

Oracle SBC integration with Cisco CUCM and Generic Sip Trunk

**Technical Application Note** 



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# **Revision History**

Version	Description of Changes	Date Revision Completed
1.0	Oracle SBC integration with Cisco CUCM and Generic SIP Trunk	10 <sup>th</sup> May 2022

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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 Call Manager (Cisco CUCM).

### 2. Document Overview

This Oracle technical application note outlines how to configure the Oracle SBC to interwork between Generic Sip Trunk with on premises Cisco CUCM. The solution contained within this document has been tested using Oracle Communication SBC with **OS900p2** 

Please find the related documentation links below:

### 2.1. Cisco Call Manager (Cisco CUCM)

Cisco Unified Call Manager provides industry-leading reliability, security, scalability, efficiency, and enterprise call and session management and is the core call control application of the collaboration portfolio.

It should be noted that while this application note focuses on the optimal configurations for the Oracle SBC in an enterprise **Cisco CUCM 12.5** environment, the same SBC configuration model can also be used for other enterprise applications with a few tweaks to the configuration for required features.

In addition, it should be noted that the SBC configuration provided in this guide focuses strictly on the Cisco CUCM Server associated parameters. Many SBC applications may have additional configuration requirements that are specific to individual customer requirements. These configuration items are not covered in this guide. Please contact your Oracle representative with any questions pertaining to this topic.

For additional information on CUCM 12.5, please visit

https://www.cisco.com/c/en/us/products/unified-communications/unified-communications-manager-version-12-5/index.html

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 CUCM 12.5 version using Oracle Enterprise SBC. There will be steps that require navigating the CUCM 12.5 server configuration, Oracle SBC GUI interface, understanding the basic concepts of TCP/UDP, IP/Routing, DNS server and SIP/RTP are also necessary to complete the configuration and for troubleshooting, if necessary.

#### 3.2. Requirements

- Fully functioning Cisco CUCM 12.5 version
- Oracle Enterprise Session Border Controller (hereafter Oracle SBC) running 9.0.0 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	Cisco CUCM Version
Revision 1	9.0.0	12.5

#### 3.3. Architecture



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

- Phase 1 Configuring the Cisco Unified Call Manager v12.5 for Oracle SBC.
- Phase 2 Configuring the Oracle SBC.

# 4. Configuring the Cisco Call Manager (Cisco CUCM)

Please login to Cisco CUCM admin web GUI with proper login credentials (Username and password). After that, perform the steps below in the given order.

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Cisco Unified CM Administration	Username admin Password Login Reset
Copyright © 1999 - 2016 Cisco Systems, Inc. All rights reserved. This product contains cryptographic features and is subject to United States and local country I products does not imply third-party authority to import, export, distribute or use encryption. Ir U.S. and local country laws. By using this product you agree to comply with applicable laws an product immediately.	laws governing import, export, transfer and use. Delivery of Cisco cryptographic mporters, exporters, distributors and users are responsible for compliance with d regulations. If you are unable to comply with U.S. and local laws, return this
A summary of U.S. laws governing Cisco cryptographic products may be found at our Export C	Compliance Product Report web site.
For information about Cisco Unified Communications Manager please visit our Unified Commun	nications System Documentation web site.

## 4.1. Configuring a new SIP Trunk

01) Go to Device ----- Trunk ----- Add New

02) Select Trunk Type – SIP Trunk and then Click Next

03) In the Device Name field, enter the SIP Trunk name and optionally provide a description.

04) In the Device Pool drop-down list, select a device pool id created already else select Default

05) Enter the Destination Address and Destination Port of the SBC under SIP Information.

06) Select appropriate SIP profile and SIP trunk security profile from the dropdown menu.

07) Click Save

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Trunk Configuration					Related Links: 🛽	Back To	Find/Lis	st 🔻	Go
Next									
– Status –									
i Status: Ready									
□ Trunk Information									
Trunk Type*	SIP Trunk		•						
Device Protocol*	SIP		•						
Trunk Service Type*	None(Default)		•						
Next indicates req	uired item.								

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Trunk Configuration			Related Links: <b>Back To Find/List</b>
🔚 Save 🗙 Delete			
Product:	SIP Trunk		
Device Protocol:	SIP		
Trunk Service Type	None(Default)		
Device Name*	CUCM-SBC		
Description			
Device Pool*	Default	¥	
Common Device Configuration	< None >	<b>v</b>	
Call Classification*	Use System Default	¥	
Media Resource Group List	< None >	¥	
Location*	Hub_None	¥	
AAR Group	< None >	¥	
Tunneled Protocol*	None	¥	
QSIG Variant*	No Changes	$\checkmark$	
ASN.1 ROSE OID Encoding*	No Changes	$\vee$	
Packet Capture Mode*	None	¥	
Packet Capture Duration	0		
Media Termination Point Required			
Retry Video Call as Audio			
Path Deplacement Cuppert			

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Destination										
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MTP Preferred Originating Codec*	711ulaw		v	1						
BLF Presence Group*	Standard Pres	ence group	Ŷ							
SIP Trunk Security Profile*	Non Secure S	IP Trunk Profile	Ý							
Rerouting Calling Search Space	< None >		v							
Out-Of-Dialog Refer Calling Search Space	< None >		×							
SUBSCRIBE Calling Search Space	< None >		v							
SIP Profile*	Standard Sip	Profile - Options Enabled ISR	~	View Details						
DTMF Signaling Method*	RFC 2833	•	~							
┌ Normalization Script										
Normalization Script < None >		~								
		•								
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////

### 4.2. Configure a new Route Pattern

01) Go to Call Routing ------ Route/Hunt ------ Route Pattern and click Add New
02) Enter a Route Pattern according to the network requirements and calling plan.
03) From the Gateway/Route List drop-down list, select the created SIP Trunk device name.

04) Click Save. We can create other route patterns in the same way as shown below.

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System      Call Routing      Media Resources	Advanced Features 👻	Device 🔻	Application •	User Ma	nagement 🔻	Bulk Administration 👻	Help 🔻	
Route Pattern Configuration								Related Links: Back To Find/List 🗸
🔜 Save 🗶 Delete 🕒 Copy 🕂 Add N	lew							
Status								
(i) Status: Ready								
Pattern Definition								
Route Pattern*	1XXXXXXXXXX							
Route Partition	< None >			v				
Description	Route to SBC							
Numbering Plan	Not Selected			V				
Route Filter	< None >			V				
MLPP Precedence*	Default			¥				
Apply Call Blocking Percentage								
Resource Priority Namespace Network Domain	< None >			¥				
Route Class*	Default			~				
Gateway/Route List*	CUCM-SBC			~	( <u>Edit</u> )			
Route Option	Route this pattern							
	$\bigcirc$ Block this pattern	No Error		۷				

The route patterns that has been created is shown below:

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Find Route	Patterns where Pattern	✓ begins with ✓	Find Clear Filter	r 🗘 🛥						
	Pattern 📩	Description	Partition	Route Filter	Associated D	evice Copy				
	<u>1XXXXXXXXXXX</u>	Route to SBC			CUCM-SBC	ß				
	<u>91XXXXXXXXXX</u>	Route to SBC			CUCM-SBC	ß				
Add Ne	w Select All Clear All Delete Selected									

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The created SIP trunk associated with the route pattern is shown below:

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					Select item (	or enter search text	tν								
		Name *	Description	Calling Search Space	Device Pool	Route Pattern	Partition	Route Group	Priority	Trunk Type	SIP Trunk Status	SIP T	runk Duration	SIP Trunk Se	curity Profile
		CUCM-			<u>Default</u>					SIP	Full Service	Time In Full Ser	vice: 9 days 16 hours	Non Secure	SIP Trunk
		CUCM- SBC			<u>Default</u>					SIP Trunk	Full Service	Time In Full Ser minutes	vice: 0 day 0 hour 41	Non Secure Profile	SIP Trunk
		<u>CUCM-</u> <u>SBC</u>			<u>Default</u>	91XXXXXXXXXXXXX				SIP Trunk	Full Service	Time In Full Ser minutes	vice: 0 day 0 hour 41	<u>Non Secure</u> <u>Profile</u>	<u>SIP Trunk</u>
		sbcce			<u>Default</u>					SIP Trunk	No Service	Time not in Full hours 33 minute	Service: 7 days 19 es	Non Secure Profile	SIP Trunk
Add Net	N Select A	Clear All	Delete Sele	cted Reset	t Selected										

# 4.3. End User Configuration

- 01) Go to User Management ---- End User and click Add New
  02) Enter in your User ID, password, pin, and Last Name
  03) You must also enter in a password in the Digest Credentials and Confirm.
  04) Click Save (remember the User ID and Password and DN of the device)

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End User (	End User Configuration Related Links: Back to Find List Users											
Save	X Delete	Add New										
Status												
(i) Statu	is: Ready											
User Info	ormation ——											
User Statu	ıs	Enabled Local User	r									
User ID*		isrvoip1										
Password		•••••	•••••	•••••	Edit Cre	edential						
Confirm Pa	assword	•••••	•••••	•••••								
Self-Servic	ce User ID	18507904044										
PIN		•••••	•••••	•••••	Edit Cre	edential						
Confirm PI	IN	•••••	•••••	•••••								
Last name	*	isrvoip1										
Middle nan	me											
First name	e											
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Title												
Directory l	URI											
Telephone	Number	18507904044										

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End User Configuration			Related Links: Back to Find List Users 🔻 Go
📊 Save 🗙 Delete 🚽	Add New		
Home Number Mobile Number Pager Number Mail ID Manager User ID Department User Locale Associated PC/Site Code Digest Credentials	< None >	▼	
User Profile	Standard (Factory Default) User Profile	View Details	
User Rank*	1-Default User Rank	•	
Service Settings	ified CM IM and Presence (Configure IM and Pr ing information in presence(Requires Exchange	esence in the associated UC Service Profi Presence Gateway to be configured on C	ile) CUCM IM and Presence server)
UC Service Profile	Use System Default	View Details	

## 4.4. Adding SIP Phone in CUCM

- 01) Go to Device ---- Phone and click Add New
- 02) Select Third Party Sip Device (Basic) and click Next
- 03) Enter in a 12 digit MAC address (any dummy MAC address)
- 04) Enter the pertinent information for the SIP DEVICE settings it should mostly be configured the same as a standard phone on your system except for the following settings
  - a) in the owner user ID field select the user you created above
  - b) in the Device Security Profile field select the security profile you created above
  - c) in the Digest User field select the user you created above
- 05) Click Save.
- 06) Configure the line settings for the SIP device the line settings should match the line settings of your standard user's Cisco IP phones

There are no special attributes that we need to worry about on the line configuration.

