.0.



Please refer to the Oracle SBC GUI Guide for more information.

 $\frac{https://docs.oracle.com/en/industries/communications/enterprise-session-border-controller/9.3.0/webgui/webgui-gui-guide.pdf$

Expert mode is used for configuration.

Tip: To make this configuration simpler, one can directly search the element to be configured, from the Objects tab available.

7.1 System-Config

To enable system level functionality for the Oracle SBC, you must first enable the system-config.

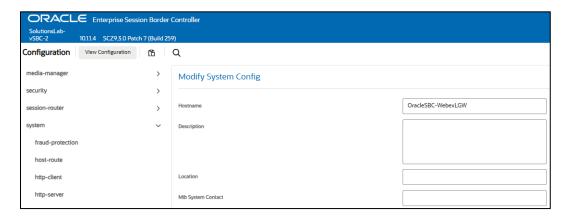
GUI Path: system/system-config

ACLI Path: config t→system→system-config

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

- Hostname
- Description
- Location
- Default Gateway (recommended to be the same as management interface gateway)

Transcoding Core (This field is only required if you have deployed a VME SBC)



For VME, transcoding cores are required. Please refer the documentation here for more information

https://docs.oracle.com/en/industries/communications/enterprise-session-border-controller/9.3.0/releasenotes/esbc-release-notes.pdf

The above step is needed only if any transcoding is used in the configuration.

7.1.1 NTP-Config

You can use the following example to connect the Oracle SBC to any network time servers you have in your network. This is an optional configuration but recommended.

GUI Path: system/ntp-config

ACLI Path: config t→system→ntp-sync



7.2 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 Cisco Webex Calling, the other to connect to PSTN Network. The slots and ports used in this example may be different from your network setup.

7.2.1 Physical Interfaces

GUI Path: system/phy-interface

ACLI Path: config t→system→phy-interface

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

Config Parameter	PSTN	Cisco Webex
Name	s0p0	S1p0
Operation Type	Media	Media
Slot	0	1
Port	0	0



7.2.2 Network Interfaces

GUI Path: system/network-interface

ACLI Path: config t→system→network-interface

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

Configuration Parameter	PSTN	Cisco Webex
Name	s0p0	s1p0
IP Address	10.1.2.4	10.1.3.4
Netmask	255.255.255.0	255.255.255.0
Gateway	10.1.2.1	10.1.3.1
DNS Primary IP		8.8.8.8
DNS Domain		cgbusolutionslab.com



Click OK at the bottom of each after entering config information

Next, we'll configure the necessary elements to secure signaling and media traffic between the Oracle SBC and Cisco Webex Calling.

7.3 Security Config

This section describes how to configure the SBC for both TLS and SRTP communication with Cisco Webex Calling.

Cisco Webex Calling supports TLS connections from SBCs for SIP signaling and uses SRTP for media traffic. It requires SBCs to present certificates signed by trusted Certificate Authorities. A list of Cisco's supported CA's can be found at:

Cisco Webex Calling Supported Root Certificate Authorities

7.3.1 Certificate Records

"Certificate-records" are configuration elements on Oracle SBC which capture 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

ACLI Path: config t→security→certificate-record

For the purposes of this application note, we'll create three certificate records. They are as follows:

- SBC Certificate (end-entity certificate)
- GoDaddy Root Cert (Root CA used to sign the SBC's end entity certificate)
- IdentTrust Commercial Root CA (Cisco Webex Presents the SBC a certficate signed by this authority)

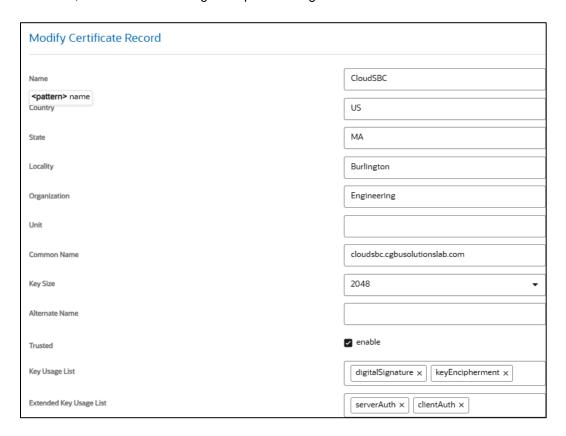
Note: The GoDaddy RootCA is only part of this example, as that is the Authority we used to sign our SBC certificate. You would replace this with the root and/or intermediate certificates used to sign the CSR generated from your SBC.

7.3.1.1 SBC End Entity Certificate

The SBC's end entity certificate is the certificate the SBC presents to Cisco to secure the connection. The only requirements when configuring this certificate is the common name must contain the SBC's FQDN and the extended key usage list must contain both serverAuth and clientAuth. In this example our common name will be cloudsbc.cgbusolutionslab.com. You must also give it a name. All other fields are optional, and can remain at default values.

To Configure the certificate record:

Click Add, and use the following example to configure the SBC certificate



· Click OK at the bottom

Next, using this same procedure, configure certificate records for the Root CA certificates

7.3.1.2 Root CA and Intermediate Certificates

7.3.1.2.1 Go Daddy Root

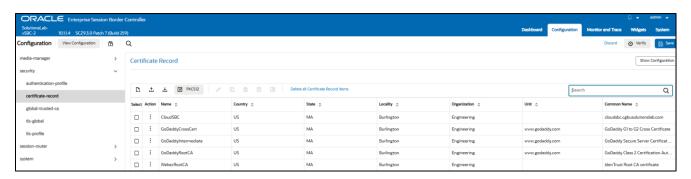
The following, GoDaddyRoot, is the root CA certificate used to sign the SBC's end entity certificate. As mentioned above, your root CA and/or intermediate certificate may differ. This is for example purposes only.

7.3.1.2.2 IdenTrust Commercial Root CA

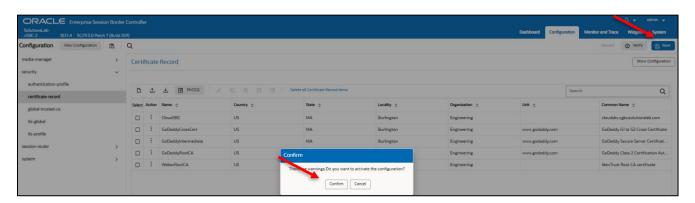
Cisco presents a certificate to the SBC which is signed by IdentTrust Commercial Root CA. To trust this certificate, your SBC must have the certificate listed as a trusted ca certificate. You can download this certificate here: IdenTrust Commercial Root CA.

Please use the following table as a configuration reference: Modify the table according to the certificates in your environment.

Config Parameter	GoDaddy Root	Identrust Commercial Root
Common Name	Go Daddy Class2 Root CA	IdenTrust Root CA Certificate
Key Size	2048	2048
Key-Usage-List	digitalSignature keyEncipherment	digitalSignature keyEncipherment
Extended Key Usage List	serverAuth	serverAuth
Key algor	rsa	rsa
Digest-algor	Sha256	Sha256



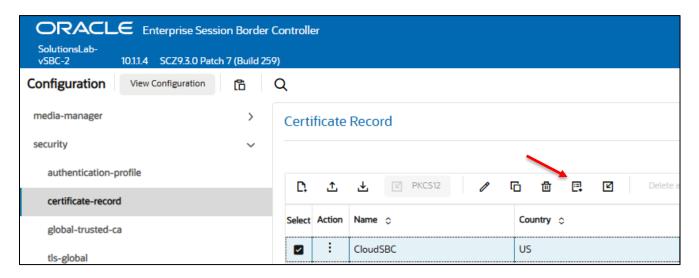
At this point, before generating a certificate signing request, or importing any of the Root CA certs, we must save and activate the configuration of the SBC.



7.3.1.2.3 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 intermidiate 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 at the top:





Copy/paste the text that gets printed on the screen as shown above and upload to your CA server for signature

Also note, **another save and activate is required** before you can import the certificates to each certificate record created above.

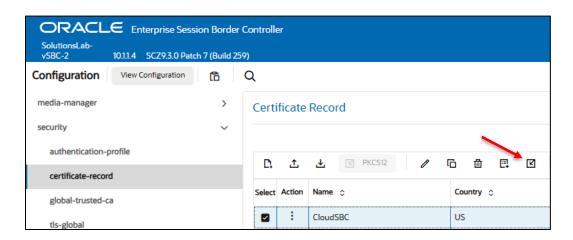
Once you have received the signed certificate back from your signing authority, we can now import all certificates to the SBC configuration.

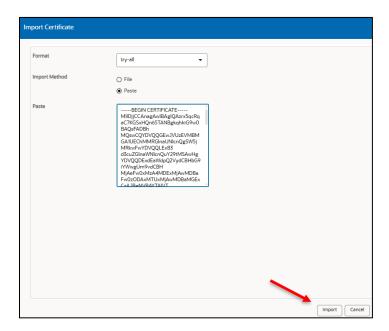
7.3.1.2.4 Import Certificates to the SBC

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 a third **save/activate** from the WebGUI to complete the configuration of certificates on the Oracle SBC.





Once pasted in the text box, select Import at the bottom, then save and activate your configuration.

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

7.3.2 TLS Profile

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

GUI Path: security/tls-profile

ACLI Path: config t→security→tls-profile

• Click Add, use the example below to configure



Select OK at the bottom

Next, we'll move to securing media between the SBC and Cisco Webex Calling

7.3.3 Media Security

This section outlines how to configure support for media security between the OCSBC and Cisco Webex Calling.

