

Oracle SBC integration with Cisco Webex Calling and Webex Contact Center (CC) as 3<sup>rd</sup> party Local Gateway (LGW).

**Technical Application Note** 



## Disclaimer

The following is intended to outline our general product direction. It is intended for information purposes only, and may not be incorporated into any contract. It is not a commitment to deliver any material, code, or functionality, and should not be relied upon in making purchasing decisions. The development, release, and timing of any features or functionality described for Oracle's products remains at the sole discretion of Oracle.

# **Revision History**

Version	<b>Description of Changes</b>	Date Revision Completed
1.0	Oracle SBC integration with Cisco Webex Calling as 3rd party Local Gateway (LGW)	30 <sup>th</sup> October 2022
1.1	Added Appendix B section to the document for the new feature which supports Cisco DTMF with OPUS codec	05 <sup>th</sup> January 2023
1.2	Added ACLI config of the SBC. Also added screenshots to import certs to the SBC.  Added section w.r.t sip options ping in multitenancy setup  ChangeContactHost sip manipulation changed for ACK method.  SBC version changed to 9.x in the whole document to keep it uniform.  Config added for Media optimization feature (is supported from 9.3,0 and later)	08 <sup>th</sup> November 2024

Added crypto attributes for SRTP which includes GCM ciphers.	
Added support for Cisco Webex Contact Center along with Cisco Webex Calling part.	

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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 3<sup>rd</sup> Party Local Gateway.

#### 2. 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.** 

Please find the related documentation links below:

#### 2.1. 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. This 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 & 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 3**<sup>rd</sup> 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.

#### 3. Introduction

#### 3.1. Audience

This is a technical document intended for telecommunications engineers with the purpose of configuring Cisco Webex Calling with 3<sup>rd</sup> 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.

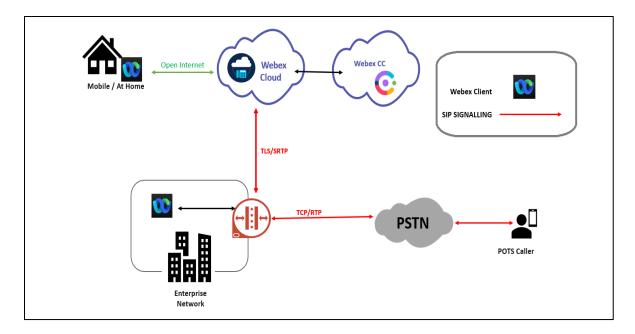
#### 3.2. Requirements

- Fully functioning Cisco Webex Control Hub (Provisioned Webex Control Hub with necessary Webex Calling licenses/Subscription and also 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

#### 3.3. Architecture



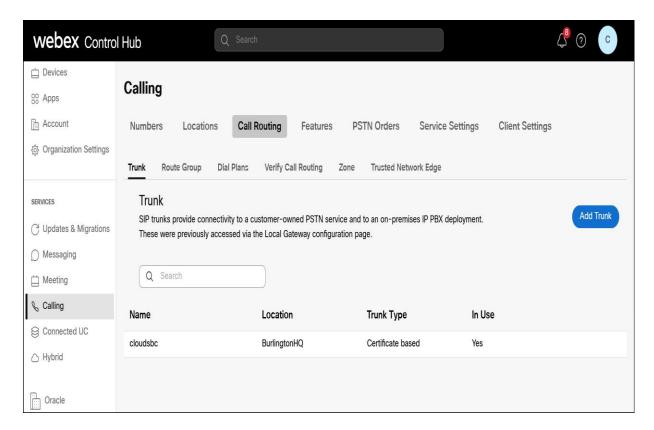
The configuration, validation and troubleshooting are the focuses of this document and will be described in three phases:

- Phase 1 Configuring the Cisco Webex calling with 3<sup>rd</sup> party Local Gateway (LGW) feature for Oracle SBC.
- Phase 2 Configuring the Oracle SBC.
- Phase 3 Configuring the Cisco Webex Contact Center.

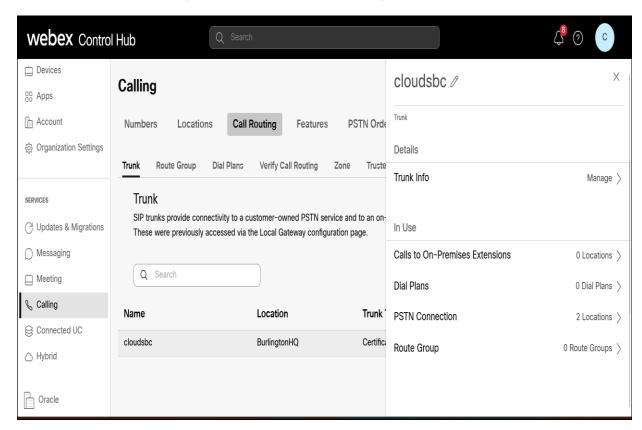
## 4. Cisco Webex Side Configuration

The configuration of Cisco Webex side 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 for Webex</u> 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.

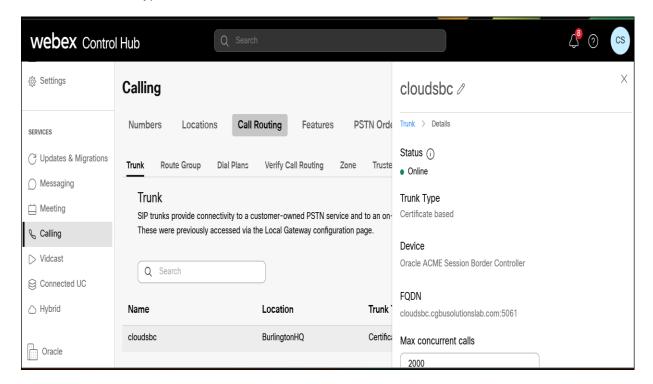
Please login to **Webex Control Hub ----- Calling ------ Call routing** and you can check the created Trunk which actually connects to the SBC.



Click on the trunk name to get more details about the configured trunk.

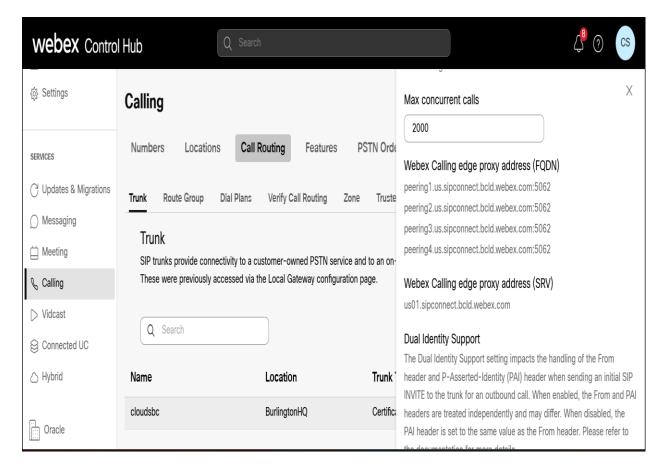


Click on the Trunk Info to get the details which is the parameters used to connect to the Oracle SBC. The trunk status shows Online which means the Webex is able to establish a connection with Oracle SBC and the trunk type are defined as Certificate based and the FQDN is also defined.



In the below screen, you can check the destination (Webex Edge proxy) Address which will be used as Session Agent in the Oracle SBC to connect to Cisco Webex side. As Cisco recommends using SRV based Webex Calling edge address, we will be using that as Session Agent in oracle SBC (This requirement is for now and may be changed in future). You can also check the created directory numbers and the locations in the same Calling page of Webex Control Hub.

Please note that Webex Calling Proxy Addresses given below is 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.



With this, Cisco side configuration is complete.

## 5. Configuring the SBC

This chapter provides step-by-step guidance on how to configure Oracle SBC for Cisco Webex Calling and PSTN SIP Trunk. In this SBC config, Cisco Webex Calling side is secure (TLS/SRTP) and PSTN Side is unsecure (UDP or TCP/RTP).

#### 5.1. 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 4600
- AP 6300
- AP 6350
- AP 3950
- AP 4900
- VME
- Oracle SBC on Public Cloud

# 6. New SBC configuration

If the customer is looking to setup a new SBC from scratch, please follow the section below.

## 6.1. Establishing a serial connection to the SBC

Connect one end of a straight-through Ethernet cable to the front console port (which is active by default) on the SBC and the other end to console adapter that ships with the SBC, connect the console adapter (a DB-9 adapter) to the DB-9 port on a workstation, running a terminal emulator application such as Putty. Start the terminal emulation application using the following settings:

- Baud Rate=115200
- Data Bits=8
- Parity=None
- Stop Bits=1
- Flow Control=None

Power on the SBC and confirm that you see the following output from the boot-up sequence

```
Starting tLemd...
Starting tServiceHealth...
Starting tCollect...
Starting tAtcpd...
Starting tAsctpd...
Starting tMbcd...
Starting tCommMonitord...
Starting tFped...
tarting tAlgd...
Starting tRadd...
Starting tEbmd...
Starting tSipd...
Starting tH323d...
Starting tbfdd...
Starting tIPTd...
Starting tSecured...
Starting tAuthd...
tarting tCertd...
Starting tIked...
Starting tTscfd...
Starting tFcgid...
Starting tauditd...
Starting tauditpusher...
Starting tSnmpd...
Starting tIFMIBd...
Start platform alarm...
Starting display manager...
Initializing /opt/ Cleaner
Starting tLogCleaner task
Bringing up shell...
Starting acliMgr...
password secure mode is enabled
Admin Security is disabled
assword:
```

Enter the default password to log in to the SBC. Note that the default SBC password is "acme" and the default super user password is "packet".

Both passwords have to be changed according to the rules shown below.

```
Password:

% Only alphabetic (upper or lower case), numeric and punctuation
% characters are allowed in the password.
% Password must be 8 - 64 characters,
% and have 3 of the 4 following character classes:
% - lower case alpha
% - upper case alpha
% - numerals
% - punctuation
%
Enter New Password:
Confirm New Password:
Password is acceptable.
```

Now set the management IP of the SBC by setting the IP address in bootparam.

To access bootparam. Go to Configure terminal->bootparam.

```
SolutionsLab-vSBC-2(configure)# bootparam
 .' = clear field; '-' = go to previous field; q = quit
Boot File
                        : /boot/nnSCZ900p4.bz
IP Address
VLAN
Netmask
Gateway
Pv6 Address
IPv6 Gateway
Host IP
TP username
                     : vxftp
TTP password
Flags
                      : 0x00000040
                   : 0x00000040
: SolutionsLab-vSBC-2
: COM1
: 115200
rarget Name
Console Device
Target Name
Console Baudrate
Other
NOTE: These changed parameters will not go into effect until reboot.
Also, be aware that some boot parameters may also be changed through
PHY and Network Interface Configurations.
        ERROR
                : space in /boot
                                       (Percent Free: 18)
SolutionsLab-vSBC-2(configure)#
SolutionsLab-vSBC-2(configure)#
```

Note: There is no management IP configured by default.

To configure product type, type in setup product in the terminal

Set product type to Enterprise Session Border Controller as shown below.

```
SolutionsLab-vSBC-2# setup product

WARNING:
Alteration of product alone or in conjunction with entitlement changes will not be complete until system reboot

Last Modified 2022-10-03 07:21:29

1: Product : Enterprise Session Border Controller

Enter 1 to modify, d' to display, 's' to save, 'q' to exit. [s]:
```

Enable the features for the ESBC using the setup entitlements command as shown

Save the changes and reboot the SBC. (The below screen is just an example and not actual config)

```
Entitlements for Enterprise Session Border Controller
Last Modified: 2022-02-23 18:18:18
 1 : Session Capacity
                                                 : enabled
      STIR/SHAKEN Client
 4 : Admin Security
  : Data Integrity (FIPS 140-2)
: IPSec Trunking Sessions
 7 : MSRP B2BUA Sessions
 8 : SRTP Sessions
 9 : Transcode Codec AMR
 10: Transcode Codec AMR Capacity
 11: Transcode Codec AMRWB
 12: Transcode Codec AMRWB Capacity
 13: Transcode Codec EVRC
 14: Transcode Codec EVRC Capacity
 16: Transcode Codec EVRCB Capacity
 17: Transcode Codec EVS
 18: Transcode Codec EVS Capacity
 19: Transcode Codec OPUS
                                                : enabled
 20: Transcode Codec OPUS Capacity
                                               : enabled
 21: Transcode Codec SILK
 22: Transcode Codec SILK Capacity
                                                : 2000
Enter 1 - 22 to modify, d' to display, 's' to save, 'q' to exit. [s]: 1
  Session Capacity (0-10000)
Enter 1 - 22 to modify, d' to display, 's' to save, 'q' to exit. [s]: 10
  Transcode Codec AMR Capacity (0-10000)
Enter 1 - 22 to modify, d' to display, 's' to save, 'q' to exit. [s]: 14
  Transcode Codec EVRC Capacity (0-10000)
Enter 1 - 22 to modify, d' to display, 's' to save, 'q' to exit. [s]:
```

The SBC comes up after reboot and is now ready for configuration.