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Phone Configuration		F	Related Links: Back To Find/List 🗸 🗸
📄 Save 🗶 Delete 📄 Copy 睯 Reset 🧷 Ap	oly Config 🕂 Add New		
Status			
i Status: Ready			
Association	Phone Type		
Modify Button Items	Product Type: Third-party SIP Dev	ice (Basic)	
1 <u>Ine [1] - 18507904044 (no partition)</u>	Device Protocol: SIP		
Unassigned Associated Items	Real-time Device Status		
2 •m: Line [2] - Add a new DN	Registration: Registered with Cisco	Unified Communications Manager CUCM-Cisco.pe.orad	cle.com
ent	IPv4 Address: 10.232.50.2		
	Active Load ID: None		
	Download Status: None		
	Device Information		
	Device is Active		
	A Device is not trusted		
	MAC Address*	00AABB11CCFF	
	Description	ISRVoip1	
	Device Pool*	Default	View Details
	Common Device Configuration	< None >	View Details
	Phone Button Template*	Third-narty SIP Device (Basic)	v

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		Phone Button Template*	Third-party SIP Device (Basic)	~	_		
		Common Phone Profile*	Standard Common Phone Profile	View Deta	ills		
		Calling Search Space	< None >	~			
		AAR Calling Search Space	< None >	~			
		Media Resource Group List	< None >	*			
		Location*	Hub_None	¥			
		AAR Group	< None >	~			
		Device Mobility Mode*	Default	View Curr	ent Device Mobility Settings		
		Owner	User O Anonymous (Public/Shared Space)				
		Owner User ID*	isrvoip1	*			
		Mobility User ID	< None >	¥			
		Use Trusted Relay Point*	Default	¥			
		Always Use Prime Line*	Default	¥			
		Always Use Prime Line for Voice Message*	Default	¥			
		Geolocation	< None >	¥			
		□ Ignore Presentation Indicators (interna	l calls only)				
		✓ Logged Into Hunt Group					
		Remote Device					

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	SIP Profile*	Standard Sip Profile - Options Enabled ISR	View Details	
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	MLPP Domain < Non Confidential Access Mode < Non	e>		
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## 4.5. Associating End User to Phone

- 01) Go to User Management ----- End Users and search for the sip user you created above, once you find it, click on it
- 02) Scroll down to Device Association and click on the Device Association button
- 03) Locate and select the sip device you created above
- 04) Check the checkbox next to this device and click Save Selected/Changes
- 05) Click Go next to the Back to User related link near the upper right-hand corner
- 06) Click Save one more time on the End User Configuration screen.

End User Configuration	× +				-	
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nd User Configuration			Rel	ated Links: Back to Find	List Use	s V G
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Manager User ID						
Department						
User Locale	< None >	•				
Associated PC/Site Code						
Digest Credentials						
Confirm Digest Credentials						
User Profile	Standard (Factory Default) User Profile	View Details				
User Rank*	1-Default User Rank	•				
Service Settings						
A Home Cluster						
Enable User for Un	ified CM IM and Presence (Configure IM and I	Presence in the associated UC	Service Profile)			
Include meet	ing information in presence/Requires Exchange	e Presence Gateway to be co	infigured on CUCM IM and I	Presence server)		
UC Service Profile	Use System Default	View Details				
Device Information						
Controlled Devices	5112000002295352133	-	Devides Association			
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With these steps, the CUCM configuration is complete.

# 5. Configuring the SBC

This chapter provides step-by-step guidance on how to configure Oracle SBC for Cisco Call Manager (Cisco CUCM) and Generic SIP Trunk. If the Oracle SBC being deployed is new, with no existing configuration, the simplest way to configure it to interface with Cisco Call Manager (Cisco CUCM) is by utilizing the <u>Configuration Assistant</u> feature.

### 5.1. Validated Oracle SBC version

Oracle conducted tests with Oracle SBC 9.0 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 (Starting from SBC 9.0 version)
- AP 4900 (Starting from SBC 9.0 version)
- VME

# 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
Starting	tAlgd
Starting	tRadd
Starting	tEbmd
Starting	tSipd
Starting	tH323d
Starting	tbfdd
Starting	tIPTd
Starting	tSecured
Starting	tAuthd
Starting	tCertd
Starting	tIked
Starting	tTscfd
Starting	tFcgid
Starting	tauditd
Starting	tauditpusher
Starting	tSnmpd
Starting	tIFMIBd
Start pla	atform alarm
Starting	display manager
[nitializ	zing /opt/ Cleaner
Starting	tLogCleaner task
Bringing	up shell
Starting	acliMgr
bassword	secure mode is enabled
Admin Sec	curity is disabled
Password	

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.



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

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

```
NN4600-139(configure) # bootparam
'.' = clear field; '-' = go to previous field; g = quit
Boot File
                  : 10.138.194.139
                          : /boot/nnSCZ900p2.bz
IP Address
                   : 0
: 255.255.255.192
: 10.138.194.129
VLAN
Netmask
Gateway
IPv6 Address
IPv6 Gateway
FTP username : vxftp
FTP password : ********
Flags:Target Name: NN4600-139Console Device: COM1Console Baudrate: 115200
Flags
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: 30)
NN4600-139(configure)#
NN4600-139(configure)#
```

Note: There is no management IP configured by default.

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

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



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

Save the changes and reboot the SBC.

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.

name	webServerInstance
state	enabled
realm	
ip-address	
http-state	enabled
http-port	80
https-state	disabled
https-port	443
http-interface-list	GUI
http-file-upload-size	0
tls-profile	
auth-profile	
last-modified-by	Q
last-modified-date	2020-10-06 00:28:26

# 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 <u>http://<SBC\_MGMT\_IP</u>>.

	0	
		Sign in to E-SBC
		Enter your details below
ORACLE		Username
Enterprise Session Border Controller		
		Password
		Required
		SIGN IN

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



Go to Configuration as shown below, to configure the SBC

			Dashboard	Configuration	Monitor and Trace	Widgets	System
🚯 Wizards 👻	Commands 👻				Save Verify	Discard	Search
media-manager	•	Configuration Objects					
security	Þ						
session-router	<b>b</b>	Name	Description				
		access-control	Configure a static or dynamic access control list				
system	•	account-config	Configure Quality of Service accounting				
		authentication-profile	Configure authentication profile				
		certificate-record	Create, generate, and import a certificate				
		class-policy	Configure classification profile policies				
		codec-policy	Create and apply a codec policy to a realm and an agent				
		filter-config	Create a custom filter for SIP monitor and trace				
		fraud-protection	Configure fraud protection				
		host-route	Insert entries into the routing table				
		http-client	Configure an HTTP client				
		http-server	Configure an HTTP server				*
		Displaying 1 - 11 of 42					
Show All							

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

https://docs.oracle.com/en/industries/communications/enterprise-session-bordercontroller/9.0.0/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

	Session Border Controller					admi
<ul> <li>C. The defined second second second field (11)</li> </ul>			Dashboard	Configuration	Monitor and Trace	Widgets Sy
🐼 Wizards 💌					Save Verify	Discard
http-client	Modify System Config					Show Configura
http-server network-interface ntp-config phy-interface redundancy-config	Hostname Description	OracleSBC				
snmp-community	Location Mib System Contact					
system-config	Mib System Name Mib System Location					
trap-receiver v	Acp TLS Profile	Delete				

Please enter the default gateway value in the system config page.

	Session Border Controller						adn
				Dashboard	Configuration	Monitor and Trace	Widgets S
🐼 Wizards 👻						Save Verify	Discard
http-client	Modify System Config						Show Configu
http-server	Displaying 0 - 0 of 0 Options						
network-interface	C-11 T						
ntp-config	Call Trace	enable					
phy-interface	Default Gateway	10.138.194.129					
redundancy-config	Restart	🖌 enable					
snmp-community	Telnet Timeout	0	(Range: 065535)				
spl-config	HTTP Timeout	0	(Range: 065535)				
system-config	Alarm Threshold	2	(Range. 0zo)				
tdm-config	Add						
trap-receiver	ОК	Delete					
Show All							

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

https://docs.oracle.com/en/industries/communications/enterprise-session-bordercontroller/9.0.0/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 M10 for Generic Trunk side and M11 for Cisco side.

Parameter Name	Generic Sip Trunk side (M00)	Cisco side (M01)
Slot	0	1
Port	0	0
Operation Mode	Media	Media

Please configure M00 interface as below.

