Configuration Guide for Google CCAI Call Recording Using Avaya Session Border Controller 10.2.0.0-86-24077



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1 Audience

This document is intended for the SIP Trunk customer's technical staff and Value-Added Reseller (VAR) having installation and operational responsibilities.

1.1 Introduction

This configuration guide describes configuration steps for Google CCAI Call Recording using Avaya Session Border Controller v10.2.0.0-86-24077.

1.1.1 TekVizion Labs

TekVizion Labs[™] is an independent testing and verification facility offered by TekVizion, Inc. TekVizion Labs offers several types of testing services including:

- Remote Testing provides secure, remote access to certain products in TekVizion Labs for pre-Verification and ad hoc testing.
- Verification Testing Verification of interoperability performed on-site at TekVizion Labs between two products or in a multi-vendor configuration.
- Product Assessment independent assessment and verification of product functionality, interface usability, assessment of differentiating features as well as suggestions for added functionality, stress, and performance testing, etc.

TekVizion is a systems integrator specifically dedicated to the telecommunications industry. Our core services include consulting/solution design, interoperability/Verification testing, integration, custom software development and solution support services. Our services help service providers achieve a smooth transition to packet-voice networks, speeding delivery of integrated services. While we have expertise covering a wide range of technologies, we have extensive experience surrounding our practice areas which include SIP Trunking, Packet Voice, Service Delivery, and Integrated Services.

The TekVizion team brings together experience from the leading service providers and vendors in telecom. Our unique expertise includes legacy switching services and platforms, and unparalleled product knowledge, interoperability, and integration experience on a vast array of VoIP and other next-generation products. We rely on this combined experience to do what we do best: help our clients advance the rollout of services that excite customers and result in new revenues for the bottom line. TekVizion leverages this real-world, multi-vendor integration and test experience and proven processes to offer services to vendors, network operators, enhanced service providers, large enterprises and other professional services firms. TekVizion's headquarters, along with a state-of-the-art test lab and Executive Briefing Center, is located in Plano, Texas.

For more information on TekVizion and its practice areas, please visit TekVizion Labs website.

2 SIP Trunking Network Components

The network for the SIP trunk reference configuration is illustrated below and is representative of Google CCAI Call Recording with Avaya Session Border Controller (ASBC) v10.2.0.0-86-24077 configuration.

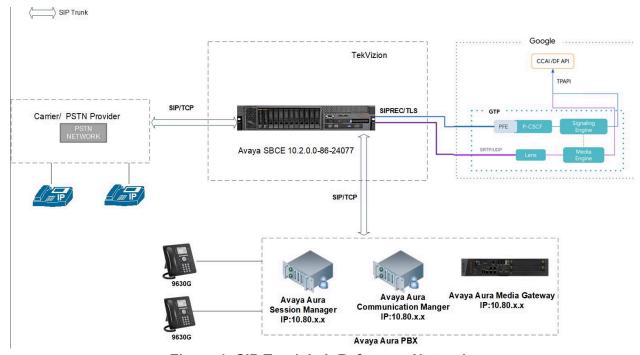


Figure 1: SIP Trunk Lab Reference Network

The lab network consists of the following components.

- Google CCAI Cloud Environment
- Avaya Session Border Controller (ASBC) v10.2.0.0-86-24077
- OnPrem PBX (Avaya Aura PBX)

3 Hardware Components

Running on ESXi- 7.0.3: Avaya SBC v10.2.0.0-86-24077

4 Software Requirements

- Avaya SBC v10.2.0.0-86-24077
- OnPrem PBX (Avaya Aura PBX)

5 Features

5.1 Features tested for Google CCAI Call Recording

- Basic Inbound calls
- Call Hold and Resume
- Call Transfer
- Conference

5.2 Features Not tested for Google CCAI Call Recording

None

5.3 Caveats and Limitations

DTLS	DTLS towards Google CCAI is not supported
Blind Transfer	Avaya PBX does not support blind transfer. This test case is performed by ringing
	transfer
Long duration call	Avaya SBC does not send session refresh RE-INVITE. Google CCAI sends session refresh every 15 minutes using UPDATE

5.4 Failed Testcase

None

6 Configuration

6.1 Configuration Checklist

Below are the steps that are required to configure Avaya SBC.

Table 1 - Avaya SBC Configuration Steps

Step	Description Description	Reference
Step 1	Avaya SBC Login	Section 6.4.1
Step 2	Server Interworking	Section 6.4.2
Step 3	SIP Servers	Section 6.4.3
Step 4	Topology Hiding	Section 6.4.4
Step 5	Routing	Section 6.4.5
Step 6	Recording Profile	Section 6.4.6
Step 7	Session Policies	Section 6.4.7
Step 8	Session Flows	Section 6.4.8
Step 9	Signaling Manipulation	Section 6.4.9
Step 10	Signaling Rules	Section 6.4.10
Step 11	End Point Policy Groups	Section 6.4.11
Step 12	Media Interface	Section 6.4.12
Step 13	Network Management	Section 6.4.13
Step 14	Signaling Interface	<u>Section 6.4.14</u>
Step 15	End Point Flow	Section 6.4.15
Step 16	TLS Configuration	Section 6.4.16

6.2 IP Address Worksheet

The specific values listed in the table below and in subsequent sections are used in the lab configuration described in this document are for **illustrative purposes only**.

Table 3 - IP Address Worksheet

Component	IP Address	
Google CCAI		
Signaling	tekvizion.telephony.goog	
Media	74.125.X.X	
OnPrem PBX		
LAN IP Address	10.70.X.X	
Avaya SBC		
LAN IP Address	10.64.X.X	
WAN IP Address	192.65.X.X	

6.3 Google CCAI API Configuration

Below link can be referred to configure Google CCAI API configuration for Call recording.
Link to be provided by Google team

6.4 Avaya ASBC Configuration

The following is the example configuration of Avaya SBC for Google CCAI Call Recording.

6.4.1 Avaya SBC Login

- Log into Avaya Session Border Controller (ASBC) web interface by typing "https://X.X.X.X/sbc".
- Enter the Username and Password
- Click Log In



Figure 2: Avaya ASBC Login

 Device, select Name(avayasbc1) from drop down to expand the configuration for Avaya SBC

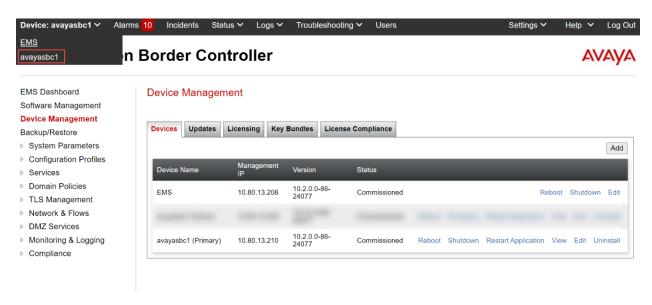


Figure 3: Selection of Avaya SBC Device

6.4.2 Server Interworking

Server Interworking for Avaya Aura Session Manager (SM)

- Navigate: Configuration Profiles > Server Interworking
- Select the default Interworking Profile avaya-ru, click Clone
- Set Clone Name: AvayaSM10.2
- Click Finish