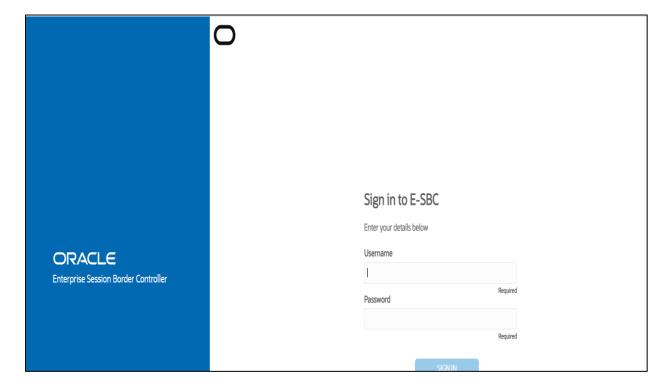
Go to configure terminal->system->http-server-config. Enable the http-server-config to access the SBC using Web GUI. Save and activate the config.

```
SolutionsLab-vSBC-2(http-server)# show
 ttp-server
                                                               webserver
          ip-address
                                                               enabled
          http-port
          HTTP-strict-transport-security-policy
                                                               disabled
                                                               disabled
          https-state
          https-port
          http-interface-list
          http-file-upload-size
          tls-profile
          auth-profile
          last-modified-by
last-modified-date
SolutionsLab-vSBC-2(http-server)#
SolutionsLab-vSBC-2(http-server)#
SolutionsLab-vSBC-2(http-server)#
```

## 6.2. Configure SBC using Web GUI

In this app note, we configure SBC using the WebGUI.

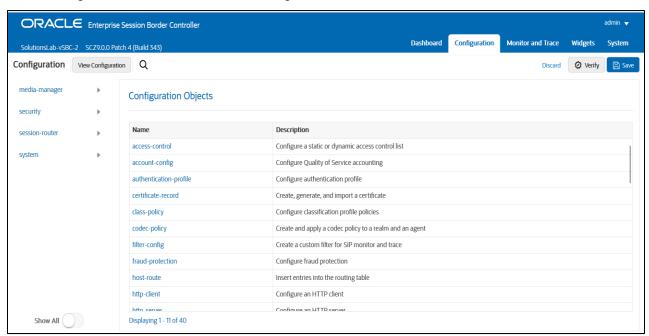
The Web GUI can be accessed through the url http://<SBC\_MGMT\_IP>.



The username and password is the same as that of CLI.



Go to Configuration as shown below, to configure the SBC



Kindly refer to the GUI User Guide given below for more information.

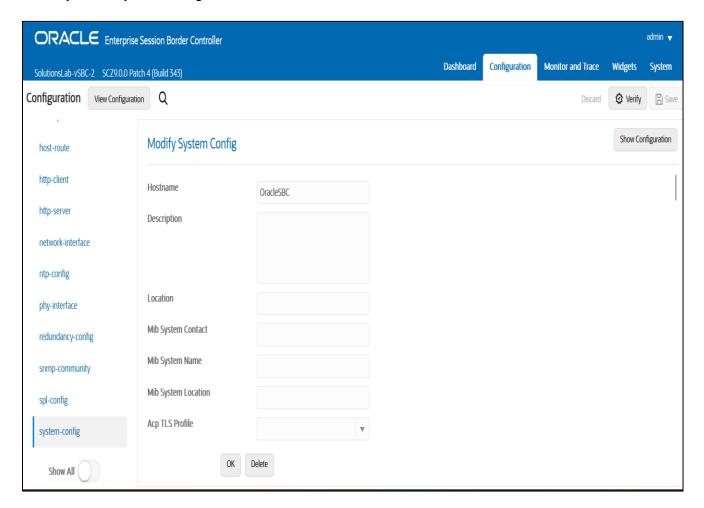
https://docs.oracle.com/en/industries/communications/enterprise-session-border-controller/9.3.0/webgui/web-gui-guide.pdf

The 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.

## 6.3. Configure system-config

Go to system-system-config



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

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

The above step is needed only if any transcoding is used in the configuration. If there is no transcoding involved, then the above step is not needed.

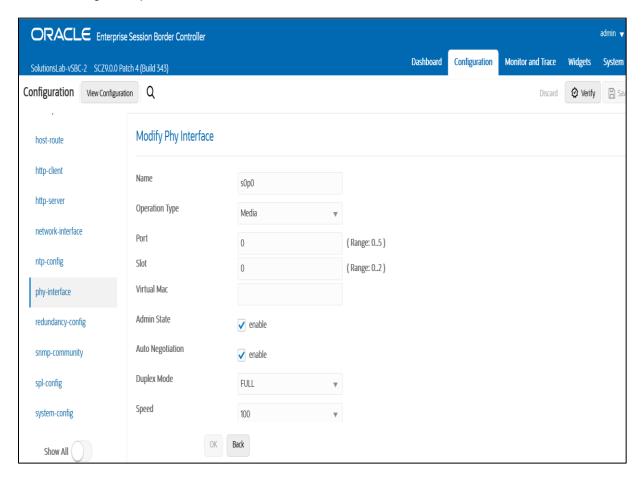
## 6.4. Configure Physical Interface values

To configure physical Interface values, go to System->phy-interface.

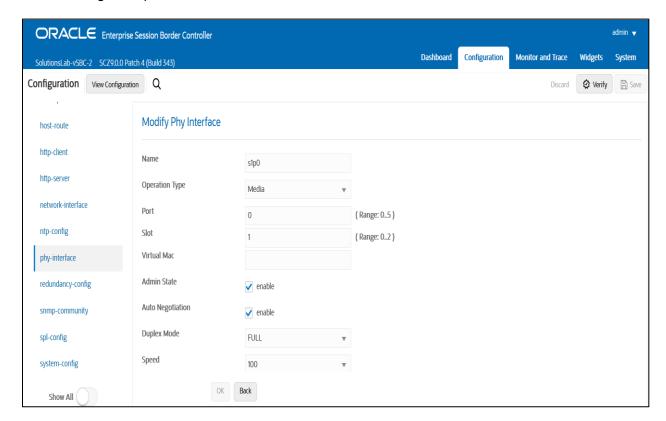
Please configure s0p0 for PSTN side and s1p0 for Cisco Webex side.

Parameter Name	PSTN Trunk side (s0p0)	Cisco Webex side (s1p0)
Slot	0	1
Port	0	0
Operation Mode	Media	Media

Please configure s0p0 interface as below.



### Please configure s1p0 interface as below



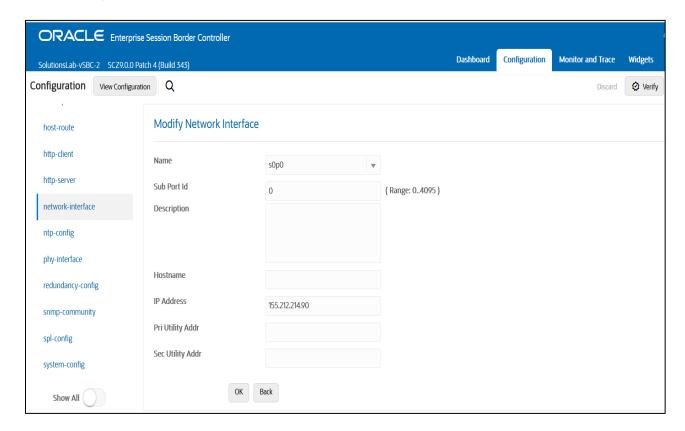
## 6.5. Configure Network Interface values

To configure network-interface, go to system->Network-Interface. Configure interface

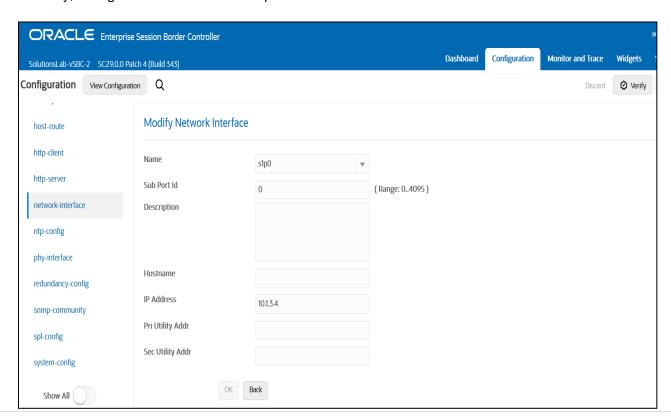
The table below lists the parameters, to be configured for both the interfaces.

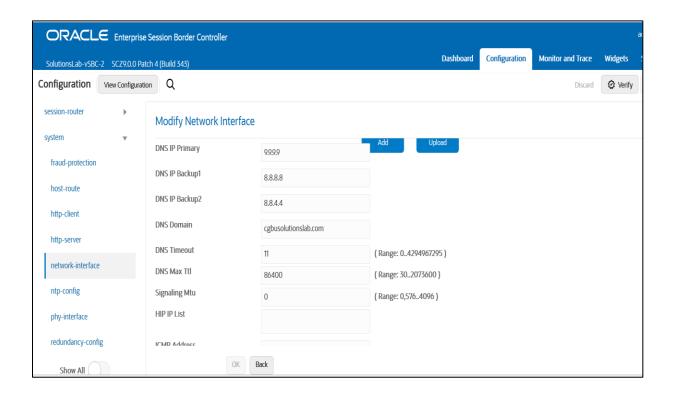
Parameter Name	PSTN Trunk Side Network Interface(s0p0)	Cisco Webex side Network Interface(s1p0)
Name	s0p0	S1p0
Host Name		
IP Address	155.212.214.90	10.1.3.4
Net Mask	255.255.255.0	255.255.255.0
Gateway	155.212.214.65	10.1.3.1

## Please configure network interface s0p0 as below



## Similarly, configure network interface s1p0 as below

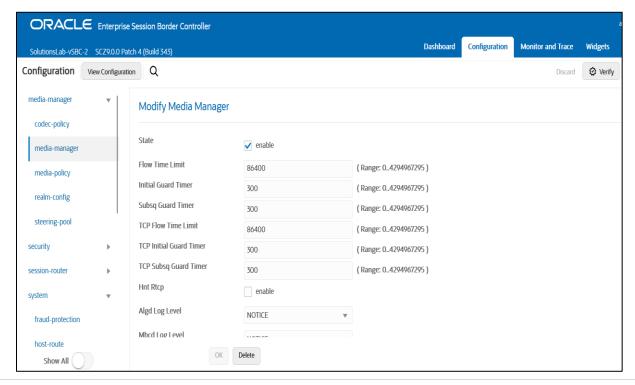


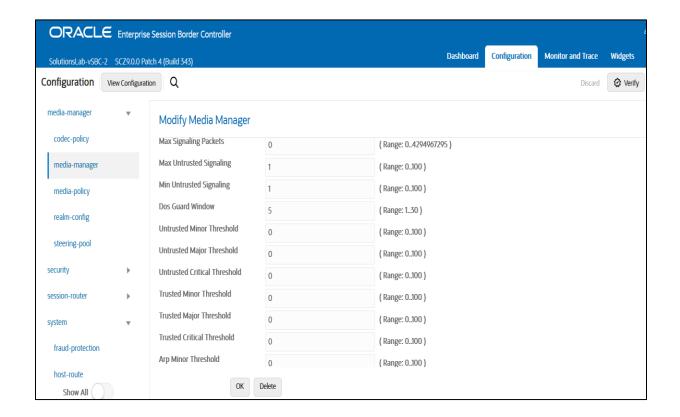


## 6.6. Enable media manager

Media-manager handles the media stack required for SIP sessions on the SBC. Enable the media manager option as below.

Go to Media-Manager->Media-Manager

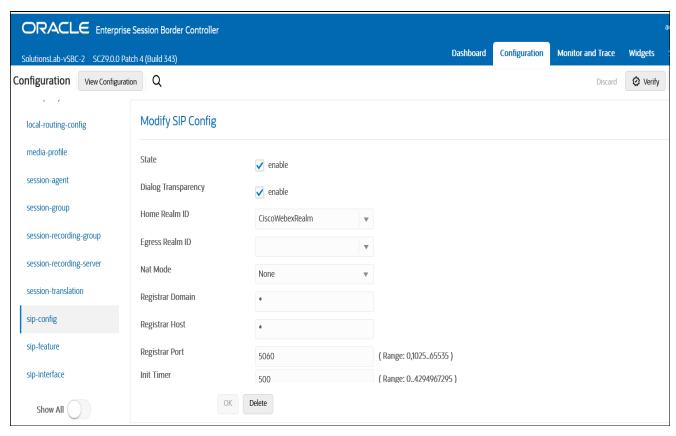


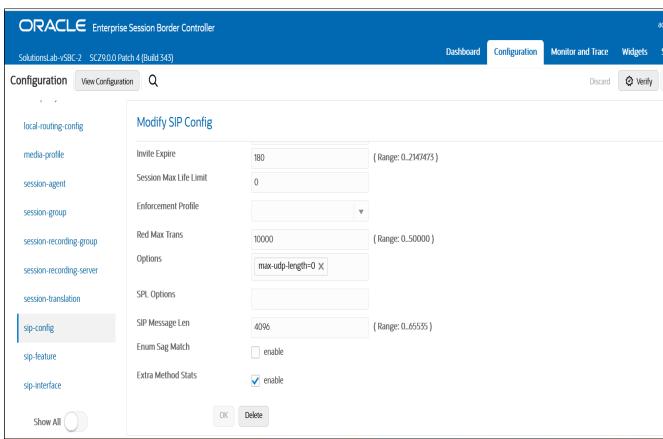


## 6.7. Enable sip-config

SIP config enables SIP handling in the SBC. To configure sip-config, Go to Session-Router->sip-config.