	ORACLE Enterprise Session Border Controller									
NN4600-139 10.138.194.139 SCZ9.0.0	NN4600-139 10.138.194.139 SCZ9.0.0 Patch 2 (Build 211) Dashboard Configuration Monitor and Trace W									
Configuration View Configuration	Q					Discard	😧 Verify	Save		
▲ host-route	Modify Phy Interface									
http-client	Name	M00						^		
http-server	Operation Type	Media	•							
network-interface	Port	0	(Range: 05)							
ntp-config	Slot	0	(Range: 0)					- 11		
phy-interface	Virtual Mac							- 11		
redundancy-config	Admin State	✓ enable								
snmp-community	Auto Negotiation	✓ enable								
spl-config	Duplex Mode	FULL	<b>•</b>					~		
system-config	OK	Bark						•		

Please configure M01 interface as below

¥r wost visiteu							Other book
	e Session Border Controller					Û 🔺	admin 🔻
NN4600-139 10.138.194.139 SCZ	9.0.0 Patch 2 (Build 211)		Dashboa	d Configuration	Monitor and Trace	Widgets	System
Configuration View Configurat	tion Q				Discard	😧 Verify	🖹 Sav
host-route	Modify Phy Interface						
http-client	Name	M01					^
http-server	Operation Type	Media 💌					
network-interface	Port	1	(Range: 05)				
ntp-config	Slot	0	(Range: 0)				
phy-interface	Virtual Mac						
redundancy-config	Admin State	✓ enable					
snmp-community	Auto Negotiation	✓ enable					
spl-config	Duplex Mode	FULL v					
system-config v Show All	ОК	Back					,

2///0

## 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	Generic Sip Trunk Side Network Interface(M00)	Cisco side Network Interface(M01)
Name	M00	M01
Host Name		
IP Address	155.212.214.105	10.232.50.79
Net Mask	255.255.255.0	255.255.255.0
Gateway	155.212.214.65	10.232.50.1

Please configure network interface M00 as below

ORACL	E Enterpris	e Session Border Controller						Q 🗸	admin 🔻
NN4600-139 10.1	38.194.139 SCZ	9.0.0 Patch 2 (Build 211)			Dashboard	Configuration	Monitor and Trace	Widgets	System
Configuration	View Configurat	tion Q					Discard	😧 Verify	Save
media-manager	•	Modify Network Interfa	асе						
security	•								^
session-router	•	Name	M00						
system		Sub Port Id	0	(Range: 04095)					- 11
fraud-protection		Description							
host-route									
http-client									
http-server		Hostname							
network-interface	2	IP Address	155.212.214.105						
		Pri Utility Addr							~
ntp-config	~								
Show All		OK	Back						

11/1/10

Similarly, configure network interface M01 as below

ORACL	.E En	terprise	Session Border Controller							Û 🔺	admin 🔻
NN4600-139 10.1	138.194.13	9 SCZ9	2.0.0 Patch 2 (Build 211)				Dashboard	Configuration	Monitor and Trace	Widgets	System
Configuration	View Co	onfiguratio	on Q						Discard	😧 Verify	B Save
media-manager	•	^	Modify Network Interface								
security	►										^
session-router	Þ		Name	M01	▼						
system			Sub Port Id	0		(Range: 04095)					
fraud-protection			Description								
host-route											
http-client											
http-server			Hostname	10.232.50.79							
network-interface	е		IP Address	10.232.50.79							
			Pri Utility Addr								~
ntp-config		v									
Show All			OK	Back							

# 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.

In addition to the above config, please set the max and min untrusted signaling values to 1. Go to Media-Manager->Media-Manager

ORACI	_E Enterprise S	Session Border Controller					ć
				Dashboard	Configuration	Monitor and Trace	Widgets
🔅 Wizards 🔻	🔅 Commands 🔻					Save Verify	Discard
media-manager codec-policy	*	Modify Media Manager					
media-manage	r	State	✓ enable				
media-policy		Flow Time Limit	86400	(Range: 04294967295)			
realm-config		Initial Guard Timer	300	(Range: 04294967295)			
reality config		Subsq Guard Timer	300	(Range: 04294967295)			
steering-pool		TCP Flow Time Limit	86400	(Range: 04294967295)			
security	•	TCP Initial Guard Timer	300	(Range: 04294967295)			
session-router	•	TCP Subsq Guard Timer	300	(Range: 04294967295)			
system	•	Hnt Rtcp	enable				
		Algd Log Level	NOTICE .				
		Mbcd Log Level	NOTICE				
		OK	Delete				
Show All							

ORACL	Enterprise	Session Border Controller				a
				Dashboard Configuration	Monitor and Trace Widgets	
🔅 Wizards 🔻	🔅 Commands 🔻				Save Verify Discard	
media-manager	•	Modify Media Manager				
codec-policy			1000	[minBerow iEr Horizio]		
media-manage		Media Policing	✓ enable			
media-policy		Max Arp Rate	10	(Range: 0100)		
media policy		Max Signaling Packets	0	(Range: 04294967295)		
realm-config		Max Untrusted Signaling	1	(Range: 0.100)		
steering-pool		Min Untrusted Signaling	1	(Range: 0.100)		
security	•	Tolerance Window	30	(Range: 04294967295)		
session-router	•	Untrusted Drop Threshold	0	(Range: 0100)		
system		Trusted Drop Threshold	0	(Range: 0100)		
5,51011		Acl Monitor Window	30	(Range: 53600)		
fraud-protection	n	Trap On Demote To Deny	enable			
host-route						
Show All	~	ОК	Delete			

# 6.7. 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	Generic SIP Trunk	Cisco Side
Identifier	SIPTrunk	CUCMRealm
Network Interface	M00	M01
Mm in realm		
FQDN		
Media Sec policy	RTP	RTP
Access Control Trust Level	High	High

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

ORACL	E Enterpris	e Session Border Controller					Û 🔺	admin 🔻
NN4600-139 10.	138.194.139 SC2	29.0.0 Patch 2 (Build 211)		Dashboard	Configuration	Monitor and Trace	Widgets	System
Configuration	View Configura	tion Q				Discard	😧 Verify	Save
media-manager		Modify Realm Config						
codec-policy								^
media-manager		Identifier	SIPTrunk					
media-policy		Description						
realm-config								
steering-pool								
security	•	Addr Prefix	0.0.0.0					
session-router	•	Network Interfaces	M00:0.4 ×					
system	•	Media Realm List						
		Mas In Doolas						~
Show All		ОК	Back					

ORACI	_E Ent	erprise	Session Border Controller				
					Dashboard	Configuration	Monitor and Trace
🔅 Wizards 🔻	🔅 Comm	ands 🔻					Save Verify
media-manager	•	^	Add Realm Config				
codec-policy			Out Translationid		•		
media-manage	r		In Manipulationid		•		
media-policy			Out Manipulationid		•		
realm-config			Average Rate Limit	0	(Range: 0., 4294967295)		
steering-pool			Access Control Trust Level	high			
security	►		Invalid Signal Threshold	0	(Range: 04294967295)		
session-router	►		Maximum Signal Threshold	0	(Range: 04294967295)		
system	•		Untrusted Signal Threshold	0	(Range: 04294967295)		
fraud-protectio	n		Nat Trust Threshold	0	(Range: 065535)		
		~	May Endpoints Dor Nat				
host-route			ОК	Back			

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

ORACL	E Enterprise	Session Border Controller					Û 🔺	admin 🔻
NN4600-139 10.1	38.194.139 SCZ9.	0.0 Patch 2 (Build 211)		Dash	board Configurat	on Monitor and Trace	Widgets	System
Configuration	View Configuratio	n Q				Discard	😧 Verify	🖹 Save
media-manager	▼	Modify Realm Config						
media-manager		Identifier	CUCMRealm					^
media-policy		Description						
realm-config								
steering-pool								
security	•	Addr Prefix	0.0.0.0					
session-router	•	Network Interfaces	M01:0.4 🗙					
system	•	Media Realm List						
		kim in Doalm						~
Show All		ОК	Back					

ORACI	_E Ent	erprise	Session Border Controller				
					Dashboard	Configuration	Monitor and Trace
🔅 Wizards 🔻	🄅 Comma	ands 🔻					Save Verify
media-manager	•	^	Add Realm Config				
codec-policy			Out Translationid		•		
media-manage	r		In Manipulationid		•		
media-policy			Out Manipulationid		•		
realm-config			Average Rate Limit	0	(Range: 04294967295)		
steering-pool			Access Control Trust Level	high	v		
security	►		Invalid Signal Threshold	0	(Range: 04294967295)		
session-router	Þ		Maximum Signal Threshold	0	(Range: 04294967295)		
system	•		Untrusted Signal Threshold	0	(Range: 04294967295)		
fraud-protectio	n		Nat Trust Threshold	0	(Range: 065535)		
host-route Show All		~	May Endnointe Dar Mat	Back			

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/9.0.0/security/securityguide.pdf

### 6.8. 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 Generic SIP Trunk side as below. Set allow-anonymous to agents-only to ensure traffic to this sip-interface only comes from the particular Session agents added to the SBC.

NN4600-139 10.138.194.139 SCZ9.0.0 Patch 2	2 (Build 211)						Dashboard	Configuration	Monitor and Trace	Widgets	System
Configuration View Configuration Q									Discard	😧 Verify	🖹 Sav
media-profile	Modify	SIP In	terface							Show Cor	figuration
session-agent	State		C								^
session-group	Realm ID		CIDTauck								
session-recording-group	Descriptio	on	SIFTURK	•							
session-recording-server											
session-translation											
sip-config	SID Dorte										
sip-feature		/ [	) D								
sip-interface	Action	Select	Address	Port	Transport Protocol	TLS Profile	Allow Anon	ymous	Multi Home Addr	5	
sip-manipulation	:		155.212.214.105	5060	UDP		agents-only	1			
sip-monitoring	:		155.212.214.105	5060	TCP		agents-only				
translation-rules											
system			OV Beak								Y
Show All			OK								

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

ORACLE Enterprise Session	Border Con	troller										Û 🔺	admin 🔻
NN4600-139 10.138.194.139 SCZ9.0.0 Patch	2 (Build 211)								Dashboard	Configuration	Monitor and Trace	Widgets	System
Configuration View Configuration Q											Discard	😧 Verify	🖹 Sa
media-profile	Modify	SIPIn	terface									Show Co	infiguration
session-agent	State												,
session-group	Poolm ID			<ul> <li>enable</li> </ul>									
session-recording-group	Redititit			CUCMRealm		Ŧ							- 1
session-recording-server	Descriptio	n											
session-translation													
sip-config													
sin-feature	SIP Ports												
apreature	D:	/ 6	) <b>1</b>										
sip-interface	Action	Select	Address		Port		Transport Protocol	TLS Profile	Allow Anor	iymous	Multi Home Addr	5	
sip-manipulation	:		10.232.50.79		5060		UDP		agents-only	1			
sip-monitoring	:		10.232.50.79		5060		TCP		agents-only	I.			
translation-rules													
system													
•			ОК	Back									

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

### 6.9. 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. 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 Generic SIP Trunk

- Host name to "68.68.117.67", port to 5060
- realm-id needs to match the realm created for the Generic SIP Trunk