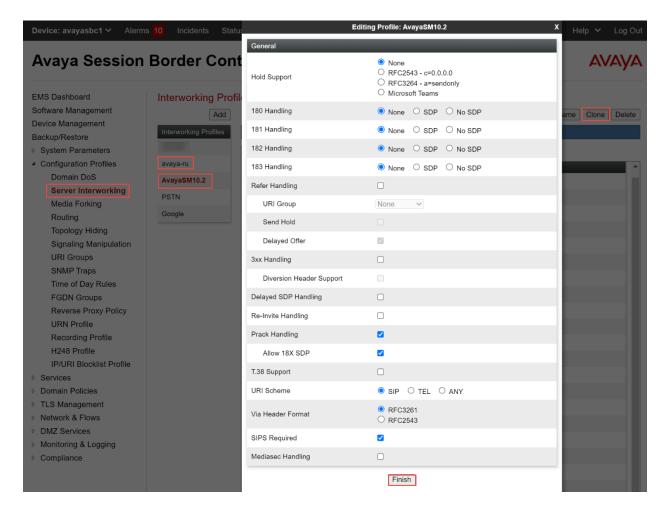


Figure 4: Server Interworking profile for Avaya Aura SM

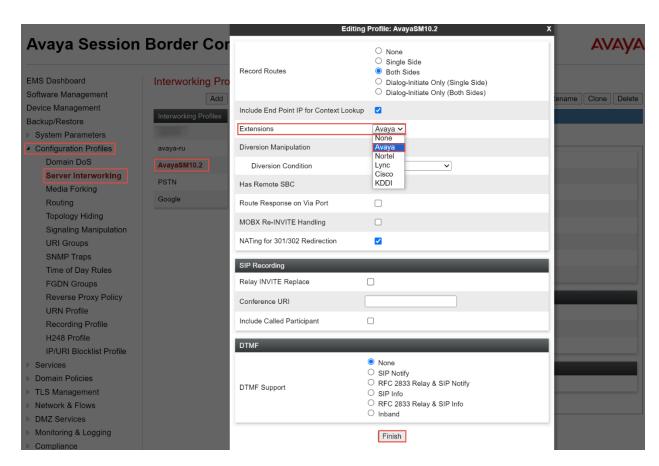


Figure 5: Server Interworking profile for Avaya Aura SM continuation

Server Interworking for Google CCAI

Repeat the same procedure to create the Interworking Profile towards Google CCAI

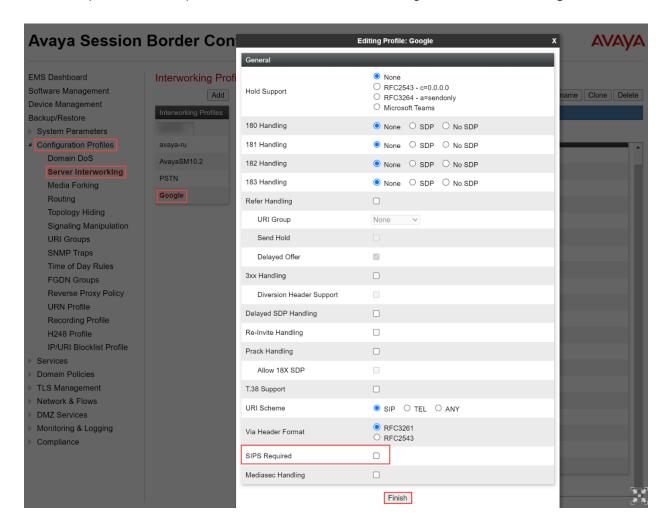


Figure 6: Server Interworking profile for Google CCAI

Avaya Session Border Controller



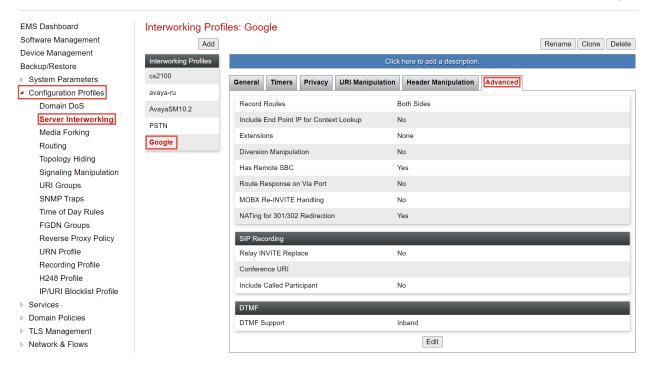


Figure 7: Server Interworking profile for Google CCAI continuation

Server Interworking for PSTN Gateway

Avaya Session Border Controller

Repeat the same procedure to create the Interworking Profile towards PSTN Gateway

AVAYA

EMS Dashboard Interworking Profiles: PSTN Software Management Rename Clone Delete Add Device Management Interworking Profiles Backup/Restore cs2100 System Parameters General Timers Privacy URI Manipulation Header Manipulation Advanced Configuration Profiles avaya-ru Domain DoS AvayaSM10.2 Server Interworking Hold Support None PSTN Media Forking 180 Handling None Google Routing 181 Handling None Topology Hiding 182 Handling Signaling Manipulation 183 Handling None **URI** Groups SNMP Traps Refer Handling Time of Day Rules URI Group None FGDN Groups Send Hold No Reverse Proxy Policy Delayed Offer URN Profile 3xx Handling No Recording Profile H248 Profile Diversion Header Support No IP/URI Blocklist Profile Delayed SDP Handling No Services Re-Invite Handling No Domain Policies Prack Handling No ▶ TLS Management Allow 18X SDP Network & Flows T.38 Support DMZ Services No Monitoring & Logging URI Scheme Compliance Via Header Format RFC3261 SIPS Required Mediasec No Edit

Figure 8: Server Interworking profile for PSTN Gateway

Avaya Session Border Controller



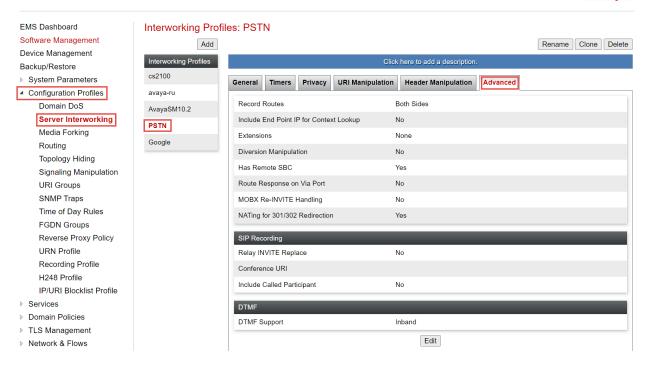


Figure 9: Server Interworking profile for PSTN Gateway continuation

6.4.3 SIP Servers

SIP Server for Avaya Aura SM

- Navigate: Services > SIP Servers
- Click Add
- Set Profile Name: AvayaSM10.2
- Click Next



Figure 10: SIP Server For Avaya Aura SM

Set Server Type: Select Trunk Server from the drop down

- Set IP Address/FQDN/CIDR Range: Enter the Avaya Aura SM IP Address
- Set Port: 5060Set Transport: TCP
- Click Finish

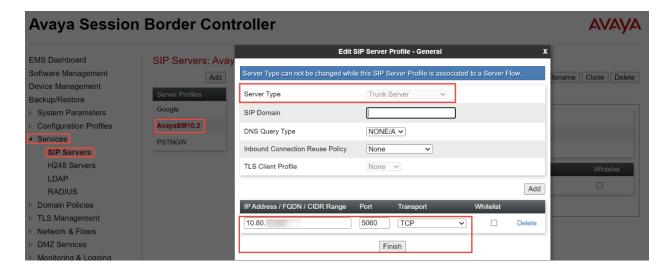


Figure 11: SIP Server For Avaya Aura SM Continuation

- Navigate: Heartbeat tab
- Set Enable Heartbeat: Checked
- Click Finish

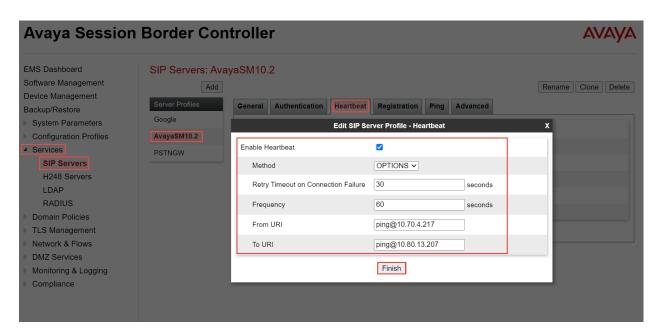


Figure 12: SIP Server For Avaya Aura SM Continuation

- Navigate: Ping tab
- Set Enable Ping: Checked

Click Finish

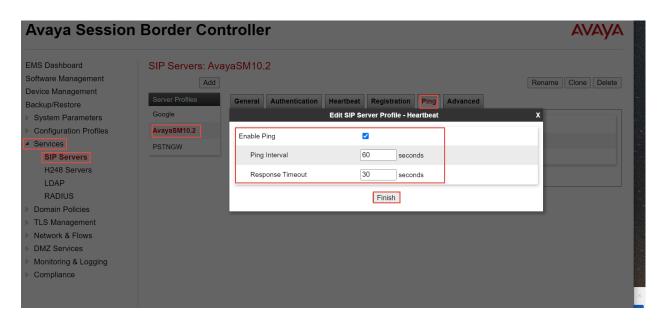


Figure 13: SIP Server For Avaya Aura SM Continuation

Navigate: Advanced tab

Set Enable Grooming: Checked

Set Interworking Profile: Select AvayaSM10.2

Avaya Session Border Controller

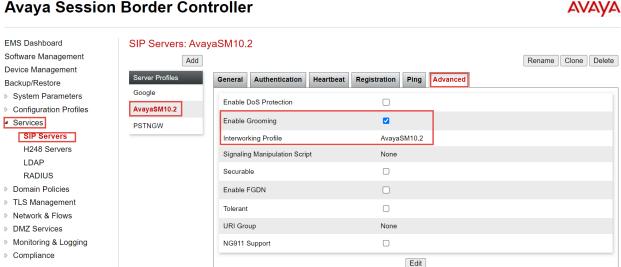


Figure 14: SIP Server For Avaya Aura SM Continuation

SIP Server for Google CCAI

- Navigate: Services > SIP Servers
- Click Add

- Set Profile Name: Google
- Click Next



Figure 15: SIP Server For Google CCAI

- Set Server Type: Select Recording Server from the drop down
- Set IP Address/FQDN: Enter the Google CCAI FQDN
- Set Port: 5672Set Transport: TLS
- Click Finish