7.3.3.1 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. Cisco Supports the following crypto's:

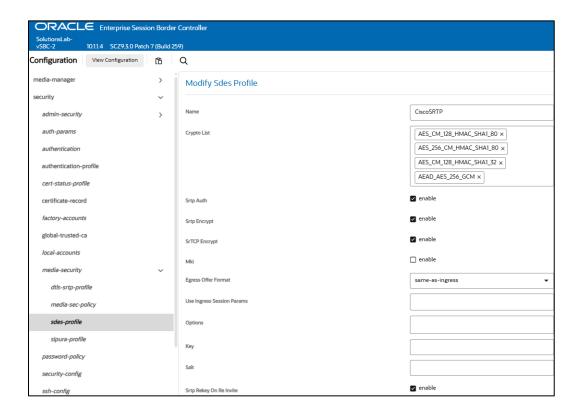
- AES_CM_256_HMAC_SHA1_80
- AES_CM_128_HMAC_SHA1_80
- AES_CM_128_HMAC_SHA1_32

In the SBC's GUI, on the bottom left, you will need to enable the switch "**Show All**" to access the media security configuration elements.

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

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

Click Add, and use the example below to configure



You may notice there is a fourth crypto in the list, AEAD_AES_256_GCM. This is only supported in Cisco for Government environments.

• Select OK at the bottom.

7.3.4 Media Security Policy

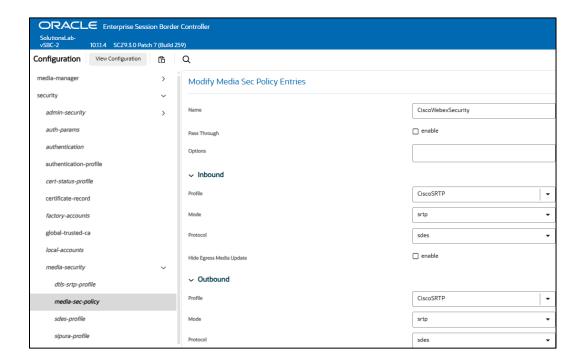
Media-sec-policy instructs the SBC how to handle the SDP received/sent under a realm (RTP, SRTP or any) and, if SRTP needs to be used, the sdes-profile that needs to be used

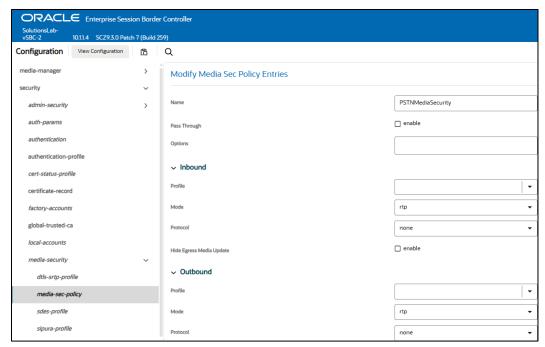
In this example, we are configuring two media security policies. One to secure and decrypt media toward Cisco Webex, the other for non-secure media facing PSTN.

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

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

• Click Add, use the examples below to configure.





• Select OK at the bottom of each when finished.

This finishes the security configuration portion of the application note. We'll now move on to configuring media and transcoding.

7.4 Transcoding Configuration (Optional)

Transcoding is the ability to convert between media streams that are based upon disparate codecs. The OCSBC supports IP-to-IP transcoding for SIP sessions and can connect two voice streams that use different coding algorithms with one another.

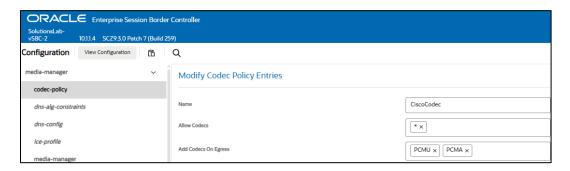
7.4.1 Codec Policies

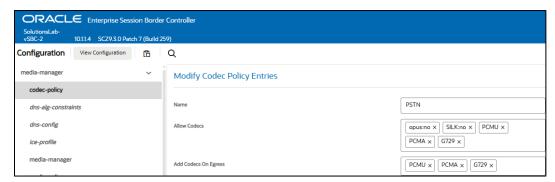
Codec policies are sets of rules that specify the manipulations to be performed on SDP offers allowing the Oracle SBC the ability to add, strip, and reorder codecs for SIP sessions.

GUI Path: media-manager/codec-policy

ACLI Path: config t→media-mangaer→codec-policy

Since some SIP Trunks may have issues with the codecs being offerened by Cisco Webex, you can create a codec policy to remove unwanted or unsupported codecs from the request/responses to your Sip Trunk provider.





• Select OK at the bottom of each when finished

7.4.2 ICE Profile

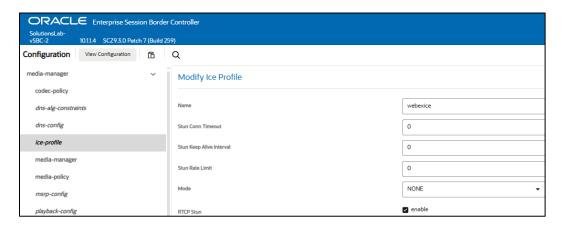
Interactive Connectivity Establishment - Session Traversal Utility for NAT (ICE STUN lite mode) enables an Advanced Media Termination client to perform connectivity checks and can provide several STUN servers to the browser. ICE STUN support requires configuring an ICE Profile.

The use of ICE is required only if using Cisco Webex Media Optimization Feature.

GUI Path: media-manager/ice-profile

ACLI Path: config t→media-manger→ice-profile

Click Add, use the example below as a guide to configure.



You must enable RTCP Stun on the ICE Profile for Cisco Media Optimization feature to be successful.

In some environments, it may be necessary to change the default values for Stun Conn Timeout, Stun Keep Alive Interval, and Stun Rate Limit to a value of 0 (zero).

Select OK at the bottom.

This concludes the configuration for transcoding and Advanced Media Termination options on the SBC. We can now move to setup Media.

7.5 Media Configuration

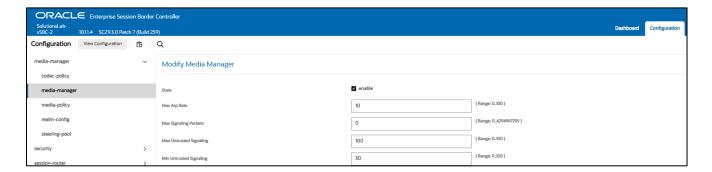
This section will guide you through the configuration of media manager, realms and steering pools, all of which are required for the SBC to handle signaling and media flows toward Cisco and PSTN.

7.5.1 Media Manager

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

GUI Path: media-manager/media-manager

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



• Click OK at the bottom.

7.5.2 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-manger/realm-config

ACLI Path: config t→media-manger→realm-config

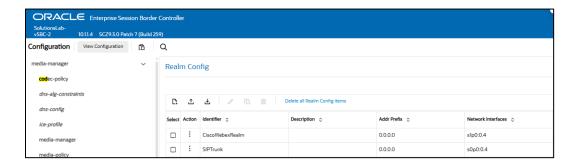
 Click Add and use the following table as a configuration example for the realms. The following parameters are all required unless mentioned as optional below.

Config Parameter	Cisco Webex	PSTN
Identifier	CiscoWebexRealm	SipTrunk
Network Interface	S1p0:0	S0p0:0
Mm in realm	V	√
Media Sec policy	CiscoWebexSecurity	PSTNMediaSecurity
ice profile	webexice (required for media opt only)	
Codec policy	CiscoCodec	PSTN
Trunk Context	Cloudsbc.cgbusolutionslab.com	
Access-control-trust-level	HIGH	HIGH

Also notice the realm configuration is where we assign some of the elements configured earlier in this document. IE...

- Network Interface
- Media Security Policy
- Ice Profile (optional, only required if using Media Opt)
- Codec Policy (optional)

Please also note the "trunk context" parameter. The value you set here should be the SBC's FQDN, which was registered earlier in the Webex Control Hub. Later in this application note, this value will be used to adjust SIP header syntax to match Cisco Webex Calling requirements.



• Click OK at the bottom of each.

7.5.3 Steering Pools

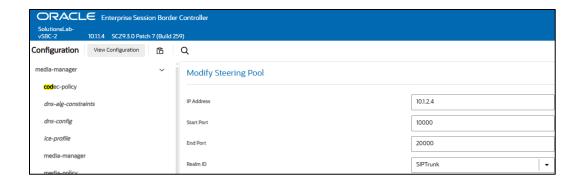
Steering pools define sets of ports that are used for steering media flows through the OCSBC. These selected ports are used to modify the SDP to cause receiving session agents to direct their media toward this system.

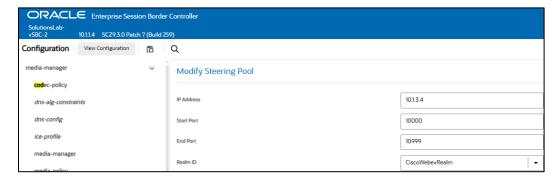
We configure one steering pool for PSTN. The other facing Cisco.

GUI Path: media-manger/steering-pool

ACLI Path: config t→media-manger→steering-pool

• Click Add, and use the below examples to configure.





Select OK at the bottom

We will now work through configuring what is needed for the SBC to handle SIP signaling.

7.6 Sip Configuration

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

7.6.1 Sip-Config

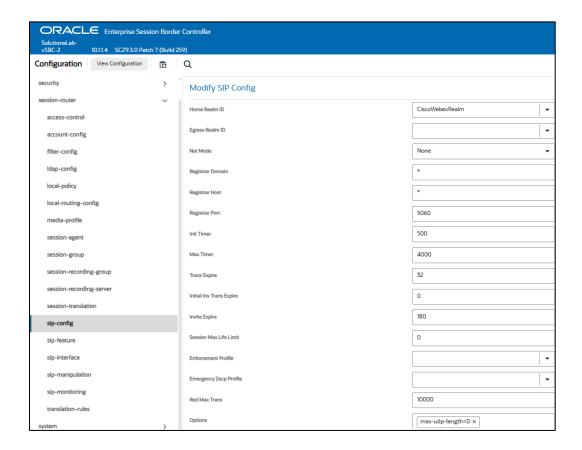
To enable sip related objects on the Oracle SBC, you must first configure the global Sip Config element:

GUI Path: session-router/sip-config

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

There are only two recommended changes/additions to the global Sip Config.

