



ORACLE

Oracle SBC InterOP with PingCo TRE and
Microsoft Teams Phone Mobile

Technical Application Note

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COMMUNICATIONS



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1 Revision History

Document Version	Description	Revision Date
1.0	Initial release	09-04-2025
1.1	Added Root CA list and ECU considerations	30-01-2026

2 Intended Audience

This document outlines the integration of Oracle SBCs with the PingCo TCAP Routing Engine (TRE) for the purpose of Microsoft Teams Phone Mobile. This paper is intended for IT or telephony professionals.

3 Validated Oracle Software Versions

All testing was successfully conducted with the Oracle Communications SBC versions:

SCZ930 or above.

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

- AP 1100
- AP 3900
- AP 3950 (Release SCZ9.0.0 or later Only)
- AP 4600
- AP 4900 (Release SCZ9.0.0 or later Only)
- AP 6350
- AP 6300
- VME

Please visit <https://docs.microsoft.com/en-us/microsoftteams/direct-routing-border-controllers> for further information.

4 Related Documentation

4.1 Oracle SBC

- <https://docs.oracle.com/en/industries/communications/session-border-controller/9.2.0/aclireference/acli-reference-guide.pdf>
- <https://docs.oracle.com/en/industries/communications/session-border-controller/9.2.0/releasenotes/sbc-release-notes.pdf>
- <https://docs.oracle.com/en/industries/communications/session-border-controller/9.2.0/configuration/sbc-configuration-guide.pdf>

4.2 Microsoft Teams

<https://cloudpartners.transform.microsoft.com/partner-gtm/operators/teams-phone-mobile>

4.3 PingCo TRE

TCAP - <https://pingco.cloud/tcap>

TRE - <https://pingco.cloud/features/tre>

5 About Teams Phone Mobile with Oracle SBC and TRE

Microsoft Teams Phone Mobile is an intuitive, mobile-first Microsoft Teams experience that allows business users to access Teams capabilities through their mobile identity on both their native dialler and any Teams endpoint. The solution delivers cellular network quality of service to Teams communications, while allowing customers to enforce business policies, reduce costs, and improve the user experience for the growing mobile workforce. Through Teams Phone Mobile, Microsoft will collaborate closely with mobile network operators (MNOs), leveraging their unique mobile assets, including 5G technologies, to build a differentiated, high-quality, connected, and immersive mobile experience that can evolve with worldwide mobility trends.

5.1 Customer Benefits

- Create a unified business communication experience. Enables an inclusive workplace for mobile, remote, hybrid, and office workers by providing a reliable communication solution to work securely from either their native dialler or from the Teams app on any device.
- Reduce costs and eliminate redundancies. Allows customers to eliminate fixed lines for their mobile and remote workers and reduce international, long-distance, and intra-company mobile costs.
- Streamline management and governance. Provisions access usage and manages telephony services for all employees from one centralized place – the Office 365 portal. Provides enterprises the ability to enforce business policies, including security, compliance, and data governance protocols, even with wireless-only, 5G users.
- Deliver a business-grade mobile communications solution. Allows mobile network operators to build on future innovation of their 5G networks through partnership with Microsoft, empowering digital transformation for enterprise customers with a high-quality and differentiated user experience.

For a list of operators participating in the Microsoft Teams Phone Mobile program and the countries or regions where their service is available, see [Microsoft 365 Teams Phone Mobile](#).

5.2 PingCo TCAP Routing Engine

TCAP is all-in one, fully automated platform for Microsoft Teams Calling registration, provisioning, number management, billing, support and more. Built by PingCo, TCAP offers a powerful automation platform to self-manage your companies phone system. You can deploy your whole company's system in 15 minutes, and then within second stand-up new employees phone numbers. TCAP allows real-time management of your phone bill and reporting to give you power to make business decisions. With the heavy lifting covered, you can manage your own system and reduce costs with no set-up fees and low monthly costs. Enable your workforce to work from anywhere with Microsoft teams Calling providing and management with TCAP.

Note: This documents only focuses on the integration of Oracle SBC with PingCo TRE for Teams Phone Mobile calling. Provisioning of Service is out of scope.

The TCAP Routing Engine (TRE) is an advanced telecommunications solution designed to enhance the functionality of Session Border Controllers (SBCs) and optimize call routing processes.

As a powerful routing engine, TRE efficiently manages and directs incoming and outgoing calls by leveraging a range of intelligent features and extensive call analysis.

5.3 Plan for Teams Phone Mobile

Please follow below Microsoft Learn article to know more about Planning and configuring Teams Phone Mobile.

<https://learn.microsoft.com/en-us/microsoftteams/operator-connect-mobile-plan>

Ensure your organization has eligible Microsoft 365 services:

- Teams Phone System SKU or E5 with Teams
- Teams Phone Mobile add-on SKU

Below link provides details about configuring your Microsoft Teams Services for enabling Teams Phone Mobile.

<https://learn.microsoft.com/en-us/microsoftteams/operator-connect-mobile-configure>

5.4 Media Bypass vs Non-Media Bypass

Teams Phone Mobile can only work in Non-Media Bypass mode.

Media bypass enables you to shorten the path of media traffic and reduce the number of hops in transit for better performance. With media bypass, media is kept between the Oracle Session Border Controller (SBC) and the client instead of sending it via the Microsoft Phone System. Media bypass leverages protocols called **Interactive Connectivity Establishment (ICE)** on the Teams client and Advanced Media Termination ICE Lite on the Oracle SBC.

5.5 Teams Phone Mobile Call Scenarios utilizing Oracle SBC and TRE

- Each user is allocated with Operator's mobility number along with a SIM (Physical or eSIM). The mobile number is also used as User's Teams Identity.
- Intra Tenant (users in the same tenants Teams to Teams) calls are handled within the MS Teams network. However, Teams to Native dialler calls are routed to the mobile network operator via Oracle SBC and TRE to enable mobile default dialler ringing.
- Outbound call – Teams client to non-Teams number: Call is handed over to mobile operator by Microsoft for termination to called party. These calls also traverse Oracle SBC and TRE for necessary manipulations to the signalling.
- Inbound call – Non-Teams number to Teams client: Call is handed over to Microsoft by mobile operator for termination to called party. These calls also traverse Oracle SBC and TRE for necessary manipulations to the signalling.
- Users can initiate or receive calls from Teams Client on desktop, laptop, mobile over-the-top (OTT) or tablet. In addition, user call also uses the Mobile Native dialler to initiate or receive calls.
- International roaming users will be able to make / receive calls under standard roaming arrangements. In some countries, due to regulatory constraints, calls between Teams client and PSTN may be restricted.
- Number portability shall be applicable as per respective Geography regulatory rules.

5.6 Reference Breakdown

OC-SBC	Oracle Operator Connect SBC supporting Teams Phone Mobile Service
VoLTE	Voice over LTE- Mobile default dialler

MSFT	Microsoft
TPM	Teams Phone Mobile
Native Dialler	Default Handset Dialler of User Client
On-Net	Users calls within the same Mobile network
Off-Net	Users calls within different operators networks
Breakout	Call towards PSTN
MO	Mobile Originated Call
MT	Mobile Terminated Call
APP	Call from Teams App
TRE	PingCo TCAP Routing Engine

6 OC-SBC Interworking & Media requirements.

To integrate Teams into an Operators IMS Core, the Oracle SBC are the most valuable option as they are proven and interworks between a 3GPP defined platform (Carriers IMS Core) and non-3GPP Voice platform (Microsoft Teams).

6.1 Signalling Interworking Requirements:

- **100rel/PRACK Interworking** – Where IMS uses PRACK exclusively, Microsoft Teams does not, and this must be interworked at the IMS border. While OC-SBC is capable of performing PRACK interworking we are doing the PRACK interworking on P-CSCF in the test bed to support merge-early-dialogs.
- **Preconditions** – IMS networks MAY choose to implement Preconditions, which is not supported by Microsoft Teams. When this is used, Preconditions Interworking must be used at the border.
- **REFER termination and Replaces interworking** – Microsoft Teams has specific requirements for Call Transfers which require interworking at the border. In normal scenarios REFER will be handled by IMS Network's TAS but OC-SBC is also capable of handling REFERs. OC-SBC is also REFERs in the test bed.
- **Encryption Interworking** – Where an IMS core can be unencrypted, Microsoft Teams can optionally be encrypted to elevate security (although this is not a requirement of Operator Connect). When needed, encryption services are applied at the border. OC-SBC can perform the interoperability between an encrypted and unencrypted network for both signalling as well as Media.
- **Teams-specific Contact** – Microsoft Teams requires the Contact header be formatted with an FQDN as opposed to IP, which is not the case within 3GPP networks. OC-SBC converts the IP Address to FQDN of the Contact Header.
- **Local Media Playback** – Oracle SBC performs can perform Local Media Playback when required for generating ringback tones and during early media scenarios.
- **SBC Interworking for handling SDP offer in a-line (call hold/waiting)** – During call hold scenarios Oracle SBC performs the conversion of SDP a line from Caller towards Microsoft to convert the attribute to inactive.
- **Transcoding of unsupported codecs** – Where IMS cores use both AMR-WB and AMR, Microsoft Teams only supports AMR-WB, so calls delivered using AMR must be transcoded.
- **RTCP** – While IMS calls may support end to end RTCP, Microsoft Teams requires it, so selective RTCP Generation at the border is required to ensure service continuity.

- **Comfort Noise** – IMS does not implicitly require Comfort Noise (CN) packets, whereas Microsoft Teams prefers this. Comfort Noise generation at the border provides a smoother experience.

6.2 PingCo TRE Header Requirements:

This section outlines the call routing process of the TRE platform through the Oracle SBC (Session Border Controller). The section describes the SIP headers used for routing purposes and their respective scenarios. The headers include:

- X-TRE-CallType
- X-TRE-TrunkType (Not used for Teams Phone Mobile, Informational Only)
- X-TRE-Trunk Prefix
- X-TRE-OperatorProfile
- X-TRE-OutboundCarrier
- X=TRE-SourcePlatform

X-TRE-CallType - The X-TRE-CallType header is used to distinguish the call as either inbound or outbound. This header provides information about the direction of the call and can be utilized for various purposes within the platform. By examining the CallType header, the SBC and other components of the system can differentiate between inbound and outbound calls and perform specific actions or apply customized routing logic based on the call direction.

X-TRE-TrunkType - The X-TRE-TrunkType header is utilized to route calls to specific applications, such as Teams. The value of this header determines the target application for call routing. For instance, if the TrunkType is defined as "Direct Routing," the SBC should identify this trunk type and route the call to the Teams application. The TrunkType header is typically a GUID, for example: 9f43b6ea-facc-4766-bec0-47b3e083b1c2.

X-TRE-TrunkType - The X-TRE-TrunkType header is utilized to route calls to specific applications, such as Teams. The value of this header determines the target application for call routing. For instance, if the TrunkType is defined as "Operator Connect," the SBC should identify this trunk type and route the call to the Teams application. The TrunkType header is typically a GUID, for example: **X-TRE-TrunkType: 9bcaa351-df28-4457-32b4-08d9e15f3200**

SIP Trunk Selection :

In the case of SIP trunks, the TrunkPrefix serves as a key element for determining the appropriate trunk to select when sending a call. When the SBC analyses the TrunkType as "SIP Trunk," it further utilizes the TrunkPrefix to make routing decisions. By considering the TrunkPrefix in conjunction with the TrunkType, the SBC can determine the specific trunk through which the call should be routed.

X-TRE-OperatorProfile - The X-TRE-OperatorProfile header is used for routing calls to a specific Operator Connect instance. When the TrunkType is defined as "Operator Connect," the SBC references the Operator Profile field in the X-TRE-OperatorProfile header to determine which trunk the calls need to be routed to. This header is particularly useful in multi-tenant environments where an SBC is working with multiple operator profiles or when an operator has multiple Operator Connect instances in Teams. By considering the Operator

Profile field, the SBC can make informed routing decisions based on the specific requirements and configurations associated with each operator.

X-TRE-OutboundCarrier - The X-TRE-OutboundCarrier header is responsible for routing outgoing calls to the appropriate carrier. This header helps the SBC identify the preferred outbound carrier for the call. By utilizing the OutboundCarrier header, the SBC can ensure that the call is sent through the designated carrier that aligns with the requirements and preferences of the platform.

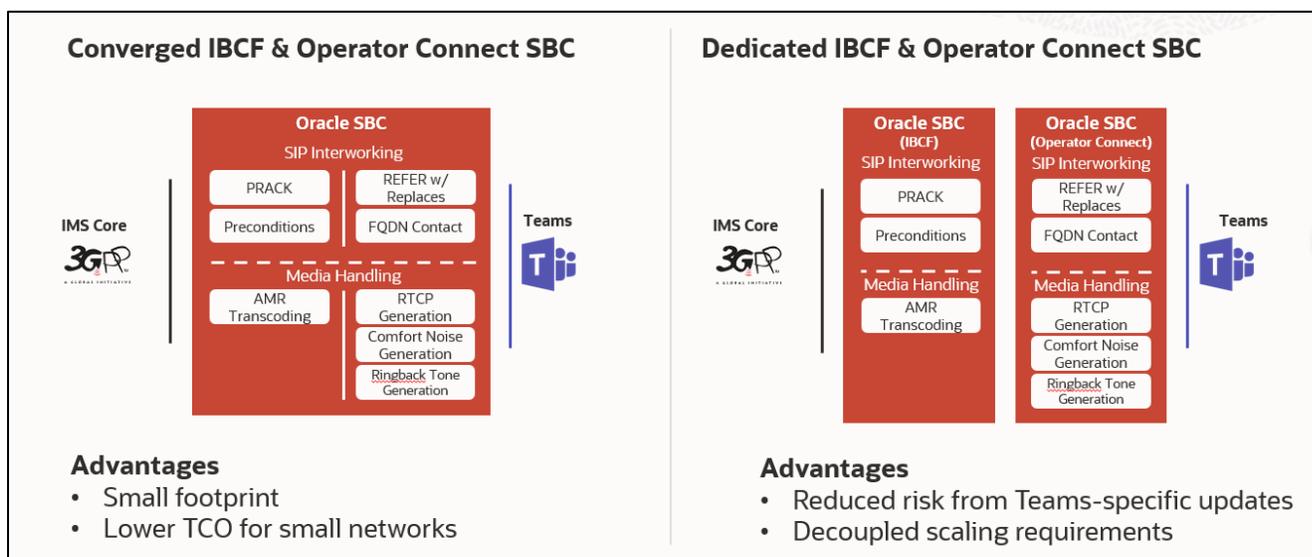
Operator Connect Routing When the TrunkType is defined as "Operator Connect," the SBC references the Operator Profile field in the X-TRE-OperatorProfile header to determine the specific trunk to which the calls should be routed. This routing mechanism is particularly useful in multi-tenant environments or when an operator has multiple Operator Connect instances in Teams. The Operator Profile field helps the SBC make routing decisions based on the specific operator requirements and configurations.