	Session Border Controller					Û 🔺	admin 🔻
NN4600-139 10.138.194.139 SCZ9.	.0.0 Patch 2 (Build 211)		Dashboard	Configuration	Monitor and Trace	Widgets	System
Configuration View Configuration	on Q				Discard	Ø Verify	🖹 Save
media-profile	Modify Session Agent					Show Cor	figuration
session-agent	Hostname	68.68.117.67					^
session-group	IP Address	68.68.117.67					
session-recording-group	Port	5060	(Range: 0,102565535)				
session-recording-server	State	✓ enable					
session-translation	App Protocol	SIP	,				
sip-config	Арр Туре		r				
sip-feature	Transport Method	UDP	·				
sip-interface	Realm ID	SIPTrunk	7				
sip-manipulation	Egress Realm ID		,				~
sip-monitoring  Show All	OK	Back					

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

- Host name to FQDN of CUCM which is "CUCM-Cisco.pe.oracle.com" in our example. We can also give Cisco CUCM IP address if there is no host name configured.
- The same FQDN value should be configured in Cisco CUCM under System ---Enterprise Parameter ----Cluster FQDN.

	e Session Border Controller						Ĥ ▲ 6
				Dashboard	Configuration	Monitor and Trace	Widgets
🔅 Wizards 🔻	•					Save Verify	Discard
media-profile	Add Session Agent						
session-agent							
session-group	Hostname	CUCM-Cisco.pe.oracle.com					
session-recording-group	IP Address	10.232.50.89					
session-recording-server	Port	5060	(Range: 0,102565535)				
session-translation	State	✓ enable					
sip-config	App Protocol	SIP					
sin_feature	Арр Туре						
sin interface	Transport Method	UDP+TCP					
sip-interface	Realm ID	CUCMRealm					
sip-manipulation	Egress Realm ID						
sip-monitoring	OK	Pack					
Show All	UK	Dack					

2/11/1/1/1722/1/185

← → C 🔺 Nct secure   10.232.50.89/ccmadr	nin/serviceParamEd t.do?service=11&showall=fals	e		Θ	:
Cisco Unified CM Administrat	tion <sup>ns</sup>	Navigation Cisco Unifed CM Administratio admin Search Docurrentation	n Atout	• Log	G0 jout
System ▼ Call Rouring ▼ Media Resources ▼ Advanced Featu	res - Device - Application - User Management -	Bulk Administration 👻 Help 👻			
interprise Parameters Configuration					
🔜 Save 🤣 Set to Default   🍟 Reset 🥒 Apply Config					
Syncing Mode for Enterprise Groups *	Differential Sync	<ul> <li>Differential Sync</li> </ul>			-
Service Manager TCP ports parameters					
Service Manager TCP Server communication port number	8883	8888			
Service Manager TCP C ient communication port number	8889	8889			
CR5 Application Parameters					
Auto Attendant Installed.*	false				
PCC Express Installed *	false				
Clusterwide Domain Configuration				_	
Organization Top Level Domain	pe.oracle.com				
Cluster Fully Qualified Domain Name	CUCM-Cisco.pe.oracle.com				
Denial-of-Service Protection					
Denial-of-Service Protection *	True	▼ True			
TLS Handshake Timer					
TLS Handshake Timer.*	60	60			
TLS Resumption Timer					
TLS Resumption Timer_*	3603	3600			
				_	1

# 6.10. 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 side to Generic	SIP trunk side, Use the	e below local –policy

Enterprise S	ession Border Controller					ΰ 🔹 e
			Dashboard	Configuration	Monitor and Trace	Widgets
🔅 Wizards 🔻					Save Verify	Discard
account-config	Add Local Policy					
filter-config						
ldap-config	From Address	**				
local-policy	To Address	* ×				
local-routing-config	Source Realm	CUCMRealm 🗙				
media-profile	Description					
session-agent						
session-group						
session-recording-group						
session-recording-server	State	✓ enable				
session-translation	Policy Priority	none 🔻				
~	ОК	Back				
Show All						

ORACL	E Ente	rprise S	ession Bo	order Co	ntroller								Û 🔺	admin 🔻
NN4600-139 10.1	138.194.139	SCZ9.0	.0 Patch 2 (	Build 211	)					Dashboard	Configuration	Monitor and Tra	ace Widgets	System
Configuration	View Conf	figuration	Q									Disc	card 😧 Verify	B Save
media-manager	+	^	Modify	/ Local	Policy									
security	•													^
session-router	Ŧ													
access-control			State			✓ enable								
account-config			Policy Pri	ority		none		V						
filter-config			Policy Att	ributes										
Idap-config			D:	1	Ē									
local-policy			Action	Select	Next Hop	Realm	Action	Terminate	Cost	State	App Protoco	l Lookup	Next Key	Auth
local-routing-cor	nfig		:		68.68.117.67	SIPTrunk	none	disabled	0	enabled		single		
media-profile														
session-agent		~												~
Show All					ОК	Back								

To route the calls from the Generic SIP Trunk side to Cisco side, Use the below local -policy

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111111

ORACL	E Ente	erprise S	Session Border Controller						Û 🔺	admin 🔻
NN4600-139 10.1	138.194.139	SCZ9.0	).0 Patch 2 (Build 211)			Dashboard	Configuration	Monitor and Trace	Widgets	System
Configuration	View Con	figuratior	Q					Discard	😧 Verify	🖹 Save
media-manager	Þ	^	Modify Local Policy							
security	•									^
session-router	Ŧ		From Address	*X						
access-control			To Address	* X						
account-config			Source Realm	SIPTrunk 🗙						
filter-config			Description							- 1
Idap-config										
local-policy										
local-routing-con	nfig		State							
media-profile			Dolicy Driority							
session-agent			Toncy Frioncy	none	•					~
Show All		Ý	OK	Back						

	erprise S	ession Border Controller										Û 🔺
								Dashboard	Configuration	Monitor an	d Trace	Widgets
🐼 Wizards 🔻	ands 🔻									Save	Verify	Discard
account-config	^	Modify Local Policy										
filter-config												
ldap-config		Description										
local-policy												
local-routing-config												
media-profile		State	🗸 enabl	e								
session-agent		Policy Priority	none		▼							
session-group	١.	Policy Attributes										
session-recording-group		Add										
session-recording-server		Next Hop	Realm	Action	Terminate Rec	ursion	Cost	State	App Protocol	Lookup	Next K	iey
session-translation		CUCM-Cisco.pe.oracle.com	CUCMRealm	replace-uri	disabled	(	0	enabled		single		
Show All	~	OK	Back									

# 6.11. Configure steering-pool

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

Cisco side steering pool.

ORACL	E Enterprise	Session Border Controller						Û 🔺	admin 🔻
NN4600-139 10.1	138.194.139 SCZ9.	0.0 Patch 2 (Build 211)			Dashboard	Configuration	Monitor and Trace	Widgets	System
Configuration	View Configuratio	n Q					Discard	😧 Verify	B Save
media-manager	٣	Modify Steering Pool							
media-manager		IP Address	10.232.50.79						
media-policy		Start Port	25000	(Range: 0,165535)					
realm-config		End Port	29999	(Range: 0,165535)					
at a star a star		Realm ID	CUCMRealm	•					
steering-poor		Network Interface		•					
security	•								
session-router	•								
system	•								
Show All		ОК	Back						

2///

Generic SIP trunk side steering pool.

ORACL	Enterprise S	ession Border Controller						Û 🔺	admin 🔻
NN4600-139 10.1	138.194.139 SCZ9.0	.0 Patch 2 (Build 211)			Dashboard	Configuration	Monitor and Trace	Widgets	System
Configuration	View Configuration	Q					Discard	😧 Verify	🖹 Save
media-manager codec-policy	v	Modify Steering Pool							
media-manager		IP Address	155.212.214.105						
media-policy		Start Port	10000	(Range: 0,165535)					
realm-config		End Port	19999	(Range: 0,165535)					
steering-pool		Realm ID Network Interface	SIPTrunk	•					
security	•			*					
session-router	×								
system	•								
Show All		OK	Back						

### 6.12. Configure Ping Response

To simplify the ORACLE SBC configuration, from GA Release SCZ830m1p7, there is a new parameter introduced under the **Session agent** configuration element. The parameter name is **Ping response**.

#### Ping Response:

When this parameter is enabled, the SBC responds with a 200 OK to all Sip Options Pings it receives from trusted agents. This takes the place of the current Sip Manipulation, RepondOptions.

	Session Border Controller						Ĥ▲ a
				Dashboard	Configuration	Monitor and Trace	Widgets
🚯 Wizards 🔻						Save Verify	Discard
media-profile	Add Session Agent						
session-agent							
session-group	Hostname	CUCM-Cisco.pe.oracle.com					
session-recording-group	IP Address	10.232.50.89					
session-recording-server	Port	5060	(Range: 0,102565535)				
session-translation	State	✓ enable					
sip-config	App Protocol	SIP	,				
sip-feature	Арр Туре		,				
sip-interface	Transport Method	UDP+TCP	,				
sip-manipulation	Realm ID	CUCMRealm	<b>,</b>				
sip-monitoring	Egress Realm ID		7				
¥	ОК	Back					
Show All							

	ssion Border Controller							Δ 🗸	admin 🕠
				Dashboard	Configuration	Monitor and T	race	Widgets	Syste
🔅 Wizards 🔻						Save Ver	ify	Discard	Se
Idap-config	Modify Session Agent							Show Con	ifiguratior
local-policy	Out Translationid	• •							
local-routing-config	Trust Me	enable							
media-profile	Local Response Map	T							
session-agent	Ping Response	🖌 enable	<u></u>						
session-group	In Manipulationid	<b>.</b>							
session-recording-group	Out Manipulationid	<b>v</b>							
session-recording-server	Manipulation String								
session-translation	Manipulation Pattern								
sip-config	Trunk Group								
sip-feature	Max Register Sustain Rate	0	(Range: 0999999999)						
sip-interface	ОК	Back							

## 6.13. SBC config for Cisco Offer less INVITE

When CUCM sends INVITE without SDP towards SBC and in that case, SBC needs to send out INVITE with SDP towards Generic SIP trunk and vice versa. To do that, please set the parameter "Add SDP Invite" as both under sip interface (both SIP interfaces) as highlighted below. When this option is enabled, codecs have to be configured under the parameter "Add SDP profiles". The configured codecs is also shown below.