Also add the options to the sip-config as shown below. In options add max-udp-length =0.





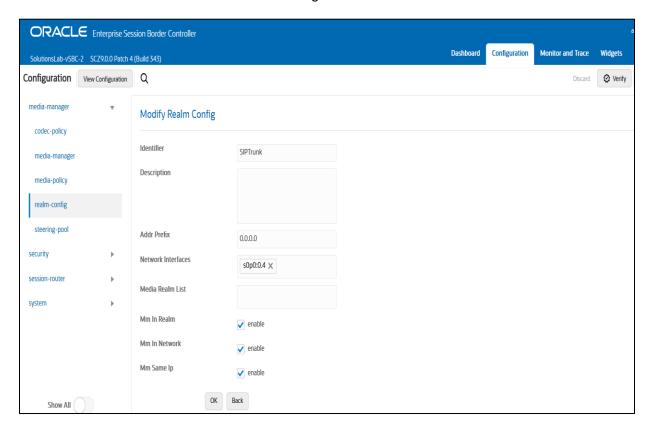
## 6.8. Configure Realms

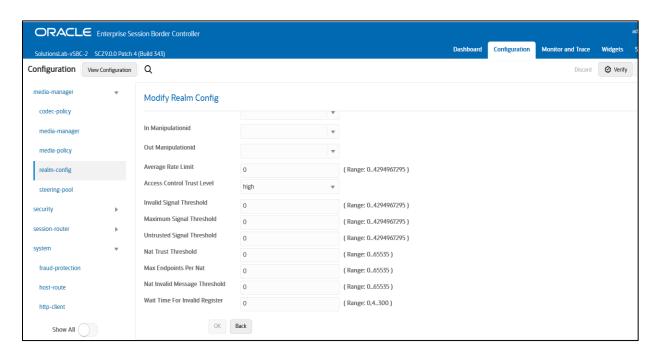
Navigate to realm-config under media-manager and configure a realm as shown below The name of the Realm can be any relevant name according to the user convenience.

Use the following table as a configuration example for the two realms used in this configuration:

Config Parameter	PSTN Side	Cisco Webex Side
Identifier	SIPTrunk	CiscoWebexRealm
Network Interface	S0p0	s1p0
Mm in realm	Ø	Ø
trunk-context		cloudsbc.cgbusolutionslab.com
Media Sec policy	CiscoWebexSecurity	PSTNSide
Access Control Trust Level	High	High

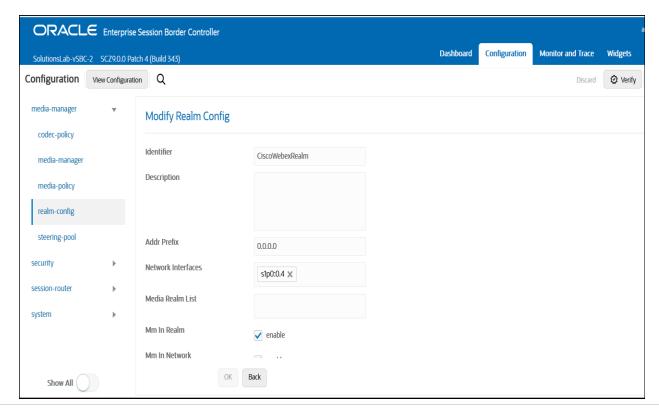
In the below case, Realm name is given as **SIPTrunk** for PSTN Side Please set the Access Control Trust Level as high for this realm

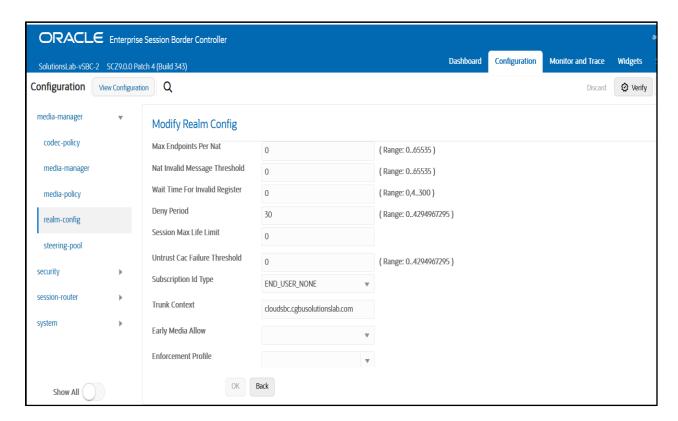




Similarly, Realm name is given as **CiscoWebexRealm** for Cisco Webex Calling side. Please set the Access Control Trust Level as high for this realm too.

Please set the parameter trunk-context to cloudsbc.cgbusolutionslab.com (Please note that this parameter value given here is an example used for our testing purposes and the user can configure this value according to their environment). This value is configured as FQDN of SBC in the Cisco Webex Admin portal, and this will be used by Cisco Webex calling side to reach SBC when making calls.





For more information on Access Control Trust Level, please refer to SBC Security guide link given below:

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

## 6.9. Configuring a certificate for SBC

This section describes how to configure the SBC for TLS and SRTP communication for Cisco Webex Calling. Cisco Webex calling side allows TLS connections from SBC's for SIP traffic, and SRTP for media traffic. It requires a certificate signed by the trusted Certificate Authorities like Go Daddy Root CA and also IdenTrust Root CA certificate as Cisco Webex has moved to a new Certificate Authority, IdenTrust Commercial Root CA from March 2021.

The links for Identrust certificate is given below:

https://help.Webex.com/en-us/article/WBX9000034330/New-Root-Certificate-Authority-for-Cisco-Webex-Services-from-March-2021

https://help.Webex.com/en-us/article/WBX9000008850/What-Root-Certificate-Authorities-are-Supported-for-Calls-to-Cisco-Webex-Audio-and-Video-Platforms?

Though the links talks about Identrust certificates used by Cisco VCS and Expressway, we can still Download the Identrust root certificate and can upload it to the Oracle SBC with the steps given below.

The process includes the following steps:

1) Create a certificate-record – "Certificate-record" are configuration elements on Oracle SBC which captures information for a TLS certificate – such as common-name, key-size, key-usage etc.

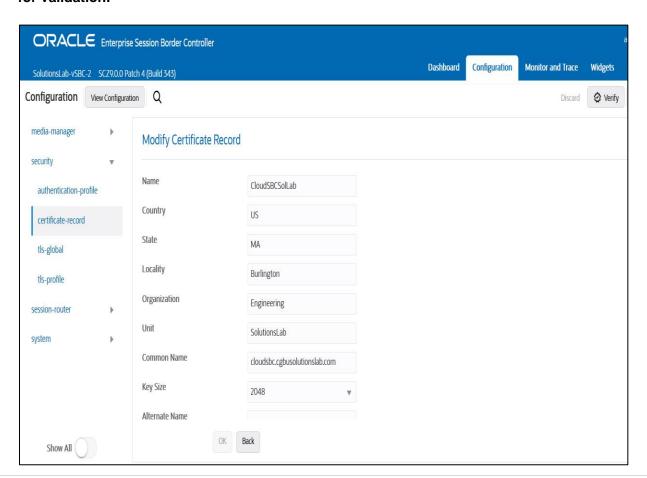
This section walks you through how to configure certificate records, create a certificate signing request and import the necessary certificates into the SBC's configuration.

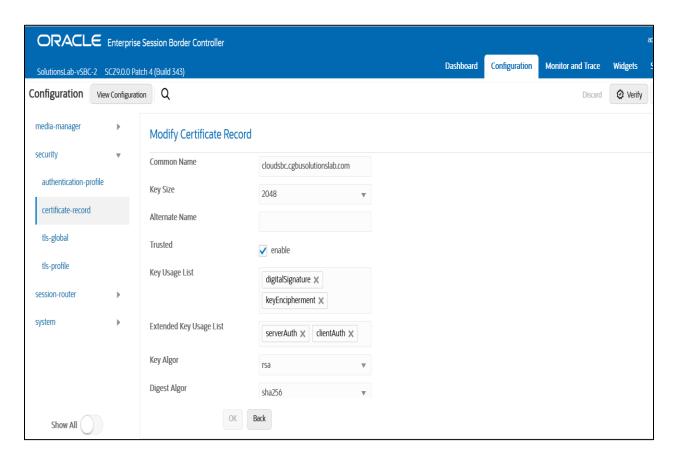
- SBC 1 certificate-record assigned to SBC
- Root 1 certificate-record for root cert
- 2) Deploy the SBC and Root certificates on the SBC

## Step 1 - Creating the certificate record

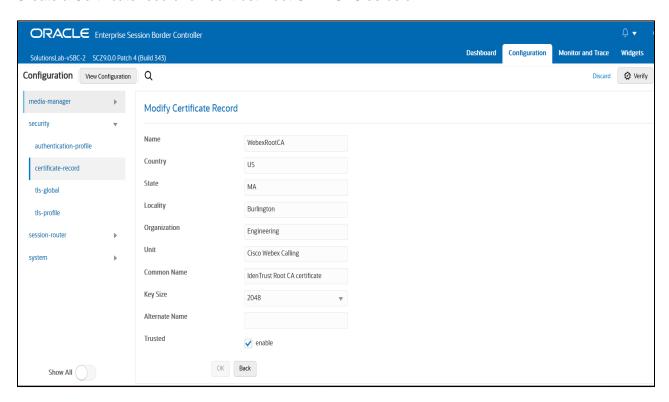
Go to security->Certificate Record and configure the SBC entity certificate for SBC as shown below.

Please note that the FQDN created on the Webex side must be the Common Name (CN) or Subject Alternative Name (SAN) of the certificate. As Cisco does an exact match and do not support wildcard certificates, each domain must be called out in CN or SAN of the certificate for validation.





Create a Certificate record for Identrust Root CA in SBC as below:



The table below specifies the parameters required for certificate configuration. Modify the configuration according to the certificates in your environment.

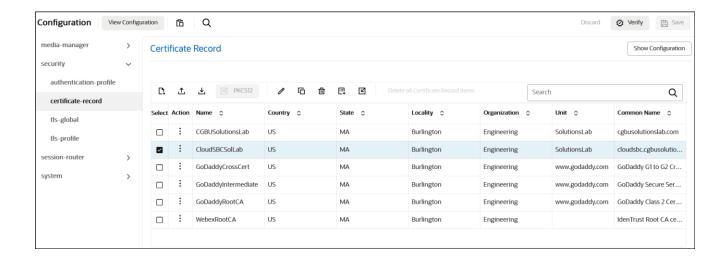
Config Parameter	Go Daddy Root	IdenTrust Root
Common Name	Go Daddy class2 Root CA	IdenTrusrt Root CA
Key Size	2048	2048
Key-Usage-List	digitalSignature keyEncipherment	digitalSignature keyEncipherment
Extended Key Usage List	serverAuth	serverAuth
Key algor	rsa	rsa
Digest-algor	Sha256	Sha256

## Step 2 – Generating a certificate signing request

(Only required for the SBC's end entity certificate, and not for root CA certs)

Please note – certificate signing request is only required to be executed for SBC Certificate – not for the root/intermediate certificates.

- Select the certificate and generate certificate on clicking the "Generate" command.
- Please copy/paste the text that gets printed on the screen as shown below and upload to your CA server for signature.