Outbound Carrier Selection The X-TRE-OutboundCarrier header plays a crucial role in routing outgoing calls. By specifying the preferred outbound carrier in this header, the SBC ensures that the call is directed through the designated carrier. The choice of carrier is based on the requirements and preferences defined within the OutboundCarrier header.

Conclusion In summary, we have outlined the SIP headers used for call routing within the TRE platform when handling calls on an SBC. The document has described the headers' purposes and how they are utilized in different scenarios. The information presented serves as a guide for implementing dynamic call routing functionality on an SBC in conjunction with TRE.

7 SBC Deployment options

7.1 Converged vs Dedicated SBC Options



When integrating an Oracle SBC with your IMS core for Microsoft Teams Phone Mobile, you have two options:

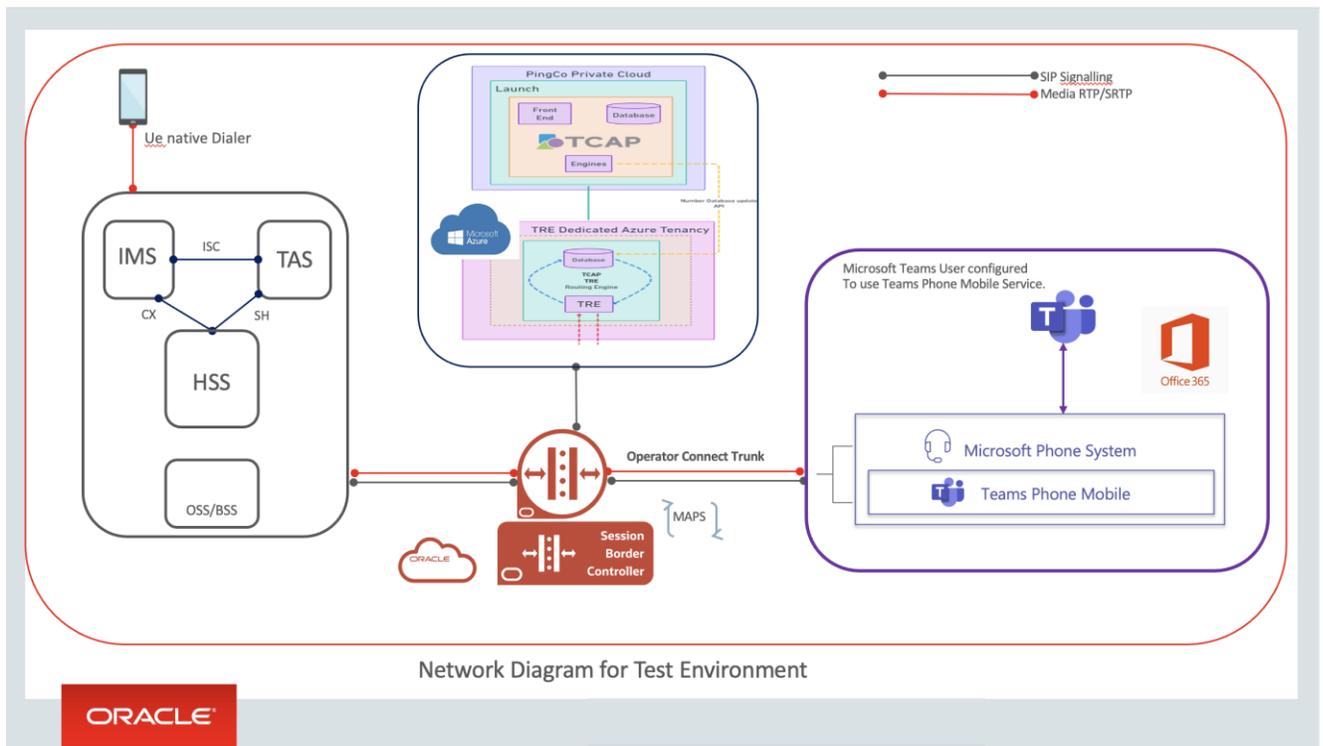
- Converged SBC: Configure your Oracle SBC to handle both IBCF and Operator Connect SBC functionalities.
- Dedicated SBC: Introduce a separate SBC dedicated to interworking with Teams Phone Mobile.

We recommend the Dedicated SBC approach for integrating an SBC with your IMS core to support Microsoft Teams Phone Mobile. This involves introducing a separate SBC to handle all required functionality. While a Converged SBC option using the Oracle SBC is also possible, we've focused on testing and documenting the Dedicated SBC approach for this application note.

8 Network topology

The following figure shows a network diagram of the test environment that is used for the Interop. Some key points to Note are –

- Oracle SBC is deployed in OCI Environment in Oracle's Development Tenant.
- Pingo TRE is hosted in Microsoft Azure and connects to Oracle SBC over UDP for the purpose of the Application Note.
- Oracle's Development Microsoft 365 Tenant is used to build the Teams Phone Mobile Service.
- All signalling traverse from Microsoft to Oracle SBC to PingCO TRE. Media from Microsoft Phone System is sent to Oracle SBC for termination to Peer Party.
- Teams Phone Mobile Trunk can also use TCP/RTP Protocol. Use of MAPS (Microsoft Azure Peering Service) Transport is a MUST for Network to Network Connection between the Oracle SBC and Operator Connect Teams Phone Mobile. Traffic sent through 3rd Part Internet is not supported. For the purpose of the Application Note we have provided TLS/SRTP method of connectivity between Oracle SBC and Microsoft Teams Phone Mobile.



9 Oracle SBC Configuration

This chapter provides step-by-step guidance on how to configure Oracle SBC for interworking with PingCo TRE for the purpose of Microsoft Teams Phone Mobile.

Note :- In the running configuration you will find configuration related to PSTN connectivity because in the current setup PSTN breakout is also terminated onto OC-SBC. While the configuration is shown in the ACLI output and the ACLI running configuration, it is not highlighted as part of the Application Note.

This guide assumes the OC-SBC has been installed, management interface has been configured, product selected and entitlements have been assigned.

If you require more information on how to install your SBC platform, please refer to the [ACLI configuration guide](#).

To access the ACLI on your OC-SBC, ssh to the management IP address or access via SBC console port:

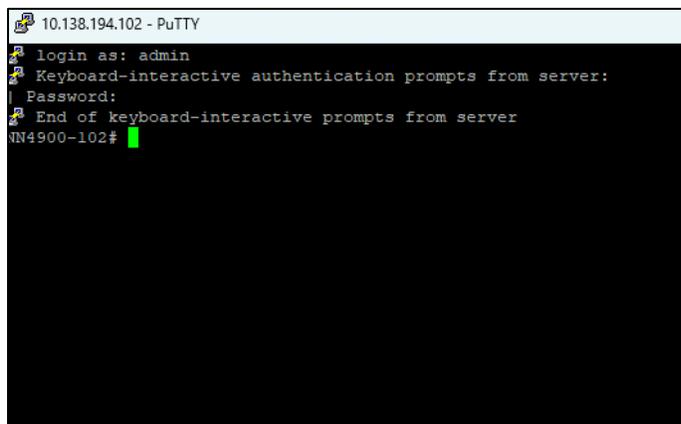
Console Settings:

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

When the login screen appears, enter the username and password to access the OC-SBC.

Any configuration parameter not specifically listed below can remain at the OC-SBC default value and does not require a change for the connection to Microsoft Teams Phone Mobile to function properly.

Note: the configuration examples below were captured from a system running the latest GA software, 9.2.0



```
10.138.194.102 - PuTTY
login as: admin
Keyboard-interactive authentication prompts from server:
Password:
End of keyboard-interactive prompts from server
VN4900-102#
```

9.1 System-Config

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

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

- Hostname
- Description
- Location
- Default Gateway (recommended to be the same as management interface gateway)
- Transcoding Core (This field is only required if you have deployed a VME SBC)

To configure system-config from ACLI –

ACLI Path: config t→system→system-config

```
system-config
hostname          oraclesbc.com
description       SBC connecting IMS to Teams Phone Mobile
location          Burlington, MA
transcoding-cores 1
```

9.1.1 NTP-Sync

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

To configure NTP from ACLI –

ACLI Path: config t→system→ntp-sync

```
ntp-config
server                216.239.35.0
```

Now we'll move on configuring network connection on the SBC.

9.2 Network Configuration

To connect the SBC to network elements, we must configure both physical and network interfaces. For the purposes of this example, we will configure two physical interfaces, and two network interfaces. One to communicate with Microsoft Teams Phone Mobile, the other to connect to an IMS Network.

The slots and ports used in this example may be different from your network setup.

9.2.1 Physical Interfaces

- Use the following table as a configuration example:

Config Parameter	Teams Phone Mobile	IMS	PingCo TRE
Name	s0p0	S1p0	S1p1
Operation Type	Media	Media	Media
Slot	0	1	1
Port	0	0	0

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

To configure Physical Interfaces from ACLI –

ACLI Path: config t→system→phy-interface

```
phy-interface
name                s0p0
operation-type      Media
phy-interface
name                s1p0
operation-type      Media
slot                1
phy-interface
name                s1p1
operation-type      Media
port                1
slot                1
```

9.2.2 Network Interfaces

- Use the following table as a configuration example:

Configuration Parameter	Teams Phone Mobile	IMS	PingCo TRE
Name	S0p0	S1p0	S1p1
IP Address	10.0.2.10	10.0.3.10	10.0.5.27
Netmask	255.255.255.0	255.255.255.0	255.255.255.0
Gateway	10.1.2.1	10.1.3.1	10.0.5.1
DNS Primary IP	8.8.8.8		
DNS Domain	cloudsbc.cgbusolutionslab.com		

To configure Network Interfaces from ACLI –

ACLI Path: config t → system → network-interface

```
network-interface
  name          s1p0
  ip-address    10.0.3.10
  netmask       255.255.255.0
  gateway       10.1.3.1
network-interface
  name          s0p0
  ip-address    10.0.2.10
  netmask       255.255.255.0
  gateway       10.1.2.1
  dns-ip-primary 8.8.8.8
  dns-ip-backup1 8.8.4.4
  dns-ip-backup2 9.9.9.9
  dns-domain    Cloudsbc.cgbusolutionslab.com
network-interface
  name          s1p1
  ip-address    10.0.5.27
  netmask       255.255.255.0
  gateway       10.0.5.1
```

Next, we'll configure the necessary elements to secure signalling and media traffic between the Oracle SBC and Microsoft Teams Phone Mobile.

9.3 Security Configuration

This section describes how to configure the SBC for both TLS and SRTP communication with Microsoft Teams Phone Mobile. For the purpose of our testing communication between Oracle SBC and PingCo TRE is UDP. TRE only handles signalling and media stays between Microsoft Network and IMS via Oracle SBC.

Note: Teams Phone Mobile Trunk can also use TCP/RTP Protocol. Use of MAPS (Microsoft Azure Peering Service) Transport is a MUST for Network to Network Connection between the Oracle SBC and Operator Connect Teams Phone Mobile. Traffic sent through 3rd Part Internet is not supported. For the purpose of the Application Note we have provided TLS/SRTP method of connectivity between Oracle SBC and Microsoft Teams Phone Mobile.

When Using TLS/SRTP Microsoft Operator Connect recommends TLS connections from SBC's for SIP traffic, and SRTP for media traffic. It requires a certificate signed by Certificate Authorities (CAs) that are part of the [Microsoft Trusted Root Certificate Program](#). A list of currently supported Certificate Authorities can be found at: [Public trusted certificate for the SBC](#). These are same as Direct Routing Supported CAs.

9.3.1 Certificate Records

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

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

GUI Path: security/certificate-record

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

- SBC Certificate (end-entity certificate)
- DigiCert RootCA Cert (Root CA used to sign the SBC's end entity certificate)
- [Microsoft Root Certificate Authorities](#) (Microsoft Presents the SBC a certificate signed by one of these authorities)

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

9.3.1.1 SBC End Entity Certificate

The SBC's end entity certificate is the certificate the SBC presents to Microsoft to secure the connection. The only requirements when configuring this certificate is the common name must contain the SBC's FQDN and the **extended key usage list** must contain **serverAuth**. Including **clientAuth** is optional for now as Microsoft Teams Direct Routing currently permits the use of SBC client certificates even if the Client Authentication EKU is not included.

However, [Microsoft has indicated](#) that in the future, all SBC client certificates will be required to include the Client Auth EKU. When this enforcement goes into effect, a list of publicly trusted certificate authorities (CAs) that can issue such certificates will be published.

It's important to note that public CAs may stop including the Client Authentication EKU in certificates due to updated [industry requirements](#) and CA policies. You should check with your CA to determine when they plan to stop including the Client Authentication EKU by default, so you can plan accordingly.

For more information, please refer to:

<https://learn.microsoft.com/en-us/microsoftteams/direct-routing-whats-new#update-on-upcoming-certificate-changes-updated-december-12-2025>

and

<https://www.oracle.com/a/otn/docs/microsoft-teams-ca-changes-and-eku-considerations.pdf>

In this example our common name will be **cloudsbc.cgbusolutionslab.com**. You must also give it a name and we have included **clientAuth** to the **extended key usage list**.