- Set the home realm ID parameter to CiscoWebexRealm, and add the following hidden option:
- Max-udp-length=0: Setting this option to zero (0) forces sipd to send fragmented UDP packets. Using
 this option, you override the default value of the maximum UDP datagram size (1500 bytes; sipd
 requires the use of SIP/TCP at 1300 bytes).



Select OK at the bottom.

7.6.2 Sip Manipulation

To ensure that the SBC generates SIP messages conforming to Cisco Webex Calling requirements, create SIP manipulation rules to:

- Change the Contact header's host URI to the SBC's FQDN.
- Add a Contact header to SIP OPTIONS messages that includes the SBC's FQDN.
- Modify the Request URI host to use the Cisco Webex hostname.
- Change the P-Asserted-Identity (P-Asserted-ID) host part to the SBC's FQDN.

GUI Path: session router/sip manipulation

ACLI Path: config t→session-router→sip-manipulation

Click Add, and use the following example to configure:

While this can be configured via the GUI, we are using the ACLI output to provide an example config for ease of viewing.

```
sip-manipulation
    name
                           CiscoOutManipulation
    header-rule
        name
                               ChangeContactHost
        header-name
                                   Contact
        action
                               manipulate
                                 ACK, INVITE
        methods
        element-rule
            name
                                   contacthost
                                  uri-host
            type
            action
                                   replace
                                     $TRUNK_GROUP_CONTEXT
            new-value
   header-rule
                               AddContactOptions
        name
                                   Contact
        header-name
        action
                               add
                                 request
        msg-type
        methods
                                 OPTIONS
                                      "<sip:ping@"+$TRUNK_GROUP_CONTEXT+":5061;transport=tls>"
        new-value
    header-rule
                               changeToUser
        name
                                   To
        header-name
                               manipulate
        action
                                 request
        msg-type
        methods
                                 INVITE
        element-rule
            name
                                   changeTOhost
                                  uri-host
            type
            action
                                   replace
            new-value
                                     us01.sipconnect.bcld.webex.com
     header-rule
        name
                               ChangePAI
                                   P-Asserted-Identity
        header-name
                               manipulate
        action
        comparison-type
                                    pattern-rule
                                 INVITE
        methods
        element-rule
                                   ChangePAI
            name
                                  uri-host
            type
            action
                                   replace
            new-value
                                     $TRUNK_GROUP_CONTEXT
```

You'll notice that, in most of these header manipulation rules, the new values are set to \$TRUNK_GROUP_CONTEXT. This variable automatically uses the value configured in the realm parameter we set up earlier—which is the SBC's FQDN. There is an additional, optional SIP manipulation that you can configure if needed. Some SIP trunk providers don't support certain SIP headers and header parameters that Cisco Webex Calling includes in its messages. To ensure smooth interoperability between your PSTN provider and Cisco, you may need to set up and apply the following SIP manipulation rule.

```
sip-manipulation
    name
                            StripCiscoHeaders
    description
                              Remove Cisco Headers to PSTN
    header-rule
        name
                                DeleteLocationInfo
        header-name
                                    X-Cisco-Location-Info
                                delete
        action
        methods
                                  BYE, INVITE, OPTIONS
    header-rule
        name
                                DeleteRecvInfo
                                    Recv-Info
        header-name
        action
                                delete
        methods
                                  BYE, INVITE, OPTIONS
    header-rule
        name
                                DeleteSessionID
        header-name
                                    Session-ID
        action
                                delete
        methods
                                  BYE, INVITE, OPTIONS
  header-rule
        name
                                StripDTG
        header-name
                                    Request-URI
                                manipulate
        action
        comparison-type
                                     case-sensitive
        msg-type
                                 request
        methods
                                  Invite
        match-value
        new-value
          element-rule
            name
                                    stripdtg
                                          dtg
            parameter-name
                                    header-param
            type
            action
                                    delete-element
            match-val-type
                                        any
            comparison-type
                                         case-sensitive
            match-value
            new-value
```

7.6.3 Sip Interface

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

Configure two sip interfaces, one associated with PSTN Realm, and the other for Cisco.

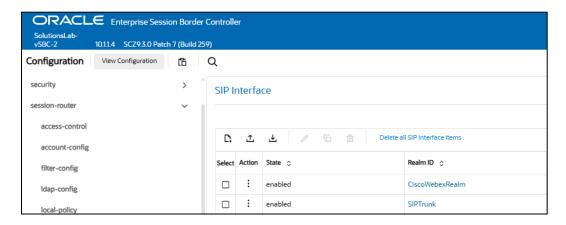
GUI Path: session-router/sip-interface

ACLI Path: config t→session-router→sip-interface

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

Config Parameter	SipTrunk	Cisco Webex
Realm ID	SipTrunk	CiscoWebexRealm
User Agent		Oracle/VM/9.3.0
Initial Inv Trans Expire		10
Out Manipulationid	StripCiscoHeaders	CiscoOutManipulation
Sip Port Config Parmeter	Sip Trunk	Cisco Webex
Address	10.1.2.4	10.1.3.4
Port	5060	5061
Transport protocol	UDP	TLS
TLS profile		TLSWebex
Allow anonymous	Agents-only	Agents-only

Here, you'll assign the TLS profile configured under the <u>Security</u> section of this guide and the sipmanipulations which ensures the SBC works smoothly with other systems. We also configure the User Agent parameter; this should reflect your specific Oracle SBC platform and software version of your SBC. Lastly, we adjust the Initial Invite Trans Expire value <Timer B> to ensure the SBC recurses properly when using Cisco's SRV agent.



• Select OK at the bottom of each when applicable.

7.6.4 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

In this example, we'll configure two session agents. The SRV agent for Cisco Webex Calling, and another for our Sip Trunk.

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

Config parameter	Session Agent 1	Session Agent 2
Hostname	us01.sipconnect.bcld.webex.com	138.3.226.40
IP Address		138.3.226.40
Port	0	5060
Transport method	StaticTLS	StaticTCP
Realm ID	CiscoWebexRealm	SipTrunk
Ping Method	OPTIONS	OPTIONS
Ping Interval	30	30
Ping Response	enabled	enabled
Ping All Addresses	V	



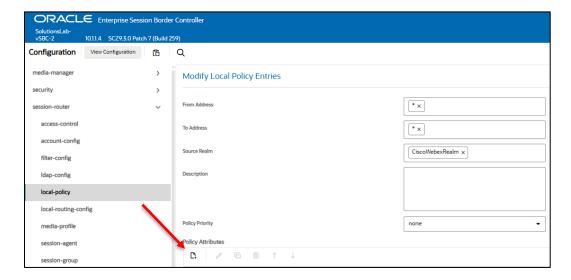
• Click OK at the bottom of each when applicable.

7.7 Routing Configuration

Now that a majority of the signaling, security and media configuration is in place, we can configure the SBC to route calls from one end of the network to the other. The SBC has multiple routing features that can be utilized, but for the purposes of this example configuration, we'll configure local policies to route calls from Cisco Webex Calling to our Sip trunk, and vice versa...

GUI Path: session-router/local-policy

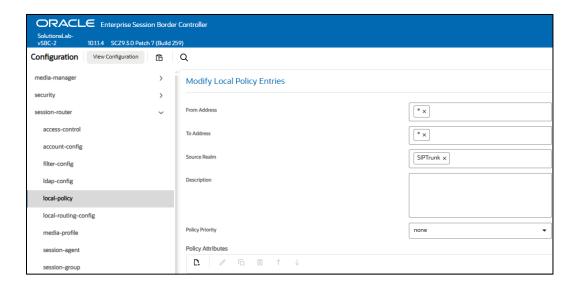
ACLI Path: config t→session-router→local-policy



After entering values for to and from address and source realm, click Add under policy attribute to configure the next hop destination.



Next, we'll setup routing from our SIP Trunk to Cisco Webex Calling:





• Select OK when applicable on each screen

7.8 Sip Access Controls

The Oracle Session Border Controller (SBC) family of products are designed to increase security when deploying Voice over IP (VoIP) or Unified Communications (UC) solutions. Properly configured, Oracle's SBC family helps protect IT assets, safeguard confidential information, and mitigate risks—all while ensuring the high service levels which users expect from the corporate phone system and the public telephone network.

Please note, DDOS values are specific to platform and environment. For more detailed information please refer to the Oracle Communications SBC Security Guide.

https://docs.oracle.com/en/industries/communications/session-border-controller/9.3.0/security/security-guide.pdf

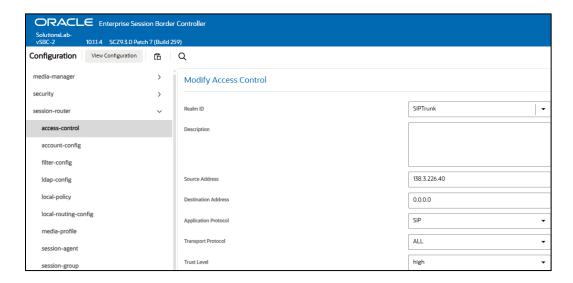
However. While some values are environment specific, there are some basic security parameters that can be implemented on the SBC that will help secure your setup.

- 1. On all public facing interfaces, create Access-Controls to only allow sip traffic from trusted IP's with a trust level of high
- 2. Set the access control trust level on public facing realms to HIGH.

Use this example to create ACL's for all Cisco and PSTN IP's. This example can be followed for any of the public facing interfaces.

GUI Path: session-router/access-control

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



Select OK at the bottom.

This concludes the required configuration of the SBC to act as a local gateway for Cisco Webex Calling.