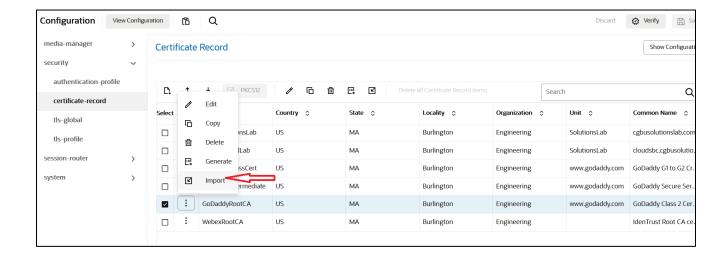


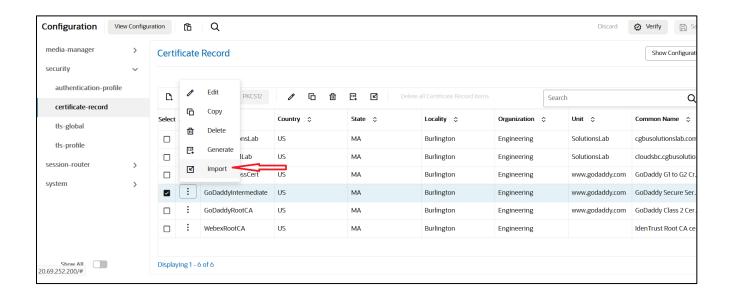


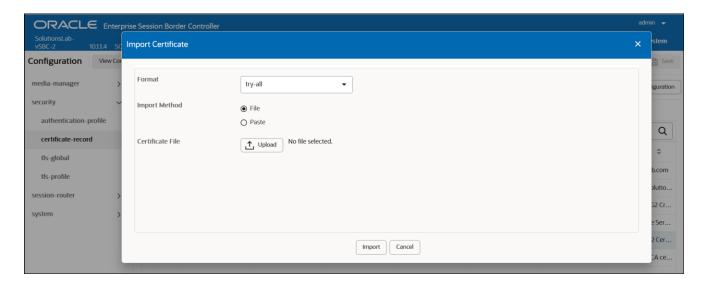
Also, note that a save/activate is required

## Step 3 - Deploy SBC & root certificates

Once certificate signing request have 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 done, issue save/activate from the WebGUI





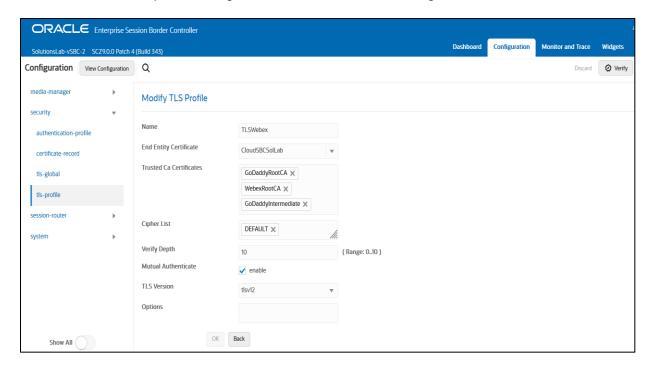


Repeat these steps to import all the root and intermediate CA certificates into the SBC:

At this stage all the required certificates have been imported to the SBC for Cisco Webex Calling.

#### 6.10. TLS-Profile

A TLS profile configuration on the SBC allows for specific certificates to be assigned. Go to security-> TLS-profile config element and configure the tls-profile as shown below The below is the TLS profile configured for the Cisco Webex calling side:

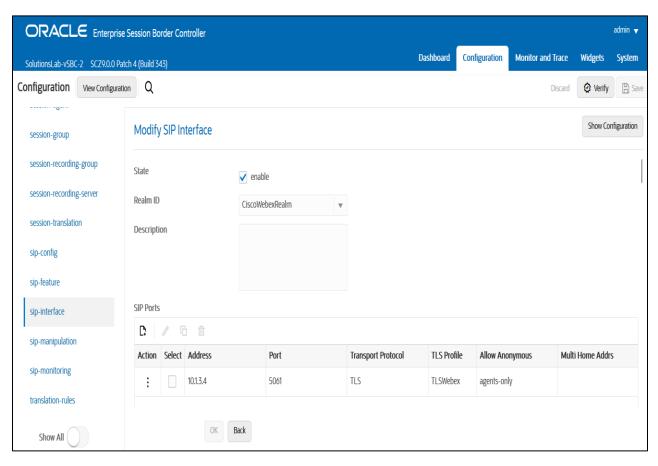


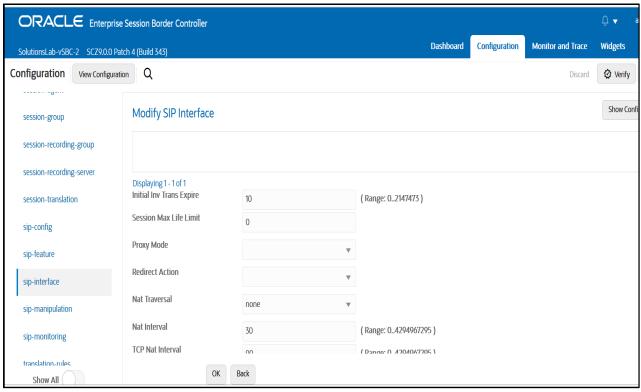
## 6.11. Configure SIP Interfaces

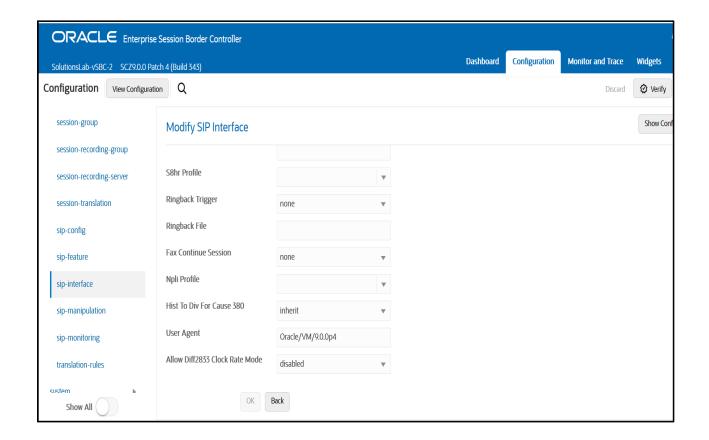
Navigate to sip-interface under session-router and configure the sip-interface as shown below. Please configure the below settings under the sip-interface.

Please Configure sip-interface for the Cisco Webex Calling side as below:

- Tls-profile needs to match the name of the tls-profile previously created
- Set allow-anonymous to agents-only to ensure traffic to this sip-interface only comes from the particular Session agents added to the SBC.
- Set user-agent parameter as Oracle/VM/9.0.0p4 (This can be the respective Oracle SBC Platform and version and these values can be updated accordingly)
- Set **initial-inv-trans-expire parameter value to 10** so the SBC will recurse on no response to SRV session agent

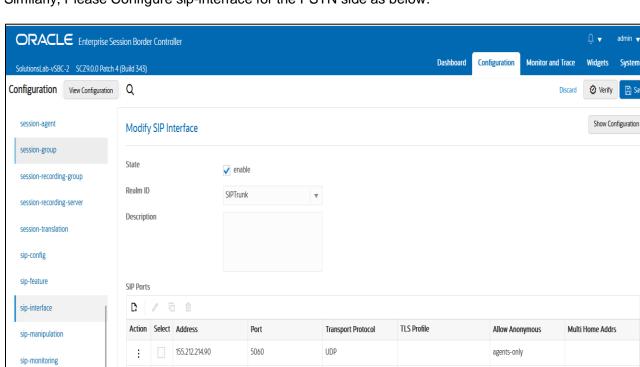






We have some mandatory sip-manipulations that needs to be used with the Oracle SBC so that call flow between Cisco Webex and PSTN will be successful. The User can add these sip manipulations to the SBC using either GUI or CLI mode and is free to decide the way they want to add the sip manipulations. As per the request of Cisco, the FQDN of the SBC needs to be added to all sip messages toward Cisco Webex. Please assign the below sip-manipulation as the out-manipulation ID in the Cisco Webex sip interface or Cisco Webex Session Agent as per customer need.

```
sip-manipulation
                                  ToCiscoWebex
    name
    header-rule
                                   addplus
         name
         header-name
                                      Contact
                                  manipulate
         action
         comparison-type
                                      pattern-rule
         msg-type
                                    request
         methods
                                    Invite
         element-rule
                                       TenDigits
             name
             type
                                      uri-user
             action
                                       replace
             comparison-type
                                           pattern-rule
                                         ^[0-9]{10}$
             match-value
                                         \+1+$ORIGINAL
             new-value
         element-rule
                                       ElevenDigits
             name
             type
                                      uri-user
                                      replace
             action
             comparison-type
                                           pattern-rule
             match-value
                                         ^[0-9]{11}$
             new-value
                                         \++$ORIGINAL
     header-rule
                                   ChangeContactHost
         name
         header-name
                                      Contact
         action
                                  manipulate
         methods
                                    ACK, INVITE
         element-rule
                                       contacthost
             name
                                      uri-host
             type
                                       replace
             action
             new-value
                                         $TRUNK_GROUP_CONTEXT
    header-rule
                                   AddContactOptions
         name
         header-name
                                      Contact
         action
                                  add
         msg-type
                                  request
         methods
                                 OPTIONS
                                <sip:ping@"+$TRUNK_GROUP_CONTEXT+":5061;transport=tls>"
         new-value
```



Similarly, Please Configure sip-interface for the PSTN side as below:

We also have a sip-manipulation for PSTN side to remove DTG parameter which comes from Cisco side which will not be accepted by some of the sip trunks. So, we use the below manipulation to remove it. Please assign the below sip-manipulation as the out-manipulation ID in the PSTN sip interface. Please note that this sip-manipulation can be used according to the needs of the user as some of the sip trunks allow this parameter by default.

TCP

agents-only

```
sip-manipulation
                               RemoveDTG
    name
    description
    split-headers
    join-headers
    header-rule
                                   StripDTG
         name
         header-name
                                    Request-URI
         action
                                  manipulate
         comparison-type
                                       case-sensitive
         msg-type
                                    request
         methods
                                    Invite
         match-value
         new-value
```

155.212.214.90

OK Back

translation-rules

Show All

system

5060

element-rule

name stripdtg parameter-name dtg

type header-param action delete-element

match-val-type any

comparison-type match-value

match-value new-value

Once sip-interface is configured – the SBC is ready to accept traffic on the allocated IP address.

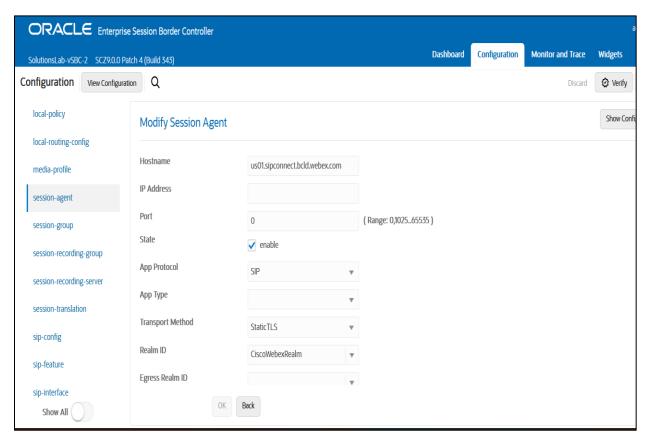
case-sensitive

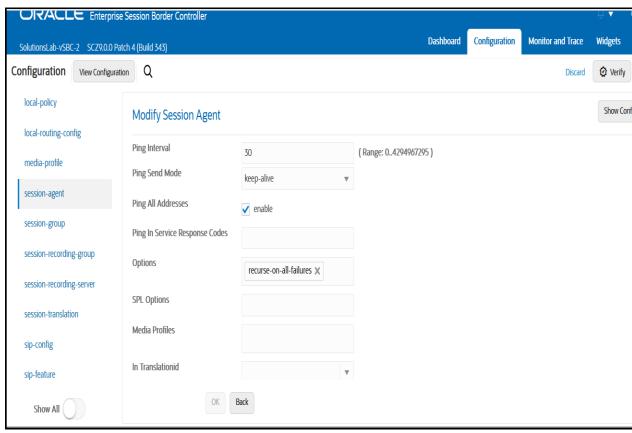
#### 6.12. Configure session-agent

Session-agents are config elements which are trusted agents who can send/receive traffic from the SBC with direct access to trusted data path.