For now, mutual TLS connections between your Oracle SBC and Microsoft Teams will continue to be established, even if the root CA removes or no longer supports the clientAuth EKU. Looking ahead, including the clientAuth EKU in your SBC's end entity certificate will be important to maintain compatibility and avoid future issues with Microsoft Teams Direct Routing. When submitting your CSR for signing, work with your CA to make sure the required EKU is maintained during the signing process.

If you generate a CSR using a certificate record that includes both serverAuth and clientAuth EKUs, but the CA removes the clientAuth EKU when signing the certificate, you can still import the resulting certificate into the SBC without any errors. The SBC will accept and present the certificate even if the clientAuth EKU is not included after signing.

All other fields are optional, and can remain at default values.

The SBC's end entity certificate is the certificate the SBC presents to Microsoft to secure the connection. There are two requirements when configuring this certificate.

1. Common name must contain the SBC's FQDN
2. extended-key-usage-list must contain both serverAuth and clientAuth.

To Configure the certificate record from ACLI:

ACLI Path: `config t` → `security` → `certificate-record`

certificate-record	
name	SBCCertificateforTPM
state	California
locality	Redwood City
organization	Oracle Corporation
unit	Oracle CGBU-LABS BOSTON
common-name	cloudsbc.cgbusolutionslab.com
key-usage-list	digitalSignature keyEncipherment
extended-key-usage-list	clientAuth serverAuth

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

9.3.1.2 Save and Activate

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

```
NN4900-102# save-config
checking configuration
-----
Results of config verification:
  3 configuration errors
  2 configuration warnings
Run 'verify-config' for more details
-----
Save-Config received, processing.
save-config waiting 120000 ms for request to finish
Request to 'SAVE-CONFIG' has Finished,
Save complete
Currently active and saved configurations do not match!
To sync & activate, run 'activate-config' or 'reboot activate'.
NN4900-102# activate-config
Activate-Config received, processing.
activate-config waiting 120000 ms for request to finish
Request to 'ACTIVATE-CONFIG' has Finished,
Activate Complete
NN4900-102#
```

9.3.1.3 Generate Certificate Signing Request

Now that the SBC's certificate has been configured, create a certificate signing request for the SBC's end entity only.

This is not required for any of the Root CA or intermediate certificates that have been created.

To perform the Steps From ACLI use the below command –

```
NN4900-102# generate-certificate-request SBCCertificateforTPM

--This Step generates a text on Screen as shown below --

-----BEGIN CERTIFICATE REQUEST-----
MIIC4zCCAcsCAQAwazELMAkGA1UEBhMCVVMxCzAJBgNVBAGTAk1BMRMwEQYDVQQH
EwpCdXJsaW5ndG9uMRQwEgYDVQQKEwtFbmdpbmVlcmluZzEkMCIGA1UEAxMbdGVs
ZWN0eXQub3R0YXZlYXN0MDYxNjE5NzcuY29tMIIBIjANBgkqhkiG9w0BAQEFAAOCAQ8A
MIIBCgKCAQEAr3AmjF15PcIcWiB/kFExUGNHQHlIbkJi28MDbcprO/KLXIHQysSnw
UWz34XLBfLQ6rS4MLyEMR8Nt8GGNSIWkiR431LsX7L+yGWvRjcBFP6DIHtH0Vuqm
ixVaUJpg5luPY6SvT1shyu26iLIBsLfem43tbKq5jz/jrvaUzyhlCvAQ23c1oS5a
D4UiF2mNOuSqvvmkx50a3/BNYbKecLNOxvKQyyTMgffNpASbZuW+eMEUKI5iB+AB
/AAoZRP4bn4qIE3wn8pJsNm8Pjxy4hbz24ySgmaN9iXpP1FdRw0TemfCsNazZRuK
DsviWJfunZYTzRfDe5pJTtoMH4u1zt2fK1QIDAQABoDMwMQYJKoZIHvcNAQkOMSQw
IjALBgNVHQ8EBAMCBaAwEwYDVR0IBAwcCgYIKwYBBQUHAWewDQYJKoZIHvcNAQEL
BQADggEBADD5Y+u08LxmTMsJ2Rjc8cgPZocTqBDXN0tp27S4FuB/01ikBBdG3YV
Ffp7/Q8ZeFHHgU/rMzeF8Gpo9Cc6JUGGux3/ws8ZkgRBxsNIG276i7pFN1vC1jEP
89AGxtryioRMc4kcdPpLJNQ10Qx1zKobHMTftGLDI6jN2pvn3zYHH8qA9V/1/yKa
3n0j33EuTrvTIQ5P4IgyVJqSBkdI29T1gXY6O8JVFLCQefTrF4TLc6teNzxXMDpw
PHoPu9hM3scGOWOHQnODXOFeq2AxBQzAa0/Cjf7Bw3l3POmMcIOawgDecZ8UjHpJ
lznX9/Gxg5X+S2QkHjNmPK+JuePqX4I=
```

-----END CERTIFICATE REQUEST-----

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

Also note, at this point, **another [save and activate](#) is required** before you can import the certificates to each certificate record created above.

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

9.3.1.4 Root CA and Intermediate Certificates

9.3.1.4.1 DigiCert Root CA

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

9.3.1.4.2 Microsoft Certificates

The DNS name of the Microsoft Teams Direct Routing interface is sip.pstnhub.microsoft.com. Microsoft presents a certificate to the SBC which is signed by one of the CA's listed in the table below. To trust this certificate, your SBC must have all the certificate listed below as a trusted CA certificate.

Download each certificate from the official source using the links provided below:

Certificate Authority	Download Link
DigiCert Global Root CA	DigiCert Global Root CA
DigiCert Global Root G2	DigiCert Global Root G2
DigiCert Global Root G3	DigiCert Global Root G3
DigiCert TLS ECC P384 Root G5	DigiCert TLS ECC P384 Root G5
DigiCert TLS RSA 4096 Root G5	DigiCert TLS RSA 4096 Root G5
Microsoft ECC Root Certificate Authority 2017	Microsoft ECC Root Certificate Authority 2017
Microsoft RSA Root Certificate Authority 2017	Microsoft RSA Root Certificate Authority 2017

The certificates listed in the table above can also be found at:

<https://learn.microsoft.com/en-us/azure/security/fundamentals/azure-ca-details?tabs=root-and-subordinate-cas-list>

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

Configuration View Configuration

media-manager > Certificate Record

security >

authentication-profile >

certificate-record

global-trusted-ca >

tls-global >

tls-profile >

session-router >

system >

PKCS12 Delete all Certificate Record items

Action	Name ^	Country ^	State ^	Locality ^
<input type="checkbox"/>	DigiCertGlobalRootG2	US	MA	Burlington
<input type="checkbox"/>	DigiCertGlobalRootG3	US	MA	Burlington
<input type="checkbox"/>	DigiCertRoot	US	MA	Burlington
<input type="checkbox"/>	DigiCertTLSECCP384RootG5	US	MA	Burlington
<input type="checkbox"/>	DigiCertTLSECCP4096RootG5	US	MA	Burlington
<input type="checkbox"/>	GoDaddyRoot	US	MA	Burlington
<input type="checkbox"/>	GoDaddyinter	US	MA	Burlington
<input type="checkbox"/>	MicrosoftECCRootCertificateAuth...	US	MA	Burlington
<input type="checkbox"/>	MicrosoftRSARootCertificateAuth...	US	MA	Burlington
<input type="checkbox"/>	Telechat2025	US	Texas	Austin

Please note – all certificates including root and intermediate certificates are required to be imported to the SBC.

9.3.1.5 Import Certificates to SBC

Once certificate signing request has been completed – import the signed certificate to the SBC. All certificates including root and intermediate certificates are required to be imported to the SBC.

After all certificates have been imported, issue a third [save/activate](#) to complete the configuration of certificates on the Oracle SBC.

To import the certificate from ACLI follow below procedure -

```
NN4900-102# import-certificate try-all SBCCertificateforTeams
```

The System will show a prompt as below -

IMPORTANT:

Please enter the certificate in the PEM format.
Terminate the certificate with ";" to exit.....

Enter the Signed Certificate text as shown below-

```
-----BEGIN CERTIFICATE REQUEST-----
MIIC4zCCAcsCAQAwazELMAkGA1UEBhMCVVMxCzAJBgNVBAGTAK1BMRMwEQYDVQQH
EwpCdXJsaW5ndG9uMRQwEgYDVQQKEwtFbmdpbmVlcmluZzEkMCIGA1UEAxMbdGVs
ZWN0YXQubW5ndG9uMRQwEgYDVQJDEjEjE5NzcuY29tMIIBIjANBgkqhkiG9w0BAQEFAAOCAQ8A
MIIBCgKCAQEAr3AmjF15PcIcWiB/kFExUGNHQHibkji28MDbcpO/KLXIHQysSnw
UWz34XLBfLQ6rS4MLyEMR8Nt8GGNSIWKiR431LsX7L+yGWvRjcBFP6DIHtH0VUqgm
ixVaUJpg5luPY6SvT1shyu26iLlBsLfem43tbKq5jz/jrvaUzyhlCvAQ23c1oS5a
D4UiF2mNOuSqxvmkx50a3/BNybKecLNOxvKQyyTMgffNpASbZuW+eMEUKI5iB+AB
/AAoZRP4bn4qIE3wn8pJsNm8Pjxy4hbz24ySgmaN9iXpP1FdRw0TemfCsNazZRuK
DsviWJfunZYTzRfDe5pJTtoMH4u1zt2fK1QIDAQABoDMwMQYJKoZIHvcNAQkOMSQw
IjALBgNVHQ8EBAMCBaAwEwYDVR0IBAwwCgYIKwYBBQUHAwEwDQYJKoZIHvcNAQEL
```

```
BQADggEBADD5Y+u08LxmTMIsJ2Rjc8cgPZocTqBDXN0tp27S4FuB/01ikBBdG3YV
Ffp7/Q8ZeFHHgU/rMzeF8Gpo9Cc6JUGGux3/ws8ZkgRBxsNIG276i7pFN1vCljEP
89AGxtryioRMc4kcdPpLJNQ10Qx1zKobHMTftGLDI6jN2pvn3zYHH8qA9V/1/yKa
3n0j33EuTrvTIQ5P4IgyVJqSBkdI29T1gXY6O8JVFLCQefTrF4TLc6teNzxXMdPw
PHoPu9hM3scGOWOHQnODXOFeq2AxBQzAa0/Cjf7Bw3l3POmMcIOawgDecZ8UjHpJ
lznX9/Gxg5X+S2QkHjNmPK+JuePqX4I=
-----END CERTIFICATE REQUEST-----;
```

[save and activate](#) your configuration.

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

9.3.2 TLS Profile

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

To configure system-config from ACLI –

ACLI Path: config t→security→tls-profile

```
tls-profile
  name TLSTeams
  end-entity-certificate SBCCertificateforTeams
  trusted-ca-certificates
    DigiCertRoot
    DigiCertGlobalRootG2
    DigiCertGlobalRootG3
    DigiCertTLSECCP384RootG5
    DigiCertTLSECCP4096RootG5
    MicrosoftECCRootCertificateAuthority2017
    MicrosoftRSARootCertificateAuthority2017
  mutual-authenticate enabled
  tls-version tlsv12
```

Next, we'll move to securing media between the SBC and Microsoft Teams Phone Mobile.

9.3.3 Media Security

This section outlines how to configure support for media security between the OC-SBC and Microsoft Teams Phone Mobile.

9.3.3.1 SDES-Profile

This is the first element to be configured for media security, where the algorithm and the crypto's to be used are configured. The only crypto-suite option supported by Microsoft is AES_CM_128_HMAC_SHA1_80 and must be included in the crypto list

To configure system-config from ACLI –

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

```

sdes-profile
  name                TeamsSRTP
  crypto-list         AES_CM_128_HMAC_SHA1_80
  srtp-auth           enabled
  srtp-encrypt        enabled
  srtpc-encrypt       enabled
  mki                 disabled
  egress-offer-format same-as-ingress
  use-ingress-session-params
  options
  key
  salt
  srtp-rekey-on-re-invite disabled
  lifetime            31

```

9.3.3.2 Media Security Policy

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

In this example, we are configuring two media security policies. One to secure and decrypt media toward Microsoft Teams, the other for non-secure media facing IMS Core.

To configure media security from ACLI.

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

```

media-sec-policy
  name                IMSNonSecure
  pass-through        disabled
  options
  inbound
    profile
    mode              rtp
    protocol          none
    hide-egress-media-update disabled
  outbound
    profile
    mode              rtp
    protocol          none
media-sec-policy
  name                TeamsMediaSecurity
  pass-through        disabled
  options
  inbound
    profile           TeamsSRTP
    mode              srtp
    protocol          sdes
    hide-egress-media-update disabled

```

outbound	
profile	TeamsSRTP
mode	srtp
protocol	sdes

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

9.4 Transcoding Configuration

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

9.4.1 Media Profiles

For different codecs and media types, you can setup customized media profiles that serve to police media values and define media bandwidth policies.

SILK & CN offered by Microsoft teams are using a payload type which is different than usual, so to support this, we configure the following media profiles on the SBC.

This is an optional configuration, and only needs to be implemented on the SBC if you are planning to use the SILK codec or wideband comfort noise between the SBC and Microsoft Teams Phone Mobile -TPM.

Configure three media profiles to support the following:

- Silk Wideband
- Silk Narrowband
- CN

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

Parameters	Silk	Silk	CN
Surname	narrowband	wideband	wideband
Payload-Type	103	104	118
Clock-rate	8000	16000	0

Besides, if the Network uses AMR the following media-profiles can be used as reference to create different media profiles for AMR-WB and AMR codec.