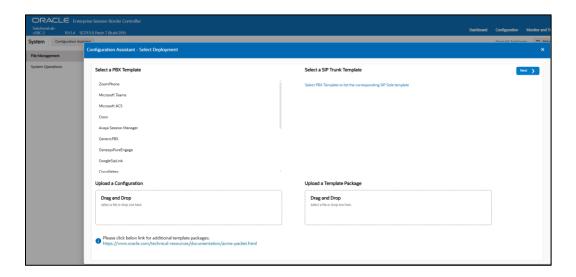
8 Oracle SBC Configuration Assistant

When you first log on to the Oracle SBC, the system requires you to set the configuration parameters necessary for basic operation. To help you set the initial configuration with minimal effort, the SBC provides the Configuration Assistant. The Configuration Assistant, which you can run from the Web GUI or the Acme Command Line Interface (ACLI), asks you questions and uses your answers to set parameters for managing and securing call traffic between the SBC and Cisco Webex Calling. You can use the Configuration Assistant for the initial set up to make to the basic configuration. See "Configuration Assistant Operations" in the Web GUI User Guide and "Run Configuration Assistant" in the ACLI Configuration Guide

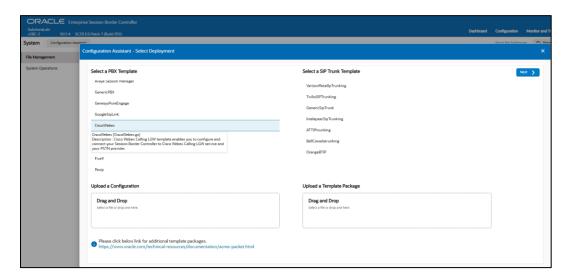
8.1 Cisco Webex Calling Configuration Assistant

The screenshots below are from an Oracle SBC GUI running 930p7.

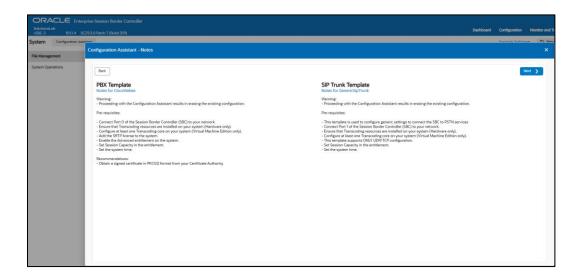
For a new SBC deployment, once access to the GUI is configured, you will see the following when logging in for the first time:



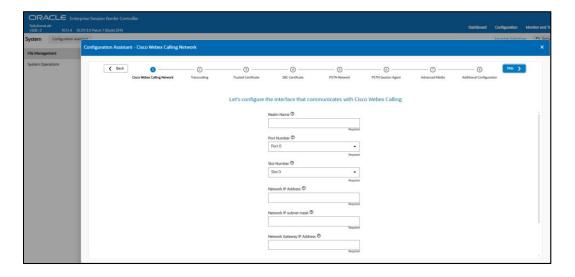
Under PBX template, we'll select Cisco Webex template. This brings up a list of available sip trunk templates.



Select a sip trunk template and click next at the top to access the Notes page. Pay close attention to the information here, as this is a list of warnings, pre-requisites, and recommendations:



Clicking "Next" on the Notes page triggers the configuration assistant to do a system check. This ensures that all the system requirements for the platform and sip trunk you have selected have been met before proceeding to configuration pages. If they have not been met, you will be greeted by a page providing the opportunity to setup entitlements, add license keys, etc...before moving on to the configuration. Once all requirements for your selected templates have been satisfied, you can proceed to the configuration pages.



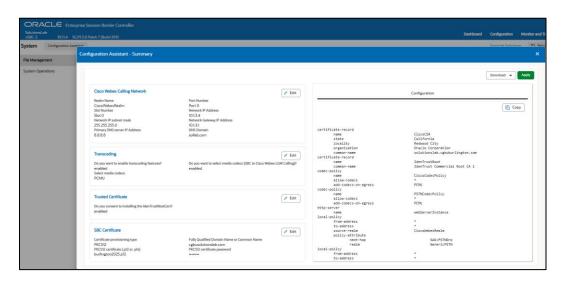
Follow the instructions on each page. Any field that is labeled required must contain an entry.

Once you have entered all information in required fields on all pages, select the option to Review in the top right of the screen:



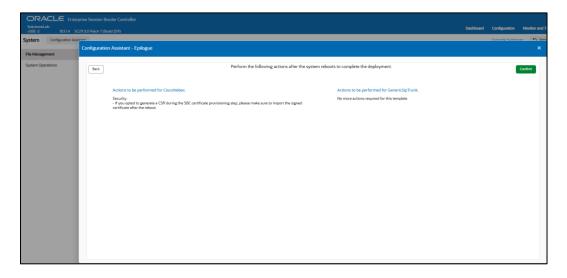
The left side of the review page contains all of the entries added on each page and allows for editing each page individually if necessary.

The right side displays the entire configuration created and when applicable, will also have a CSR tab that contains a certificate that can be signed by a CA authority.

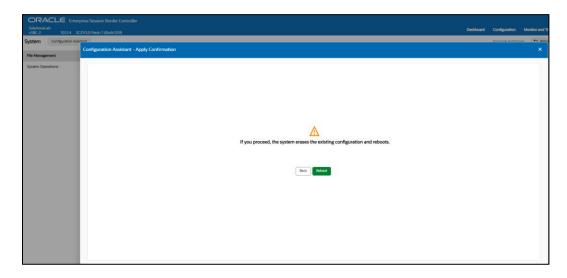


Once all the information has been reviewed and accepted, click Apply.

The SBC now presents the Epilogue.



Confirm, and then select reboot to apply the new configuration to the SBC.



When the SBC comes back up, you can verify connectivity between the SBC and Cisco WebEx Calling UCaaS platform.

9 Verify Connectivity

After you've paired the Oracle SBC with Cisco Webex Calling platform, validate that the SBC can successfully exchange SIP Options with Cisco Webex

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: Signaling/SIP/Method/Method Option



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

10 SBC Scaling

For SBC scaling, Oracle has released the below values recently and these values are derived based on certain conditions and the table is given below with the values of each platform. These values can be taken as reference and these values may differ when the users are using specific conditions like integrating with Cisco Webex with single tenancy, multi-tenancy, etc.

Feature	Virtualized SBC*	AP1100	AP3950	AP4900	AP6350
Form factor	Virtualized	1U System	1U System	1U System	3U System
System Architecture	Data Centre /COTS	Purpose Built	Purpose Built	Purpose Built	Purpose Built
Max. Media Sessions	60,000	360	10,000	40,000	160,000
Max. SRTP Call Legs	19,000	360	10,000	16,000	120,000
Max. SIPREC Sessions	19,000	180	7,500	12,000	40,000
Max. Transcoded Sessions (G711 <-> G729)	3,200**	360	6,500	6,500	58,000
Max. Calls Per Second	2,000	30	100	600	1,700
* VM configuration dependent ** Software transcoding					

11 Oracle SBC integration with Cisco Webex Contact Center

Cisco Webex Contact Center is a Software-as-a-Service (SaaS) offering that provides the significant advantages of cloud delivery. Cisco Webex Contact Center is a cloud-based enterprise Contact Center solution that can help any organization unlock higher levels of agility, flexibility, scalability, innovation, and customer success.

Cisco Webex Contact Center gives you control over every incoming and outgoing interaction from a central point, regardless of organization, technology, or location. The voice processing is performed in the cloud, and we need to route calls in and out of the cloud. It knows which agents, teams, sites, and partners are available at any given time and sends each interaction to the agent with the best identified skills for handling an issue.

The Key advantages of Cisco Webex CC are listed below:

- Native cloud
- Omnichannel
- Skills-based routing
- Agent and expert collaboration etc

For additional information on Cisco Webex Contact Center, please check the below links:

https://help.webex.com/en-us/article/nee1mb6/Get-started-with-Webex-Contact-Center

https://help.webex.com/en-us/article/utqcm7/Webex-Contact-Center-Architecture

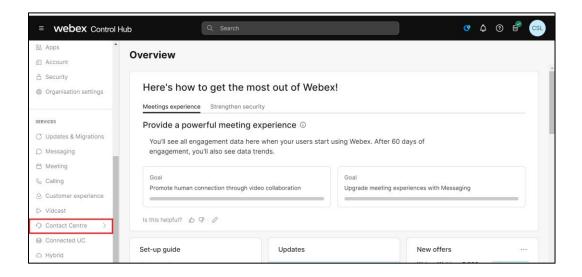
https://help.webex.com/en-us/article/n5595zd/Webex-Contact-Center-Setup-and-Administration-Guide

The Oracle SBC is fully certified to seamlessly integrate with Cisco Webex Contact Center. If your Oracle SBC is already configured for Cisco Webex Calling LGW, no additional SBC configuration is required. To leverage Cisco Webex Contact Center, customers simply need to obtain the necessary licenses. Once activated, the Contact Centre feature set will be accessible through the existing Cisco Webex admin portal.

While Cisco Webex Contact Center supports voice, email, and chat, this document will primarily focus on the voice integration between the Oracle SBC and Cisco Webex Contact Center.

Once Webex CC license is enabled, we will have additional tab for Contact center in Cisco Webex admin portal as shown below. After you click the tab, we will see options to configure Webex CC configuration in the next page. This App note focusses on the basic configuration of Cisco Webex contact center which can be configured on the Cisco Admin portal as shown below. Additional configuration of Cisco Webex Contact Center may be necessary to meet specific customer requirements and ensure optimal operation. For advanced configuration needs, please consult your Cisco representative, as these topics are beyond the scope of this document.