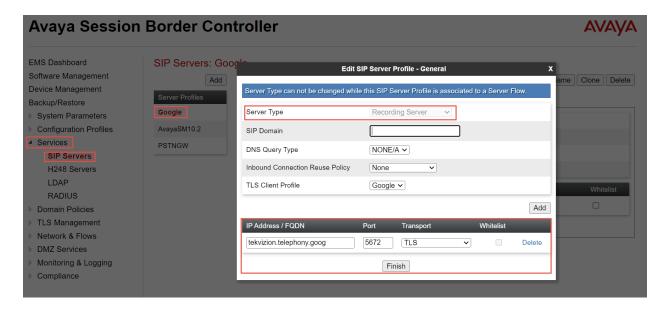


Figure 16: SIP Server For Google CCAI Continuation

• Navigate: **Heartbeat** tab

Set Enable Heartbeat: Checked

Click Finish

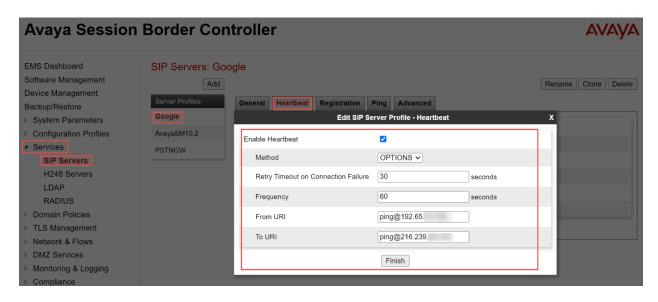


Figure 17: SIP Server For Google CCAI Continuation

- Navigate to Ping tab
- Set Enable Ping: Checked
- Click Finish

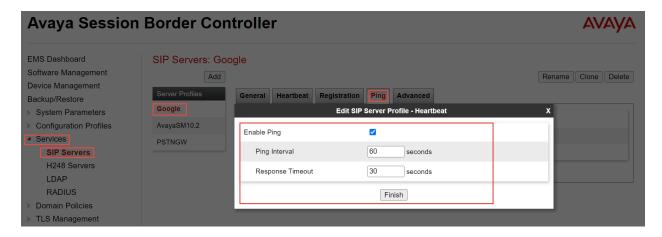


Figure 18: SIP Server For Google CCAI Continuation

Navigate: Advanced tab

• Set Enable Grooming: Checked

• Set Interworking Profile: Select Google

Click Finish

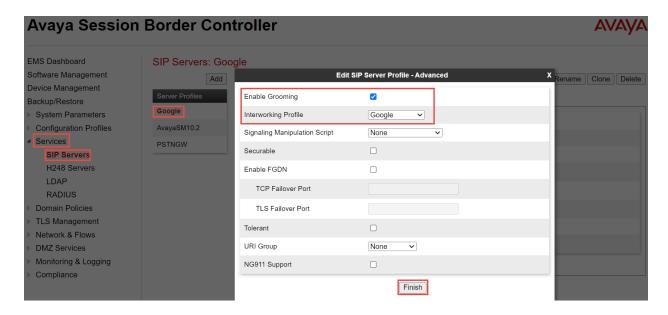


Figure 19: SIP Server For Google CCAI Continuation

SIP Server for PSTN Gateway

• Navigate: Services > SIP Servers

Click Add

• Set Profile Name: PSTNGW

Click Next



Figure 20: SIP Server For PSTN Gateway

Set Server Type: Select Trunk Server from the drop down

- Set IP Address/FQDN: Enter the PSTN IP address.
- Set Port: 5060Set Transport: TCP
- Click Finish

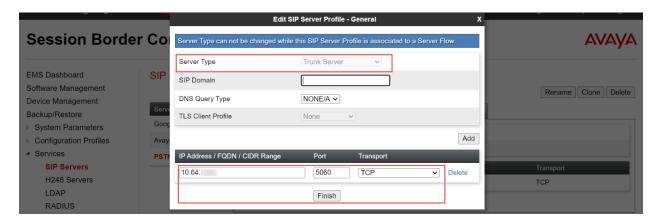


Figure 21: SIP Server For PSTN Gateway Continuation

• Navigate: Heartbeat tab

Set Enable Heartbeat: Checked

Click Finish

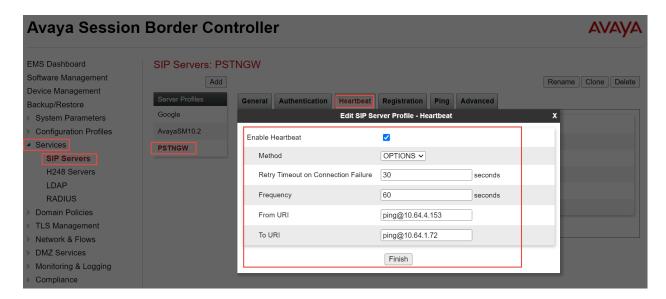


Figure 22: SIP Server For PSTN Gateway Continuation

Navigate: Ping tab

Set Enable Ping: Checked

• Click Finish

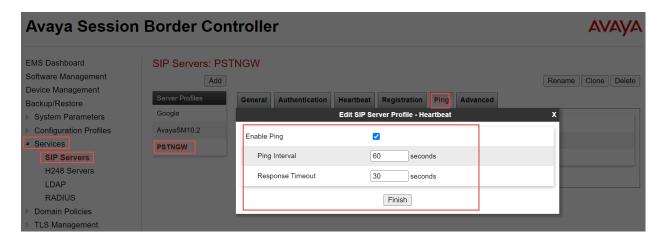


Figure 23: SIP Server For PSTN Gateway Continuation

Navigate: Advanced tab

• Set Enable Grooming: Checked

• Set Interworking Profile: Select PSTN

Click Finish

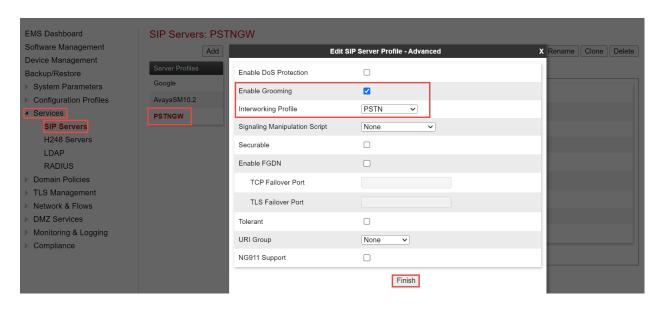


Figure 24: SIP Server For PSTN Gateway Continuation

6.4.4 Topology Hiding

Topology Hiding profile for Google

- Topology Hiding profiles are added for Google CCAI to overwrite and hide certain headers
- Navigate: Configuration Profiles > Topology Hiding
- Click Add
- Set Profile Name: Google
- Click Next

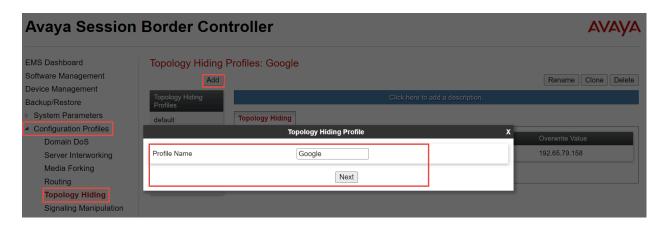


Figure 25: SIP Server For Google CCAI

- Select the newly created profile Google and Click Edit
- Overwrite Value: Replace the From header with Google CCAI Facing Public IP
- Click Finish

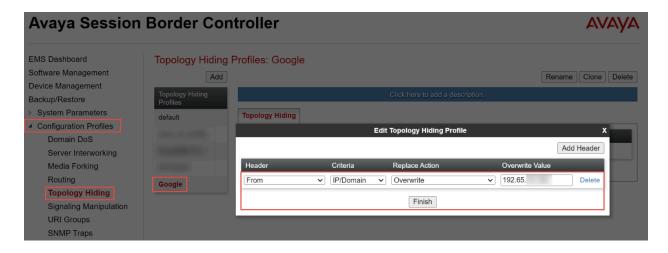


Figure 26: SIP Server For Google CCAI Continuation

6.4.5 Routing

Routing for Avaya Aura SM

- Navigate: Configuration Profiles > Routing
- Click Add
- Set Profile Name: AvayaSM10.2
- Click Next



Figure 27: Routing for Avaya Aura SM



Figure 28: Routing for Avaya Aura SM Continuation

- At Routing Profile Window, Click Add
- Set Priority/Weight: 1
- Select SIP Server Profile, Next Hop Address from the drop-down menu
- Click Finish

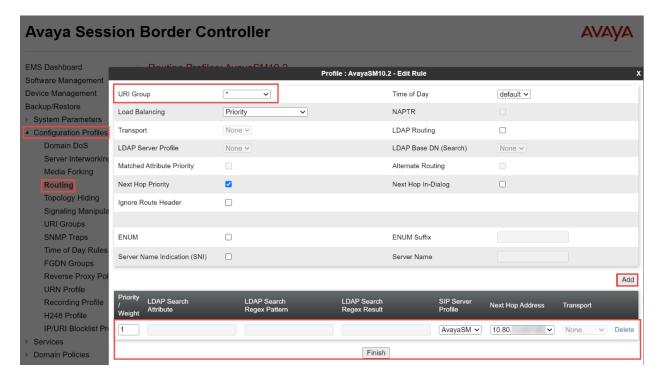


Figure 29: Routing for Avaya Aura SM Continuation

Routing for **PSTN Gateway**

- Navigate: Configuration Profiles > Routing
- Click Add
- Set Profile Name: PSTNGW
- Click Next

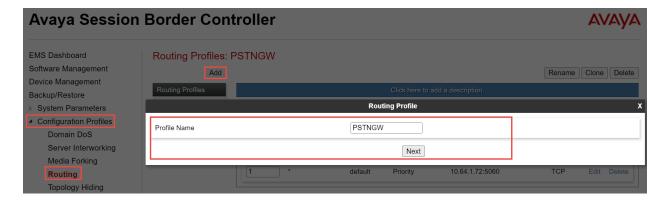


Figure 30: Routing for PSTN Gateway

- At Routing Profile Window, Click Add
- Set Priority/Weight: 1
- Select SIP Server Profile, Next Hop Address from the drop-down menu
- Click Finish

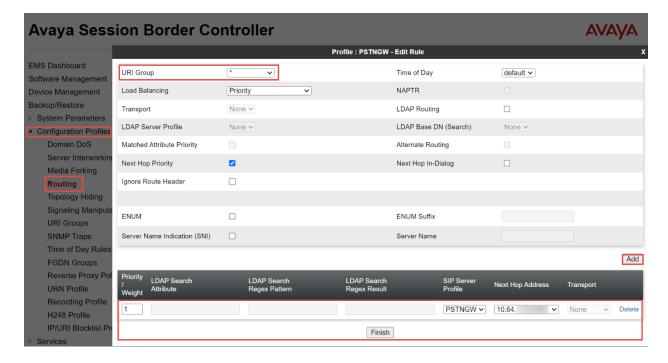


Figure 31: Routing for PSTN Gateway Continuation

Routing for Google CCAI

- Navigate: Configuration Profiles > Routing
- Click Add
- Set Profile Name: Google
- Click Next