To configure system-config from ACLI –

ACLI Path: config t → session-router → media-profile

media-profile	
name	CN

subname	wideband
payload-type	118
media-profile	
name	SILK
subname	narrowband
payload-type	103
clock-rate	8000
media-profile	
name	SILK
subname	wideband
payload-type	104
clock-rate	16000
media-profile	
name	AMR-WB
subname	LOW
payload-type	98
parameters	max-red=0 mode-change-capability=2 mode-change-neighbor=1 mode-change-period=2 mode-set="0,1,2"
media-profile	
name	AMR-WB
subname	MSFT
payload-type	121
parameters	max-red=0 mode-change-capability=2 mode-set="0,1,2"
media-profile	
name	AMR-WB
subname	NMS
payload-type	116
parameters	max-red=220 mode-change-capability=2
media-profile	
name	AMR
subname	NMS96
payload-type	96
parameters	max-red=0 mode-change-capability=2 mode-change-neighbor=1 mode-change-period=2 mode-set="0,2,4,7"
media-profile	
name	AMR
subname	NMS97
payload-type	97
parameters	max-red=0 mode-set="7"

9.4.2 Codec Policies

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

While transcoding media codecs is optional, Microsoft does require the SBC generate Comfort Noise and RTCP packets towards Teams if the connection on the other side of the SBC does not support either. Microsoft does not support AMR narrowband (AMR) but does support AMR:WB so AMR narrowband must be stripped from the IMS offer towards Microsoft.

To satisfy this requirement, the SBC uses transcoding resources to generate those packets, which does require a codec policy be configured and assigned.

Here is an example config of a codec policy used for the SBC to generate CN packets towards Teams.

```
codec-policy
  name          TPMCodecPolicy
  allow-codecs  * AMR:no
  add-codecs-on-egress  CN
  order-codecs
  packetization-time  20
```

If you have chosen to configure the [media profiles](#) in the previous section to use SILK or wideband CN, you would set your codec policy to add them on egress. Here is an example:

```
codec-policy
  name          TPMCodecPolicy
  allow-codecs  *
  add-codecs-on-egress  CN::wideband SILK::wideband
  order-codecs
  packetization-time  20
```

Lastly, since some IMS networks may have issues with the codecs being offered by Teams Phone Mobile, you can create another codec policy to remove unwanted or unsupported codecs from the request/responses to your Sip Trunk provider.

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

```
codec-policy
  name          IMSCoreCodecs
  allow-codecs  PCMU G729 telephone-event AMR
  add-codecs-on-egress  PCMU AMR
```

The below reference codec-policy can be used to optimise and allow AMR and AMR-WB codec.

```
codec-policy
  name          TPMIMS
```

```

allow-codecs          AMR-WB::NMS:no AMR-WB::LOW AMR::NMS96
AMR::NMS97 EVS AMR-WB::MSFT:no *
add-codecs-on-egress  AMR-WB::LOW AMR::NMS96 AMR::NMS97 EVS CN
order-codecs          AMR-WB::LOW AMR::NMS96 AMR::NMS97 EVS telephone-
event *

```

We have applied below codec-policy towards Teams which is specific to our Test requirements.

```

codec-policy
name          addCN
allow-codecs  *
add-codecs-on-egress  CN
order-codecs  PCMU G729 *

```

The below reference codec-policy can be used to optimise the AMR-WB usage towards Teams.

```

codec-policy
name          MSTPM
allow-codecs  *
add-codecs-on-egress  AMR-WB::MSFT CN
order-codecs  *
packetization-time  20
force-ptime    disabled
secure-dtmf-cancellation  disabled
dtmf-in-audio  disabled
tone-detect-renegotiate-timer  500
reverse-fax-tone-detection-reinvite  disabled
evrc-tty-baudot-transcode  disabled

```

9.4.3 RTCP Policy

The following RTCP policy needs to be configured for the Oracle SBC to generate RTCP sender reports toward Microsoft Teams.

FYI, for the SBC to generate RTCP sender reports to Teams, the realm in which this policy is assigned must also have a codec policy assigned. This is to evoke the required transcoding resources needed to generate RTCP packets.

To configure system-config from ACLI –

ACLI Path: config t → media-manger → rtcp-policy

```

rtcp-policy
name          rtcpGen
rtcp-generate  all-calls
hide-cname    disabled

```

9.5 Media Configuration

This section will guide you through the configuration of media manager, realms, and steering pools, all of which are required for the SBC to handle signalling and media flows toward Teams and IMS Core.

9.5.1 Media Manager

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

The following two hidden options are recommended for the global media manager when interfacing with Microsoft Teams Phone Mobile.

- **audio-allow-asymmetric-pt**: Provides transcoding support for asymmetric dynamic payload types enables the Oracle® Session Border Controller to perform transcoding when the RTP is offered with one payload type and is answered with another payload type.
- **xcode-gratuitous-rtcp-report-generation**: This option allows the Oracle SBC to generate a Real-Time Transport Control Protocol (RTCP) Receiver Report separately from the default Sender-Receiver Report (RFC 3550). This option requires a reboot to take effect.

To configure system-config from ACLI –

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

media-manager	
state	enabled
options	audio-allow-asymmetric-pt xcode-gratuitous-rtcp-report-generation

9.5.2 Realm Config

Realms are a logical distinction representing routes (or groups of routes) reachable by the Oracle® Session Border Controller and what kinds of resources and special functions apply to those routes.

Realms are used as a basis for determining ingress and egress associations to network interfaces.

Use the following table as a configuration example for the realms. The following parameters are all required unless mentioned as optional below.

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

- Network Interface
- Media Security Policy
- Codec Policy (optional on the PSTN Realm)
- RTCP Policy

Config Parameter	Teams Phone Mobile Realm	IMS Realm	Config Parameter	Teams Phone Mobile Realm	PingCo Realm
Identifier	Teams	ims	Identifier	Teams	PingCo
Network Interface	s0p0:0	s1p0:0	Network Interface	s0p0:0	s1p1:0.4
Mm in realm	enabled	enabled	Mm in realm	enabled	
Media Sec policy	TeamsSecurityPolicy	PSTNNonSecure	Media Sec policy	TeamsSecurityPolicy	
Teams-FQDN	cloudsbc.cgbusolutionslab.com		Teams-FQDN	cloudsbc.cgbusolutionslab.com	
Teams-fqdn-in-uri	enabled		Teams-fqdn-in-uri	enabled	
Sdp-inactive-only	enabled		Sdp-inactive-only	enabled	
RTCP mux	enabled		RTCP mux	enabled	
Codec policy	TPMCodecPolicy	IMSCoreCodecs	Codec policy	TPMCodecPolicy	
RTCP policy	rtcpGen		RTCP policy	rtcpGen	
Access-control-trust-level	HIGH	HIGH	Access-control-trust-level	HIGH	
ringback-trigger		183	ringback-trigger		
ringback-file		US_Ringback_tone.raw	ringback-file		
merge-early-dialogs		enabled	merge-early-dialogs		
hide-egress-media-update		enabled	hide-egress-media-update		

- Ringback trigger,ringback-file, merge-early-dialogs and hide-egress-media-update are required on IMS Ream and are explained in Section [12.2](#) of the document.

To configure realm-config from ACLI –

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

realm-config identifier	Teams
----------------------------	-------

description	Realm Facing Teams Direct Routing
network-interfaces	s0p0:0.4
mm-in-realm	enabled
qos-enable	enabled
media-sec-policy	sdesPolicy
rtcp-mux	enabled
teams-fqdn	cloudsbc.cgbusolutionslab.com
teams-fqdn-in-uri	enabled
sdp-inactive-only	enabled
access-control-trust-level	high
codec-policy	TPMCodecPolicy
rtcp-policy	rtcpGen
realm-config	
identifier	ims
network-interfaces	s1p0:0.4
media-sec-policy	RTP
access-control-trust-level	high
options	merge-early-dialogs enable
codec-policy	IMSCoreCodecs
hide-egress-media-update	enabled
ringback-trigger	183
ringback-file	US_Ringback_tone.raw
merge-early-dialogs	enabled
realm-config	
identifier	PingCo
network-interfaces	s1p1:0.4

9.5.3 Steering Pools

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

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

GUI Path: media-manger/steering-pool

- Click Add, and use the below examples to configure steering-pool.

To configure steering pool from ACLI

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

steering-pool	
ip-address	10.0.2.10
start-port	20000
end-port	40000
realm-id	Teams
steering-pool	

ip-address	10.0.3.10
start-port	20000
end-port	40000
realm-id	ims
steering-pool	
ip-address	10.0.5.27
start-port	20000
end-port	30000
realm-id	PingCo

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

9.6 Sip Configuration

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

9.6.1 Sip-Config

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

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

- Set the home realm ID parameter to Teams Realm, and add the following hidden option:
- **Max-udp-length=0**: Setting this option to zero (0) forces sip to send fragmented UDP packets. Using this option, you override the default value of the maximum UDP datagram size (1500 bytes; sipd requires the use of SIP/TCP at 1300 bytes).
- **inmanip-before-validate** (optional) allows the header rules in a sip-manipulation to apply before the message is parsed.
- **sip-message-len** has been increased in the setup to 65535 to allow large sip packets from the Network.
- **multiple-dialogs-enhancement** applied on the sip-config to enable multiple early dialog support. To Allow the merging of early dialogs within forking scenarios, "merge-early-dialogs" should be enabled on the caller side realm-config.
- dialog-transparency is disabled to support merge-early-dialogs feature as explained in [Section 12.2.3](#) of the document.

To configure sip config from ACLI.

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

```

sip-config
  dialog-transparency      disabled
  home-realm-id            Teams
  options                  inmanip-before-validate
                          max-udp-length=0
                          multiple-dialogs-enhancement
  sip-message-len          65535
  extra-method-stats       enabled
  npli-upon-register       disabled

```

9.6.2 Replaces Header Support

The Oracle® Session Border Controller supports the Replaces header in SIP messages according to RFC 3891. The header, included within SIP INVITE messages, provides a mechanism to replace an existing early or established dialog with a different dialog. The different dialog can be used for Microsoft Teams services such as call parking, attended call transfer and various conferencing features.

The Oracle SBC's support for Replaces header is required to properly interwork with Microsoft Teams, but Microsoft Teams does not support the use of Replaces header. In other words, Microsoft sends Replaces to the SBC, the SBC should not send Replaces to Microsoft.

To configure support for Replaces, we configure the following:

9.6.2.1 Sip Feature

The sip feature configuration element allows the SBC to support the Replaces value in the SIP Require and Supported Headers to and from Microsoft Teams.

To configure sip feature from ACLI

ALCI Path: `config t→session-router→sip-feature`

```

sip-feature
  name          replaces
  realm         Teams
  require-mode-inbound    Pass
  require-mode-outbound   Pass

```

9.6.2.2 Sip Profile

Sip Profile, once configured and assigned to a sip interface, will act on a Replaces header when received by Microsoft teams to replace a dialog.

To configure sip profile from ACLI

ALCI Path: `config t→session-router→sip-profile`

MS will add X-MS-FMC: APP
Towards SBC advising on call origin

- Mobile Terminating calls.

Microsoft adds Headers X-MS-FMC: MT which PingCo TRE uses to route the call back to Oracle SBC to terminate to Native Dialler.

Inbound Call (From PSTN)

- X-TRE-Source-Platform : PSTN
 - Oracle SBC sends this header in the INVITE to TRE for inbound calls coming from Carrier.

PingCo TRE also adds following headers as described in Section

TRE Headers

- X-TRE-CallType: Outbound
- X-TRE-OutboundCarrier: <Outbound Carrier ID>
- X-TRE-OperatorProfile: <Operator Profile ID>
- X-TRE-TrunkType: < Operator Connect Trunk Type Guid>

To configure the sip manipulation via ACLI:

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

sip-manipulation	
name	TPMlogic
header-rule	
name	NativeDiallerlogic
header-name	P-Served-User
action	manipulate
msg-type	request
methods	INVITE
element-rule	
name	matchorigval
parameter-name	sescase
type	header-param
action	store
comparison-type	boolean
match-value	orig
header-rule	
name	addXTRESourcePlatform
header-name	X-TRE-Source-Platform
action	add
comparison-type	boolean
msg-type	request
methods	INVITE
match-value	\$NativeDiallerlogic.\$0
new-value	Mobile
header-rule	

name	removesupported
header-name	Supported
action	delete
msg-type	request
methods	INVITE
header-rule	
name	ModPAI
header-name	P-Asserted-Identity
action	manipulate
msg-type	request
methods	INVITE
element-rule	
name	ModUserPAI
type	uri-host
action	replace
comparison-type	pattern-rule
new-value	\$FROM_HOST.\$0
header-rule	
name	removePVNI
header-name	P-Visited-Network-ID
action	delete
msg-type	request
methods	INVITE
header-rule	
name	RemoveUserAgent
header-name	User-Agent
action	delete
msg-type	request
methods	INVITE
header-rule	
name	StoreHost
header-name	request-uri
action	store
comparison-type	pattern-rule
msg-type	out-of-dialog
methods	INVITE
element-rule	
name	storeurihost
type	uri-host
action	store
header-rule	
name	CopyHost
header-name	To
action	manipulate
methods	INVITE
element-rule	
name	replacehost
type	uri-host
action	replace
comparison-type	boolean

match-value	\$StoreHost.\$storeurihost
new-value	\$StoreHost.\$storeurihost.\$0
header-rule	
name	addPSTNlogic
header-name	From
action	manipulate
msg-type	request
methods	INVITE
element-rule	
name	matchfromhost
type	uri-host
action	store
comparison-type	boolean
match-value	pstn.com
header-rule	
name	AddSourcePlatformPSTN
header-name	X-TRE-Source-Platform
action	add
comparison-type	boolean
msg-type	request
methods	INVITE
match-value	\$addPSTNlogic.\$matchfromhost
new-value	PSTN
header-rule	
name	TeamsClientcalllogic
header-name	X-MS-FMC
action	manipulate
msg-type	request
methods	INVITE
element-rule	
name	matchapp
type	header-value
action	store
comparison-type	boolean
match-value	APP
header-rule	
name	AddSourcePlatformTeamsPhoneMobile
header-name	X-TRE-Source-Platform
action	add
comparison-type	boolean
msg-type	request
methods	INVITE
match-value	\$TeamsClientcalllogic.\$matchapp.\$0
new-value	TeamsPhoneMobile

Note: We have additional header rules in this sip-manipulation which may or may not be required in your implementation. The rules are provided for reference –

- NativeDiallerlogic – based on P-Served-User:orig adds X-TRE-Source-Platform:Mobile towards TRE.
- Removesupported – Removes supported header when sending the Sip Invite towards Microsoft.