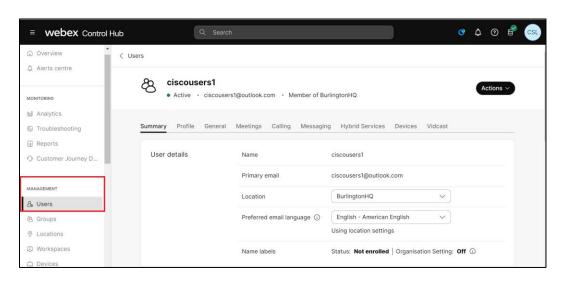
Webex admin page with Contact Center tab enabled:

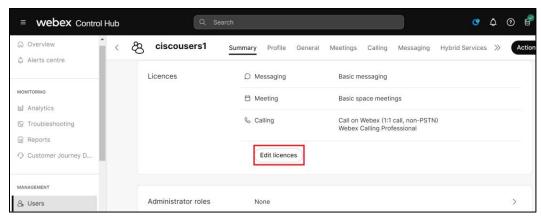


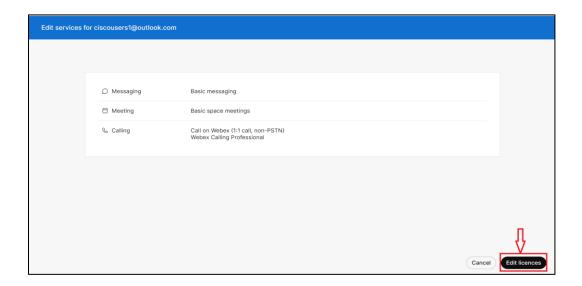
11.1 Assigning Webex Contact Center Licenses to Users

The first step is to assign the Webex Contact Center license to users.

Log in to the **Cisco Webex Control Hub** portal and navigate to **Management > Users**. From there, enable the Webex CC license for the desired users as shown below.

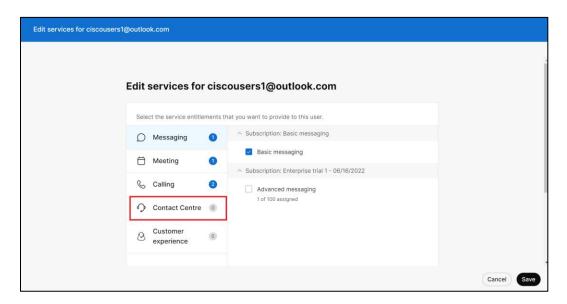


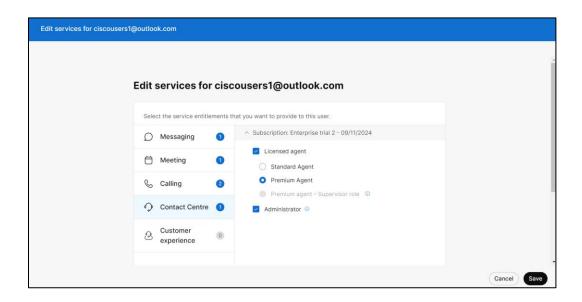




Click on the **Contact Center** tab and choose the appropriate agent type: Standard Agent, Premium Agent, or Premium Agent with Supervisor role. Select the agent type that best fits your requirements. You can also designate an agent as an Admin for Webex Contact Center if needed.

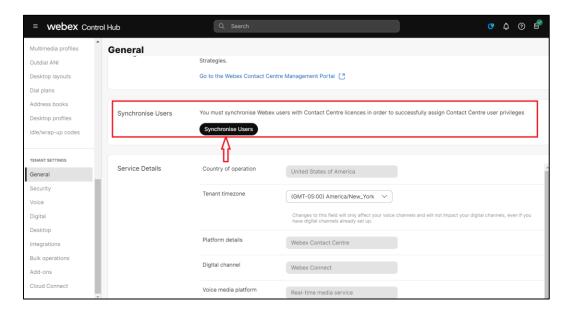
After making your selections, click Save to apply the changes. Repeat this process for any additional users who will serve as agents in Webex Contact Center.





11.2 Synchronize Users with the Webex Contact Center Tenant

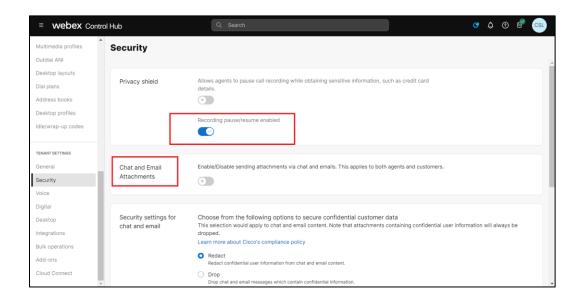
Navigate to Cisco Webex Control Hub portal > Services > Contact Center > Tenant Settings > General, and click on the Synchronize Users tab. This ensures that any recent changes to user accounts are reflected in the Cisco Webex Contact Center page. You can also update the time zone from this page; other settings can typically be left at their default values.



11.3 Configure Settings in the Security Tab.

Go to Cisco Webex Control Hub portal > Services > Contact Center > Tenant Settings > Security and configure the following settings:

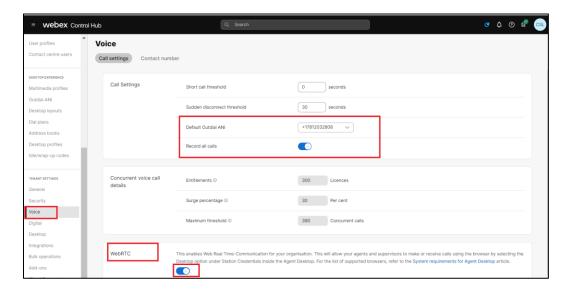
- Under the Privacy Shield tab, enable Recording Pause/Resume.
- Disable Chat, Email, and Attachments features, as this setup is focused solely on the Calling option.



11.4 Configure Settings in the Voice Tab.

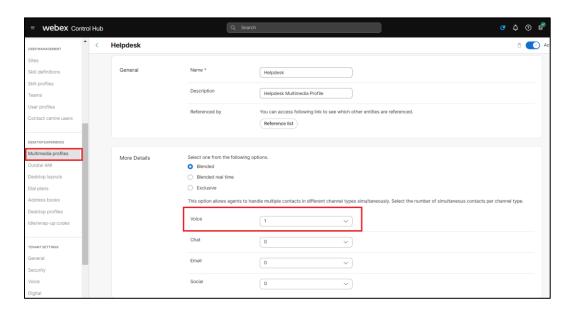
Go to **Cisco Webex Control Hub portal > Services > Contact Center > Tenant Settings > Voice** and enter a DID for the default out-dial ANI. This number will be used for inbound calls to Webex Contact Center from external sources and will route callers to the IVR prompt.

Additionally, enable **WebRTC** to provide agents with access to the Webex Contact Center Agent Desktop option.



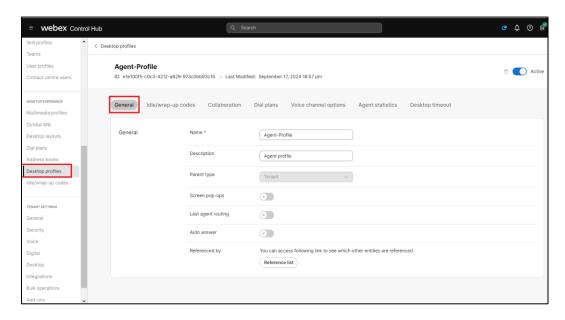
11.5 Configure the Multimedia Profile Tab.

Go to Cisco Webex Control Hub portal > Services > Contact Center > Desktop Experience > Multimedia Profile and create a multimedia profile for the agents. This configuration allows agents to manage multiple contact types across various channels simultaneously. For this setup, set the number of simultaneous calls to 1 and leave the other channel options at zero, as they are not required for our use case.

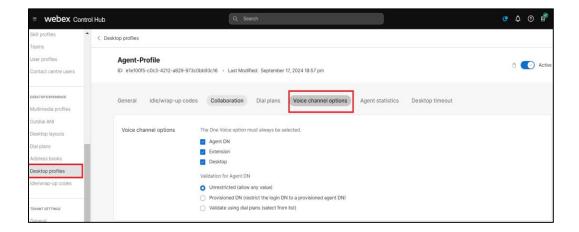


11.6 Configure the Desktop Profile Tab.

Go to Cisco Webex Control Hub portal > Services > Contact Center > Desktop Experience > Desktop Profile and create a desktop profile for the agents as shown below.

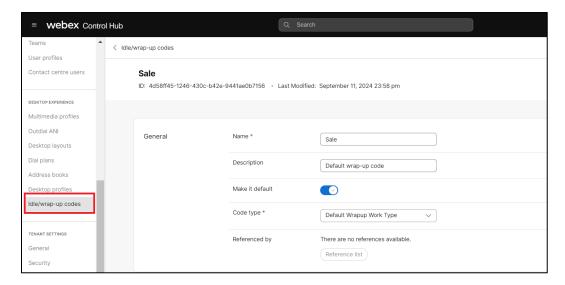


Click on **Voice Channel Options** and select the configurations as indicated below.



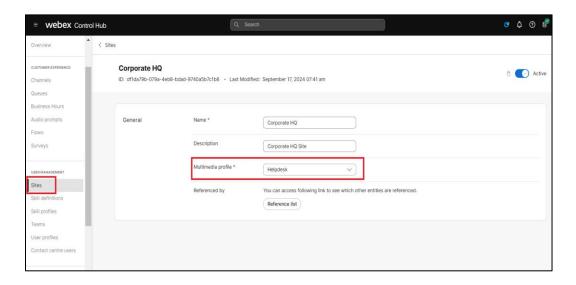
11.7 Configure the Idle/Wrap-up Codes Tab.

Go to Cisco Webex Control Hub portal > Services > Contact Center > Desktop Experience > Idle/Wrap-up Codes and create a new profile for the agents as shown below.