Note: this is an optional config – configure this only if CUCM sends offer less INVITE towards SBC.

NN4600-139 10.138.194.139 SCZ9.0.0 Patch	2 (Build 211)						Dashboard	Configuration	Monitor and Trace	Widgets	System
Configuration View Configuration Q									Discard	😧 Verify	🕒 Sav
media-profile	Modify	/ SIP In	terface							Show Con	figuration
session-agent	State										^
session-group	Poolm ID		✓ enable								
session-recording-group	Descripti	00	SIPTrunk	V							1
session-recording-server	Descripti	UII									
session-translation											
sip-config											
sip-feature	SIP Ports										- 1
sip-interface	D:	/ 6									_
	Action	Select	Address	Port	Transport Protocol	TLS Profile	Allow Anon	ymous	Multi Home Addr	5	
sip-manipulation	:		155.212.214.105	5060	UDP		agents-only				
sip-monitoring	:		155.212.214.105	5060	тср		agents-only				
translation-rules											
system											~
Show All			OK Back								

						,	0
	Session Border Controller			Dashboard	Configuration	Monitor and Trace	u v a Widgets
🔅 Wizards 💌						Save Verify	Discard
session-group	Modify SIP Interface						Show Confi
session-recording-group	TCP Keepalive	nono	•				
session-translation	Add SDP Invite	both	•				
sip-config	Add SDP In Msg						
sip-feature	P Early Media Header	disabled					
sip-interface	P Early Media Direction						
sip-monitoring	Add SDP Profiles	PCMU X PCMA X					
sti-server	Add SDP Profiles In Msg						
Show All	ОК	Back					

# 6.14. Configure Translation Rules

The translation rules sub-element is where the actual translation rules are created. Go to Session router  $\rightarrow$  translation-rules and create the below rule.

ORACLE Enterprise Set	ssion Border Controller						Û ▲ ac
				Dashboard	Configuration	Monitor and Trace	Widgets
Wizards v Ö Commands v session-group						Save Verify	Discard
session-recording-group	Add Translation Rules						
session-recording-server	Id	addplus					
session-translation	Туре	replace v					
sip-config	Add String	+					
sip-feature	Add Index	0					
sip-interface	Delete String						
sip-manipulation	Delete Index	0	(Range: 0999999999)				
sip-monitoring							
sti-server							
translation-rules							
system							
Show All	ОК	Back					

///

ORACLE Enterprise Se	ession Border Controller						Û 🔺	admi
				Dashboard	Configuration	Monitor and Trace	Widgets	Sy
🔯 Wizards 🔻						Save Verify	Discard	
session-group	Add Translation Rules							
session-recording-group								
session-recording-server	ld	removeplus						
session-translation	Туре	delete 🔻						
sip-config	Add String							
sip-feature	Add Index	0						
sip-interface	Delete String	*						
sip-manipulation	Delete Index	0	(Range: 0999999999)					
sip-monitoring								
sti-server								
translation-rules								
system								
Show All	OK E	Back						

# 6.15. Configure Session Translation Rules

A session translation defines how translation rules are applied to calling and called numbers. Go to Session Router  $\rightarrow$  session-translation and configure the below translation rules.

Add the below translation rule to Cisco side.

	Session Border Controller					- <del>4</del> - 7
			Dashboard	Configuration	Monitor and Trace	Widgets
🔅 Wizards 🔻 🏠 Commands 🔻					Save Verify	Discard
Idap-contig	Add Session Translation					
local-policy						
local-routing-config	ld	toCUCM				
media-profile	Rules Calling	removeplus 🗙				
session-agent	Rules Called	removeplus 🗙				
session-group	Rules Asserted Id					
session-recording-server	Rules Redirect					
session-translation	Rules Isup Cdpn					
sip-config	Rules Isup Cgpn					
sip-feature	Dulae Icun Gn					
sip-interface 🗸	ОК	Back				
Show All						

Add the below translation rule to SIP trunk side as PSTN expects call with + sign.

	Session Border Controller					Û 🔺	admin 🔻
NN4600-139 10.138.194.139 SCZ9	2.0.0 Patch 2 (Build 211)		Dashboard	Configuration	Monitor and Trace	Widgets	System
Configuration View Configuration	on Q				Discard	Ø Verify	🖹 Save
session-group	Modify Session Transla	tion					
session-recording-group	ld						^
session-recording-server		toPSTN					
session-translation	Rules Calling	addPlus 🗙					
sip-config	Rules Called	addPlus 🗙					
sip-feature	Rules Asserted Id						
sip-interface	<pattern> Manipulates the SIP P</pattern>	P-Asserted-Id header					. 1
sip-manipulation	Rules History Info						
sip-monitoring							
translation-rules	Rules Isup Cdpn						
system 🕨 🗸	D. J I C						~
Show All	ОК	Back					

Please add the above session translation rules to Cisco realm as shown below

ORACL	E Enterprise	Session Border Controller					Û 🔺	admin 🔻
NN4600-139 10.1	38.194.139 SCZ9.	.0.0 Patch 2 (Build 211)		Dashboar	Configuration	Monitor and Trace	Widgets	System
Configuration	View Configuratio	n Q				Discard	😧 Verify	🖹 Save
media-manager	•	Modify Realm Config						
codec-policy								^
media-manager		Identifier	CUCMRealm					
media-policy		Description						
realm-config								
steering-pool								
security	•	Addr Prefix	0.0.0.0					
session-router	Þ	Network Interfaces	M01:0.4 🗙					
system	•	Media Realm List						
		Men in Donies						~
Show All		OK B	ack					

11110

2///0

ORACL	E Enterpris	se Session Border Controller						Û 🔺	admin 🔻
NN4600-139 10.1	138.194.139 SC	Z9.0.0 Patch 2 (Build 211)			Dashboard	Configuration	Monitor and Trace	Widgets	System
Configuration	View Configura	ation Q					Discard	😧 Verify	🖹 Save
media-manager	*	Modify Realm Config							
codec-policy		SUP INACTIVE UNIV	enable						^
media-manager		DTLS Srtp Profile		v					
media-policy		Srtp Msm Passthrough	enable						
realm-config		Class Profile		₹					
steering-pool		In Translationid	toPSTN	Ŧ					
security	•	Out Translationid	toCUCM	v					
session-router	v	In Manipulationid							
access-control		Out Manipulationid		Ŧ					
account-config		Average Rate Limit	0		(Range: 04294967295)				
filter-config	~	Access Control Trust Level	hiah	_					~
Show All		ОК	Back						

With this, SBC configuration is complete

# 7. SBC configuration for Cisco Remote Worker

This section of Cisco Remote Worker configuration is included for Cisco remote endpoints that register through the Oracle SBC to the Cisco Call Manager (Cisco CUCM). This would require additional configuration to be configured on the Oracle SBC along with the SIP trunking config as mentioned in the earlier description of the test bed. To complete the particular testing, we have configured Cisco endpoints which will register to Cisco CUCM through the SBC. SBC will handle the calls based on the registration information present in the cache. **Please note that Cisco Remote worker Access side is secured (TLS/SRTP) and Cisco Core side is unsecured (UDP or TCP/RTP)** 

### Note: Remote worker configuration through TLS for Jabber clients is not supported by CUCM.

In order to achieve the requirement, we have made below configuration on the Oracle SBC

Access and Core Realm for Cisco Remote worker Steering Pool associated with the Realm for Cisco Remote worker Sip-interface associated with the Realm for Cisco Remote worker (Optional) A local policy to route the registration requests from this Realm to the SIP Server.

Note -The local-policy element is optional as we can enable the Route to registrar parameter on the sipinterface config to route the requests to the Registrar.

The registrar host and port are configured in the sip-config element on the SBC. The remote endpoint sends register requests from Cisco Access Realm onto the SBC and then SBC registers these endpoints onto the Cisco Core Realm maintaining the registration cache in its database to route inbound calls to these endpoints.

Below are the snippets from the Oracle SBC Web GUI for the Remote worker configuration.

### 7.1. 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	Cisco Access Side	Cisco Core Side
Identifier	CUCMpublicRealm	CUCMCoreRealm
Network Interface	M10	M11
Mm in realm		
FQDN		
Media Sec policy	sdespolicy	RTP
Access Control Trust Level	High	High

In the below example, Realm name is given as CUCMpublicRealm for Cisco Access Side. Please set the Access Control Trust Level as medium for this realm

2///8

ORACI	_E Enterprise	Session Border Controller					Û.▲ a
				Dashboard	Configuration	Monitor and Trace	Widgets
🔅 Wizards 🔻	🔅 Commands 🔻					Save Verify	Discard
media-manager	•	Modify Realm Config					
codec-policy							
media-manage	r	Identifier	CUCMpublicRealm				
media-policy		Description					
realm-config							
steering-pool							
security	Þ	Addr Prefix	0.0.0.0				
session-router	►	Network Interfaces	M10:0.4 🗙				
system	•	Media Realm List					
		Mm In Realm	✓ enable				
		ОК	Back				
Show All	$\sum$						

	LE Enterprise	Session Border Controller					<b>▼</b> (
				Dashboard	Configuration	Monitor and Trace	Widgets
🔅 Wizards 🔻	Commands 🔻					Save Verify	Discard
media-manager	•	Modify Realm Config					
codec-policy		Out Iransiationid					
media-manage	r	In Manipulationid					
media-policy		Out Manipulationid					
realm-config		Average Rate Limit	0	(Range: 04294967295)			
steering-pool		Access Control Trust Level	medium				
security	►	Invalid Signal Threshold	10	(Range: 04294967295)			
session-router	►	Maximum Signal Threshold	30	(Range: 04294967295)			
austom		Untrusted Signal Threshold	10	(Range: 04294967295)			
system	•	Nat Trust Threshold	0	(Range: 065535)			
		Max Endpoints Per Nat	0	(Range: 065535)			
Show All		ОК	Back				

Similarly, Realm name is given as CUCMCoreRealm for Cisco Core side

ORACI	_E Enterprise	Session Border Controller			8		ļ, ▲ a
				Dashboard	Configuration	Monitor and Trace	Widgets
🚯 Wizards 🔻	Commands 🔻					Save Verify	Discard
media-manager	•	Modify Realm Config					
media-manage	r	Identifier	CUCMCoreRealm				
media-policy		Description					
realm-config							
steering-pool							
security	►	Addr Prefix	0.0.0.0				
session-router	•	Network Interfaces	M11:0.4 🗙				
system	•	Media Realm List					
		Mm In Realm	✓ enable				
Show All		ОК	Back				

### 7.2. Enable sip-config

SIP config enables SIP handling in the SBC. Make sure the home realm-id, registrar-domain and registrar-host are configured. Also add the options to the sip-config as shown below.