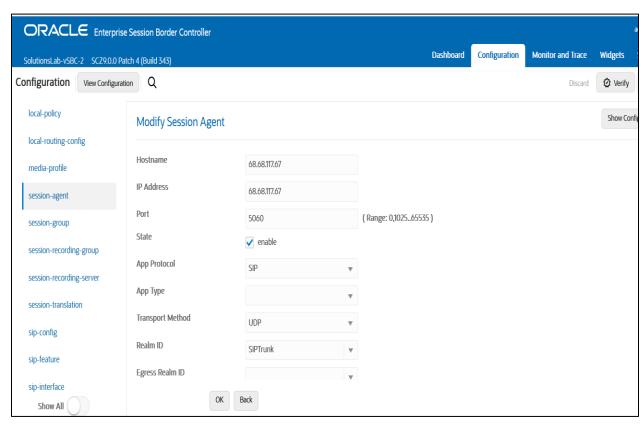
Go to session-router->Session-Agent and Configure the session-agents for the Cisco Webex side

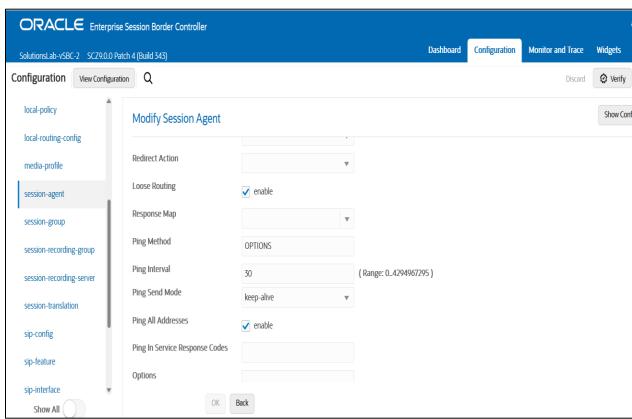
- Host name to "us01.sipconnect.bcld.Webex.com", which is SRV based SA.
- When Using SRV as session agent, please make **port as 0** so that SRV will work properly.
- realm-id needs to match the realm created for the Cisco Webex side.
- transport set to "staticTLS"
- Please enable the parameters **ping all addresses**, **ping-response**,
- Please enable hidden option **load-balance-dns-query** and **recurse-on-all- failures and set out-service-response-codes parameter to 408,503**
- Please set ping method to OPTIONS and ping-interval duration in secs.





Similarly, configure the session-agents for the PSTN Side as below:





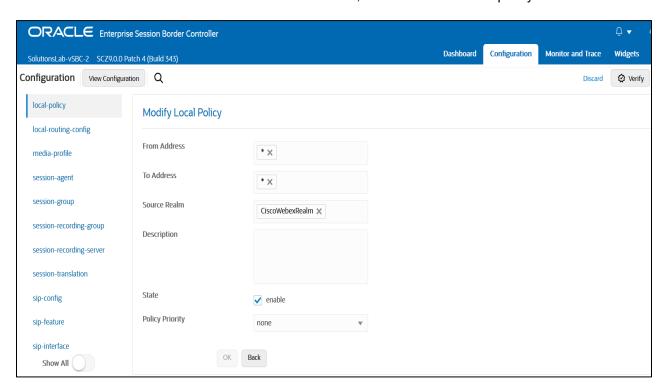
Please assign the below mandatory sip-manipulation as the out-manipulation ID in PSTN sip interface or PSTN Session Agent as per customer need.

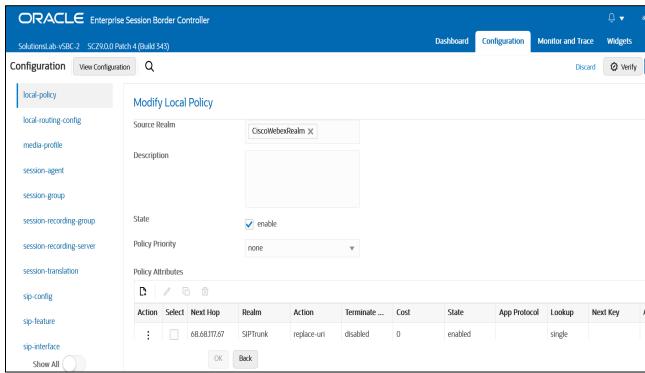
```
sip-manipulation
    name
                               ToPSTN
    description
    split-headers
    join-headers
    header-rule
                                   StripDTG
         name
                                      Request-URI
         header-name
                                  manipulate
         action
         comparison-type
                                       case-sensitive
         msg-type
                                    request
         methods
                                    Invite
         match-value
         new-value
         element-rule
                                        stripdtg
              name
              parameter-name
                                            dtg
                                       header-param
              type
              action
                                       delete-element
              match-val-type
                                           any
              comparison-type
                                            case-sensitive
             match-value
              new-value
      header-rule
                                   DeleteXBroadworks
         name
                                    X-BroadWorks-Correlation-Info
         header-name
         action
                                  delete
         comparison-type
                                    case-sensitive
         msg-type
                                    any
         methods
                                    BYE, INVITE, OPTIONS
         match-value
         new-value
    header-rule
         name
                                   DeleteSessionID
         header-name
                                   Session-ID
         action
                                  delete
         comparison-type
                                  case-sensitive
         msg-type
                                   any
         methods
                                   BYE, INVITE, OPTIONS
         match-value
         new-value
    header-rule
                                   DeleteRecvInfo
         name
         header-name
                                   Recv-Info
         action
                                  delete
         comparison-type
                                  case-sensitive
         msg-type
         methods
                                    BYE, INVITE, OPTIONS
```

### 6.13. Configure local-policy

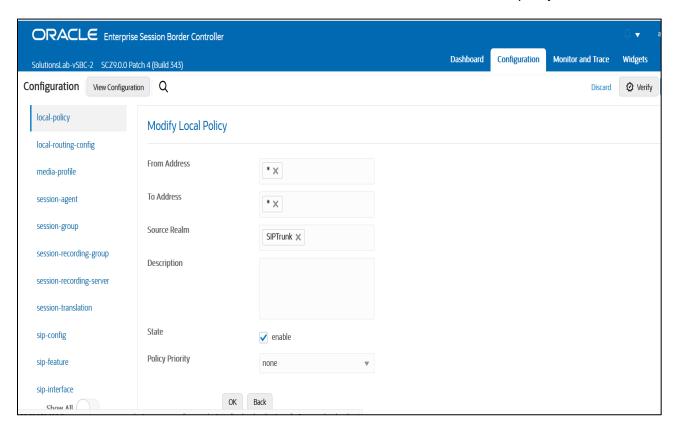
Local policy config allows for the SBC to route calls from one end of the network to the other based on routing criteria. To configure local-policy, go to Session-Router->local-policy.

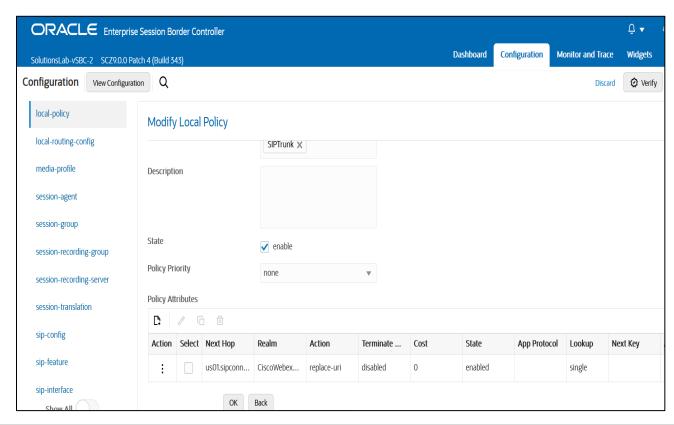
To route the calls from Cisco Webex side to PSTN side, Use the below local -policy





To route the calls from the PSTN side to Cisco Webex side, Use the below local-policy



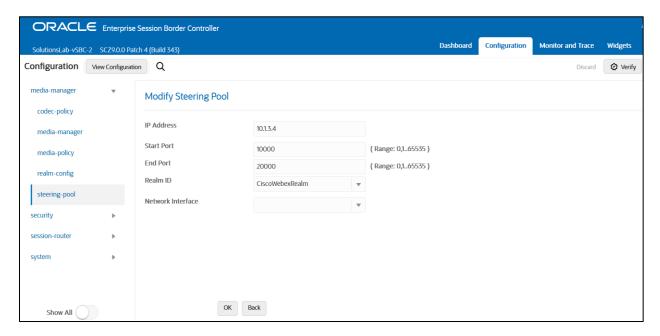


#### 6.14. Configure steering-pool

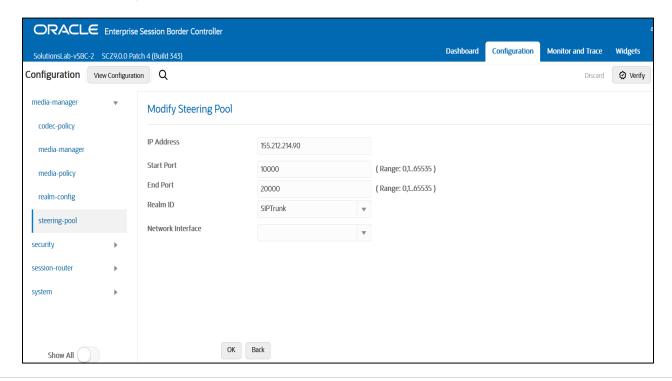
Steering-pool allows configuration to assign IP address(es), ports & a realm.

The port configuration for Webex Calling as the media ports on LGW side is allowed/advertised from port 8000 to 48000 as per Cisco and the End user can define this port range on the Oracle SBC.

Cisco Webex side steering pool.



PSTN side steering pool.

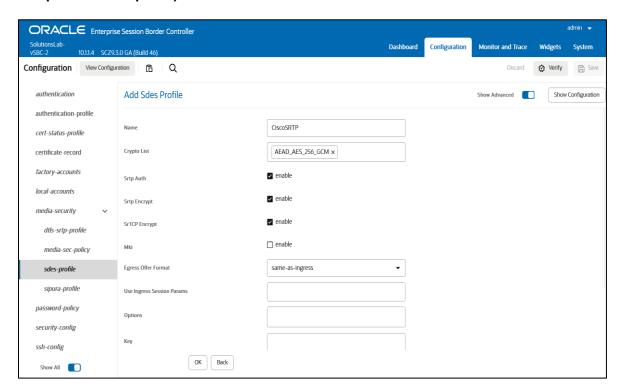


#### 6.15. Configure sdes profile

Oracle SBC and Cisco Webex Calling Support the following ciphers for SRTP:

Please go to →Security → Media Security →sdes profile and create the policy as below.

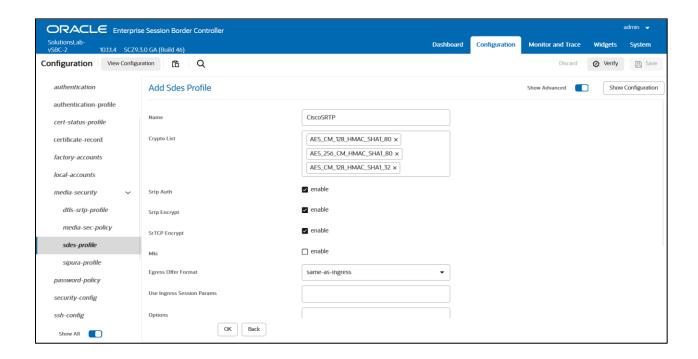
# AEAD\_AES\_256\_GCM (This cipher is applicable only for Webex for Government as it is FIPS-compliant GCM ciphers)



Add the below ciphers to the SDES profile as shown below.

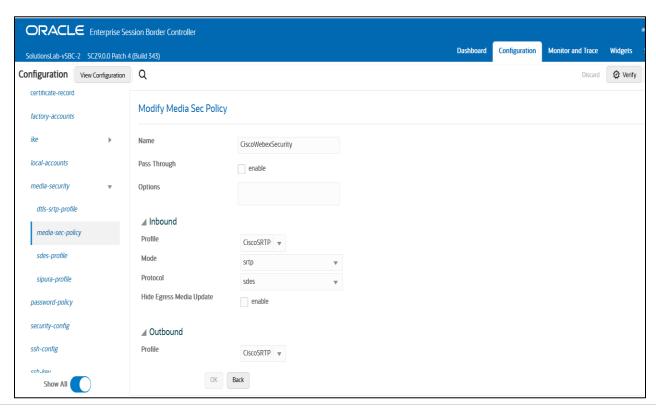
AES\_CM\_256\_HMAC\_SHA1\_80 AES\_CM\_128\_HMAC\_SHA1\_80

AES\_CM\_128\_HMAC\_SHA1\_32 (These 3 ciphers is applicable only for Cisco Webex Calling)

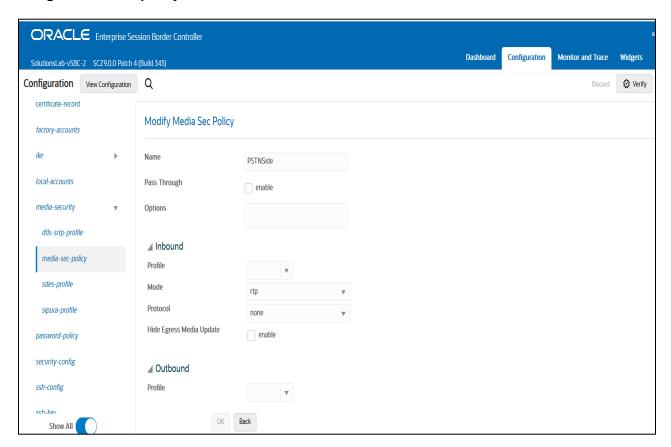


#### 6.16. Configure Media Security Profile

Please go to →Security → Media Security →media Sec policy and create the policy as below: Create Media Sec policy with name CiscoWebexSecurity which will have the sdes profile created above. Assign this media policy to the Cisco Webex Realm