- ModPAI – Modifies the P-Asserted-Identity to format it as per Microsoft requirements.
- removePVNI, RemoveUserAgent,– Remove the P-VisitedNetwork ID, User Agent when sending the Invite towards TRE.
- StoreHost – Formats the To header as per the request-URI parameters.
- addPSTNlogic -Checks Inbound calls from Carrier and adds AddSourcePlatform:PSTN towards TRE
- TeamsClientcalllogic – When calls are originated from Teams Client adds AddSourcePlatform: TeamsPhoneMobile towards TRE.

The above sip-manipulation is applied as out-manipulationid on the TRE facing sip-interface.

Sip-manipulation towards IMS

Striprouthead – Sip-manipulation named Striprouthead is applied as in-manipulationid on the IMS's sip-interface.

- The header rules - striprouthead1, striprouthead0 strip the IMS added route headers towards Microsoft.
- ChangeCLine – is a requirement for the test bed and can be ignored.

sip-manipulation	
name	striprouthead
header-rule	
name	striprouthead1
header-name	Route[1]
action	delete
msg-type	request
methods	INVITE
header-rule	
name	striprouthead0
header-name	Route[0]
action	delete
msg-type	request
methods	INVITE
mime-sdp-rule	
name	ChangeCLine
msg-type	request
methods	INVITE
action	manipulate
sdp-session-rule	
name	Cline
action	manipulate
sdp-line-rule	
name	modcline
type	c
action	replace
comparison-type	pattern-rule
match-value	IN IP4 129.158.200.139
new-value	"IN IP4 10.0.3.10"

9.6.3.2 Sip-manipulation to change error 487 to 603

- Header rule - check487 converts 487 Request terminated Error response to 603 Decline. As per Microsoft requirement When a TPM user rejects the call via their default mobile dialler with a SIP error response of 487, the operator network must send towards Teams network a 603, so that the call can be redirected to the User's voicemail as well as stop ringing any registered Teams endpoints.

Note: The sip-manipulation converts 480 to 603 in the test environment as the Native Dialler rejected calls with an error 480 instead of 487.

sip-manipulation	
name	check480
header-rule	
name	check480
header-name	@status-line
action	manipulate
msg-type	reply
methods	INVITE
element-rule	
name	make603
type	status-code
action	replace
match-value	480
new-value	603
element-rule	
name	changeReason
type	reason-phrase
action	replace
comparison-type	boolean
new-value	"Decline"

9.6.3.3 Sip-Manipulation for P-Early-media header.

Sip-manipulation named E164 which is applied as out-manipulationid on the IMS sip-interface serves below purpose –

- Header-rule addPlus formats the Number to E.164 format.
- Header-rule PEMAdd calls a sip-manipulation Add_PEM_to_183 which calls another sip-manipulation Ins_PEM183 is created to add P-early Media header with a value of “send only” on the 183 Message from Microsoft towards IMS.

The requirement for this sip-manipulation is explained in [Section 12.2](#) of the document.

```

sip-manipulation
  name          E164
  header-rule
    name        addPlus
    header-name Request-URI
    action      manipulate
    comparison-type  pattern-rule
    msg-type    request
    methods     INVITE
  element-rule
    name        tendigits
    type        uri-user
    action      replace
    comparison-type  pattern-rule
    match-value ^[0-9]{10}$
    new-value   \+1+$ORIGINAL
  element-rule
    name        elevendigits
    type        uri-user
    action      replace
    comparison-type  pattern-rule
    match-value ^[0-9]{11}$
    new-value   \++$ORIGINAL
  header-rule
    name        PEMAdd
    header-name FROM
    action      sip-manip
    msg-type    reply
    methods     INVITE
    new-value   Add_PEM_to_183
pri-tpm-sbc# sh ru sip-manipulation Add_PEM_to_183 short
sip-manipulation
  name          Add_PEM_to_183
  header-rule
    name        Detect_183
    header-name @status-line
    action      manipulate
    comparison-type  pattern-rule
  element-rule
    name        detect183
    type        status-code
    action      sip-manip
    comparison-type  pattern-rule
    match-value  183
    new-value   Ins_PEM183
pri-tpm-sbc# sh ru sip-manipulation Ins_PEM183 short
sip-manipulation
  name          Ins_PEM183
  header-rule

```

name	Ins_PEM_Field
header-name	P-Early-Media
action	add
new-value	sendonly

Sip-manipulation towards Carrier.

We have created the below sip-manipulation towards Carrier to remove the custom headers by different platforms.

sip-manipulation	
name	RemoveCustomHeaders
description	Removes specified X- headers from INVITE
header-rule	
name	RemoveXMSFMC
header-name	X-MS-FMC
action	delete
msg-type	request
methods	INVITE
header-rule	
name	RemoveXMSTenantId
header-name	X-MS-TenantId
action	delete
msg-type	request
methods	INVITE
header-rule	
name	RemoveXTRESourcePlatform
header-name	X-TRE-Source-Platform
action	delete
msg-type	request
methods	INVITE
header-rule	
name	RemoveXTRECallType
header-name	X-TRE-CallType
action	delete
msg-type	request
methods	INVITE
header-rule	
name	RemoveXTREChainLinkID
header-name	X-TRE-ChainLinkID
action	delete
msg-type	request
methods	INVITE
header-rule	
name	RemoveXTRETrunkType
header-name	X-TRE-TrunkType
action	delete
msg-type	request
methods	INVITE

header-rule	
name	RemoveXTRECompanyID
header-name	X-TRE-CompanyID
action	delete
msg-type	request
methods	INVITE
header-rule	
name	RemoveXTREOperatorProfile
header-name	X-TRE-OperatorProfile
action	delete
msg-type	request
methods	INVITE
header-rule	
name	RemoveXTREOutboundCarrier
header-name	X-TRE-OutboundCarrier
action	delete
msg-type	request
methods	INVITE

9.6.4 Sip Interface

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

Configure three sip interfaces for TRE , IMS Realm, and for Teams Phone Mobile.

Use the table below as an example to configure:

Config Parameter	IMS	Teams	PingCo TRE
Realm ID	ims	Teams	PingCo
Sip-Profile		forreplaces	
out-manipulationid	striprouthead	check480	TPMlogic
in-manipulationid	E164		
Sip Port Config Parameter	IMS	Teams	Teams
Address	10.0.3.10	10.0.2.10	10.0.5.27
Port	5060	5061	5060
Transport protocol	TCP	TLS	UDP
TLS profile		TeamsTLSProfile	
Allow anonymous	agents-only	all	agents-only

Notice this is where we assign the TLS profile configured under the [Security](#) section of this guide, and the sip-profile which allows the SBC to act on the Replaces header when received by Microsoft Teams.

To configure sip interface from ACLI

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

```

pri-tpm-sbc# sh ru sip-interface short
sip-interface
  realm-id          Teams
  sip-port
    address         10.0.2.10
    port            5061
    transport-protocol TLS
    tls-profile      tlsteams
    allow-anonymous agents-only
  spl-options
HeaderNatPublicSipIfIp=129.80.211.181,HeaderNatPrivateSipIfIp=10.0.2.10
  out-manipulationid check480
  sip-profile        forreplaces
sip-interface
  realm-id          ims
  sip-port
    address         10.0.3.10
    allow-anonymous agents-only
  sip-port
    address         10.0.3.10
    transport-protocol TCP
    allow-anonymous agents-only
  spl-options
HeaderNatPublicSipIfIp=129.158.200.139,HeaderNatPrivateSipIfIp=10.0.3.10
  stop-recurse      401,407,480
  in-manipulationid stripouteheader
  out-manipulationid E164
sip-interface
  realm-id          pingco
  sip-port
    address         10.0.5.27
  spl-options
HeaderNatPublicSipIfIp=150.136.176.126,HeaderNatPrivateSipIfIp=10.0.5.27
  out-manipulationid TPMlogic
sip-interface
  realm-id          siptrunk
  sip-port
    address         10.0.4.10
    allow-anonymous agents-only
  spl-options
HeaderNatPublicSipIfIp=129.80.186.157,HeaderNatPrivateSipIfIp=10.0.4.10
  out-manipulationid RemoveCustomHeaders
pri-tpm-sbc#

```

9.6.5 Session Agents

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

Microsoft provides four (4) regional FQDN's for PSTN Hub (NOAM, EMEA, APAC, OCEA), These FQDNs must be configured as Session-Agents in the order of the served market. For e.g. If SBC primarily serves NOAM market(s) you MUST configure their environment to target the NOAM FQDN first.

Following 4 FQDNs must be configured as Session-Agents on Oracle SBC.

NOAM: sip-us.gcs.pstnhub.microsoft.com

EMEA: sip-eu.gcs.pstnhub.microsoft.com

APAC: sip-as.gcs.pstnhub.microsoft.com

OCEA: sip-au.gcs.pstnhub.microsoft.com

Use the table below to configure Session Agents:

Config parameter	Session Agent 1	Session Agent 2	Session Agent 3	Session Agent 3
Hostname	sip-us.gcs.pstnhub.microsoft.com	sip-eu.gcs.pstnhub.microsoft.com	sip-as.gcs.pstnhub.microsoft.com	sip-au.gcs.pstnhub.microsoft.com
Port	5061	5061	5061	5061
Transport method	StaticTLS	StaticTLS	StaticTLS	StaticTLS
Realm ID	Teams	Teams	Teams	Teams
Ping Method	OPTIONS	OPTIONS	OPTIONS	OPTIONS

Ping Interval	60	60	60	60
Refer Call Transfer	enabled	enabled	enabled	enabled
Ping Response	enabled	enabled	enabled	enabled

Note: In the test setup OC-SBC is handling REFERs for call transfers hence the Refer Call Transfer parameter is enabled on the Session-Agents. This will not be required if the REFER messages for call transfers are handled by the IMS Network.

We'll also configure a session agent for the IMS Core.

To configure session agents from ACLI

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

```

session-agent
  hostname          sip-as.gcs.pstnhub.microsoft.com
  port              5061
  transport-method  StaticTLS
  realm-id          Teams
  ping-method       OPTIONS
  ping-interval     60
  ping-response     enabled
  refer-call-transfer enabled
session-agent
  hostname          sip-au.gcs.pstnhub.microsoft.com
  port              5061
  transport-method  StaticTLS
  realm-id          Teams
  ping-method       OPTIONS
  ping-interval     60
  ping-response     enabled
  refer-call-transfer enabled
session-agent
  hostname          sip-eu.gcs.pstnhub.microsoft.com
  port              5061
  transport-method  StaticTLS
  realm-id          Teams
  ping-method       OPTIONS
  ping-interval     60
  ping-response     enabled

```

refer-call-transfer	enabled
session-agent	
hostname	sip-us.gcs.pstnhub.microsoft.com
port	5061
transport-method	StaticTLS
realm-id	Teams
ping-method	OPTIONS
ping-interval	60
ping-response	enabled
refer-call-transfer	enabled

We have defined Session Agents SCSCF and PCSCF for IMS Core as per the requirement of the Test Environment. You may have to define additional IMS components based on your network setup and requirements.

session-agent	
hostname	129.213.187.4
transport-method	StaticTCP
realm-id	ims
ping-method	OPTIONS
ping-interval	30
ping-response	enabled
refer-call-transfer	enabled
session-agent	
hostname	volte.oraclecgbupoc.co.uk
port	5063
realm-id	ims

Session agent for TRE-

We have configured following session-agent for PingCo TRE.

session-agent	
hostname	13.93.229.25
realm-id	PingCo
ping-method	OPTIONS
ping-interval	10
ping-response	enabled

9.6.6 Session Group

A session agent group allows the SBC to create a load balancing model:

All four Teams session agents configured above will be added to the group. The session agents listed under destination must be in this order, and the strategy must be set to HUNT.

- Use the following as an example to configure:

To configure session group from ACLI

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

```
session-group
  group-name      TeamsPhoneMobile
  dest            sip-us.gcs.pstnhub.microsoft.com
                 sip-eu.gcs.pstnhub.microsoft.com
                 sip-as.gcs.pstnhub.microsoft.com
                 sip-au.gcs.pstnhub.microsoft.com
```

9.7 Routing Configuration

Now that a majority of the signalling, security and media configuration is in place, we can configure the SBC to route calls from one end of the network to the other. As per the reference architecture both MO and MT leg traverse through PingCo TRE for necessary modifications to the signalling.

The SBC has multiple routing features that can be utilized, but for the purposes of this example configuration, we'll configure local policies to route calls from :-

- MO Teams calls Microsoft Teams to PingCo TRE
- MO native dialler calls to PingCo TRE
- MT calls from Microsoft to PingCo TRE for termination to native dialler.
- Route calls from TRE to PSTN.

To configure local policy from ACLI:

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

```
# Route calls from Carrier trunk to TRE
local-policy
  from-address      *
  to-address        *
  source-realm      siptrunk
  policy-attribute
  next-hop          13.93.229.25
  realm             PingCo

# Routes call from TRE towards PSTN
local-policy
  from-address      *
  to-address        16174261400
```

```

17815321400
18004444444
6174261400
7815321400
80044444444
+16174261400
+17815321400
+18004444444
source-realm          PingCo
policy-attribute
  next-hop            138.3.226.40
  realm              siptrunk
  action             replace-uri

#Routes call to TPM Destination numbers from TRE originating from IMS network
local-policy
  from-address        volte.oraclecgbupoc.co.uk
  to-address          17812032798
                    17812032799
                    7812032798
                    7812032799
                    +17812032798
                    +17812032799
source-realm          PingCo
policy-attribute
  next-hop            sag:ocsag
  realm              Teams
  action             replace-uri

#Routes call from Teams to PingCo TRE for TPM users for the Native Dialler termination.
local-policy
  from-address        *
  to-address          17812032798
                    17812032799
                    7812032798
                    7812032799
                    +17812032798
                    +17812032799
source-realm          Teams
policy-attribute
  next-hop            13.93.229.25
  realm              PingCo
  action             replace-uri

```

As we are handling Transfers on OC-SBC an additional Local Policy is created for call transfers towards Microsoft TPM users which routes back the REFERS to a TPM user towards Microsoft.

```

local-policy
  from-address        *

```

to-address	sip.gcs.pstnhub.microsoft.com
source-realm	Teams
policy-attribute	
next-hop	sag:ocsag
realm	Teams
action	replace-uri

9.8 SIP Access Controls

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

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

<https://docs.oracle.com/en/industries/communications/session-border-controller/9.2.0/security/security-guide.pdf>

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

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

Microsoft Teams has two subnets, 52.112.0.0/14 and 52.120.0.0/14 that must be allowed to send traffic to the SBC. Both must be configured as an access control on the Oracle SBC and associated with the realm facing Teams.