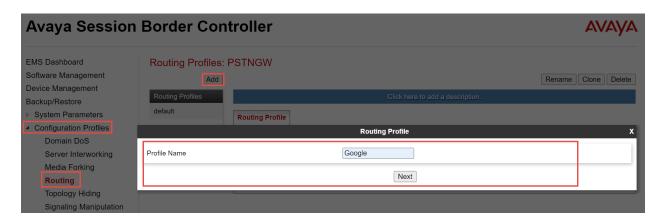


Figure 32: Routing for Google CCAI

At Routing Profile Window, Click Add

- Set Priority/Weight: 1
- Select SIP Server Profile, Next Hop Address from the drop-down menu
- Click Finish

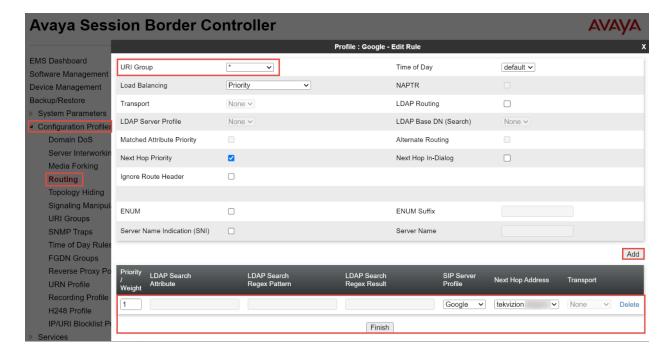


Figure 33: Routing for Google CCAI Continuation

6.4.6 Recording Profile

Navigate: Configuration> Recording Profile

Click Add

Set Profile Name: Google_RP

Click Next

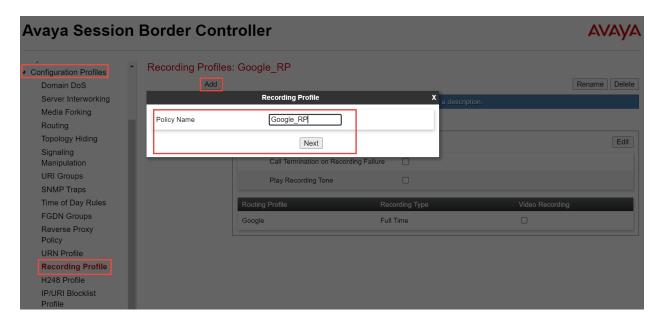


Figure 34: Recording Profile for Google CCAI

Set Routing Profile: Select Google

Set Recording Type: Select Full Time from the dropdown

Click Finish

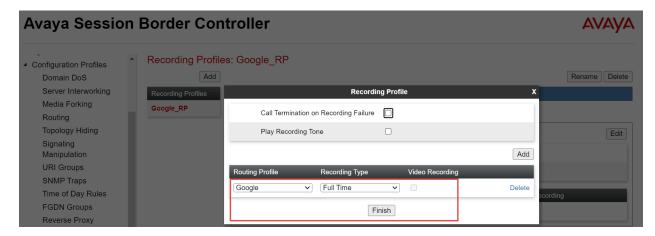


Figure 35: Recording Profile for Google CCAI Continuation

6.4.7 Session Policies

- Navigate: Domain Policies > Session Policies
- Select default under Session Policies, Click Clone
- Set Profile Name: Google_SP
- Click Next

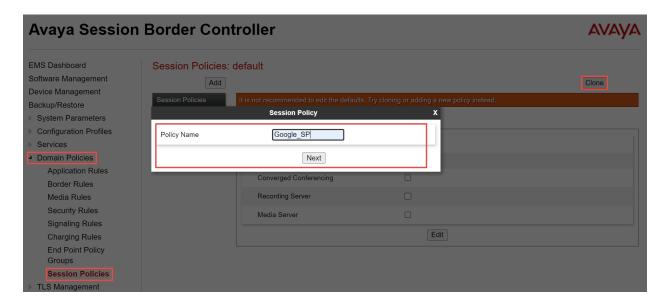


Figure 36: Session Policies for Google CCAI

- Media Anchoring: CheckedRecording Server: Checked
- Set Routing Profile: Select the route profile Google RP
- Click Finish

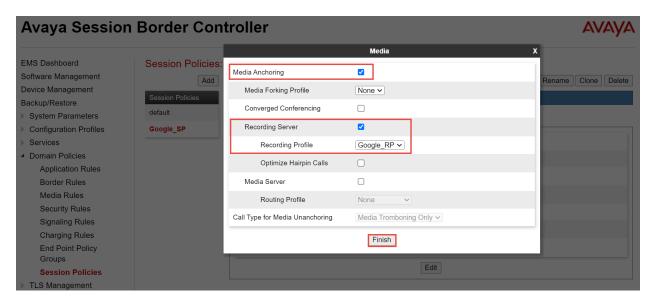


Figure 37: Session Policies for Google CCAI Continuation

6.4.8 Session Flows

- Navigate: Network and Flows> Session Flows
- Click Add
- Set Name: Google_SF
- Select Session Policy: Google_SP
- Click Finish

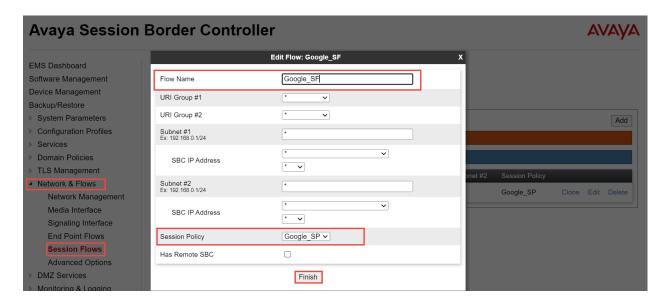


Figure 38: Session flow for Google CCAI

6.4.9 Signaling Manipulation

- Navigate: Configuration Profiles > Signaling Manipulation
- Click AddTitle: Google
- Click Save
- Below sigma script is created to add Call-Info header towards Google CCAI with the Dialog Flow API request along with the Conversation ID.
- Avaya signaling manipulation does not allow to add double slash (http://) in the
 manipulation, hence "&slash" is added to the %baseURI as shown below. Later
 "&slash" is replaced with symbol "/" using manipulations.
- %baseUri value provided below is a reference value. Project
 name("ccai-38XXXXXconversations") present in the call-info header will vary according to the
 project created by user. Ab_ is just an identifier, you can use any values which matches the regex
 pattern requirement of call info header.

```
within session "all"
       act on request where %DIRECTION="OUTBOUND" and
     %ENTRY POINT="POST ROUTING" and %METHOD="INVITE"
          %aor = %HEADERS["Call-ID"][1];
          %baseUri =
     "<a href="http:&slash/dialogflow.googleapis.com/v2beta1/projects/ccai-38XXXXX/conversation">http:&slash/dialogflow.googleapis.com/v2beta1/projects/ccai-38XXXXX/conversation</a>
     s/Ab_";
          append( %baseUri, %aor);
          %newUri1 = ">;purpose=Goog-ContactCenter-Conversation";
          append( %baseUri, %newUri1);
          %HEADERS["Call-Info"][1] = %baseUri;
          %HEADERS["Call-Info"][1].URI.regex_replace("&slash","/");
          %HEADERS["Request_Line"][1].URI.USER.regex_replace("(.*)",
     "+1833449XXXX);
          %HEADERS["TO"][1].URI.USER.regex_replace("^.....", "+1833449XXXX");
          %HEADERS["Allow"][1].regex_replace(", UPDATE,", "");
       act on request where %DIRECTION="OUTBOUND" and
     %ENTRY POINT="POST ROUTING" and %METHOD="ACK"
          %HEADERS["TO"][1].URI.USER.regex_replace("^.....", "+1833449XXXX");
       act on request where %DIRECTION="OUTBOUND" and
     %ENTRY POINT="POST ROUTING" and %METHOD="UPDATE"
          %HEADERS["TO"][1].URI.USER.regex_replace("^.....", "+1833449XXXX");
          %HEADERS["Content-Type"][1].regex_replace("application/rs-metadata",
     "application/rs-metadata+xml");
       act on request where %DIRECTION="OUTBOUND" and
     %ENTRY_POINT="POST_ROUTING" and %METHOD="BYE"
          %HEADERS["TO"][1].URI.USER.regex_replace("^.....", "+1833449XXXX");
}
```

Figure 39: Signaling Manipulation- Google CCAI

6.4.10 Signaling Rules

- Configure Navigate: **Domain Policies > Signaling Rules**
- Select default under Signaling Rules, Click Clone
- Set Rule Name: Avaya SM
- Click Finish



Figure 40: Signaling Rules for Avaya Aura SM

- Select the newly cloned Signaling Rule Avaya_SM, under tab Request Headers, Click Add In Header Control
- Set Proprietary Request Header: Checked
- Set Header Name: AV-Global-Session-ID
- Set Method Name: Select ALL from the drop down
- Set Header Criteria: Forbidden
- Set Presence Action: Remove header is selected from the drop down
- Click Finish