To configure sip-config, Go to Session-Router->sip-config and in options, add the below

- add max-udp-length =0
- reg-cache-mode=from

This option is used so that the userinfo from the From header is copied to the userinfo of the forwarded Contact header

	Session Border Controller					û ▼ a
			Dashboard	Configuration	Monitor and Trace	Widgets
🔅 Wizards 🔻					Save Verify	Discard
local-routing-config	Modify SIP Config					
media-profile	State	✓ enable				
session-agent	Dialog Transparency	✓ enable				
session-group	Home Realm ID	CUCMCoreRealm				
session-recording-group	Egress Realm ID					
session-recording-server	Nat Mode	None				
session-translation	Registrar Domain	*				
sip-config	Registrar Host	*				
sip-feature	Registrar Port	5060	(Range: 0,102565535)			
sip-interface	Init Timer	500	(Range: 04294967295)			
sip-manipulation	ОК	Delete				

	Session Border Controller				Û.▲ a
			Dashboard Config	guration Monitor and Trace	Widgets
🔅 Wizards 🔻 🧔 Commands 🔻				Save Verify	Discard
session-agent	Modify SIP Config				
session-group	Mouly Sir Coring				
sossion recording group	Trans Expire	32	(Range: 04294967295)		
session-recording-group	Initial Inv Trans Expire	0	(Range: 0999999999)		
session-recording-server	Invite Expire	180	(Range: 04294967295)		
session-translation	Session Max Life Limit	0			
sip-config	Enforcement Profile	•			
sip-feature	Red Max Trans	10000	(Range: 050000)		
sip-interface	Options	max-udp-length=0 🗙			
sip-manipulation		reg-cache-mode=from X			
sip-monitoring	SPL Options				
sti-server	SIP Message Len	4096	(Range: 065535)		
Show All	ОК Е	lelete			

### 7.3. Enable media manager

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

In addition to the above config, please set the max and min untrusted signaling values to 9 which takes care of Access Realm. Go to Media-Manager->Media-Manager

ORACI	_E Enterprise S	Session Border Controller					ā
				Dashboard	Configuration	Monitor and Trace	Widgets
🔅 Wizards 🔻	🚯 Commands 👻					Save Verify	Discard
media-manager codec-policy	•	Modify Media Manager					
media-manage	r-	State	✓ enable				
media-policy		Flow Time Limit	86400	(Range: 04294967295)			
roalm config		Initial Guard Timer	300	(Range: 04294967295)			
realiti-coring		Subsq Guard Timer	300	(Range: 04294967295)			
steering-pool		TCP Flow Time Limit	86400	(Range: 04294967295)			
security	•	TCP Initial Guard Timer	300	(Range: 04294967295)			
session-router	•	TCP Subsq Guard Timer	300	(Range: 04294967295)			
system	•	Hnt Rtcp	enable				
		Algd Log Level	NOTICE 💌				
		Mbcd Log Level	NOTICE				
		ОК	Delete				
Show All							

ORACI	LE Enterprise	Session Border Controller					Û 🔺
				Dashboard	Configuration	Monitor and Trace	Widgets
🔯 Wizards 🔻	Commands v					Save Verify	Discard
media-manager	v	Modify Media Manager	r.				
codec-policy		Red Sync Comp Time	1000	(Range: 04294967295)			
media-manage	er	Media Policing	✓ enable				
media-policy		Max Signaling Bandwidth	10000000	(Range: 7100010000000)			
realm-config		Max Untrusted Signaling	9	(Range: 0100)			
steering-pool		Min Untrusted Signaling	9	(Range: 0100)			
		Tolerance Window	30	(Range: 04294967295)			
security	÷	Untrusted Drop Threshold	0	(Range: 0100)			
session-router	+	Trusted Drop Threshold	0	(Range: 0100)			
system	•	Acl Monitor Window	30	(Range: 53600)			
		Trap On Demote To Deny	enable				
		OK	Delete				
Show All	$\mathbf{D}$						

## 7.4. 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 for Cisco Access side as below:

- Set allow-anonymous to Registered to ensure traffic to this sip-interface only comes from the registered user.
- Set NAT traversal to always for the remote workers to register.
- Enable Registration Caching and Route to Register

	Session Border Con	troller						Û 🗕 🧧
					Dashboard	Configuration	Monitor and Trace	Widgets
🔅 Wizards 👻							Save Verify	Discard
local-routing-config	Modify SIP Int	erface						Show Confi
media-profile	State		enable					
session-agent	Realm ID		CUCMpublicRealm	T				
session-group	Description							
session-recording-group								
session-translation								
sip-config	SIP Ports							
sip-feature	Add							
sip-interface	Address	Port	Transport Protocol	TLS Profile	Allow Anonymous		Multi Home Addrs	
sip-manipulation		5061 OK	TLS	TLSProfile	registered			
Show All		OK	LOUR					

	Session Border Controller					<u></u> . Ĥ ▲ _ 6
			Dashboard	Configuration	Monitor and Trace	Widgets
🔅 Wizards 🔻					Save Verify	Discard
session-agent	Modify SIP Interface					Show Conf
session-recording-group	Nat Traversal	always 💌				
session-recording-server	Nat Interval	30	(Range: 04294967295)			
session-translation	TCP Nat Interval	90	(Range: 04294967295)			
sip-config	Registration Caching	✓ enable				
sip-feature	Registration Interval	300	(Range: 0999999999)			
sip-interface	Route To Registrar	3600 ✓ enable	(Range: 04294967295)			
sip-manipulation	Secured Network	enable				
sip-monitoring	Uri Fqdn Domain					
sti-server	Options					
Show All	ОК	Back				

////

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

	Session Border Cont	roller						Û 🔺 🧯
					Dashboard	Configuration	Monitor and Trace	Widgets
🔅 Wizards 🔻							Save Verify	Discard
Idap-config	Modify SIP Inte	erface						Show Confi
local-routing-config	State		✓ enable					
media-profile	Realm ID		CUCMCoreRealm	v				
session-agent	Description							
session-group								
session-recording-group								
session-recording-server	SIP Ports							
session-translation								
sip-config	Add							
	Address	Port	Transport Protocol	TLS Profile	Allow Anonymous		Multi Home Addrs	
sip-reature	10.232.50.85	5060	UDP		agents-only			
sip-interface 🗸 🗸		ОК	Back					

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

# 7.5. Configure steering-pool

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

Cisco Access side steering pool.

ORACI	_E Enterprise	Session Border Controller								Q. ▼ a
						Dashboard	Configuration	Monito	r and Trace	Widgets
🔯 Wizards 👻	🔅 Commands 👻							Save	Verify	Discard
media-manager		Add Steering Pool								
media-manage	ri -	IP Address								
media-policy		Start Port	40000		(Range: 165535)					
realm-config		End Port	49999		(Range: 165535)					
steering-pool		Realm ID	CUCMpublicRealm	Ŧ						
security	÷.	Network Interface		Ŧ						
session-router	•									
system	È.									
Show All		ОК	Back							

Cisco Core side steering pool.

ORACL	_E Enterprise	Session Border Controller						Û ▲ a
					Dashboard	Configuration	Monitor and Trace	Widgets
🔯 Wizards 🔻	🔅 Commands 🔻						Save Verify	Discard
media-manager	•	Add Steering Pool						
media-manager	r	IP Address	10.232.50.85					
media-policy		Start Port	30000	(Range: 165535)				
realm-config		End Port	35000	(Range: 165535)				
steering-pool		Realm ID	CUCMCoreRealm 🔹					
security	Þ	Network Interface						
session-router	•							
system	►							
		ОКВ	ack					
Show All	$\sum$							

# 7.6. Configure local policy (Optional)

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 Access side to Cisco Core side and vice versa, Use the below local -policy

ORACI	_E En	terprise !	Session Border Controller					Û 🗕 g
					Dashboard	Configuration	Monitor and Trace	Widgets
🔅 Wizards 🔻	Comr	nands 🔻					Save Verify	Discard
media-manager	►	^	Modify Local Policy					
security	►							
session-router	•		From Address	* X				
access-control			To Address	* ×				
account-config			Source Realm	CUCMpublicRealm 🗙				
filter-config			Description					
ldap-config								
local-policy								
local-routing-co	onfig							
media-profile			State	✓ enable				
session-agent			Policy Priority	none 🔻				
Show All		v	OK	Back				

ORACI	_E Ent	erprise S	Session Border Controller								Û 🔺
								Dashboard	Configuration	Monitor and T	race Widgets
🔅 Wizards 🔻	Comm	ands 💌								Save Ver	ify Discard
media-manager	►	^	Modify Local Policy								
security session-router	•		Description								
access-control											
account-config											
filter-config			State	🗸 enable							
ldap-config			Policy Priority	none		•					
local-policy			Policy Attributes								
local-routing-co	onfig		Add								
media-profile			Next Hop	Realm	Action	Terminate Recursion	Cost	State	App Protocol	Lookup	Next Key
session-agent		~	CUCM-Cisco.pe.oracle.com	CUCMCoreRealm Back	replace-uri	disabled	0	enabled	SIP	single	

Cisco Offer less INVITE can happen in the Remote worker scenarios too. In that case, please set the parameter "**Add SDP Invite**" as both and "**Add SDP profiles**" under <u>Cisco</u> <u>Access side sip-interface</u>. The configuration is similar to what we have done in <u>Sec 6.13</u>.