We also created static ACLs for TRE and IMS.

To configure access control from ACLI

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

Use this example to create ACL’s for both Microsoft Teams subnets, 52.112.0.0/14, and 52.120.0.0/14.

```

access-control
  realm-id          ims
  source-address    129.213.136.120
  application-protocol SIP
  trust-level       high
access-control
  realm-id          ims
  source-address    129.213.187.4
  application-protocol SIP
  trust-level       high
access-control

```

realm-id	PingCo
source-address	13.93.229.25
application-protocol	SIP
trust-level	high
access-control	
realm-id	siptrunk
source-address	138.3.226.40
application-protocol	SIP
trust-level	high
access-control	
realm-id	ims
source-address	158.101.98.101
application-protocol	SIP
trust-level	high
access-control	
realm-id	Teams
source-address	52.112.0.0/14
application-protocol	SIP
trust-level	high
access-control	
realm-id	Teams
source-address	52.120.0.0/14
application-protocol	SIP
trust-level	high

This concludes the required configuration of the SBC to properly interface with Microsoft Teams Phone Mobile.

You'll need to [save and activate](#) your configuration!

10 Verify Connectivity

10.1 Oracle SBC Options Pings

While in the Oracle SBC CLI, Utilize the “show sipd options” command to check for OPTIONS to and from the SBC.

```

NN4900-102# show sipd options
OPTIONS (22:17:05-116)
----- Server -----
Message/Event  Recent   Total  PerMax  Recent   Total  PerMax
----- Client -----
OPTIONS Requests      10    80976    10      7    59979     9
Retransmissions       0         0     0       0         0     0
200 OK                10    80928    10      7    59979     9
403 Forbidden         0         4     0       0         0     0
Transaction Timeouts -         -     -       0         0     0
Locally Throttled    -         -     -       0         0     0

Avg Latency=0.001 for 7
Max Latency=0.002
NN4900-102#

```

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

11 Syntax Requirements for SIP Invite and SIP Options:

This section covers high-level requirements to SIP syntax of Invite and Options messages. The information can be used as a first step during troubleshooting when calls don't go through. From our experience most of the issues are related to the wrong syntax of SIP messages.

Microsoft includes two customer headers **X-MS-TenantId** and **X-MS-FMC**: that contains the specific customer's O365 Tenant ID and type of call which can be MO,MT or App (Call originated from Teams Client)

Note: The information is masked in the below example for security purpose.

11.1 Terminology

- Recommended – not required, but to simplify the troubleshooting, it is recommended to configure as in examples as follow.
- Must – strict requirement, the system does not work without the configuration of these parameters.

11.2 Requirements for INVITE Messages and Final Responses.

Contact Header-Invite and Final Response

- Must have the FQDN sub-domain of the Oracle SBC.
- **Syntax: Contact:** <phone number>@< subdomain FQDN >:<SBC Port>;<transport type>

Picture 1 Shows the INVITE from TRE for a MO call originated from Native Dialler.

```
INVITE sip:+17812032798@13.93.229.25:5060;user=phone SIP/2.0
Via: SIP/2.0/UDP 13.93.229.25:5060;branch=z9hG4bK756b.d6784b8.0
Max-Forwards: 66
Contact: <sip:X.did.026.31e77592@13.93.229.25:5060>
To: <sip:+17812032798@13.93.229.25:5061>
From:
"+17812032799"<sip:+17812032799@volte.oraclecgbupoc.co.uk;transport=TCP>;
tag=SDk5guf01-0ad94042
Call-ID:
DLGCH_HDQOWzQ8NmN+XVAKNXBoMH0RAQ81K2dqe0MDDTB6YmUuEwMNMS9pNXxCSA9jeWRjekB
VW2M-
CSeq: 1 INVITE
Allow: INVITE, ACK, CANCEL, BYE, NOTIFY, REFER, MESSAGE, OPTIONS, INFO,
SUBSCRIBE
Content-Type: application/sdp
Allow-Events: presence, kpml, talk, as-feature-event
Content-Length: 323
P-Asserted-Identity: <sip:+17812032799@volte.oraclecgbupoc.co.uk>
P-Asserted-Identity: <tel:+17812032799>
```

```

P-Early-Media: supported
P-Served-User:
<sip:+17812032799@volte.oraclecgbupoc.co.uk>;sescase=orig;regstate=reg
X-MS-SBC: Oracle/VM/8.4.5p2
X-TRE-Source-Platform: Mobile
X-MS-FMC: MO
X-TRE-CallType: Inbound
X-TRE-InboundCarrier: 3076b66f-1690-4b4b-e04a-08da9a92b44e_tcap
X-TRE-TrunkType: 9bcaa351-df28-4457-32b4-08d9e15f3200
X-TRE-OperatorProfile: 712b6d41-adb9-4079-3424-08daa7756d02_tcap

```

Picture 2 Shows the INVITE from TRE to IMS for the MT Leg to Native Dialler.

```

INVITE sip:+17812032798@13.93.229.25:5060;user=phone SIP/2.0
Via: SIP/2.0/UDP
192.168.100.10:5060;received=13.93.229.25;branch=z9hG4bK6c88.cdf7aa44.0
FROM:
"+17812032799"<sip:+17812032799@sip.gcs.pstnhub.microsoft.com:5061;user=p
hone>;tag=SDnm3bd01-3a3a4bb83b26456d996b8e30c066c64d
TO: <sip:+17812032798@13.93.229.25:5061;user=phone>
CSEQ: 1 INVITE
CALL-ID:
DLGCH_HDQLA2ArNGN+XQFZZH81Y3ZJV1s1e2BqK0VWW2N4aGF8Q1wNY3Bia3ZBSA9jeWRjekB
VXWM-
MAX-FORWARDS: 68
CONTACT: <sip:X.did.f88.aff80e55@192.168.100.10>
CONTENT-LENGTH: 347
CONTENT-TYPE: application/sdp
ALLOW: INVITE,ACK,OPTIONS,CANCEL,BYE,NOTIFY
PRIVACY: id
X-MS-FMC: MT
X-MS-TenantId: d8d754c2-108b-4816-9f4b-79fb12d5ca28
X-TRE-Source-Platform: TeamsPhoneMobile
X-TRE-OutboundCarrier: 3076b66f-1690-4b4b-e04a-08da9a92b44e
X-TRE-CallType: Outbound

```

11.3 Requirements for SIP Options.

Below are the Microsoft requirements for SIP Options Message.

- The SBC MUST support the SIP OPTIONS method and respond to an incoming SIP OPTIONS request based on RFC 3261.
- The SBC MUST NOT respond with SIP/2.0 405 Method Not Supported or 215 SIP/2.0 501 Not Implemented.

- The OPTIONS pings from SBC MUST NOT exceed a frequency of one transaction every 60 seconds for each configured trunk and MUST NOT be more less frequent than one 229 transaction every 180 seconds for each configured trunk.
- Microsoft will not initiate OPTIONS pings to SBC until it receives OPTIONS pings from the SBC.
- The CONTACT header MUST contain the FQDN of the trunk and MUST specify both the port and protocol (e.g., 5061 and TLS)
- **Syntax: Contact: <phone number>@< subdomain FQDN >:<SBC Port>;<transport type>**
- Microsoft will not include the ACCEPT header and will ignore any body text in the response.

Picture 3 - Example of SIP OPTIONS message from Oracle SBC to Microsoft.

```

OPTIONS sip:sip-us.gcs.pstnhub.microsoft.com:5061;transport=tls SIP/2.0
Via: SIP/2.0/TLS 20.65.42.129:5061;branch=z9hG4bKdik418206025aqb9v510
Call-ID: c75cbb319998591b44c2c7e20e8f717b0000g30@10.1.4.4
To: sip:ping@sip-us.gcs.pstnhub.microsoft.com
From:
sip:ping@cloudsbc.cgbusolutionslab.com;tag=bba52bd57d6bd688fde828d05f2a71830000g30
Max-Forwards: 70
CSeq: 7 OPTIONS
Contact: sip:ping@cloudsbc.cgbusolutionslab.com:5061;transport=tls;sip.ice
Expires: 60
Route: sip:52.115.54.0:5061;transport=tls;lr
X-MS-SBC: Oracle/VM/9.2.0p2
Content-Length: 0

```

Picture 4 - Example of SIP OPTIONS message from Microsoft to Oracle SBC.

```

OPTIONS sip:ping@cloudsbc.cgbusolutionslab.com:5061;transport=tls SIP/2.0
FROM: <sip:sip-us.gcs.pstnhub.microsoft.com:5061>;tag=89a53e30-276b-4596-a761-0ac7c919a859
TO: <sip:ping@cloudsbc.cgbusolutionslab.com>
CSEQ: 1 OPTIONS
CALL-ID: 92542534-cad5-4501-a418-b9f6304bf45b
MAX-FORWARDS: 70
VIA: SIP/2.0/TLS 52.115.54.0:5061;branch=z9hG4bK728aa3f0
CONTACT: <sip:sip-us.gcs.pstnhub.microsoft.com:5061>
CONTENT-LENGTH: 0
USER-AGENT: Microsoft.PSTNHub.SIPProxy v.2022.2.14.2 i.USEA.3
ALLOW: INVITE,ACK,OPTIONS,CANCEL,BYE,NOTIFY

```

12 Appendix A

12.1 Oracle SBC deployed behind NAT

The Support for SBC Behind NAT SPL plug-in changes information in SIP messages to hide the end point located inside the private network.

The specific information that the Support for SBC Behind NAT SPL plug-in changes depends on the direction of the call, for example, from the NAT device to the SBC or from the SBC to the NAT device.

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

- The private IP address must be the same IP as configured on both the SIP Interface and Steering Pool
- The public IP address must be the public IP address of the NAT device

Here is an example configuration with SBC Behind NAT SPL config.

The SPL is applied to the Teams side SIP interface.

```
HeaderNatPublicSipIfIp= 129.80.211.18,HeaderNatPrivateSipIfIp=10.0.2.10
```

HeaderNatPublicSipIfIp is the public interface ip

HeaderNatPrivateSipIfIp is the private ip.

To configure header NAT SPL from ACLI

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

Choose the sip interface on which the header NAT SPL needs to be applied. Under spl-options add the entry as per example shared below.

```
spl-options          HeaderNatPublicSipIfIp= 129.80.211.18,HeaderNatPrivateSipIfIp=10.0.2.10
```

- Perform a [save and activate](#) configuration for changes to take effect.

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

12.2 Early Media handling, Local Media Playback and Merge Dialogs -

For certain call flows with early Media Microsoft expects OC-SBC to merge early dialogs sent by Teams and generates a PEM header towards the IMS Core and play Ringback Tone Locally. Microsoft also requires OC-SBC to merge multiple 183 Session Progress Messages from Teams backend and make it a single fork.

We have achieved this by configuring some additional parameters onto the SBC and through sip-manipulations.

12.2.1 Early Media handling

For the requirement of OC-SBC generates a PEM header towards the IMS Core and play Ring back Tone Locally. We have created the sip-manipulation explained in [section 9.6.3](#)

The HMR works as below on the 183 Session progress Message from Microsoft.

```
Inbound 183 Session Progress from Microsoft towards Oracle SBC.
```

SIP/2.0 183 Session Progress

FROM:

<sip:+918130313388@cloudsbc.cgbusolutionslab.com;user=phone>;tag=SD4dthf02-861130111-1706516226676-

TO: "ORACLESOLLAB ."<sip:+17812032798@sip-us.gcs.pstnhub.microsoft.com:5061;user=phone>;tag=1c370fd74fa848f180e1cdb5e1b03172

CSEQ: 707105083 INVITE

CALL-ID: SD4dthf02-66926c0021824938f6f2de24181dbaf4-a004050

VIA: SIP/2.0/TLS 10.0.2.10:5061;branch=z9hG4bKijge4u00c81tudqgtau0.1

RECORD-ROUTE: <sip:sip-us.gcs.pstnhub.microsoft.com:5061;transport=tls;lr>

CONTACT: <sip:api-du-b-usea.pstnhub.microsoft.com:443;x-i=38d12233-f46d-4215-adb9-c5d52b97b0f5;x-

c=b6721950dcf157ae9168ba217afeb25a/s/1/bbef7d3473084ab6b4d4687af54d063b>

CONTENT-LENGTH: 465

CONTENT-TYPE: application/sdp

ALLOW: INVITE,ACK,OPTIONS,CANCEL,BYE,NOTIFY

SERVER: Microsoft.PSTNHub.SIPProxy v.2024.1.22.1 i.USEA.1

X-MS-TenantId: XXXXXXXXXXXX

Outbound 183 Session progress from SBC towards IMS

SIP/2.0 183 Session Progress

Via: SIP/2.0/TCP

10.0.17.20:5060;received=129.213.187.4;branch=z9hG4bK9pi2kq20e8fli111qan0.1

Via: SIP/2.0/UDP 10.0.17.22:5060;branch=z9hG4bKjva37b20agt0hpru2910.1

Via: SIP/2.0/UDP

129.158.200.139:5060;branch=z9hG4bKh3ad2100cok0vf8848r0.1

From: <sip:+918130313388@63.77.76.250;user=phone>;tag=SD4dthf01-861130111-1706516226676-