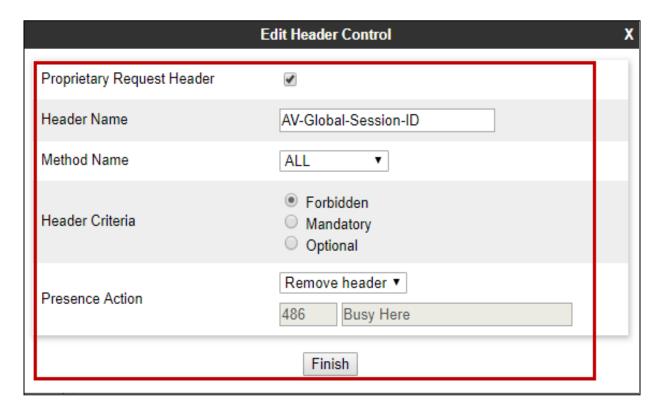


Figure 41: Signaling Rules for Avaya Aura SM Continuation

• Repeat the same steps for all other required headers

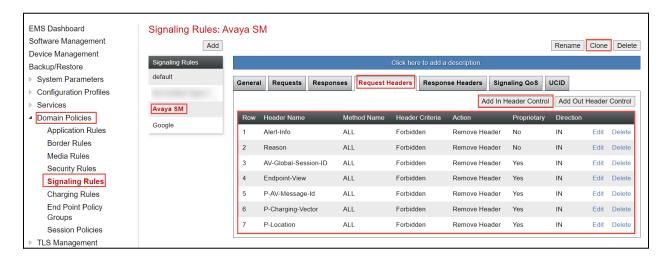


Figure 42: Signaling Rules for Avaya Aura SM Continuation

Repeat the same steps for Response Headers

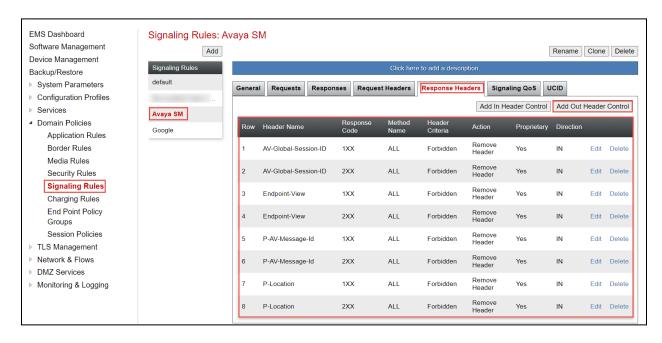


Figure 43: Signaling Rules for Avaya Aura SM Continuation

6.4.11 End Point Policy Groups

End Point Policy Group for Avaya Aura SM

- A new End Point Policy Group is created for Avaya Aura Session Manager.
- Navigate: Domain Policies > End Point Policy Groups
- Select default-low under Policy Groups
- Click Clone
- Set Group Name: Avaya SM
- Click Next

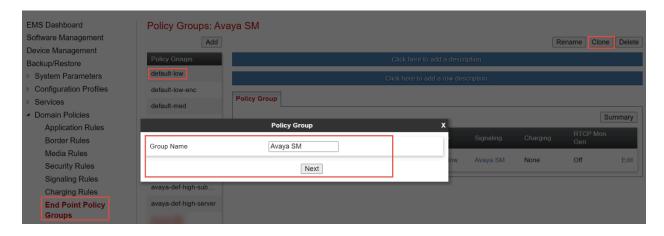


Figure 44: End Point Policy Group for Avaya Aura SM

- Select the newly created Group Avaya SM, Click Edit
- Set Signaling Rule: Avaya SM
- Click Finish



Figure 45: End Point Policy Group for Avaya Aura SM Continuation

End Point Policy Group for Google CCAI

• Repeat the same steps to create End Policy Group for Google CCAI

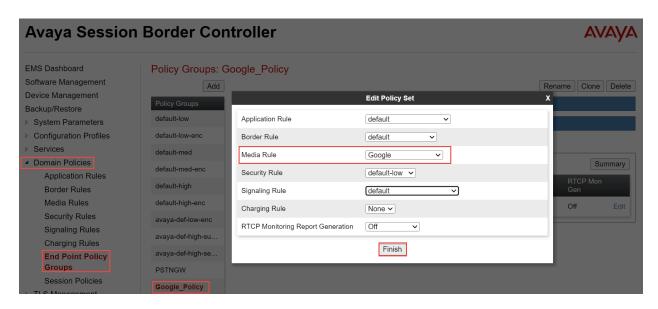


Figure 46: End Point Policy Group for Google CCAI

End Point Policy Group for PSTN Gateway

Repeat the same steps to create End Policy Group for PSTNGW

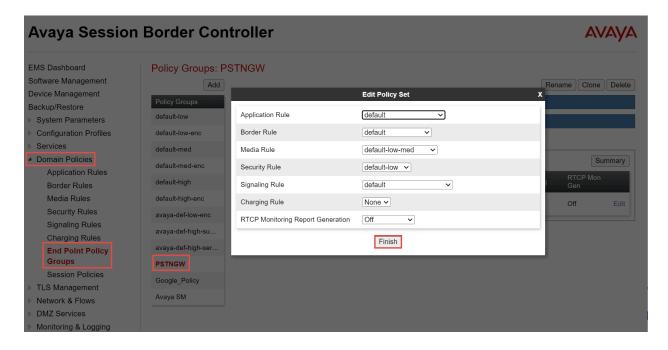


Figure 47: End Point Policy Group for PSTN Gateway

6.4.12 Media Interface

- Navigate: Network & Flows > Media Interface. Click Add
- Set Name: AvayaSM10.2 is given here
- Set IP Address: Select LAN_PBX from the drop down and the IP address populates automatically. The IP address for Interface facing Avaya Aura SM is 10.70.X.X
- Set Port Range: 35000-40000
- Click Finish

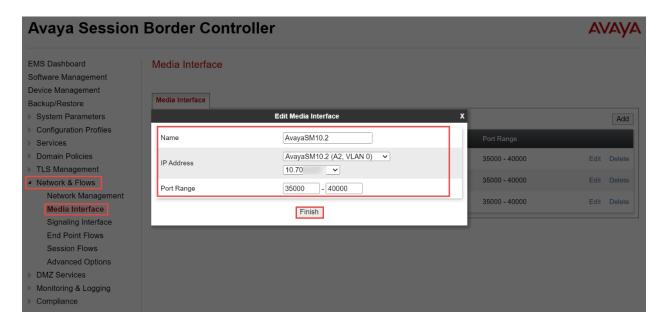


Figure 48: Media Interface Facing Avaya Aura SM

Repeat the same steps to create a Media Interface facing Google CCAI

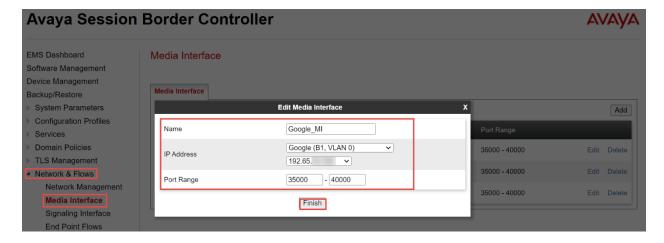


Figure 49: Media Interface Facing Google CCAI

Repeat the same steps to create a Media Interface facing PSTN Gateway



Figure 50: Media Interface Facing PSTN Gateway

6.4.13 Network Management

Network Management for Avaya Aura SM

- Navigate: Network & Flows > Network Management. Click Add, new Add Network Interface window appears
- Set Name: AvayaSM10.2 is given for the network facing Avaya Aura SM
- Set default Gateway IP Address
- Set Network Prefix or Subnet Mask
- Set Interface
- Set IP Address facing Avaya Aura SM
- Click Finish

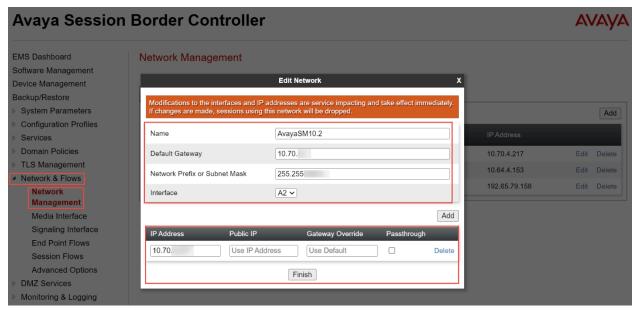


Figure 51: Network Management Facing Avaya Aura SM

Network Interface for Google CCAI

Repeat the same steps to create the Signaling Interface facing Google CCAI.

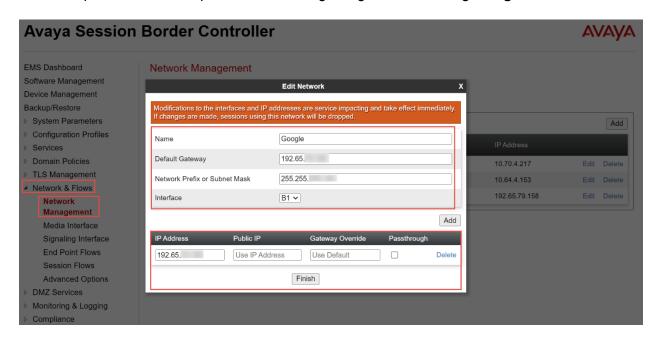


Figure 52: Network Management Facing Google CCAI

Network Interface for **PSTN Gateway**

Repeat the same steps to create the Signaling Interface facing PSTN.

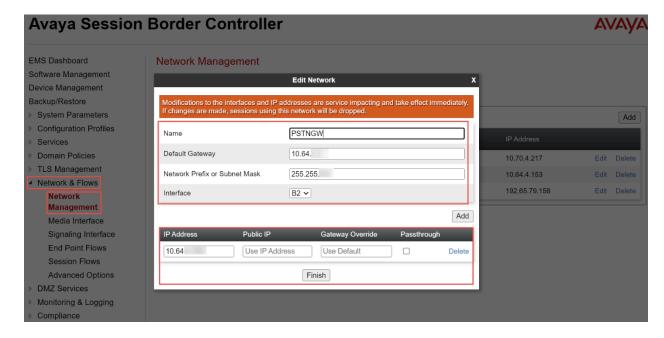


Figure 53: Network Management Facing PSTN Gateway

6.4.14 Signaling Interface

Signaling Interface for Avaya Aura SM

- Navigate to: Network & Flows > Signaling Interface. Click Add, new Add Signaling Interface window appears
- Set Name: AvayaSM10.2 is given for the interface facing Avaya Aura SM
- Set IP Address: Select LAN PBX
- Set TCP Port: 5060
- Click Finish

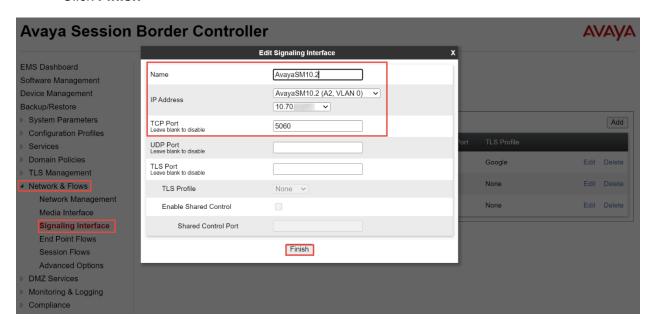


Figure 54: Signaling Interface Facing Avaya Aura SM

Signaling Interface for Google CCAI

 Repeat the same steps to create the Signaling Interface facing Google CCAI. TLS is used between Avaya SBC and Google CCAI.