# 8. New SBC config/Deployment Using Configuration Assistant

When you first log on to the E-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 E-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. You can use the Configuration Assistant for the initial set up to make to the basic configuration. Please check "Configuration Assistant Operations" in the <u>Web GUI User Guide</u> and "Configuration Assistant Workflow and Checklist" in the <u>ACLI Configuration Guide</u>

Please note, applying a configuration to the SBC via the Configuration Assistant will overwrite any existing configuration currently applied to the SBC. We highly recommend this only be used for initial setup of the SBC. This feature is not recommended to be used to make changes to existing configurations.

### 8.1. Section Overview and Requirements

This section describes how to use our Configuration Assistant feature as a quick and simple way to configure the Oracle SBC for integration with Cisco Call Manager and Generic SIP Trunking. The pre-requisite is given below.

• SBC running release SCZ840p7 or later which will have this template package by default added to the SBC code.

The following outline assumes you have established initial access to the SBC via console and completed the following steps:

- Configured boot parameters for management access
- Setup Product
- Set Entitlements
- Configured HTTP-Server to establish access to SBC GUI

### 8.2. Initial GUI Access

The Oracle SBC WebGui can be accessed by entering the following in your web browser: http(s)://<SBC Management IP>.

The username and password are the same as that of the CLI. If there is no configuration on the SBC, the configuration assistant will show immediately upon login to the SBC GUI as shown below

Select a PBX Template	Select a SIP Trunk Template
ZoomPhone	Select PBX Template to list the corresponding SIP Side template
Microsoft Teams	
Microsoft ACS	
Cisco	
Avaya Session Manager	
Upload a Configuration	Upload a Template Package
Drag and Drop	Drag and Drop
Select a file or drop one here.	Select a file or drop one here.

////

As we can see, there are some templates of PBX populated in the template and we can select the PBX template that we want to use with Generic SIP trunk and for this document, we have selected Cisco template and once we select that, it asks us to select the SIP trunk template. After we select Generic SIP trunk template, the Next option would be enabled.

Select a PBX Template	Select a SIP Trunk Template	Next 💙
Microsoft ACS	VerizonRetaillpTrunking	
Cisco	TwilioSIPTrunking	
Avaya Session Manager	GenericSipTrunk	
GenericPBX	IntelepeerSipTrunking	
GenesysPureEngage	ATTIPtrunking	
Upload a Configuration	Upload a Template Package	
Drag and Drop	Drag and Drop	
Select a file or drop one here.	E Select a file or drop one here.	

## Click *Next*: The following "Notes" will be displayed related to pre-requisite

Configuration Assistant - Notes	×
Back PBX Template	SIP Trunk Template
Notes for Cisco	Notes for GenericSipTrunk
Warning: - Proceeding with the Configuration Assistant results in erasing the existing configuration.	Warning: - Proceeding with the Configuration Assistant results in erasing the existing configuration.
<ul> <li>Pre-requisites:</li> <li>This template is used to configure Cisco CUCM with Oracle SBC and any SIP trunk</li> <li>Connect Port 0 of the Session Border Controller (SBC) to your network.</li> <li>Ensure that Transcoding resources are installed on your system (Hardware only).</li> <li>Configure at least one Transcoding core on your system (Virtual Machine Edition only).</li> <li>This template supports ONLY UDP/TCP configuration.</li> <li>Enable the Advanced entitlement on the system.</li> <li>Set Session Capacity in the entitlement.</li> <li>Set the system time.</li> </ul>	<ul> <li>Pre-requisites:</li> <li>This template is used to configure generic settings to connect the SBC to PSTN services</li> <li>Connect Port 1 of the Session Border Controller (SBC) to your network.</li> <li>Ensure that Transcoding resources are installed on your system (Hardware only).</li> <li>Configure at least one Transcoding core on your system (Virtual Machine Edition only).</li> <li>This template supports ONLY UDP/TCP configuration.</li> <li>Set Session Capacity in the entitlement.</li> <li>Set the system time.</li> </ul>

////

## Click *Next* and we get the below screen where we need to enter the details for SBC configuration.

figuration A	Assistant - Co	onfigure CUC	M Network	here						
<b>〈</b> Back	1	2	3	4	5	6	7	8	Skip 🖒	
	Configure CUCM Network here	Offerless SDP configuration	Transcoding	Additional Configuration	PSTN Network	PSTN Session Agent	Transcoding	Additional Configuration		
		Let's cc	nfigure the	interface that	at communic	ates with yo	ur CUCM			
			Realm Na	me 🕐						
						Required				
			Enter CUC	.M hostname her	eO					
			Enter the	CUCM IP here 🕲		Required				
			Enter the	CLICM nort here	0					

## 8.3. Configuration Assistant Template Navigation

#### 8.3.1. Page 1-Cisco Call Manager (CUCM) Network

Page 1 of the template is where you will configure the network information to connect Cisco Call Manager. On this page, we will enter the CUCM hostname, IP and port which will be the next hop IP address/hostname for sip signaling to and from your CUCM

Configuration .	Assistant - Co	onfigure CUC	M Network	here						×
<b>K</b> Back	Configure CUCM Network here	Offerless SDP configuration	3 Transcoding	Additional Configuration	PSTN Network	PSTN Session Agent	7 Transcoding	Additional Configuration	Skip 义	<b>^</b>
		Let's co	nfigure the	interface tha	at communic	ates with yo	ur CUCM			
			Realm Nai	me Ø						^
			Enter CUC	M hostname her	e Ø	Required				t.
			Enter the	CUCM ID horo (2)		Required				
			Enter the	CLICM nort here (	0					¥

Next to each field is a help icon. If you hover over the icon, you will be provided with a description or definition of each filed. Also, pay close attention to which fields are listed as "required".

#### 8.3.2. Page 2-Offerless SDP Invite

Page 2 of the template is where you will configure the information related to Cisco's offer less SDP Invite configuration. You can enable or disable the configuration through the Yes/No Radio Button.

Note Click on the ? icon to know more about the configuration parameters and their usage.

Confi	guration ,	Assistant - Of	ferless SDP	configuratio	n						×
	<b>く</b> Back	<b>O</b>	2	3	4	5	6	7	8	Next 📏	^
		Configure CUCM Network here	Offerless SDP configuration	Transcoding	Additional Configuration	PSTN Network	PSTN Session Agent	Transcoding	Additional Configuration		>
			Let's	configure th	e SBC to ha	ndle offerles	s invite from	CUCM			
			E	)o you want SBC CUCM?	to handle offerle	ss INVITE from	© No 🕐 Y	es			
			C	)o you want to a	ld Media Profiles	? 🕲 No 🦲	Yes				

#### 8.3.3. Page 3 - Cisco side Transcoding

Page 3 is where you will be able to configure transcoding between the SBC and Cisco Call Manager. Once transcoding features is set to "yes", you will then have an option to select additional media codecs you want included in offers/answers towards Cisco Call Manger. If you select yes to either question regarding media codecs, you will be presented with a required drop down. You can select as many codecs from the list presented.

Con	figu	ration	Assistant - Tr	anscoding								×
	<	Back	Configure CUCM Network here	Offerless SDP configuration	3 Transcoding	Additional Configuration	5 PSTN Network	6 PSTN Session Agent	Transcoding	8 Additional Configuration	Next	
						Let's configu	ure transcodi	ng				
					Do you want to e the SBC?	nable transcoding	g features on	1 No 🕐 Y	′es			
					Do you want to s CUCM?	elect media code	cs for your	1 No	′es			
					Select media cod	ecs 🕐						
					G729 🗙 PC	MU 🗙						
							Requi	ired				

#### 8.3.4. Page 4 - Cisco side Additional Configuration

Page 4 is where you will be able to configure Session Agent Capabilities towards CUCM side. This includes enabling OPTIONS, enabling session translation etc towards CUCM side as shown below. . You can enable or disable the configuration through the Yes/No Radio Button

Con	figuratio	on Assistant - Ao	dditional Co	nfiguration							×
	< Bac	k 🕑	<b>~</b>	<b></b>	4	5	6	7	8	Next 💙	^
		Configure CUCM Network here	Offerless SDP configuration	Transcoding	Additional Configuration	PSTN Network	PSTN Session Agent	Transcoding	Additional Configuration		~
				Let's co	onfigure Sess	sion Agent ca	apabilities				
			Di	o you want to ena JCM	able OPTIONS pir	ng towards 🤇 🤆	No Yes	5			^
			Do	o you want SBC to our CUCM?	o handle call tran	sfer from 🤅	No Yes	5			
			Dotto	o you want to ena wards CUCM?	able session trans	slation	No 🚺 Yes	5			
			D	o you want to add	l or remove a stri	ing? ®					
				Remove 🗙							
			г.	ntar tha atrina 19							~

#### 8.3.5. Page 5 - Generic SIP Trunk Network

Page 5 of the template is where you will configure the network information to connect to Generic SIP trunk Network (PSTN side). Please fill the required fields and Press Next.

guration /	Assistant - PS	STN Network	(						
<b>〈</b> Back	<b>O</b>	•	<b>•</b>	<b>Ø</b>	5	6	7	8	Skip 🖒
	Configure CUCM Network here	Offerless SDP configuration	Transcoding	Additional Configuration	PSTN Network	PSTN Session Agent	Transcoding	Additional Configuration	
		Let's c	onfigure the	interface th	at communi	cates with th	e PSTN		
			Realm Na	ne Ø					
						Required			
			Port Num	oer 🕐					
			Port 1						
						Required			
			Slot Numb	er 🕐					
			Slot 0						
						Doguirod			

#### 8.3.6. Page 6 – PSTN Session Agent

Page 6 of the template is where you will configure the PSTN Session Agent details where you will enter the next hop IP address and port for sip signaling to and from your Generic SIP trunk. Please fill the required fields and click Next.