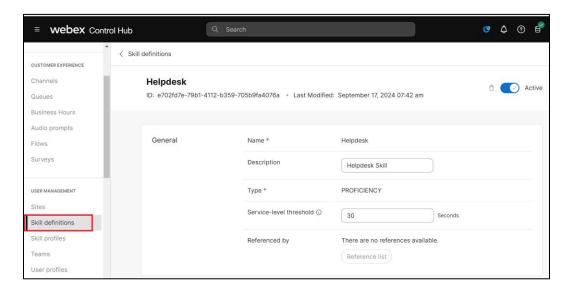
11.8 Configure the Sites Tab.

Go to Cisco Webex Control Hub portal > Services > Contact Center > User Management > Sites and create a new site. Assign the previously created Multimedia Profile to this site as shown below.



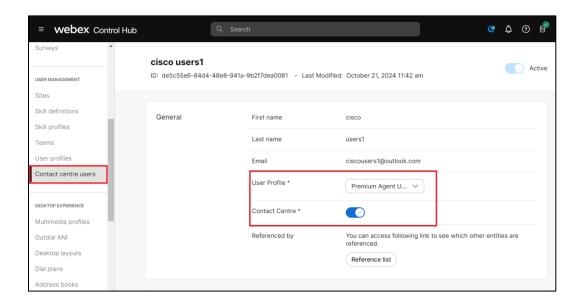
11.9 Configure the Skill Definitions Tab.

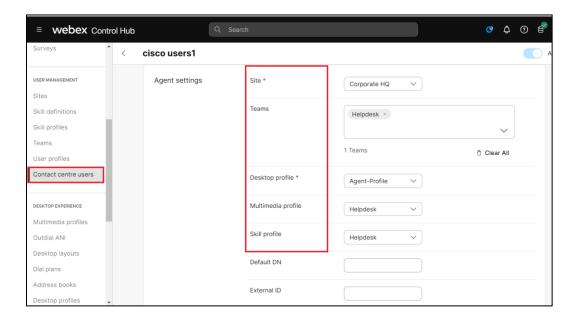
Go to Cisco Webex Control Hub portal > Services > Contact Center > User Management > Skill Definitions and create a new skill profile as shown below.



11.10 Configure the Contact Center Users Tab.

Go to Cisco Webex Control Hub portal > Services > Contact Center > User Management > Contact Center Users. Here you will see the users who have been assigned a Webex CC license and synchronized with Webex Contact Center. You can edit these users and assign the previously created profiles as shown below.

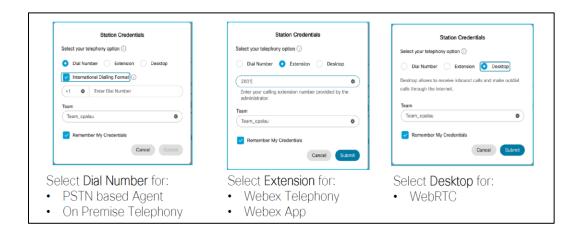


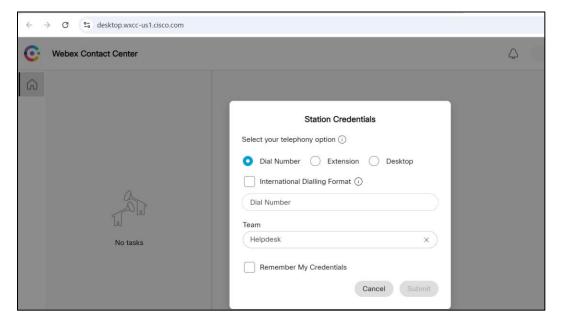


With these steps, the basic configuration of Webex Contact Center is complete. After completing the initial configuration, agents can log in using the link provided below.

https://desktop.wxcc-us1.cisco.com/

After logging in, agents in Cisco Webex Contact Center typically operate in one of three modes, as shown below.





The following are key test cases performed for Webex Contact Center, in addition to the comprehensive tests conducted during the SBC certification with Cisco Webex LGW. We have verified that voice calls are successfully routed to the agent using Oracle SBC, and the test cases listed below have passed for all three agent modes.

Test Case	Description
1	Basic Call w/ 2way Audio
2	Hold/Resume MOH from WxCC
3	Hold/Resume from ENT IP Phone
4	Mute/Unmute from ENT IP Phone
5	Consult Conference to a 2 nd Agent
6	Consult Transfer to a 2 nd Agent
7	Blind Transfer to a 2 nd Agent

12 Appendix A

12.1 Multi-Tenancy

This section describes the requirement for the Oracle SBC to support multi-tenancy. Multi-tenancy essentially means configuring an Oracle SBC for hosting multiple trunks for the same customer or multiple customers.

Cisco Webex Calling has 4 models for multitenancy. They are as follows:

- Model 1: Trunk address is an FQDN, Unique IP address per trunk, same listen port.
- Model 2: Trunk address is an FQDN, shared IP on the Oracle SBC but different listen ports.
- Model 3: Trunk address is an SRV, Unique IP address per trunk, same listen port.
- Model 4: Trunk address is an SRV, shared IP address per trunk, different listen port.

The following configuration applies to all four deployment models described below. We are highlighting this requirement up front to ensure consistent configuration across all scenarios.

12.1.1 Security Configuration

To support multitenancy in a Webex Calling environment, the TLS configuration on the Oracle SBC must include the IP address or FQDN for each trunk as either a Common Name (CN) or Subject Alternative Name (SAN) in the SBC's end-entity certificate. This certificate, which is presented by the Oracle SBC to Cisco to secure the trunk, ensures proper authentication and connectivity for all configured trunks.

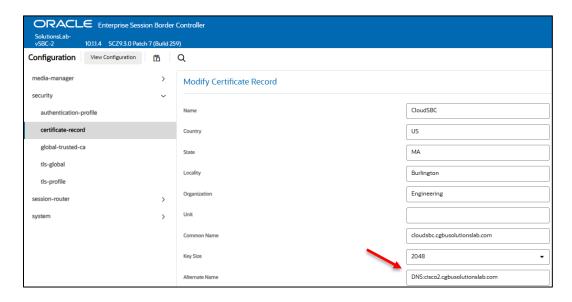
12.1.1.1 Certificate Record

You can create a separate TLS certificate to secure each FQDN or IP address or use a single certificate that includes all necessary FQDNs and IPs in the CN or SAN fields. Each environment may have unique requirements; however, for the purposes of this example, we will illustrate using a single certificate with the CN/SAN method to support multiple trunks.

GUI Path: security/certificate-record

ACLI Path: config t→security→certificate-record

Use the following example to configure a certificate record to support your multitenant environment.



Please notice, when adding an alternate name to the SBC's end entity certificate, you must use the following format for each type of entry:

- IP:<IP Address> Example → IP:10.2.2.2
- DNS:<FQDN> Example → DNS:bar.example.com

Each entry must be comma separated.

Save and active your config.

Next, following the steps outlined previously in the <u>Security chapter</u> of this document that outlines the procedure to generate a CSR and imported your signed certificate.

12.1.1.2 Sip Manipulation

To ensure that the SBC generates SIP messages conforming to Cisco Webex Calling requirements across both Realms in this multitenancy example, create SIP manipulation rules to:

- Change the Contact header's host URI to the SBC's FQDN.
- Add a Contact header to SIP OPTIONS messages that includes the SBC's FQDN and Port.
- Modify the From URI host to the SBC's FQDN
- Change the P-Asserted-Identity (P-Asserted-ID) host part to the SBC's FQDN.
- Modify the To URI Host to the Cisco Webex Calling hostname

GUI Path: session router/sip manipulation

ACLI Path: config t→session-router→sip-manipulation

Click Add, and use the following example to configure:

While this can be configured via the GUI, we are using the ACLI output to provide an example config for ease of viewing.

```
sip-manipulation
    name
                          To_Webex
   header-rule
                               AddContactInOptions
        name
        header-name
                                  Contact
        action
                              add
       msg-type
                                request
       methods
                                OPTION
       new-value
                               "sip:ping@"+$TRUNK_GROUP_CONTEXT+":"+$TRUNK_GROUP+";transport=tls"
    header-rule
       name
                               ChangeContactHost
       header-name
                                  Contact
       action
                              manipulate
                                out-of-dialog
       msg-type
                                INVITE
        methods
        element-rule
           name
                                  contacthost
           type
                                 uri-host
           action
                                  replace
                                    $TRUNK_GROUP_CONTEXT
           new-value
    header-rule
                               ChangePAI
       name
       header-name
                                  P-Asserted-Identity
        action
                              manipulate
                                   pattern-rule
        comparison-type
                                INVITE
        methods
        element-rule
           name
                                  ChangePAI
           type
                                 uri-host
           action
                                  replace
           new-value
                                    $TRUNK_GROUP_CONTEXT
    header-rule
                               ChangeFromIP
       name
                                  FROM
       header-name
        action
                              manipulate
        msg-type
                                out-of-dialog
        methods
                                INVITE
        element-rule
           name
                                  ChangeFrom
           type
                                 uri-host
           action
                                  replace
                                    $TRUNK GROUP CONTEXT
           new-value
    header-rule
                               ChangeToIP
        name
                                  TO
        header-name
        action
                              manipulate
        comparison-type
                                   pattern-rule
                                out-of-dialog
        msg-type
        methods
                                INVITE
        element-rule
           name
                                  ChangeTo
           type
                                 uri-host
           action
                                  replace
           new-value
                                    $MANIP_STRING
```

12.1.2 Additional Configuration Elements for Multitenancy

The following example assumes the foundational configuration steps outlined earlier in this document and builds upon them to demonstrate multitenancy-specific elements.

12.1.2.1 Realm Config

Nested Realm for Webex

Nested Realms is an Oracle SBC feature that supports hierarchical realm groups, allowing one or more realms to be nested within a higher-order (parent) realm. This structure enables the Oracle SBC to logically separate each tenant in a multitenancy environment.

In this example, we will create an additional realm for interfacing with Cisco Webex. The configuration will consist of a parent realm and a child realm, with each realm containing the FQDN for the respective tenant serviced by the Oracle SBC.