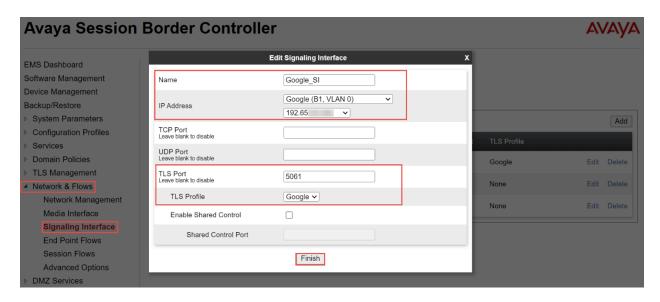


Figure 55: Signaling Interface Facing Google CCAI

Signaling Interface for PSTN Gateway

 Repeat the same steps to create the Signaling Interface facing PSTN. TCP is used between Avaya SBC and PSTN.

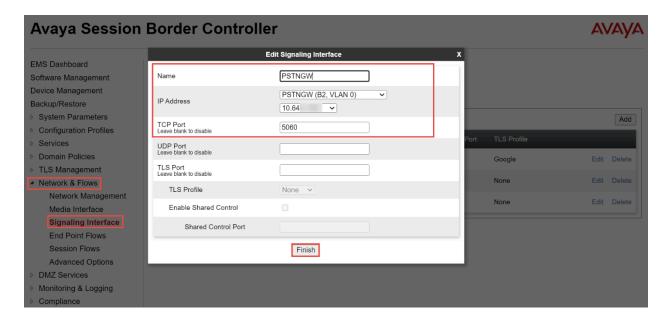


Figure 56: Signaling Interface Facing PSTN Gateway

6.4.15 End Point Flow

End Point Flow for PSTN Gateway

- Navigate: Network & Flows > End Point Flows > Server Flows Click Add
- Set SIP Server: AvayaSM10.2
- Select the required section: Received Interface, Signaling Interface, Routing Profile and Topology Hiding Profile

AVAYA

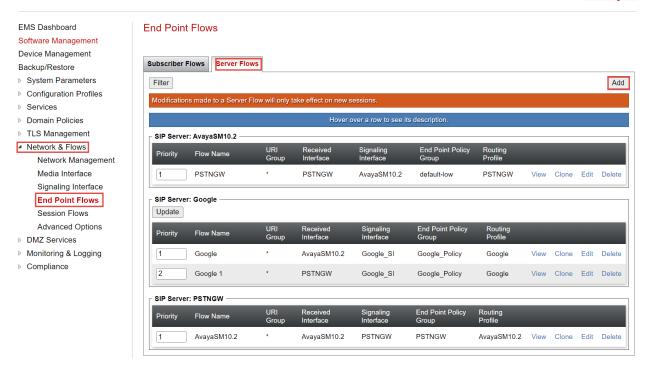


Figure 57: Server Flow for PSTN Gateway

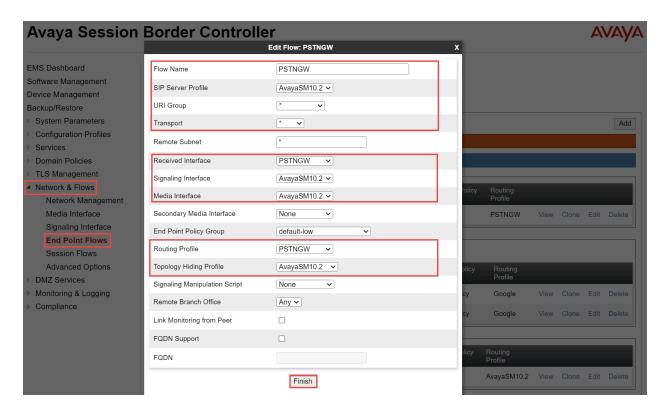


Figure 58: Server Flow for PSTN Gateway Continuation

End point flow for Google CCAI

- Navigate: Network & Flows > End Point Flows > Server Flows Click Add
- Set SIP Server: Google
- Select the required section: Received Interface, Signaling Interface, Routing Profile,
 End Point Policy Group, Topology Hiding Profile and Signaling Manipulation script



Figure 59: Server Flow for Google CCAI

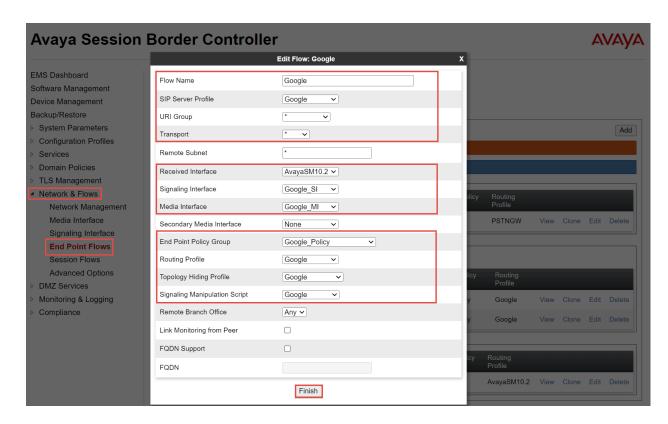


Figure 60: Server Flow for Google CCAI Continuation

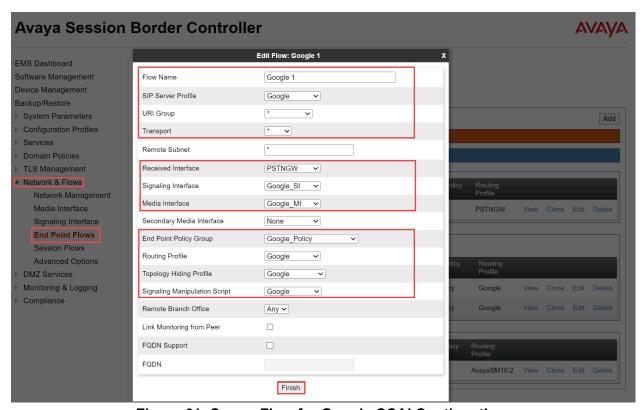


Figure 61: Server Flow for Google CCAI Continuation

End point flow for Avaya Aura SM

- Navigate: Network & Flows > End Point Flows > Server Flows Click Add
- Set SIP Server: PSTNGW
- Select the required section: URI Group, Received Interface, Signaling Interface, Routing Profile, Topology Hiding Profile



Figure 62: Server Flow for Avaya Aura SM

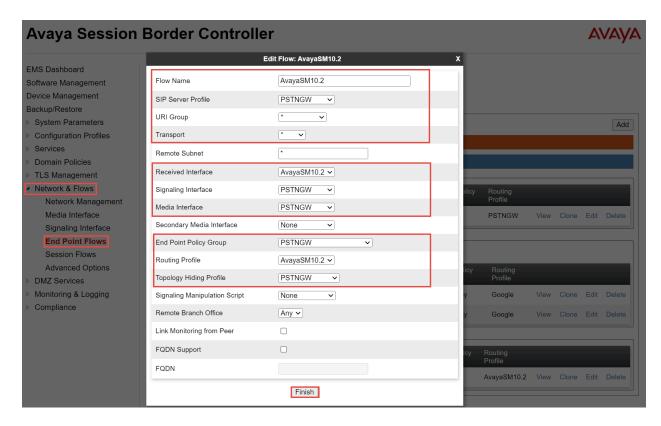


Figure 63: Server Flow for Avaya Aura SM Continuation

6.4.16 TLS Configuration

Creating SBC Certificate

• Navigate: TLS management > Certificates. Click Generate CSR

Avaya Session Border Controller





Figure 64: Generate CSR

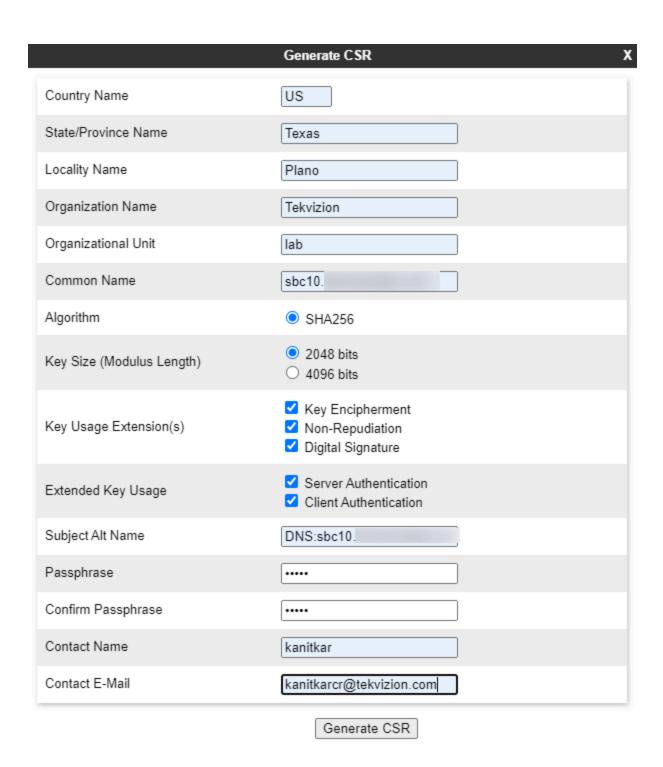


Figure 65: Generate CSR Continuation

- Navigate: TLS management > Certificates. Click Install
- Set Type: Select CA Certificate
- Set Name: GoogleRoot1CA (GTS Root R1)
- Set Allow weak Certificate/Key: Checked
- Set Certificate File: Click Choose File to select Google Root CA
- Click Upload
- Repeat the same steps to upload the GTS Root2.pem, GTS Root3.pem, GTS Root4.pem