Configuration Assistant - PS	STN Session Ag	gent							×	
K Back	<b>Ø</b>	<b>•</b>	•	<b>•</b>	6	- 7 -	8	Next 💙	^	
Configure CUCM Network here	Offerless SDP T configuration	ranscoding	Additional Configuration	PSTN Network	PSTN Session Agent	Transcoding	Additional Configuration		*	
		Let's cor	ifigure the S	ession Agen	t for PSTN					
	PSTN	Session Ager	it hostname						^	
	68.68	3.117.67								
	PSTN	Required PSTN Session Agent IP Address ②								
	68.68	3.117.67								
	PSTN	Session Ager	it Port 🕐							
	5060	5060								
		Required								
	Πορει	ιους σενίεο η	rovidor hava a c	acond @	No Va				~	

#### 8.3.7. Page 7 – PSTN side Transcoding

Page 7 is where you will be able to configure transcoding between the SBC and Generic SIP Trunk. Once transcoding features is set to "yes", you will then have an option to select additional media codecs you want included in offers/answers towars PSTN side. If you select yes to either question regarding media codecs, you will be presented with a required drop down. You can select as many codecs from the list presented.

onfiguration	Assistant - Tr	anscoding								×
<b>く</b> Back	<b>•</b>	<ul> <li>✓</li> </ul>	<ul> <li>Image: A start of the start of</li></ul>	<ul> <li>✓</li> </ul>	<ul> <li>Image: A start of the start of</li></ul>	•	7	8	Next 📏	^
	Configure CUCM Network here	Offerless SDP configuration	Transcoding	Additional Configuration	PSTN Network	PSTN Session Agent	Transcoding	Additional Configuration		~
			I	_et's configu	ıre transcodi	ng				
			Do you want to er	able transcodin	g on the SBC? 🤅	No Ves	5			
			Do you want to se PSTN)?	lect media code	cs (SBC to	O No Y	′es			
			Select media cod	ecs (SBC to PSTN	l) ®					
			PCMA 🗙 G7	22 🗙						

8.3.8. Page 8 – PSTN side Additional Configuration

Page 8 is where you will be able to configure Session Agent Capabilities towards PSTN side. This includes enabling OPTIONS, enabling session translation etc towards PSTN side as shown below. You can enable or disable the configuration through the Yes/No Radio Button

Configuration A	ssistant - Ad	ditional Conf	iguration							×
<b>く</b> Back	<b>~</b>	<b>·····································</b>	•	•	<b></b>	<b>·····································</b>	<ul> <li>Image: A start of the start of</li></ul>	8	Review	^
	Configure CUCM Network here	Offerless SDP configuration	Transcoding	Additional Configuration	PSTN Network	PSTN Session Agent	Transcoding	Additional Configuration		*
			Lets con	figure Sessi	on Agent Cap	oabilities				
		Do y	ou want to enab	le OPTIONS tow	ards PSTN?   🕅 N	lo 🚺 Yes				^
		Do y PST	ou want SBC to N?	handle call trans	fer from 🛛 🕐	No Yes				
		Do y towa	ou want to enab ard your PSTN p	le session transl rovider?	ation on 🛛 🧿	No Yes				
		Sele	ct your operatio	n 0						
		A	× bb							
		Do y	ou want to enab	le session transl	ation					*

### 8.4. Review

At the end of the template, you will notice in the top right, a "*Review*" tab. If all 8 pages presented across the top are showing green, indicting there are no errors with the information entered, click on the "Review" tab.

////

Configuration A	ssistant - Ad	ditional Conf	figuration							×
<b>K</b> Back	<b>_</b>				<b>_</b>		<b>_</b>	8	Review	^
	Configure CUCM Network here	Offerless SDP configuration	Transcoding	Additional Configuration	PSTN Network	PSTN Session Agent	Transcoding	Additional Configuration		*
			Lets con	ifigure Sessi	on Agent Cap	oabilities				
		Doy	you want to enat	ole OPTIONS tow	ards PSTN?   N	lo 🚺 Yes				^
		Do y PST	you want SBC to 'N?	handle call trans	ifer from 🛛 🕐	No Yes				
		Do y tow	you want to enat ard your PSTN p	ole session trans) provider?	ation on 🛛 💿	No Yes				
		Sele	ect your operatio	n Ø						
		A	.dd 🗙							
		Doy	you want to enat	ole session transl	lation 🕲					~

The screen looks like below after clicking the Review Tab.

			Download 🔻 Apply
Configure CUCM Network here	🥒 Edit	Configuration	
Realm Name			
CUCM			Сору
Enter CUCM hostname here			
cum pe oracle com		codec-policy	dia and an an alian
cuch.pe.oracle.com		allow-codecs	*
Enter the CUCM IP here		add-codecs-on-egress	G729 PCMU
10.232.50.75		codec-policy	
		allow-codecs	*
Enter the CUCM port here		add-codecs-on-egress	PCMA G722
5060		http-server	
Dort Number		name	webServerInstanc
Port Number		from-address	*
Port 0		to-address	*
Slot Number		source-realm	CUCM
SIOF INTELLE		policy-attribute	07.0 - D.0 mm
Slot 0		next-nop	SAG: PSTN STPTrunk

On the left side of the review contains the entries for each page. Each page has an "*Edit*" tab that can be used to make changes to the information entered on that specific page without having to go through the entire template again.

On the right side of the review page, under the "*Configuration*" tab is the ACLI output from the SBC. This is the complete configuration of the SBC based on the information entered throughout the template.

### 8.5. Download and/or Apply

Now that the entries provided throughout the template have been reviewed, the template provides you with the ability to "Download" the config by clicking the "*Download*" tab on the top right. Next, click the "*Apply*" button on the top right, and you will see the following pop-up box appear.

Configuration Assistant - Apply Confirmation	×
If you proceed, the system erases the existing configuration and reboots.	

Now you can click "*Reboot*" to confirm you want to apply the configuration to the SBC. The SBC will reboot. When it comes back up, the SBC will have a basic configuration in place for Cisco Call Manager with Generic SIP trunking.

## 8.6. Configuration Assistant Access

Upon initial login, if the Configuration Assistant Template does not immediately appear on the screen, you can access by clicking on the "SYSTEM" tab, top right of your screen. After that, click on the "Configuration Assistant" tab, top left. This allows end users to access the Configuration Assistance at any time through the SBC GUI.

	Session Border Controller				Û 🔺	admin 🔻
NN4600-139 10.138.194.139 SCZ9.0	0.0 Patch 3 (Build 290)	Dashboard	Configuration	Monitor and Trace	Widgets	System
System Configuration Assistant			Force HA Swite	thover 🗘 Reboot	🛃 Suppo	rt informat or
File Management	File Management Objects					ľ
Configuration CSV	Name	Description				
Level Dente Telle	Audit Log	Audit changes by all users on the system.				
Local Route Table	Backup Configuration	Manage backup configurations.				
Fraud Protection Table	Configuration CSV	Upload/Download/Delete configuration CSVs.				
Log	Configuration Template	Upload/Download/Delete configuration templates.				
Audit Log	Fraud Protection Table	Manage fraud protection table.				
Addit Log	Local Route Table	Manage Local route table.				
Playback Media	Log	System logs.				
Software Image	Playback Media	Upload/Download/Delete playback media.				
SPL Plug In	SPL Plug In	Upload/Download/Delete SPL plugins.				
0. 2. mp	Software Image	Upload/Download/Delete software images				
Configuration Template						

# 9. Existing SBC configuration

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

- New realm-config
- <u>New sip-interface</u>
- <u>New session-agent</u>
- New steering-pools
- New local-policy
- <u>New Translation Rules</u>
- Session Translation Rules

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

# Appendix A

Following are the test cases that are executed between Cisco User with the Generic SIP Trunk (PSTN user). Please note that Cisco User here refers both Cisco User inside Enterprise network as well as Cisco Remote worker.

///

Serial Number	Test Cases Executed	Result
1	Cisco user disconnects an inbound connected call	Pass
2	Cisco user disconnects an outbound connected call	Pass
3	Generic SIP Trunk user disconnects an inbound connected call	Pass
4	Generic SIP Trunk User disconnects an outbound connected call	Pass
5	Cisco user places inbound call from Generic SIP Trunk user on hold and then resumes	Pass
6	Cisco user makes outbound call to Generic SIP Trunk user and put that call on hold and then resumes	Pass
7	Generic SIP Trunk user places inbound call from Cisco user on hold and then resumes	Pass
8	Generic SIP Trunk user makes outbound call to Cisco user and put that call on hold and then resumes	Pass
9	Cisco user places inbound call from Generic SIP Trunk user on hold for over 15/30 minutes and then resumes	Pass
10	Cisco user makes outbound call to Generic SIP Trunk user and places the call on hold for over 15/30 minutes and then resumes	Pass
11	Inbound Generic SIP Trunk call to Cisco blind transferred to second Cisco/ PSTN User	Pass
12	Outbound Generic SIP Trunk call from Cisco user blind transferred to second Cisco/ PSTN User	Pass
13	Inbound Generic SIP Trunk Call to Cisco consultatively transferred to Cisco/ PSTN User	Pass
14	Outbound Generic SIP Trunk call from Cisco user consultatively transferred to Cisco/ PSTN User	Pass
15	Cisco user makes outbound call to Generic SIP Trunk user and makes a conference call by adding another Cisco/ PSTN user.	Pass
16	Generic SIP Trunk user makes outbound call to Cisco user and Cisco user makes a conference call by adding another Cisco/ PSTN user.	Pass

17	Cisco user mutes inbound call from Generic SIP Trunk user and then unmutes	Pass
18	Cisco user mutes outbound call made to Generic SIP Trunk user and then unmutes	Pass
19	Generic SIP Trunk user mutes inbound call from Cisco user and then unmutes	Pass
20	Generic SIP Trunk user mutes outbound call made to Cisco user and then unmutes	Pass
21	Generic SIP Trunk User disconnects outbound call to Cisco user before it is answered	Pass
22	Cisco user disconnects outbound call to Generic SIP Trunk user before it is answered	Pass







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Integrated Cloud Applications & Platform Services

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