Similarly, Create Media Sec policy with name PSTNSide to convert srtp to rtp for the PSTN side. **Assign this media policy to the PSTN Realm.** 



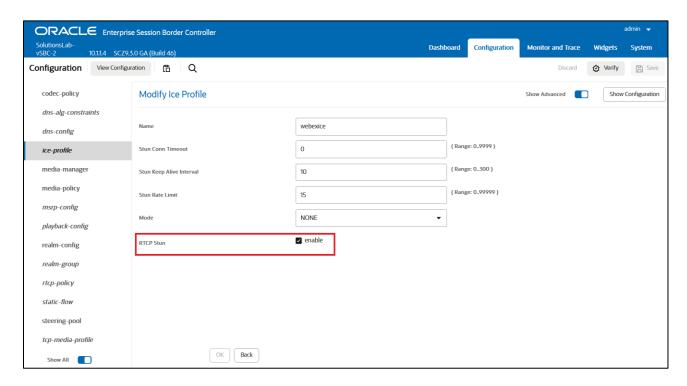
#### 6.17. Configure Media Optimization (ICE-profile)

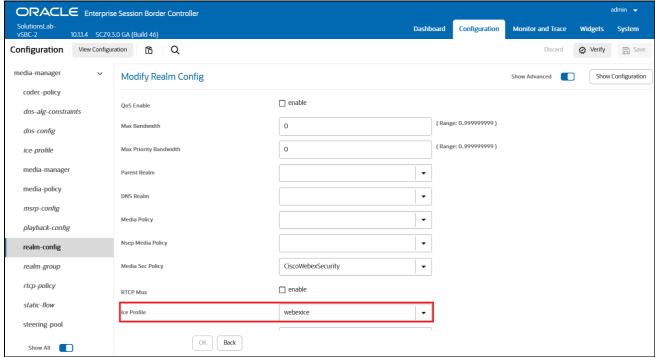
Please go to → media-manager → Select show all option →ICE-profile and create a new profile as below:

Please enable the parameter **rtcp-stun** which is disabled by default.

This is the new parameter introduced in 9.3.0 release to support media optimization feature and this is supported from release 9.3 and later. **Assign this profile to the Cisco Webex Realm**.

Please note that this configuration is used only for media optimization feature.





With this, SBC configuration is complete.

# 7. Existing SBC configuration

If the SBC being used is an existing SBC with functional configuration, following configuration elements are required:

- New realm-config
- Configuring a certificate for SBC Interface
- TLS-Profile
- New sip-interface
- New session-agent
- New steering-pools
- New local-policy
- SDES Profile
- Media-sec-Policy
- Media-Optimization

Please follow the steps mentioned in the above chapters to configure these elements.

# 8. 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

<sup>\*\*</sup> Software transcoding

# 9. 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

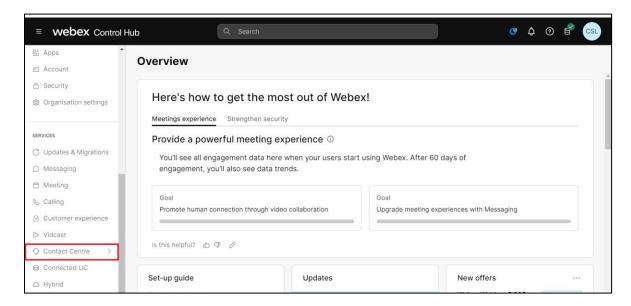
https://help.webex.com/en-us/article/n5595zd/Webex-Contact-Center-Setup-and-Administration-Guide

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 SIP trunking, 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. More detailed configuration of Cisco Webex CC may be required based on the customer needs for the proper working of Webex contact center. For such configuration, please consult your Cisco representative which will be out of scope of this document.

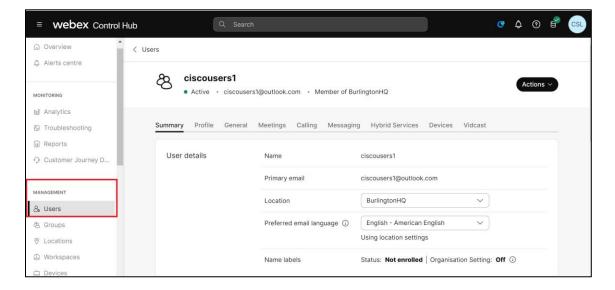
Webex admin page with Contact Center tab enabled:

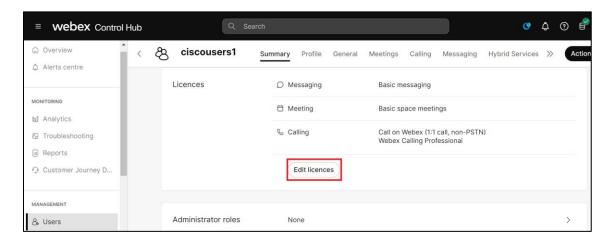


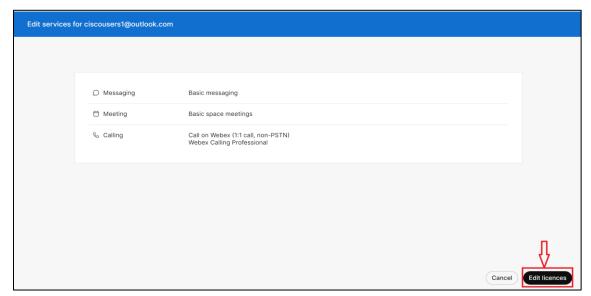
#### 9.1. Enable the Users with Webex CC license

After The first step is to enable the Webex CC license for the users.

Please login to **Cisco Webex control hub portal – Management ---- Users** and enable the license for the users that you wish to as shown below:





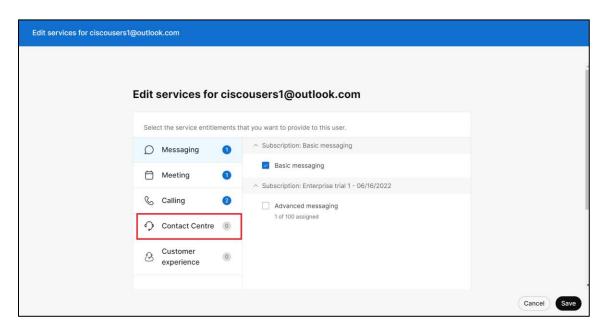


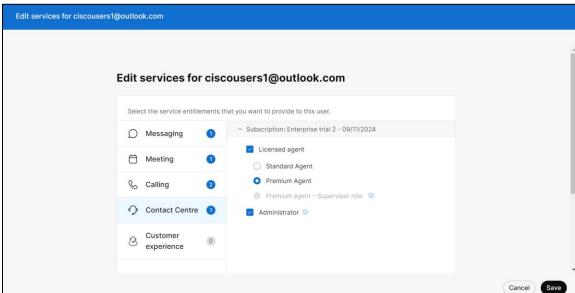
Please click on Contact Center Tab and Enable the Agent type as shown below.

The Agent types are Standard Agent, Premium Agent and Premium Agent with Supervisor role.

Please select the appropriate agent as per your requirements and you can also select the Agent who can also be the Admin for the Webex CC.

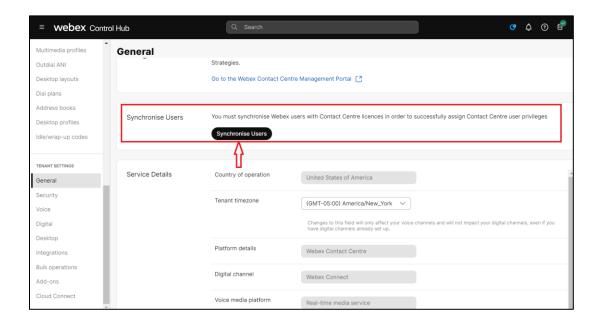
Click Save to enable the changes made and you can do the same procedure for other users which can be the Agents for the Webex CC.





### 9.2. Synchronize the Users with Webex CC tenant.

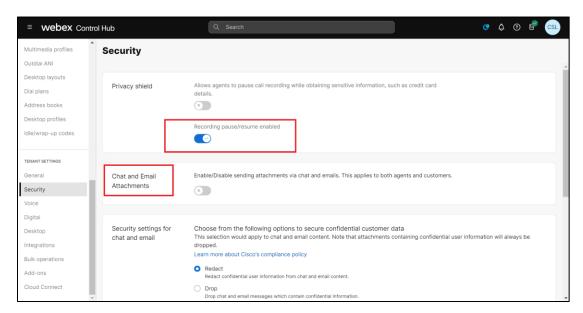
Please go to Cisco Webex control hub portal – Services ---- Contact Center ---- Tenant Settings --- General and click on Synchronize Users tab so that the changes made to Users will be reflected in Cisco Webex CC page. We can also change the time zone from this page and other options can be left default in this page.



#### 9.3. Configure the settings in Security Tab.

Please go to Cisco Webex control hub portal – Services ---- Contact Center ---- Tenant Settings --- Security and do the following settings.

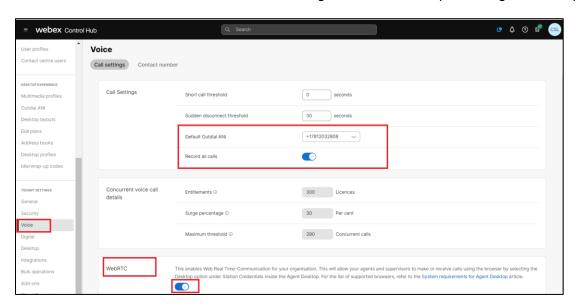
- Enable the Recording Pause/Resume Enabled under Privacy shield tab
- Disable the Chat and Email and Attachments as we are dealing only with Calling option here.



#### 9.4. Configure the settings in Voice Tab.

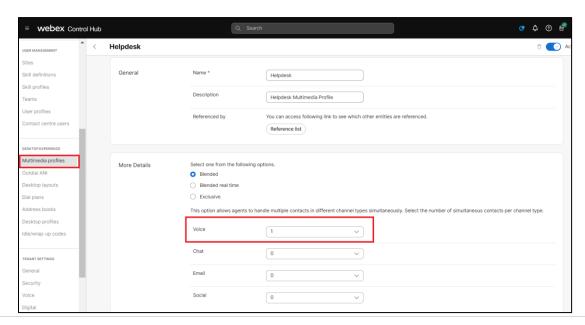
Please go to Cisco Webex control hub portal – Services ---- Contact Center ---- Tenant Settings --- Voice and provide a DID for default out dial ANI. This is the default number which will be used to call Webex CC from outside and will reach the IVR prompt.

We also need to enable WebRTC so that we will get an Webex CC option of Agent Desktop.



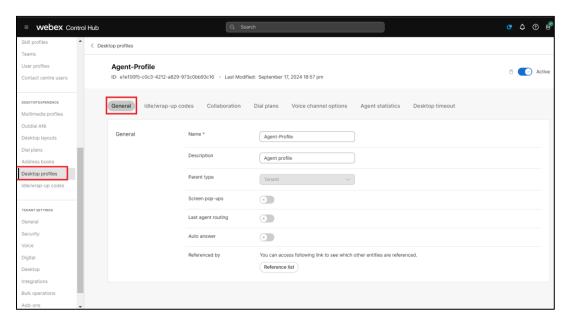
## 9.5. Configure the Multimedia Profile Tab.

Please go to Cisco Webex control hub portal – Services ---- Contact Center ---- Desktop Experience --- Multimedia Profile and create a multimedia profile for the Agents. The configuration in this tab allows agents to handle multiple contacts in different channel types simultaneously. For our profile, we have selected the simultaneous calls as 1 and we do not deal with other options and hence the options are set to zero for those options.

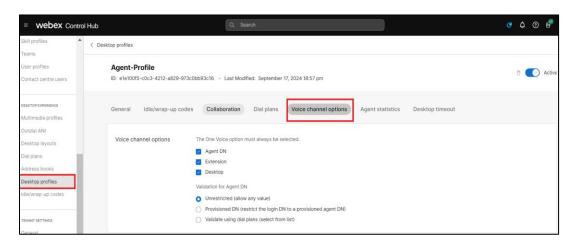


#### 9.6. Configure the Desktop Profile Tab.

Please 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:

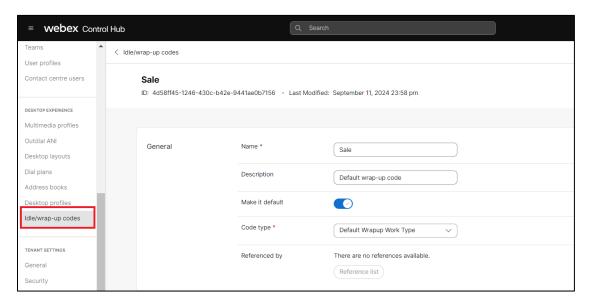


Please click Voice channel options and select the options as shown below.