GUI Path: media-manger/realm-config

ACLI Path: config t→media-manger→realm-config

• Click Add and use the following table as a configuration example for the two realms used in this configuration example.

Config Parameter	Parent Realm	Child Realm	
Identifier	CiscoWebexRealm	CiscoCust1	
Network Interface	S1p0:0	S1p0:0	
Mm in realm	V	✓	
Media Sec policy	CiscoWebexSecurity	CiscoWebexSecurity	
Codec policy	CiscoCodec	CiscoCodec	
Trunk Context	Cloudsbc.cgbusolutionslab.com	Cisco2.cgbusolutionslab.com	
Access-control-trust-level	HIGH	HIGH	
Parent Realm		CiscoWebexRealm	

Please also note the "trunk context" parameter. The value you set here should be the SBC's FQDN for each interface, which was registered earlier in the Webex Control Hub. Previously in this chapter, this value is used to adjust SIP header syntax to match Cisco Webex Calling requirements.



12.1.2.2 Sip Interface

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

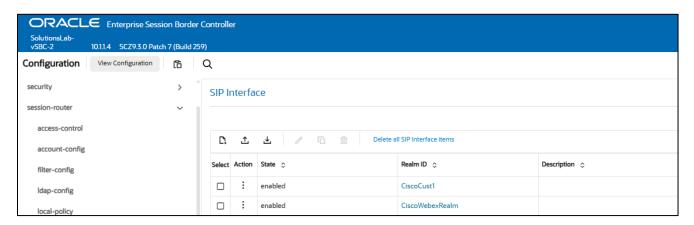
Configure two sip interfaces facing Cisco Webex, one for each tenant the SBC is servicing. Please pay close attention to the differences between Models 1 and 3, and models 2 and 4.

GUI Path: session-router/sip-interface

ACLI Path: config t→session-router→sip-interface

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

Config Parameter	CiscoWebexRealm	CiscoCust1	
Realm ID	CiscoWebexRealm	CiscoCust1	
User Agent	Oracle/VM/9.3.0	Oracle/VM/9.3.0	
Initial Inv Trans Expire	10	10	
Out Manipulationid	To_Webex	To_Webex	
Sip Port Config Parmeter	Models 1 and 3		
Address	10.1.2.4	10.1.2.5	
Port	5061	5061	
Transport protocol	TLS	TLS	
TLS profile	TLSWebex	TLSWebex	
Allow anonymous	Agents-only	Agents-only	
Sip Port Config Parameter	Models 2 and 4		
Address	10.1.2.4	10.1.2.4	
Port	5061	5062	
Transport protocol	TLS	TLS	
TLS profile	TLSWebex	TLSWebex	
Allow anonymous	Agents-only	Agents-only	



12.1.2.3 Session Agent

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. In Cisco multitenancy environments, we set up two SRV session agents that interface with Cisco Webex Calling.

GUI Path: session-router/session-agent

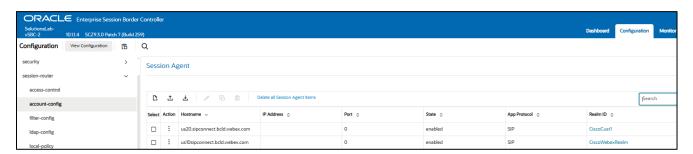
ACLI Path: config t→session-router→session-agent

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

Config parameter	Session Agent 1	Session Agent 2		
Hostname	us10.sipconnect.bcld.webex.com	us20.sipconnect.bcld.webex.com		
Port	0	0		
Transport method	StaticTLS	StaticTLS		
Realm ID	CiscoWebexRealm	CiscoCust1		
Ping Method	OPTIONS	OPTIONS		
Ping Interval	30	30		
Ping Response	enabled	enabled		
Ping All Addresses	\checkmark	\checkmark		
Manipulation-string	us10.sipconnect.bcld.webex.com	us20.sipconnect.bcld.webex.com		
Models 1 and 3				
Trunk-group	5061	5061		
Models 2 and 4				
Trunk-group	5061	5062		

Notice, for session agent configuration, the only difference between the models lies in the trunk group parameter, which specifies the SIP port for each interface according to the model.

Note: When deployed to support Cisco Webex multitenancy, the SBC has a limitation where it will only send SIP OPTIONS messages from a single realm to a global session agent. This can impact environments requiring OPTIONS from multiple realms to monitor the health of each trunk. Please see SIP OPTIONS ping from multiple Realms to global session agents under Known Issues and Limitations chapter for more information.

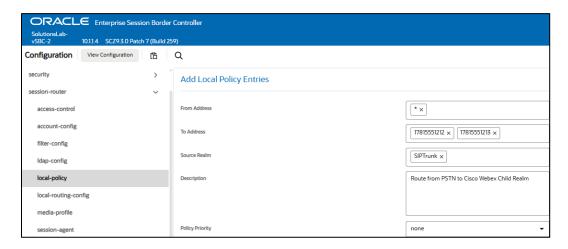


12.1.2.4 Routing Configuration

Next, we will create a new routing policy to direct traffic from the PSTN to the appropriate child tenant. For simplicity, the local policy will use the TO address field for DID separation and route calls to the child realm configured in the previous step.

GUI Path: session-router/local-policy

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



After entering values for to and from address and source realm, click Add under policy attribute to configure the next hop destination.



• Select OK when applicable on each screen.

Save and Active your config.

This completes the basic configuration necessary to support Cisco Webex Calling multitenancy across all four models specified by Cisco.

13 Appendix B

13.1 Oracle SBC deployed behind NAT

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 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 IP as configured on both the SIP Interface and Steering Pool
- 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.

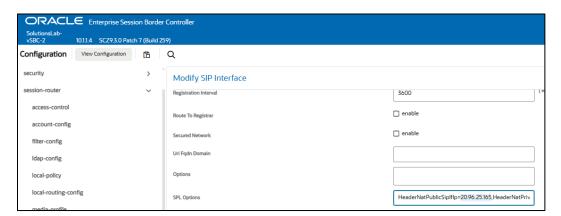
The SPL is applied to the Webex side SIP interface.

GUI Path: session-router/sip-interface

ACLI Path: config t→session-router→sip-interface

Header Nat Public Sip If Ip=20.96.25.165, Header Nat Private Sip If Ip=10.1.3.4

- HeaderNatPublicSipIfIp is the public interface ip
- HeaderNatPrivateSipIfIp is the private ip.



You will need to apply these options to every sip interface on the SBC that is connected through a NAT.

14 Known Issues and Limitations

14.1 SIP OPTIONS ping from multiple Realms to global session agents.

When deployed to support Cisco Webex multitenancy, the SBC has a limitation where it will only send SIP OPTIONS messages from a single realm to a global session agent. This can impact environments requiring OPTIONS from multiple realms to monitor the health of each trunk.

For customers running SBC versions prior to 10.0, workarounds are available, please refer to OCSBC-SIP
OCSBC-SIP
OPTIONS monitoring in Multi-Tenant environments
BCP
GUIDE guide
<a hr

14.2 Video Call Issues When Calls Originate from Cisco CUCM Towards Cisco Webex

Some customers experienced issues establishing video calls between on-premises CUCM and Webex Calling when using Oracle SBC as the Local Gateway (LGW). This problem is related to how video negotiation is handled between the systems.

The issue is resolved by removing the following headers from the SDP video attribute in messages sent from Cisco CUCM to Cisco Webex:

- a=rtcp-fb:* nack pli
- a=rtcp-fb:* ccm fir
- a=rtcp-fb:* ccm tmmbr

We have created the SIP manipulation mime-rule below to remove these headers. This SIP manipulation must be applied to traffic directed towards the Cisco Webex. This would be added to the CiscoOutManipulation configured early in this document.

```
mime-rule
 name
                         Changealine
 msg-type
                           any
                           Invite
 methods
                         manipulate
 action
 comparison-type
                              pattern-rule
 match-value
 new-value
 sdp-media-rule
                              deleteattributes
     name
     media-type
                                video
     action
                             manipulate
     comparison-type
                                  pattern-rule
     match-value
     new-value
     sdp-line-rule
          name
                                  deletertcp
          type
          action
                                 delete
          comparison-type
                                       pattern-rule
                                     (rtcp)(.*)
          match-value
          new-value
```

14.3 ICE Candidate Priority Attribute Interoperability with Cisco MPP Handphones

When using Cisco MPP hardphones with Webex media optimization, the Oracle SBC may assign a priority value on the candidate line attributes in the SDP, leading to interoperability or ICE negotiation issues with Cisco Webex Calling. This issue is specific to deployments with Cisco MPP hardphones. It can be resolved by applying the SIP manipulation below to adjust the priority value of each host candidate. This modification should be added to the CiscoOutManipulation rule configured earlier in this document. The issue is permanently fixed in SBC software release 10.0p4.

```
mime-sdp-rule
                            modpriorityhost
    name
    methods
                             Invite
                           manipulate
    action
    sdp-media-rule
        name
                                modecandidatepriority
                                  audio
        media-type
        action
                                manipulate
        comparison-type
                                     pattern-rule
        sdp-line-rule
            name
                                    modrtp
            type
                                   а
            action
                                    replace
                                         pattern-rule
            comparison-type
                                       (candidate.*)(659136)(.*)
            match-value
            new-value
                                      $1+2130706431+$3
        sdp-line-rule
            name
                                    modrtcp
            type
                                   а
            action
                                    replace
            comparison-type
                                         pattern-rule
                                      (candidate.*)(659134)(.*)
            match-value
                                      $1+2130706430+$3
            new-value
```

14.4 One-Way Audio After Call Transfer with Media Optimization

When using media optimization, inbound calls to Cisco Webex Calling may experience one-way audio following a call transfer. The only known workaround at this time is for the transferee to place the call on hold and then resume it, which restores two-way audio. A permanent resolution for this issue is currently under investigation by both Oracle and Cisco development teams.