To: "ORACLESOLLAB

."<sip:+17812032798@141.146.36.101:5061;user=phone>;tag=SD4dthf99-1c370fd74fa848f180e1cdb5e1b03172

Call-ID: SD4dthf01-66926c0021824938f6f2de24181dbaf4-a004050

CSeq: 707105083 INVITE

Record-Route: <sip:SDgdc09+fpebnfmuvif67u3p5p8frlubf-

gl5p7bvvoeuctgh5p7b10ocud5@129.213.187.4:5060;lr;transport=udp>

Contact: <sip:129.158.200.139:5060;x-i=38d12233-f46d-4215-adb9-c5d52b97b0f5;x-

c=b6721950dcf157ae9168ba217afeb25a/s/1/bbef7d3473084ab6b4d4687af54d063b;t ransport=tcp>

CONTENT-LENGTH: 345

CONTENT-TYPE: application/sdp

ALLOW: INVITE,ACK,OPTIONS,CANCEL,BYE,NOTIFY

SERVER: Microsoft.PSTNHub.SIPProxy v.2024.1.22.1 i.USEA.1

X-MS-TenantId: XXXXXXXXXXXX

P-Early-Media: sendonly

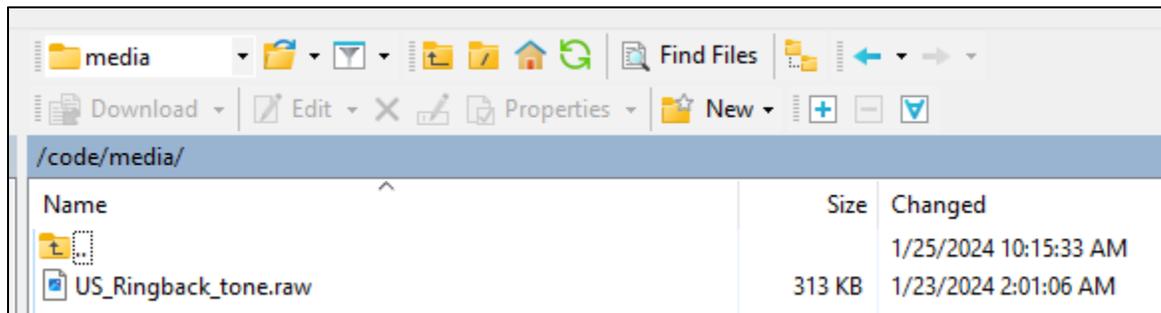
12.2.2 Oracle SBC Local Media Playback

Oracle SBC has the capability of playing local media on certain triggers. For this case we are playing the Local Media on the 183 Session progress from the Microsoft towards the Caller.

12.2.2.1 Media Files

Media files of ring back tones are uploaded to /code/media to the Oracle SBC. This file differs based on your media generation method and must be raw media binary. For Transcoding based RBT, ensure that the files RAW PCM 16-bit MONO samples, sampled at 8-khz encapsulated with little-endian formatting and cannot exceed 4.8 MB.

Next, load the file to the /code/media directory on the Oracle SBC. SFTP to the SBC management IP address to securely transfer the file into this directory. In this example, we're using a common SFTP client, WinSCP.



Lastly, we'll assign this file to the realm facing the IMS Core and set the trigger for the SBC to generate local ring back.

12.2.2.2 Local Media Playback Config

To assign the ring back file on the realm through ACLI, navigate to below path and provide the name of the ring back file at the ringback-file config object.

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

realm-config	
identifier	ims
network-interfaces	s1p0:0.4
media-sec-policy	RTP
access-control-trust-level	high
ringback-trigger	183
ringback-file	US_Ringback_tone.raw

- Perform a [save and activate](#) configuration for changes to take effect.

12.2.3 Merge Early Dialogs

Microsoft requires OC-SBC to merge multiple 183 Session Progress Messages from Teams backend and make it a single fork. To handle this requirement, we are creating below configuration on the SBC.

```

sip-config
  dialog-transparency      disabled
  home-realm-id            Teams
  registrar-domain        *
  registrar-host           *
  registrar-port           5060
  options                  inmanip-before-validate
                          max-udp-length=0
                          multiple-dialogs-enhancement

realm-config
  identifier               ims
  network-interfaces       s1p0:0.4
  media-sec-policy         RTP
  access-control-trust-level high
  options                  merge-early-dialogs enable
  hide-egress-media-update enabled
  ringback-trigger         183
  ringback-file            US_Ringback_tone.raw
  merge-early-dialogs     enabled
  
```

Please follow [Oracle SBC documentation](#) Section Merge Function within Early Dialog Support on Page 647 for detailed understanding on Merge-early-dialogs feature.

Note Merge-early-dialogs does not work with –

- Offerless call
- Preconditions interworking
- SRVCC • multiple audio or video m-line
- p-early-media-header with 'add' and 'modify' options.

- You should configure HMU to maintain RTP consistency. •
- Dialog transparency should be disabled.

13 Appendix B

13.1 Test Cases

This version of Application Note is created as per below conducted tests-

S no.	Case	Status
1	Intra Tenant Call	PASSED

3	TPM Origination Call to Non-Teams user	PASSED
4	MT Call to TPM User	PASSED
5	SBC Sip and Media Interworking towards Teams User	PASSED
6	SBC Interworking Offer in a-line	PASSED
7	Cold Transfer to TPM Enabled User	PASSED
8	Call Transfer to External PSTN	PASSED
10	Consultative Transfer to TPM Enabled User	PASSED
11	Call Uplift via Teams App for the Teams Native Dialler in Call	PASSED
13	User not Assigned in TPM	PASSED
14	Call Forking and Local Reject (Orig PSTN or Volte User)	PASSED
15	Local User Rejects incoming call needs to route to TPM VM (global reject SIP Error)	PASSED
16	Call Forking and Cancel	PASSED

14 ACLI Running Configuration

Below is a complete output of the running configuration used to create this application note. This output includes all of the configuration elements used in our examples and may also include other configuration that may be relevant to the purpose of this document. Be aware that not all parameters may be applicable to every Oracle SBC setup, so please take this into consideration if planning to copy and paste this output into your SBC.

```

pri-tpm-sbc#
pri-tpm-sbc# sh ru short
access-control
  realm-id          ims
  source-address    129.213.136.120
  application-protocol SIP
  trust-level       high
access-control
  realm-id          ims
  source-address    129.213.187.4
  application-protocol SIP
  trust-level       high
access-control
  realm-id          pingco
  source-address    13.93.229.25
  application-protocol SIP
  trust-level       high
access-control
  realm-id          siptrunk
  source-address    138.3.226.40
  application-protocol SIP
  trust-level       high
access-control
  realm-id          ims

```

source-address	158.101.98.101
application-protocol	SIP
trust-level	high
access-control	
realm-id	Teams
source-address	52.112.0.0/14
application-protocol	SIP
trust-level	high
access-control	
realm-id	Teams
source-address	52.120.0.0/14
application-protocol	SIP
trust-level	high
access-control	
realm-id	Teams
source-address	52.121.0.0/14
application-protocol	SIP
trust-level	high
certificate-record	
name	DigiCertGlobalRootCA
common-name	DigiCert Global Root CA
certificate-record	
name	DigiCertGlobalRootG2
common-name	DigiCert Global Root G2
certificate-record	
name	DigiCertGlobalRootG3
common-name	DigiCert Global Root G3
certificate-record	
name	DigiCertTLSRSA4096RootG5
common-name	DigiCert TLS RSA4096 Root G5
certificate-record	
name	MicrosoftECCRootCertificateAuthority2017
common-name	Microsoft ECC Root Certificate Authority 2017
certificate-record	
name	MicrosoftRSARootCertificateAuthority2017
common-name	Microsoft RSA Root Certificate Authority 2017
certificate-record	
name	SBCCertificateforTPM
common-name	cloudsbc.cgbusolutionslab.com
certificate-record	
name	gd_bundle-g2-g1
common-name	gd_bundle-g2-g1
codec-policy	
name	addCN
allow-codecs	*
add-codecs-on-egress	CN
order-codecs	PCMU G729 *
codec-policy	
name	TPMCodecPolicy
allow-codecs	*

```

add-codecs-on-egress          CN::wideband SILK::wideband
order-codecs
packetization-time            20
codec-policy
name                          IMSCoreCodecs
allow-codecs                  PCMU G729 telephone-event AMR
add-codecs-on-egress         PCMU AMR
codec-policy
name                          TPMIMS
allow-codecs                  AMR-WB::NMS:no AMR-WB::LOW AMR::NMS96 AMR::NMS97
EVS AMR-WB::MSFT:no *
add-codecs-on-egress         AMR-WB::LOW AMR::NMS96 AMR::NMS97 EVS CN
order-codecs                  AMR-WB::LOW AMR::NMS96 AMR::NMS97 EVS telephone-event *
codec-policy
name                          MSTPM
allow-codecs                  *
add-codecs-on-egress         AMR-WB::MSFT CN
order-codecs                  *
http-server
name                          web
ice-profile
name                          ice
stun-conn-timeout            0
stun-keep-alive-interval     0
local-policy
from-address                  *
to-address                    16174261400
                              17815321400
                              18004444444
                              6174261400
                              7815321400
                              80044444444
                              +16174261400
                              +17815321400
                              +18004444444
source-realm                  pingco
policy-attribute
next-hop                      138.3.226.40
realm                         siptrunk
action                         replace-uri
local-policy
from-address                  volte.oraclecgbupoc.co.uk
to-address                    17812032798
                              17812032799
                              7812032798
                              7812032799
                              +17812032798
                              +17812032799
source-realm                  pingco
policy-attribute

```

next-hop	sag:ocsag
realm	Teams
action	replace-uri
local-policy	
from-address	*
to-address	17812032798
	17812032799
	7812032798
	7812032799
	+17812032798
	+17812032799
source-realm	Teams
policy-attribute	
next-hop	13.93.229.25
realm	pingco
action	replace-uri
local-policy	
from-address	*
to-address	*
source-realm	Teams
policy-attribute	
next-hop	13.93.229.25
realm	pingco
local-policy	
from-address	*
to-address	sip.gcs.pstnhub.microsoft.com
source-realm	Teams
policy-attribute	
next-hop	sag:ocsag
realm	Teams
action	replace-uri
local-policy	
from-address	*
to-address	*
source-realm	ims
policy-attribute	
next-hop	13.93.229.25
realm	pingco
action	replace-uri
local-policy	
from-address	*
to-address	pstn.com
source-realm	ims
policy-attribute	
next-hop	13.93.229.25
realm	pingco
action	replace-uri
local-policy	
from-address	sip.gcs.pstnhub.microsoft.com
to-address	*

source-realm	pingco
policy-attribute	
next-hop	129.213.187.4
realm	ims
action	replace-uri
local-policy	
from-address	*
to-address	*
source-realm	pingco
policy-attribute	
next-hop	sag:ocsag
realm	Teams
action	replace-uri
local-policy	
from-address	*
to-address	*
source-realm	siptrunk
policy-attribute	
next-hop	13.93.229.25
realm	pingco
media-manager	
options	dtls-trace webrtc-trace
media-profile	
name	CN
subname	wideband
payload-type	118
clock-rate	16000
media-profile	
name	SILK
subname	narrowband
payload-type	103
clock-rate	8000
media-profile	
name	SILK
subname	wideband
payload-type	104
clock-rate	16000
media-profile	
name	AMR-WB
subname	LOW
payload-type	98
parameters	max-red=0 mode-change-capability=2 mode-change-neighbor=1 mode-change-period=2 mode-set="0,1,2"
media-profile	
name	AMR-WB
subname	MSFT

payload-type	121
parameters	max-red=0 mode-change-capability=2 mode-set="0,1,2"
media-profile	
name	AMR-WB
subname	NMS
payload-type	116
parameters	max-red=220 mode-change-capability=2
media-profile	
name	AMR
subname	NMS96
payload-type	96
parameters	max-red=0 mode-change-capability=2 mode-change-neighbor=1 mode-change-period=2 mode-set="0,2,4,7"
media-profile	
name	AMR
subname	NMS97
payload-type	97
parameters	max-red=0 mode-set="7"
media-sec-policy	
name	RTP
media-sec-policy	
name	sdesPolicy
inbound	
profile	SDES
mode	srtp
protocol	sdes
outbound	
profile	SDES
mode	srtp
protocol	sdes
network-interface	
name	s0p0
hostname	cloudsbc.cgbusolutionslab.com
ip-address	10.0.2.10
netmask	255.255.255.0
gateway	10.0.2.1
dns-ip-primary	8.8.8.8
dns-domain	cloudsbc.cgbusolutionslab.com
network-interface	
name	s0p1
ip-address	10.0.4.10
netmask	255.255.255.0
gateway	10.0.4.1

```

network-interface
  name          s1p0
  ip-address    10.0.3.10
  netmask       255.255.255.0
  gateway       10.0.3.1
network-interface
  name          s1p1
  ip-address    10.0.5.27
  netmask       255.255.255.0
  gateway       10.0.5.1
phy-interface
  name          s0p0
  operation-type Media
phy-interface
  name          s0p1
  operation-type Media
  port          1
phy-interface
  name          s1p0
  operation-type Media
  slot          1
phy-interface
  name          s1p1
  operation-type Media
  port          1
  slot          1
realm-config
  identifier     Teams
  description    Realm Facing Teams Direct Routing
  network-interfaces s0p0:0.4
  mm-in-realm    enabled
  qos-enable     enabled
  media-sec-policy sdesPolicy
  rtcp-mux       enabled
  teams-fqdn     cloudsbc.cgbusolutionslab.com
  teams-fqdn-in-uri enabled
  sdp-inactive-only enabled
  access-control-trust-level high
  codec-policy   addCN
  rtcp-policy    rtcpGen
realm-config
  identifier     ims
  network-interfaces s1p0:0.4
  media-sec-policy RTP
  access-control-trust-level high
  ringback-trigger 183
  ringback-file    US_Ringback_tone.raw
realm-config
  identifier     pingco
  network-interfaces s1p1:0.4