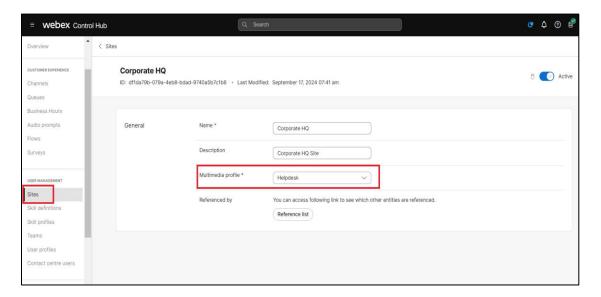
#### 9.7. Configure the Idle/Wrap-up codes Tab.

Please 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.



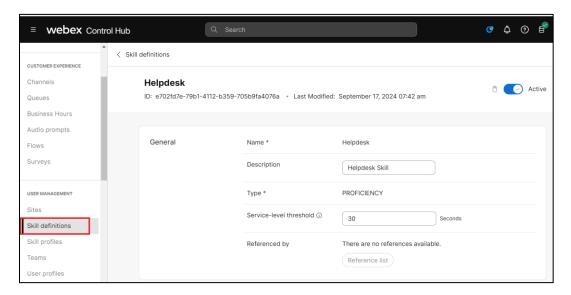
#### 9.8. Configure the Sites Tab.

Please go to Cisco Webex control hub portal – Services ---- Contact Center ---- User Management --- Sites and create a new site. Please assign the Multimedia profile which is created previously to the created site as shown below.



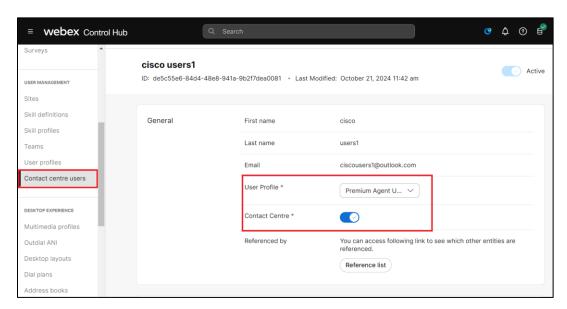
#### 9.9. Configure the Skill Definitions Tab.

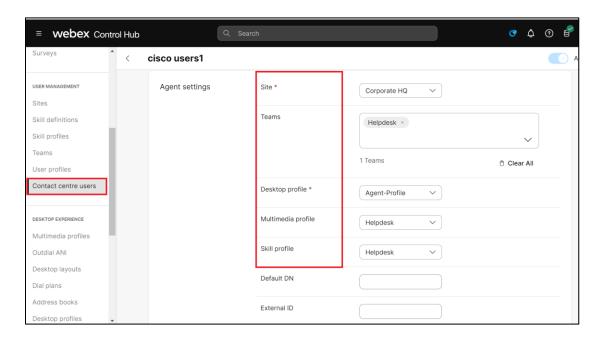
Please go to Cisco Webex control hub portal – Services ---- Contact Center ---- User Management --- Skill Definitions and create a new Skill profile as shown below.



#### 9.10. Configure the Contact Center Users Tab.

Please go to Cisco Webex control hub portal – Services ---- Contact Center ---- User Management --- Contact center users and you will see the users that has Webex CC license enabled and synchronized with Webex CC listed here. We can go ahead and edit the users and can assign the profiles which we have created previously to the users as shown below.



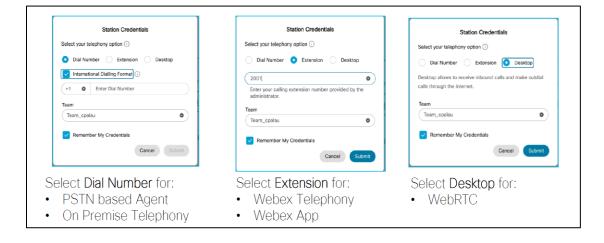


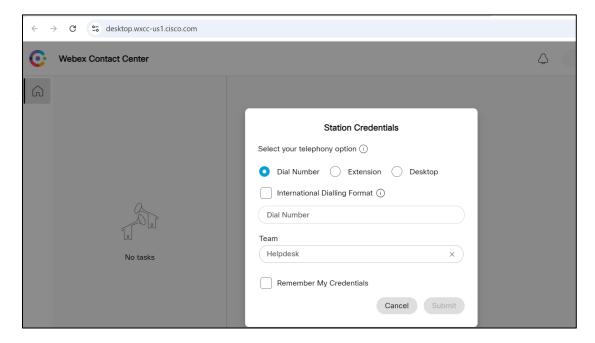
With this, the basic configuration steps of Webex CC are complete.

After the basic Cisco Webex CC configuration is complete, agent login can be performed using the below link.

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

Agents of Cisco Webex CC mainly works in 3 modes after login which is shown below.





Following are the important test cases that have performed for Webex CC on top of the extensive test cases used for certifying the SBC with Cisco Webex LGW. We have tested the voice calls getting routed to the Agent using Oracle SBC and the below test cases are working fine for all the above 3 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 <sup>nd</sup> Agent
6	Consult Transfer to a 2 <sup>nd</sup> Agent
7	Blind Transfer to a 2 <sup>nd</sup> Agent

# Appendix A

# **Configure Multi-Tenancy**

Multi-tenant configuration is primarily to host more than one trunk or locations on the given LGW or in the SBC. There are 2 types of configuration here which is given below:

- Different IP different FQDN
- Same IP different FQDN

These are optional configuration, and the customer can configure this configuration based on their needs.

The configuration steps and the screenshots for the 1st type is shared below

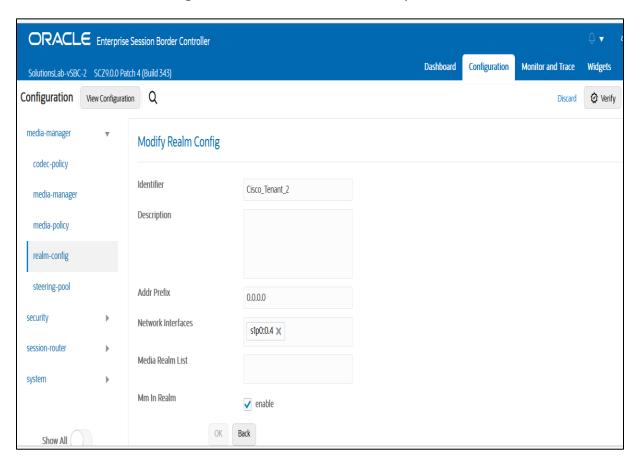
#### Add New Child Realm:

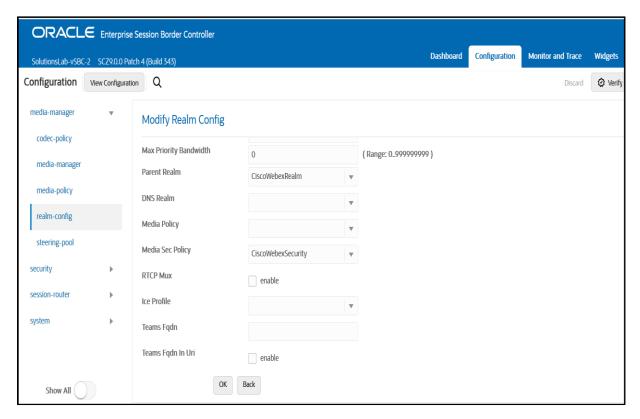
\_\_\_\_\_

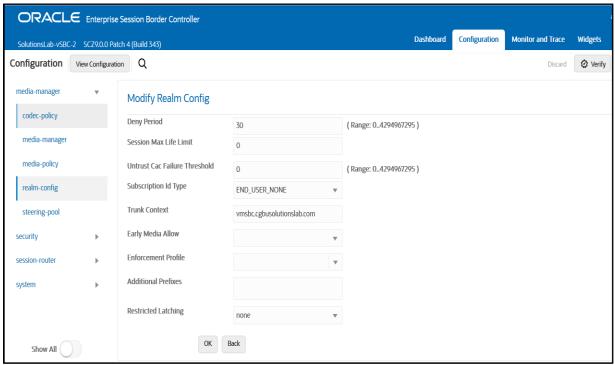
Add another realm to the configuration, identical to the CiscoWebexRealm.

Add the trunk-context field which will now have the hostname of the new tenant which is created.

You will also need to assign the CiscoWebexRealm as the parent-realm



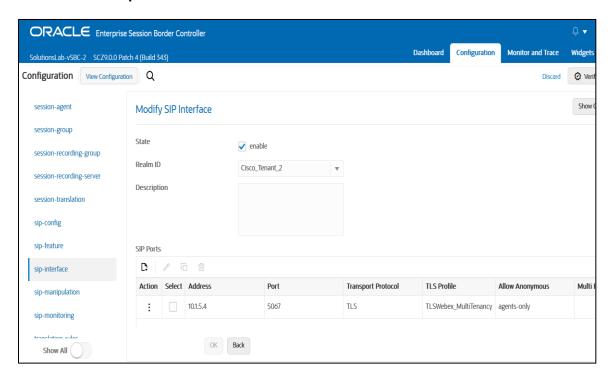




\_\_\_\_\_

Please add a new sip-interface for the child realm.

Create a new tls-profile that includes the certificate for new tenant created above.



The End user can use the below sip manipulation to change certain parameters when configuring multitenancy and the scenarios should work fine without any issues. **Please assign this as out-manipulation ID to the sip-interface created above**. The User can add these sip manipulations to the SBC using either GUI or CLI mode and is free to decide the way they want to add the sip manipulation.

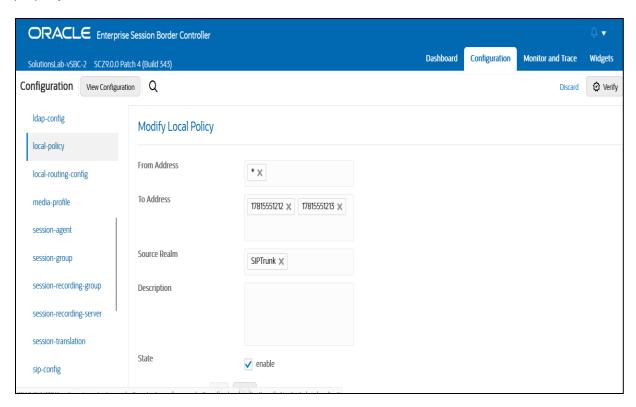
```
sip-manipulation
                                 To_Webex
    name
    header-rule
                                  ChangePAI
         name
         header-name
                                     P-Asserted-Identity
         action
                                 manipulate
         comparison-type
                                      pattern-rule
         methods
                                   INVITE
     element-rule
             name
                                       ChangePAI
                                      uri-host
             type
             action
                                      replace
             new-value
                                        $TRUNK_GROUP_CONTEXT
     header-rule
                                  ChangeToIP
         name
         header-name
                                     TO
         action
                                 manipulate
                                      pattern-rule
         comparison-type
         msg-type
                                   any
         methods
                                   INVITE
```

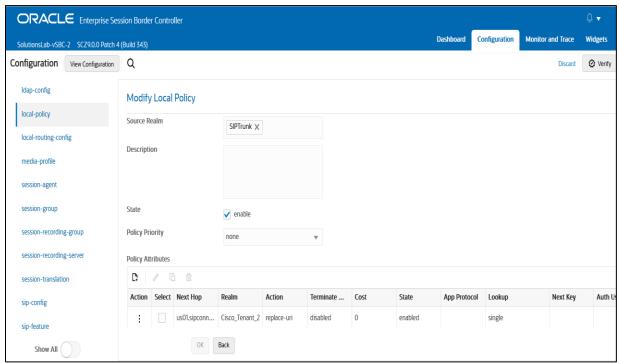
```
element-rule
             name
                                       ChangeTo
             type
                                     uri-host
             action
                                      replace
             new-value
                                      "us01.sipconnect.bcld.Webex.com"
header-rule
        name
                                  ChangeContactHost
        header-name
                                     Contact
        action
                                 manipulate
        msg-type
                                  any
        methods
                                  INVITE, ACK
     element-rule
             name
                                      contacthost
             type
                                     uri-host
             action
                                      replace
                                      $TRUNK_GROUP_CONTEXT
             new-value
    header-rule
                                  AddContactOptions
        name
                                   Contact
        header-name
        action
                                   add
        msg-type
                                   request
        methods
                                   OPTIONS
        new-value
                             <sip:ping@"+$TRUNK_GROUP_CONTEXT+":5061;transport=tls>
 header-rule
        name
                                  ChangeFromIP
                                     FROM
        header-name
        action
                                 manipulate
        msg-type
                                   any
        methods
                                   INVITE
      element-rule
                                       ChangeFrom
             name
                                     uri-host
             type
             action
                                      replace
             new-value
                                        $TRUNK_GROUP_CONTEXT
    header-rule
                                  Addplus1Contact
        name
        header-name
                                     Contact
        action
                                 manipulate
        comparison-type
                                      pattern-rule
    element-rule
                                      Tendigits
             name
             type
                                     uri-user
             action
                                      replace
             comparison-type
                                          pattern-rule
             match-value
                                         ^[0-9]{10}$
                                        \+1+$ORIGINAL
             new-value
    element-rule
             name
                                       ElevenDigits
             type
                                     uri-user
             action
                                      replace
             comparison-type
                                          pattern-rule
                                         ^[0-9]{11}$
             match-value
                                        \++$ORIGINAL
             new-value
```

#### Add New Local Policy

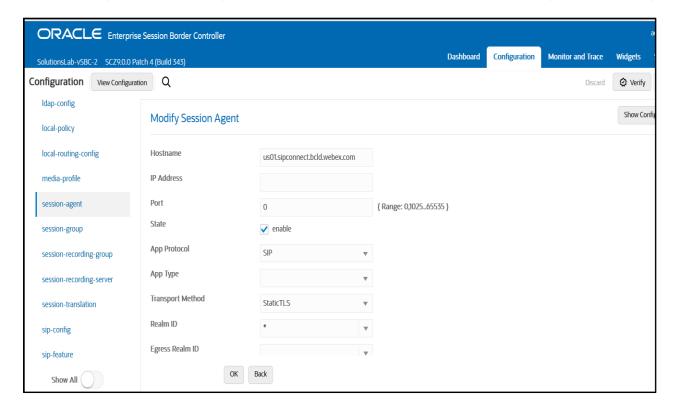
\_\_\_\_\_

Add new local policy which is matching on the DID's assigned to the users in the second tenant to properly route inbound calls as below:





Finally, change the realm ID in the **SRV session agent to** \* for multitenancy to work in both types. When making the above change, please make sure to add a home realm to the global sip config



With this, the SBC config for Multitenancy with Different IP different FQDN is complete.