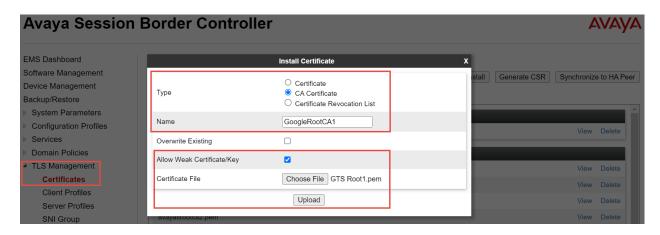


Figure 66: Upload Google Root CA

- Set Name: GoDaddy_Root
- Set Allow weak Certificate/Key: Checked
- Set Certificate File: Click Choose File to select Go_Daddy_Root.cer
- Click Upload

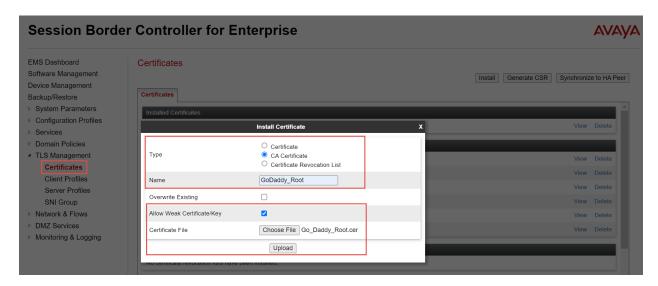


Figure 67: Upload GoDaddy Root CA

- Set Name: Go_Daddy_Secure
- Set Allow weak Certificate/Key: Checked
- Set Certificate File: Click Choose File to select Go_Daddy_Secure.cer
- Click Upload

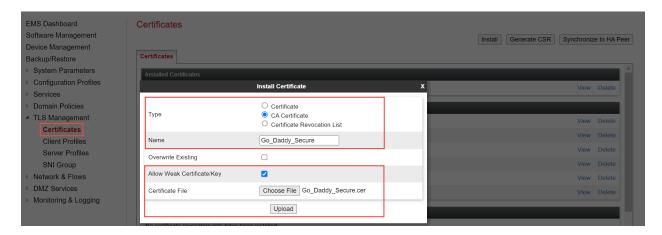


Figure 68: Upload GoDaddy Secure CA

- Navigate: TLS management > Certificates. Click Install
- Set Type: Select Certificate
- Set Name: sbc10
- Set Allow weak Certificate/Key: Checked
- Set Certificate File: Click Choose File to select sbc10.pem
- Select Use Existing Key
- Click Upload

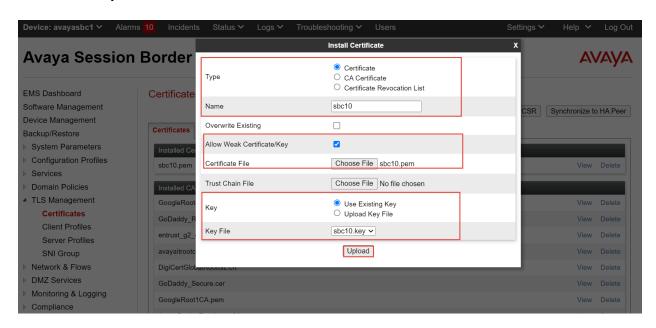


Figure 69: Upload SBC Certificate

Client Profile for Google CCAI

- Navigate: TLS management > Client Profiles. Click Add
- Set Profile Name: Google is given for interface facing Google
- Set Certificate: select server certificate sbc10.pem for Avaya SBC interface facing Google
- Set Peer Certificate Authorities: Select GoogleRoot1CA.pem, GoogleRoot2CA.pem,
 GoogleRoot3CA.pem, GoogleRoot4CA.pem which is uploaded in previous step
- Set Verification Depth: 5



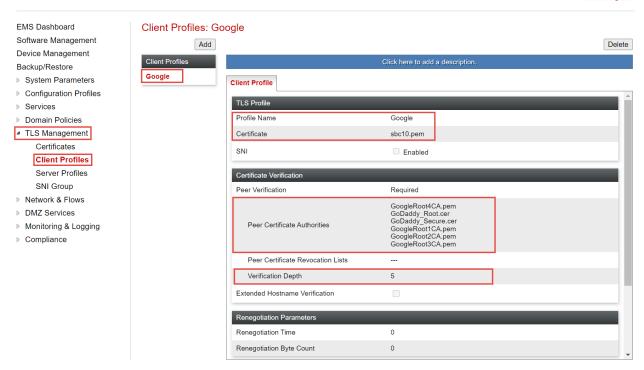


Figure 70: Client Profile facing Google CCAI

Set Version: Select TLS 1.2 versions



Figure 71: Client Profile facing Google CCAI continuation

- Navigate: TLS management > Server Profiles. Click Add
- Set Profile Name: Google is given for interface facing Google
- Set Certificate: Select server certificate sbc10.pem for Avaya SBCE interface facing Google
- Set Version: Select TLS 1.2 versions



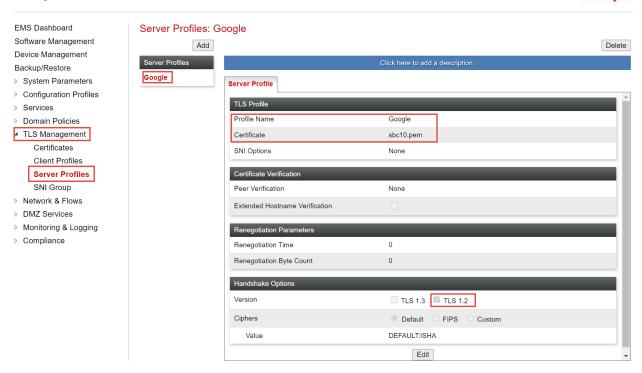


Figure 72: Client Profile facing Google CCAI continuation

Edit SIP Server

- Navigate: Services > SIP Servers
- Select Server Profile Google
- Under General tab, Click Edit
- Set Transport: Select **TLS** from Dropdown
- Set Port: **5672**
- Set TLS Client Profile: Select Client Profile Google
- Click Finish

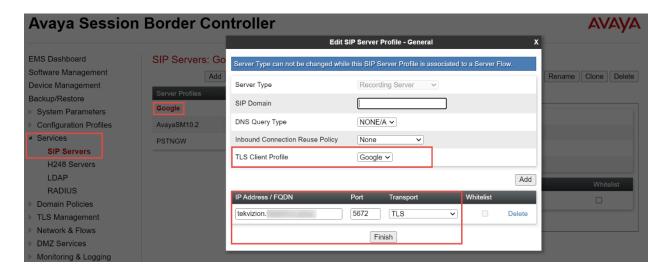


Figure 73: SIP Server Profile - Google CCAI

Configure SRTP

- Navigate: Domain Policies > Media Rules
- Select Media Rule default-low-med Click Clone
- Set Clone Name: Google_MR
- Click Next

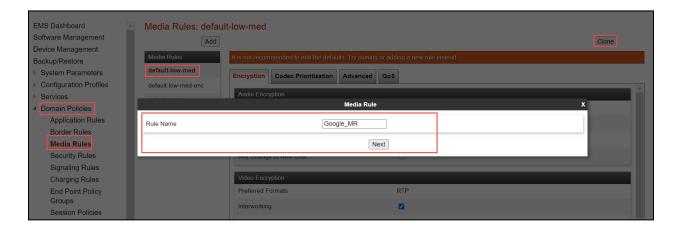


Figure 74: Media Rule - Google CCAI

- Select newly created Media Rule Google
- Set Preferred Format SRTP_AES_CM_128_HMAC_SHA1_80
- Set Encrypted RTCP: checked



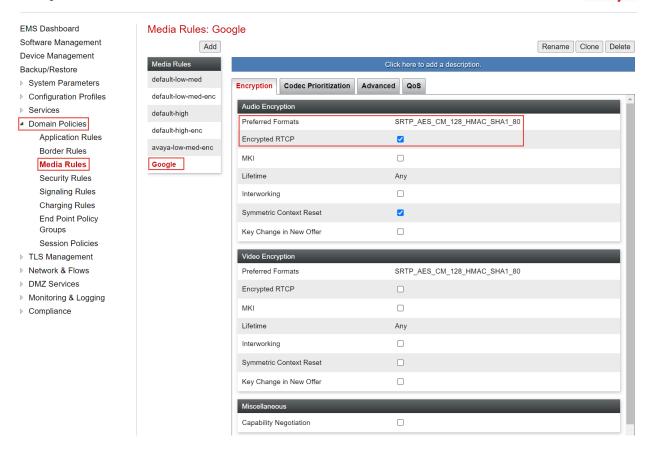


Figure 75: Media Rule-Google CCAI Continuation

Edit End Point Policy Groups

- Navigate to: **Domain Policies > End Point Policy Groups**
- Select Google under Policy Groups
- Click Edit