15 ACLI Running Configuration

Below is a complete output of the running configuration used to create this application note. This output includes all the configuration elements used in our examples, including some of the optional configuration features outlined throughout this document. Be aware this configuration does not include multitenancy configuration and not all parameters may be applicable to every Oracle SBC setup, so please take this into consideration if planning to copy and paste this output into your SBC.

```
access-control
    realm-id
                            SipTrunk
    source-address
                               138.3.226.40
    application-protocol
                                 SIP
    trust-level
                            high
certificate-record
                            CloudSBC
    name
                                 cloudsbc.cgbusolutionslab.com
    common-name
certificate-record
                            GoDaddyCrossCert
    name
    unit
                          www.godaddy.com
    common-name
                                 GoDaddy G1 to G2 Cross Certificate
certificate-record
    name
                            GoDaddyIntermediate
    unit
                          www.godaddy.com
                                 GoDaddy Secure Server Certificate - G2
    common-name
certificate-record
    name
                            GoDaddyRootCA
    unit
                          www.godaddy.com
                                 GoDaddy Class 2 Certification Authority Root Certificate
    common-name
certificate-record
                            WebexRootCA
    name
                                 IdenTrust Root CA certificate
    common-name
codec-policy
    name
                            CiscoCodec
    allow-codecs
                              PCMU Telephone-Event
                                   PCMU Telephone-Event
    add-codecs-on-egress
codec-policy
                            PSTN
    name
    allow-codecs
    add-codecs-on-egress
                                   PCMU
http-server
                            webserver
    name
ice-profile
                            webexice
    name
    stun-conn-timeout
                                 0
    stun-keep-alive-interval
                                  0
    stun-rate-limit
                              0
    rtcp-stun
                            enabled
local-policy
    from-address
    to-address
```

```
CiscoWebexRealm
    source-realm
    policy-attribute
                                 138.3.226.40
        next-hop
        realm
                               SipTrunk
        action
                               replace-uri
local-policy
    from-address
    to-address
   source-realm
                               SipTrunk
    policy-attribute
        next-hop
                                 us01.sipconnect.bcld.webex.com
                               CiscoWebexRealm
        realm
        action
                               replace-uri
media-manager
media-sec-policy
    name
                            CiscoWebexSecurity
    inbound
        profile
                               CiscoSRTP
        mode
                                srtp
        protocol
                                sdes
    outbound
        profile
                               CiscoSRTP
        mode
                                srtp
        protocol
                                sdes
media-sec-policy
    name
                            PSTN
network-interface
                            s0p0
   name
                             10.1.2.4
    ip-address
    netmask
                             255.255.255.0
                             10.1.2.1
    gateway
network-interface
                            s1p0
    name
    ip-address
                             10.1.3.4
    netmask
                             255.255.255.0
    gateway
                             10.1.3.1
    dns-ip-primary
                               9.9.9.9
    dns-ip-backup1
                                8.8.8.8
    dns-ip-backup2
                                8.8.4.4
    dns-domain
                              cgbusolutionslab.com
phy-interface
    name
                            s0p0
                               Media
    operation-type
phy-interface
    name
                            s1p0
                               Media
    operation-type
                          1
    slot
```

```
realm-config
    identifier
                            CiscoWebexRealm
    network-interfaces
                                 s1p0:0.4
    mm-in-realm
                               enabled
    media-sec-policy
                                CiscoWebexSecurity
    ice-profile
                            webexice
    access-control-trust-level
                                   high
    trunk-context
                              cloudsbc.cgbusolutionslab.com
    codec-policy
                              CiscoCodec
    rtcp-policy
                             CiscoRTCP
realm-config
   identifier
                            SipTrunk
    network-interfaces
                                 s0p0:0.4
   mm-in-realm
                               enabled
    media-sec-policy
                                PSTN
    access-control-trust-level
                                   high
rtcp-policy
    name
                            CiscoRTCP
                              all-calls
    rtcp-generate
sdes-profile
                            CiscoSRTP
    name
                        AES_CM_128_HMAC_SHA1_80
    crypto-list
                        AES_256_CM_HMAC_SHA1_80
                        AES CM 128_HMAC_SHA1_32
                        AEAD_AES_256_GCM
    srtp-rekey-on-re-invite
                                  enabled
session-agent
                              138.3.226.40
    hostname
                             138.3.226.40
    ip-address
    transport-method
                                 StaticTCP
    realm-id
                            SipTrunk
    ping-interval
                             30
                               enabled
    ping-response
    reuse-connections
                                 TCP
session-agent
    hostname
                              us01.sipconnect.bcld.webex.com
   port
                          0
    transport-method
                                 StaticTLS
    realm-id
                            CiscoWebexRealm
                               OPTIONS
    ping-method
    ping-interval
                             30
    ping-all-addresses
                                enabled
    ping-response
                               enabled
sip-config
    home-realm-id
                                CiscoWebexRealm
   registrar-domain
    registrar-host
                              5060
    registrar-port
    options
                            max-udp-length=0
```

```
sip-interface
    realm-id
                            CiscoWebexRealm
    sip-port
        address
                                10.1.3.4
        port
                              5061
        transport-protocol
                                    TLS
        tls-profile
                                TLSWebex
        allow-anonymous
                                     agents-only
    out-manipulationid
                                 CiscoOutManipulation
                             Oracle/VM/9.3.0
    user-agent
sip-interface
    realm-id
                            SipTrunk
    sip-port
        address
                                10.1.2.4
        transport-protocol
                                    TCP
        allow-anonymous
                                     agents-only
    sip-port
        address
                                10.1.2.4
        allow-anonymous
                                     agents-only
    options
                            reuse-connections=latest
    out-manipulationid
                                 StripCiscoHeaders
sip-manipulation
    name
                           CiscoOutManipulation
    header-rule
        name
                                ChangeContactHost
        header-name
                                   Contact
        action
                               manipulate
                                 ACK,INVITE
        methods
        element-rule
            name
                                    contacthost
                                   uri-host
            type
            action
                                   replace
            new-value
                                     $TRUNK_GROUP_CONTEXT
    header-rule
        name
                               AddContactOptions
        header-name
                                   Contact
        action
                               add
        msg-type
                                 request
        methods
                                 OPTIONS
                                  "<sip:ping@"+$TRUNK_GROUP_CONTEXT+":5061;transport=tls>"
        new-value
    header-rule
        name
                               changeToUser
        header-name
                                   То
        action
                               manipulate
                                 request
        msg-type
                                 INVITE
        methods
```

```
element-rule
         name
                                 changeTOhost
                                uri-host
         type
                                 replace
         action
         new-value
                                   us01.sipconnect.bcld.webex.com
  header-rule
                             ChangePAI
     name
     header-name
                                 P-Asserted-Identity
     action
                             manipulate
     comparison-type
                                  pattern-rule
     methods
                               INVITE
     element-rule
         name
                                 ChangePAI
                                uri-host
         type
         action
                                 replace
         new-value
                                   $TRUNK_GROUP_CONTEXT
 mime-sdp-rule
     name
                             modpriorityhost
                              Invite
     methods
     action
                            manipulate
     sdp-media-rule
                                 modecandidatepriority
         name
         media-type
                                   audio
                                 manipulate
         action
         comparison-type
                                      pattern-rule
         sdp-line-rule
             name
                                     modrtp
             type
             action
                                     replace
             comparison-type
                                          pattern-rule
             match-value
                                        (candidate.*)(659136)(.*)
             new-value
                                       $1+2130706431+$3
         sdp-line-rule
             name
                                     modrtcp
             type
                                    a
             action
                                     replace
             comparison-type
                                          pattern-rule
             match-value
                                        (candidate.*)(659134)(.*)
                                       $1+2130706430+$3
             new-value
mime-sdp-rule
     name
                             Changealine
                              any
     msg-type
     methods
                              Invite
     action
                             manipulate
     comparison-type
                                  pattern-rule
     match-value
     new-value
```

```
sdp-media-rule
            name
                                     deleteattributes
            media-type
                                       video
            action
                                    manipulate
            comparison-type
                                          pattern-rule
            match-value
            new-value
            sdp-line-rule
                name
                                         deletertcp
                type
                action
                                        delete
                                              pattern-rule
                comparison-type
                match-value
                                           (rtcp)(.*)
                new-value
sip-manipulation
                            StripCiscoHeaders
    name
    description
                              Remove Cisco Headers to PSTN
    header-rule
                                DeleteLocationInfo
        name
        header-name
                                    X-Cisco-Location-Info
        action
                                delete
        methods
                                  BYE, INVITE, OPTIONS
    header-rule
        name
                                DeleteRecvInfo
        header-name
                                    Recv-Info
        action
                                delete
                                  BYE, INVITE, OPTIONS
        methods
    header-rule
        name
                                DeleteSessionID
        header-name
                                    Session-ID
        action
                                delete
        methods
                                  BYE, INVITE, OPTIONS
  header-rule
        name
                                StripDTG
        header-name
                                    Request-URI
        action
                                manipulate
        comparison-type
                                     case-sensitive
        msg-type
                                  request
        methods
                                  Invite
        match-value
        new-value
          element-rule
            name
                                     stripdtg
            parameter-name
                                          dtg
            type
                                    header-param
            action
                                    delete-element
            match-val-type
                                        any
            comparison-type
                                          case-sensitive
```

sip-monitoring
match-any-filter
enabled
monitoring-filters
*

steering-pool

ip-address10.1.2.4start-port10000end-port20000realm-idSipTrunk

steering-pool

ip-address 10.1.3.4 start-port 10000 end-port 10999

realm-id CiscoWebexRealm

system-config

transcoding-cores 1

tls-profile

name TLSWebex end-entity-certificate CloudSBC

trusted-ca-certificates GoDaddyRootCA

WebexRootCA

GoDaddyIntermediate

tls-version tlsv12



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Integrated Cloud Applications & Platform Services

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