For multitenancy with same IP which is the 2nd type, the user **just has to remove the second sip interface that was configured for the new tenant**. Leave the local policy that was configured earlier for 1st type, and that should work since the calls will egress over the same IP but using a child realm as next hop.

Also, leave the session agent with \* as the realm ID as shown in the above screenshot.

For this model, please create a single certificate with 2 SAN entries using the parameter -----> alternate-name.

The details of this parameter is given below with an example:

-->The alternate name of the certificate holder which can be expressed as an IP address, DNS host, or email address. Configure this parameter using the following syntax to express each of these 3 forms.

ORACLESBC(certificate-record)# alternate-name

IP:10.2.2.2, IP:10.3.3.3, DNS:bar.example.com, DNS:foo.example.com

(Note each entry is comma separated)

### Appendix B

As Cisco doesn't support any other clock rate other than 8K for DTMF and if the customer wants to use OPUS codec(48K) with Cisco WebEx Calling, there is a new feature added to the SBC 9.1p2 version or higher which will solve this problem. The feature name is **Separate Clock Rates for Audio and Telephone Events** and this feature is applied only when using OPUS codec with Cisco WebEx Calling.

If the customer is using SBC version other than 9.1p2 or higher, (For Ex, SBC 9.0 or 8.4 versions etc) then they may face DTMF issue as the feature is not available in those releases. For more information about this feature, please check the given link with the feature name given above.

# 10. Caveat

#### Issue 1: SIP OPTIONS ping from multiple Realms to global session agents.

Cisco requires SBC vendors to send SIP OPTION ping (keepalives) from all realms in multi-tenant UCaaS environment that contains the FQDN of each trunk to monitor the connection health between the SBC and customer tenant.

The Oracle SBC has a limitation of the above requirement as of now and the SBC can only successfully monitor a single customer tenant based on the current behavior of SIP OPTIONS ping. SBC still will respond locally to all OPTIONS sent from Cisco to the SBC on all trunks in a multitenancy environment, and our testing showed no interruption in calling service due to this limitation.

Oracle Engineering is working on an enhancement request to create the ability for the SBC to send SIP OPTIONS ping (keepalives) from multiple Realms to global session agents. This enhancement will be available in future SBC release (exact release not identified as of now) and this app note will be updated accordingly once the release is available. There are some workarounds that have been successful in customer environments. Please reach out to your account team to discuss what available options may be best suited for your particular environment.

#### Issue 2: Video Call issues when call comes from Cisco CUCM towards Cisco WebEx.

Some of the customer was having issues with establishing video between on prem CUCM and Webex Calling while using Oracle SBC as LGW and this issue happens because of how video starts or is handled. This issue is resolved after removing the below headers from the SDP video attribute coming from Cisco CUCM side and going towards Cisco WebEx side.

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

We have created the below sip-manipulation which will remove these headers and this sip-manipulation needs to be applied towards Cisco WebEx side.

```
mime-sdp-rule
         name
                                    Changealine
         msg-type
                                     any
         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
                                             а
                   action
                                             delete
                   comparison-type
                                                  pattern-rule
                   match-value
                                                (rtcp)(.*)
                   new-value
```

# 11. ACLI Running Configuration

Below is a complete output of the running configuration used to create this application note. This output includes all of the configuration elements used in our examples, including some of the optional configuration features outlined throughout this document. Be aware that 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.

certificate-record name **CGBUSolutionsLab** SolutionsLab unit cgbusolutionslab.com common-name extended-key-usage-list serverAuth clientAuth certificate-record CloudSBCSolLab name unit SolutionsLab cloudsbc.cgbusolutionslab.com common-name serverAuth extended-key-usage-list clientAuth certificate-record name GoDaddyCrossCert www.godaddy.com unit GoDaddy G1 to G2 Cross Certificate common-name certificate-record name GoDaddyIntermediate www.godaddy.com unit 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 http-server webserver name local-policy from-address to-address source-realm CiscoWebexRealm policy-attribute 68.68.117.67 next-hop realm SIPTrunk replace-uri action local-policy from-address to-address source-realm **SIPTrunk** policy-attribute next-hop us01.sipconnect.bcld.webex.com realm CiscoWebexRealm action replace-uri

```
media-manager
media-sec-policy
    name
                               CiscoWebexSecurity
    inbound
                                   CiscoSRTP
         profile
         mode
                                    srtp
         protocol
                                    sdes
    outbound
                                   CiscoSRTP
         profile
         mode
                                    srtp
         protocol
                                    sdes
media-sec-policy
                               PSTNSide
    name
network-interface
    name
                               s0p0
                                155.212.214.90
    ip-address
    netmask
                                255.255.255.0
    gateway
                                155.212.214.65
network-interface
    name
                               s1p0
    ip-address
                                10.1.3.4
    netmask
                                255.255.255.0
                                10.1.3.1
    gateway
    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
                               s1p0
    name
                                 Media
    operation-type
    slot
realm-config
    identifier
                              CiscoWebexRealm
    network-interfaces
                                   s1p0:0.4
    mm-in-realm
                                 enabled
    media-sec-policy
                                   CiscoWebexSecurity
    access-control-trust-level
    trunk-context
                                 cloudsbc.cgbusolutionslab.com
    ice-profile
                              webexice
realm-config
    identifier
                              SIPTrunk
    network-interfaces
                                   s0p0:0.4
    mm-in-realm
                                 enabled
    media-sec-policy
                                   PSTNSide
```

```
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
    hostname
                                68.68.117.67
    ip-address
                               68.68.117.67
    realm-id
                              SIPTrunk
                                 OPTIONS
    ping-method
    ping-interval
                               30
    ping-response
                                 enabled
session-agent
    hostname
                                us01.sipconnect.bcld.webex.com
                             0
    port
    transport-method
                                  StaticTLS
    realm-id
                              CiscoWebexRealm
    ping-method
                                 OPTIONS
    ping-interval
                               30
    ping-all-addresses
                                  enabled
    ping-response
                                 enabled
sip-config
    home-realm-id
                                 CiscoWebexRealm
    registrar-domain
    registrar-host
    registrar-port
                               5060
    options
                              max-udp-length=0
                                  enabled
    extra-method-stats
sip-interface
                              CiscoWebexRealm
    realm-id
    sip-port
         address
                                   10.1.3.4
         port
                                 5061
         transport-protocol
                                      TLS
         tls-profile
                                  TLSWebex
         allow-anonymous
                                       agents-only
    spl-options
HeaderNatPublicSiplflp=20.96.25.165, HeaderNatPrivateSiplflp=10.1.3.4
    out-manipulationid
                                  ToCiscoWebex
    user-agent
                                Oracle/VM/9.0.0p4
sip-interface
                              SIPTrunk
    realm-id
    sip-port
         address
                                    155.212.214.90
         allow-anonymous
                                        agents-only
```

```
sip-port
         address
                                     155.212.214.90
         transport-protocol
                                       TCP
         allow-anonymous
                                         agents-only
         out-manipulationid
                                        ToPSTN
sip-monitoring
    match-any-filter
                                  enabled
    monitoring-filters
    ladder-diagram-rows
                                     500
steering-pool
    ip-address
                                 10.1.3.4
    start-port
                               10000
    end-port
                                20000
    realm-id
                                CiscoWebexRealm
steering-pool
    ip-address
                                 155.212.214.90
    start-port
                               10000
    end-port
                                20000
    realm-id
                                SIPTrunk
system-config
    transcoding-cores
                                    1
tls-profile
                                TLSWebex
    name
                                   CloudSBCSolLab
    end-entity-certificate
    trusted-ca-certificates
                                   GoDaddyRootCA
                            WebexRootCA
                            GoDaddyIntermediate
    mutual-authenticate
                                    enabled
sip-manipulation
                                    ToCiscoWebex
    name
    header-rule
                                    addplus
         name
         header-name
                                        Contact
         action
                                   manipulate
         comparison-type
                                        pattern-rule
         msg-type
                                     request
                                     Invite
         methods
         element-rule
                                         TenDigits
              name
                                        uri-user
              type
              action
                                        replace
              comparison-type
                                             pattern-rule
              match-value
                                           ^[0-9]{10}$
                                           \+1+$ORIGINAL
              new-value
         element-rule
              name
                                         ElevenDigits
                                        uri-user
              type
              action
                                        replace
              comparison-type
                                             pattern-rule
              match-value
                                           ^[0-9]{11}$
```

```
header-rule
                                  ChangeContactHost
        name
                                      Contact
        header-name
                                  manipulate
        action
        msg-type
                                   any
        methods
                                   ACK, INVITE
        element-rule
             name
                                       contacthost
                                      uri-host
             type
             action
                                      replace
                                        $TRUNK_GROUP_CONTEXT
             new-value
    header-rule
                                  AddContactOptions
        name
        header-name
                                      Contact
        action
                                  add
                                  request
        msg-type
        methods
                                 OPTIONS
                               <sip:ping@"+$TRUNK_GROUP_CONTEXT+":5061;transport=tls>"
        new-value
sip-manipulation
    name
                              RemoveDTG
    description
    split-headers
    join-headers
    header-rule
                                   StripDTG
        name
                                   Request-URI
        header-name
        action
                                  manipulate
        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
```

```
mime-sdp-rule
                                    Changealine
         name
         msg-type
                                     any
                                     Invite
         methods
         action
                                   manipulate
         comparison-type
                                        pattern-rule
         match-value
         new-value
         sdp-media-rule
                                         deleteattributes
              name
              media-type
                                           video
                                        manipulate
              action
              comparison-type
                                             pattern-rule
              match-value
              new-value
              sdp-line-rule
                   name
                                              deletertcp
                   type
                   action
                                             delete
                   comparison-type
                                                  pattern-rule
                   match-value
                                                (rtcp)(.*)
                   new-value
sip-manipulation
    name
                               ToPSTN
    description
    split-headers
    join-headers
    header-rule
         name
                                    StripDTG
                                        Request-URI
         header-name
                                   manipulate
         action
                                        case-sensitive
         comparison-type
         msg-type
                                     request
         methods
                                     Invite
         match-value
         new-value
        element-rule
                                         stripdtg
              name
              parameter-name
                                             dtg
                                        header-param
              type
              action
                                        delete-element
              match-val-type
                                            any
              comparison-type
                                             case-sensitive
              match-value
              new-value
```

header-rule name DeleteXBroadworks X-BroadWorks-Correlation-Info header-name action delete case-sensitive comparison-type msg-type any BYE, INVITE, OPTIONS methods match-value new-value header-rule **DeleteSessionID** name Session-ID header-name delete action comparison-type case-sensitive msg-type any BYE, INVITE, OPTIONS methods match-value new-value header-rule DeleteRecvInfo name Recv-Info header-name delete action comparison-type case-sensitive msg-type any methods BYE, INVITE, OPTIONS ice-profile webexice name stun-conn-timeout 0 stun-keep-alive-interval 10 stun-rate-limit 15 **NONE** mode rtcp-stun enabled



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Oracle Corporation, World Headquarters 2300 Oracle Way Austin, TX 78741, USA Worldwide Inquiries

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