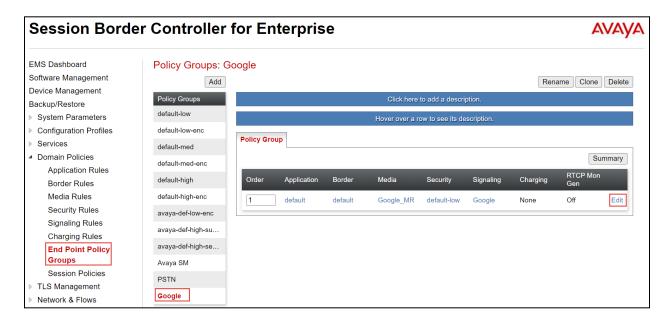


Figure 76:End Point Policy Group - Google CCAI

- Set Media Rule: Select Google
- Click Finish

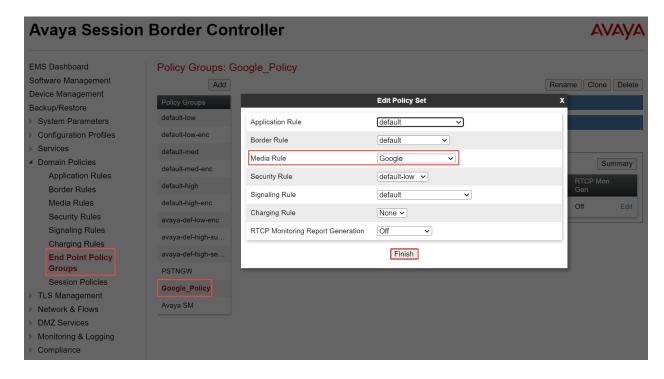


Figure 77:End Point Policy Group - Google CCAI Continuation

Edit Signaling Interface

- Navigate: Network & Flows > Signaling Interface
- Select interface Google_SI
- Click Edit

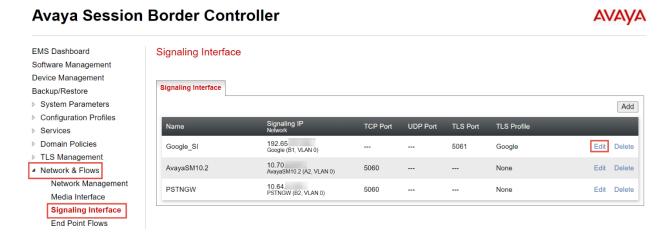


Figure 78: Signaling Interface - Google CCAI

Set TLS Port: 5061

- Set TLS Profile: Select Google from the drop-down menu
- Click Finish

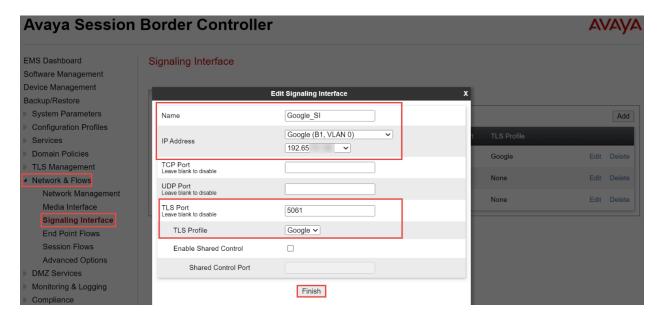


Figure 79:Signaling Interface – Google CCAI Continuation

7 Summary of Tests and Results

ID	Title	Description	Expected Results	Status (Passed or Failed etc)	Observations		
SBC	SBC Configuration Verification						
1	SBC Configuration Verification	TLS connection setup. SBC initiates TLS connection with CCAI	Successful 4way handshake with Google CCAI. Validate the right certificates are being negotiated. SBC should be loaded with GTSR1 cert for Google. SBC should also send the certificate chain when sending its cert.	PASSED			
2	SBC Configuration Verification	TCP Keep Alive. SBC will perform monitoring checks by attempting TCP Keep Alive to ensure Network Connectivity	Successful 3way handshake and thereafter termination	PASSED	TCP Keep-alive packets are sent to the SIPREC Trunk		
3	SBC Configuration Verification	TCP link is persistent. Establish call, send multiple calls that should all use the same TCP transport connection	Persistent TCP connection, we should establish a single connection and multiplex all calls over that connection.	PASSED			
4	SBC Configuration Verification	Session Timer support. SBC should be initiator for the Session Refresh timer using Update or Re-Invite	every 900 secs the SBC should refresh the SIP session.	PASSED	Avaya SBCE does not send session refresh re-invite. So Google sends session refresh every 15 minutes using RE-INVITE		

ID	Title	Description	Expected Results	Status (Passed or Failed etc)	Observations
5	SBC Configuration Verification	SIP Header Manipulation (call-info header)	Validate if the Google requested header manipulation is present in the SIP INVITE. Ensure every SDP media has a label.	PASSED	
6	SBC Configuration Verification	*SBCs may need further Header manipulations based on SIP stack constraints. Verify required manipulation are added in SBC to support Google CCAI Example: FROM, TO header manipulations HOST part change in headers etc,	All signaling in e.164 format	PASSED	
7	SBC Configuration Verification	SDES for SRTP. Configure the SDES parameters for crypto negotiation for the BYOT trunk	Validate the crypto is successfully negotiated and media is encrypted. All SBCs should support SDES for media encryption.	PASSED	
8	SBC Configuration Verification	DTLS for Media Encryption. Configure the DTLS parameters for crypto negotiation for the BYOT trunk, certificate for DTLS must be self-signed by the SBC.	Validate the crypto is successfully negotiated and media is encrypted.	NOT SUPPOR TED	Avaya SBC does not support DTLS
Inbo	ound				
9	SIP OPTIONS	SBC send SIP options every 60 seconds	Verify SBC sends SIP OPTIONS every 60 seconds and responded with 200 OK	PASSED	

ID	Title	Description	Expected Results	Status (Passed or Failed etc)	Observations
10	Inbound	Inbound call: Calling Party disconnects the call. Inbound siprec call, ensure recording are present, disconnect call from calling party and confirm proper disconnect	Verify Call is established with audio and transcripts from both participants Verify call is disconnected properly	PASSED	
11	Inbound	Inbound call: Called Party disconnects the call. Inbound siprec call, ensure recording are present, disconnect call from called party and confirm proper disconnect	Verify Call is established with audio and transcripts from both participants Verify call is disconnected properly	PASSED	
12	Inbound	Long duration call-Outbound Call- 1 hour max. Long duration siprec call	Ensure siprec calls stay up for an hour, confirm transcripts are present for entire duration	PASSED	Avaya SBCE does not send session refresh re-invite. So Google sends session refresh every 15 minutes using RE-INVITE
13	Inbound	Long duration hold and resume (wait until session audit\session refresh occurs from DUT). Long duration siprec call, have the call placed on hold by agent, have call resume. Have customer place on hold then have call resume.	Call is connected, we have two active streams, confirm once a stream goes on hold, we receive corresponding signaling events, and that we no longer record transcripts for the participant on hold.	PASSED	Avaya SBCE does not send session refresh re-invite. So Google sends session refresh every 15 minutes using RE-INVITE

ID	Title	Description	Expected Results	Status (Passed or Failed etc)	Observations
14	Inbound	Handling Error codes 603 decline. User A Calls PSTN A PSTN A rejects the incoming call	Verify SBC handles Call rejected properly	PASSED	
15	Inbound	Inbound call hold scenarios. Call starts out inactive for both participants, session moves to active	Validate if media is present when expected, confirm signaling events modify sdp properly, once call is move to active validate media and transcripts	PASSED	
16	Inbound	Inbound call hold scenarios. call starts out as active for both participants, session move to inactive, and transitions back to active	Validate if media is present when expected, confirm signaling events modify sdp properly, once call is moved to active validate media and transcripts	PASSED	This test case is tested with Skype for Business (SFB) as PBX to simulate sending of media attribute "sendonly" during Hold from SFB. When SFB user puts the call on hold, it sends "sendonly", PSTN hears MOH, MOH is recorded.
17	Inbound	Update. Validate that update sent prior to call establishment do not contain SDP	Validate that update prior to call establishment do not contain SDP as expected	PASSED	UPDATE is sent from the SBC

ID	Title	Description	Expected Results	Status (Passed or Failed etc)	Observations
18	Inbound	Update. Validate that updates post call establishment contain SDP to modify session	If SBC uses update to modify session, ensure SDP is included	NOT SUPPOR TED	
19	Inbound	re-invites. Ensure re-invites that modify session include SDP	Ensure re-invites that modify session include SDP	PASSED	Re-INVITE is sent to Google CCAI as part of session refresh, hold scenarios
20	Inbound	Codec negotiation. Ensure that g711 u-law is preferred codec	Ensure we can prioritize g711 as preferred codec, note where SBC configures preferred codec	PASSED	
21	Inbound	3 way conference. Determine requirements, record all leg.	Determine requirements, record all legs	PASSED	
22	Inbound	CCAI cloud project setup. Establish CCAI cloud project, provision the project with a GTP phone number for access (Create conversations/particip ants on the fly through SIP headers)	Verify project is setup, functional test to confirm you can connect to the GTP access phone number	PASSED	
23	Inbound	CCAI cloud project setup. Establish CCAI cloud project, provision the project with a GTP phone number for access (Pre-creation of conversations/particip ants)	Verify project is setup, functional test to confirm you can connect to the GTP access phone number	NOT APPLICA BLE	This test case is not applicable for call recording

ID	Title	Description	Expected Results	Status (Passed or Failed etc)	Observations
24	Inbound	Consultative transfer. Consultative transfer from 1. PSTN > User1 > User2 2. PSTN > User1 > PSTN user2		PASSED	
25	Inbound	Blind transfer. Blind transfer from 1. PSTN > User1 > User2 2. PSTN > User1 > PSTN user2		PASSED	Avaya PBX does not support blind transfer. This test case performed by ringing transfer