```

media-sec-policy	RTP
realm-config	
identifier	siptrunk
network-interfaces	s0p1:0.4
media-sec-policy	RTP
options	merge-early-dialogs enable
hide-egress-media-update	enabled
merge-early-dialogs	enabled
rtcp-policy	
name	rtcpGen
rtcp-generate	all-calls
sdes-profile	
name	SDES
lifetime	31
session-agent	
hostname	129.213.187.4
transport-method	StaticTCP
realm-id	ims
ping-method	OPTIONS
ping-interval	30
ping-response	enabled
refer-call-transfer	enabled
session-agent	
hostname	13.93.229.25
realm-id	pingco
ping-method	OPTIONS
ping-interval	10
ping-response	enabled
session-agent	
hostname	138.3.226.40
ip-address	138.3.226.40
realm-id	siptrunk
ping-method	OPTIONS
ping-interval	30
ping-response	enabled
session-agent	
hostname	158.101.98.101
ip-address	158.101.98.101
realm-id	ims
refer-call-transfer	enabled
session-agent	
hostname	sip-as.gcs.pstnhub.microsoft.com
port	5061
transport-method	StaticTLS
realm-id	Teams
ping-method	OPTIONS
ping-interval	30
ping-response	enabled
refer-call-transfer	enabled
session-agent	

hostname	sip-au.gcs.pstnhub.microsoft.com
port	5061
transport-method	StaticTLS
realm-id	Teams
ping-method	OPTIONS
ping-interval	30
ping-response	enabled
refer-call-transfer	enabled
session-agent	
hostname	sip-eu.gcs.pstnhub.microsoft.com
port	5061
transport-method	StaticTLS
realm-id	Teams
ping-method	OPTIONS
ping-interval	30
ping-response	enabled
refer-call-transfer	enabled
session-agent	
hostname	sip-us.gcs.pstnhub.microsoft.com
port	5061
transport-method	StaticTLS
realm-id	Teams
ping-method	OPTIONS
ping-interval	30
ping-response	enabled
refer-call-transfer	enabled
session-agent	
hostname	sip.gcs.pstnhub.microsoft.com
port	5061
transport-method	StaticTLS
realm-id	Teams
ping-method	OPTIONS
ping-interval	30
ping-response	enabled
refer-call-transfer	enabled
session-agent	
hostname	volte.oraclecgbupoc.co.uk
port	5063
realm-id	ims
session-group	
group-name	ocsag
dest	sip-us.gcs.pstnhub.microsoft.com sip-eu.gcs.pstnhub.microsoft.com sip-au.gcs.pstnhub.microsoft.com sip-as.gcs.pstnhub.microsoft.com
session-timer-profile	
name	ToTeams
force-reinvite	enabled
request-refresher	uas
session-translation	

```

id toPSTN
sip-config
  dialog-transparency disabled
  home-realm-id Teams
  registrar-domain *
  registrar-host *
  registrar-port 5060
  options inmanip-before-validate
           max-udp-length=0
           multiple-dialogs-enhancement
  sip-message-len 65535
  extra-method-stats enabled
  allow-pani-for-trusted-only disabled
  add-ue-location-in-pani disabled
  npli-upon-register disabled
sip-feature
  name replaces
  realm Teams
  require-mode-inbound Pass
  require-mode-outbound Pass
sip-interface
  realm-id Teams
  sip-port
    address 10.0.2.10
    port 5061
    transport-protocol TLS
    tls-profile tlsteams
    allow-anonymous agents-only
  spl-options
  HeaderNatPublicSipIfIp=129.80.211.181,HeaderNatPrivateSipIfIp=10.0.2.10
  out-manipulationid check480
  sip-profile forreplaces
sip-interface
  realm-id ims
  sip-port
    address 10.0.3.10
    allow-anonymous agents-only
  sip-port
    address 10.0.3.10
    transport-protocol TCP
    allow-anonymous agents-only
  spl-options
  HeaderNatPublicSipIfIp=129.158.200.139,HeaderNatPrivateSipIfIp=10.0.3.10
  stop-recurse 401,407,480
  in-manipulationid striprouthead
  out-manipulationid E164
sip-interface
  realm-id pingco
  sip-port
    address 10.0.5.27

```

```

spl-options
HeaderNatPublicSipIfIp=150.136.176.126,HeaderNatPrivateSipIfIp=10.0.5.27
  out-manipulationid          TPMlogic
sip-interface
  realm-id                    siptrunk
  sip-port
    address                    10.0.4.10
    allow-anonymous            agents-only
  spl-options
HeaderNatPublicSipIfIp=129.80.186.157,HeaderNatPrivateSipIfIp=10.0.4.10
  out-manipulationid          RemoveCustomHeaders

sip-manipulation
  name                        AddPACmePlayback
  header-rule
    name                      CheckForSDPInactive
    header-name                Content-Type
    action                     store
    comparison-type            pattern-rule
    methods                    INVITE
    element-rule
      name                    Inactive
      parameter-name          application/SDP
      type                    mime
      action                  store
      match-value              a=inactive
  header-rule
    name                      StartMoH
    header-name                P-Acme-Playback
    action                     add
    comparison-type            boolean
    methods                    INVITE
    match-value                $CheckForSDPInactive.$Inactive
    new-value                  "start;duration=continuous;direction=both;stop-on-final-resp=false"

sip-manipulation
  name                        AddRefertoAllow
  header-rule
    name                      AllowRefer
    header-name                Allow
    action                     manipulate
    methods                    Invite
    new-value                  $ORIGINAL+",REFER"

sip-manipulation
  name                        Add_PEM_to_183
  header-rule
    name                      Detect_183
    header-name                @status-line
    action                     manipulate
    comparison-type            pattern-rule
    element-rule

```

```

        name                detect183
        type                 status-code
        action               sip-manip
        comparison-type      pattern-rule
        match-value          183
        new-value            Ins_PEM183
sip-manipulation
  name                      DeleteSDP
  header-rule
    name                    Store180
    header-name              @status-line
    action                   store
    msg-type                 reply
    methods                  Invite
    element-rule
      name                   store180
      type                   status-code
      action                  store
      match-value            180
  mime-sdp-rule
    name                     sdpStrip
    msg-type                 reply
    methods                  Invite
    action                   delete
    comparison-type          boolean
    match-value              $Store180.$store180
sip-manipulation
  name                      E164
  header-rule
    name                     addPlus
    header-name              Request-URI
    action                    manipulate
    comparison-type          pattern-rule
    msg-type                 request
    methods                  INVITE
    element-rule
      name                   tendigits
      type                   uri-user
      action                  replace
      comparison-type          pattern-rule
      match-value              ^[0-9]{10}$
      new-value                \+1+$ORIGINAL
    element-rule
      name                   elevendigits
      type                   uri-user
      action                  replace
      comparison-type          pattern-rule
      match-value              ^[0-9]{11}$
      new-value                \++$ORIGINAL
  header-rule

```

name	PEMAdd
header-name	FROM
action	sip-manip
msg-type	reply
methods	INVITE
new-value	Add_PEM_to_183
sip-manipulation	
name	Ins_PEM183
header-rule	
name	Ins_PEM_Field
header-name	P-Early-Media
action	add
new-value	sendonly
sip-manipulation	
name	RemoveCustomHeaders
description	Removes specified X- headers from INVITE
header-rule	
name	RemoveXMSFMC
header-name	X-MS-FMC
action	delete
msg-type	request
methods	INVITE
header-rule	
name	RemoveXMSTenantId
header-name	X-MS-TenantId
action	delete
msg-type	request
methods	INVITE
header-rule	
name	RemoveXTRESourcePlatform
header-name	X-TRE-Source-Platform
action	delete
msg-type	request
methods	INVITE
header-rule	
name	RemoveXTRECallType
header-name	X-TRE-CallType
action	delete
msg-type	request
methods	INVITE
header-rule	
name	RemoveXTREChainLinkID
header-name	X-TRE-ChainLinkID
action	delete
msg-type	request
methods	INVITE
header-rule	
name	RemoveXTRETrunkType
header-name	X-TRE-TrunkType
action	delete

msg-type	request
methods	INVITE
header-rule	
name	RemoveXTRECompanyID
header-name	X-TRE-CompanyID
action	delete
msg-type	request
methods	INVITE
header-rule	
name	RemoveXTREOperatorProfile
header-name	X-TRE-OperatorProfile
action	delete
msg-type	request
methods	INVITE
header-rule	
name	RemoveXTREOutboundCarrier
header-name	X-TRE-OutboundCarrier
action	delete
msg-type	request
methods	INVITE
sip-manipulation	
name	TPMlogic
header-rule	
name	NativeDialerlogic
header-name	P-Served-User
action	manipulate
msg-type	request
methods	INVITE
element-rule	
name	matchorigval
parameter-name	sescase
type	header-param
action	store
comparison-type	boolean
match-value	orig
header-rule	
name	addXTRESorcePlatform
header-name	X-TRE-Source-Platform
action	add
comparison-type	boolean
msg-type	request
methods	INVITE
match-value	\$NativeDialerlogic.\$0
new-value	Mobile
header-rule	
name	removesupported
header-name	Supported
action	delete
msg-type	request
methods	INVITE

```

header-rule
  name          ModPAI
  header-name   P-Asserted-Identity
  action        manipulate
  msg-type      request
  methods       INVITE
  element-rule
    name        ModUserPAI
    type        uri-host
    action      replace
    comparison-type  pattern-rule
    new-value   $FROM_HOST.$0
header-rule
  name          removePVNI
  header-name   P-Visited-Network-ID
  action        delete
  msg-type      request
  methods       INVITE
header-rule
  name          RemoveUserAgent
  header-name   User-Agent
  action        delete
  msg-type      request
  methods       INVITE
header-rule
  name          StoreHost
  header-name   request-uri
  action        store
  comparison-type  pattern-rule
  msg-type      out-of-dialog
  methods       INVITE
  element-rule
    name        storeurihost
    type        uri-host
    action      store
header-rule
  name          CopyHost
  header-name   To
  action        manipulate
  methods       INVITE
  element-rule
    name        replacehost
    type        uri-host
    action      replace
    comparison-type  boolean
    match-value  $StoreHost.$storeurihost
    new-value   $StoreHost.$storeurihost.$0
header-rule
  name          addPSTNlogic
  header-name   From

```

action	manipulate
msg-type	request
methods	INVITE
element-rule	
name	matchfromhost
type	uri-host
action	store
comparison-type	boolean
match-value	pstn.com
header-rule	
name	AddSourcePlatformPSTN
header-name	X-TRE-Source-Platform
action	add
comparison-type	boolean
msg-type	request
methods	INVITE
match-value	\$addPSTNlogic.\$matchfromhost
new-value	PSTN
header-rule	
name	TeamsClientcalllogic
header-name	X-MS-FMC
action	manipulate
msg-type	request
methods	INVITE
element-rule	
name	matchapp
type	header-value
action	store
comparison-type	boolean
match-value	APP
header-rule	
name	AddSourcePlatformTeamsPhoneMobile
header-name	X-TRE-Source-Platform
action	add
comparison-type	boolean
msg-type	request
methods	INVITE
match-value	\$TeamsClientcalllogic.\$matchapp.\$0
new-value	TeamsPhoneMobile
sip-manipulation	
name	add100reltosupported
header-rule	
name	add100rel
header-name	Supported
action	manipulate
comparison-type	pattern-rule
msg-type	request
methods	INVITE
new-value	100rel
sip-manipulation	

```

name                check480
header-rule
  name              check480
  header-name       @status-line
  action            manipulate
  msg-type          reply
  methods           INVITE
  element-rule
    name            make603
    type            status-code
    action          replace
    match-value     480
    new-value       603
  element-rule
    name            changeReason
    type            reason-phrase
    action          replace
    comparison-type boolean
    new-value       "Decline"
sip-manipulation
  name              striprouteheader
  header-rule
    name            striproute1
    header-name     Route[1]
    action          delete
    msg-type        request
    methods         INVITE
  header-rule
    name            striproute0
    header-name     Route[0]
    action          delete
    msg-type        request
    methods         INVITE
  mime-sdp-rule
    name            ChangeCLine
    msg-type        request
    methods         INVITE
    action          manipulate
  sdp-session-rule
    name            Cline
    action          manipulate
  sdp-line-rule
    name            modcline
    type            c
    action          replace
    comparison-type pattern-rule
    match-value     IN IP4 129.158.200.139
    new-value       "IN IP4 10.0.3.10"
sip-monitoring
  match-any-filter  enabled

```

monitoring-filters	*
sip-profile	
name	forreplaces
replace-dialogs	enabled
ssh-key	
name	admin
type	authorized-key
size	2048
steering-pool	
ip-address	10.0.2.10
start-port	20000
end-port	40000
realm-id	Teams
steering-pool	
ip-address	10.0.3.10
start-port	20000
end-port	40000
realm-id	ims
steering-pool	
ip-address	10.0.4.10
start-port	20000
end-port	40000
realm-id	siptrunk
steering-pool	
ip-address	10.0.5.27
start-port	20000
end-port	30000
realm-id	pingco
system-config	
hostname	cloudsbc.cgbusolutionslab.com
dos-cores	1
transcoding-cores	1
tls-global	
session-caching	enabled
diffie-hellman-key-size	DH_KeySize_2048
tls-profile	
name	TLSTeams
end-entity-certificate	SBCCertificateforTeams
trusted-ca-certificates	DigiCertRoot DigiCertGlobalRootG2 DigiCertGlobalRootG3 DigiCertTLSECCP384RootG5 DigiCertTLSECCP4096RootG5 MicrosoftECCRootCertificateAuthority2017 MicrosoftRSARootCertificateAuthority2017
mutual-authenticate	enabled
tls-version	tlsv12
pri-tpm-sbc#	



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Oracle Corporation, World Headquarters
2300 Cloud Way
Austin, TX 78741, USA

Worldwide Inquiries
Phone: +1.650.506.7000
Fax: +1.650.506.7200

Integrated Cloud Applications & Platform Services

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