

Configuration Example for Secure SIP Integration Between CUCM and CUC based on Next Generation Encryption (NGE)

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Introduction

This document describes the configuration and verification of the secure SIP connection between the Cisco Unified Communication Manager (CUCM) and Cisco Unity Connection (CUC) server using Next Generation Encryption.

Next Generation Security over SIP interface restricts SIP interface to use Suite B ciphers based on TLS 1.2, SHA-2 and AES256 protocols. It allows the various combinations of ciphers based on the priority order of RSA or ECDSA ciphers. During the communication between Unity Connection and Cisco Unified CM, both ciphers and third party certificates are verified at both the ends. Below is the configuration for Next Generation Encryption support.

If you plan to use the certificates signed by third party Certification Authority then start with Certificate signing at the end of the configuration section (Configure - Signing the EC key based certificates by third party CA)

Prerequisites

Requirements

The information in this document is based on these software and hardware versions:

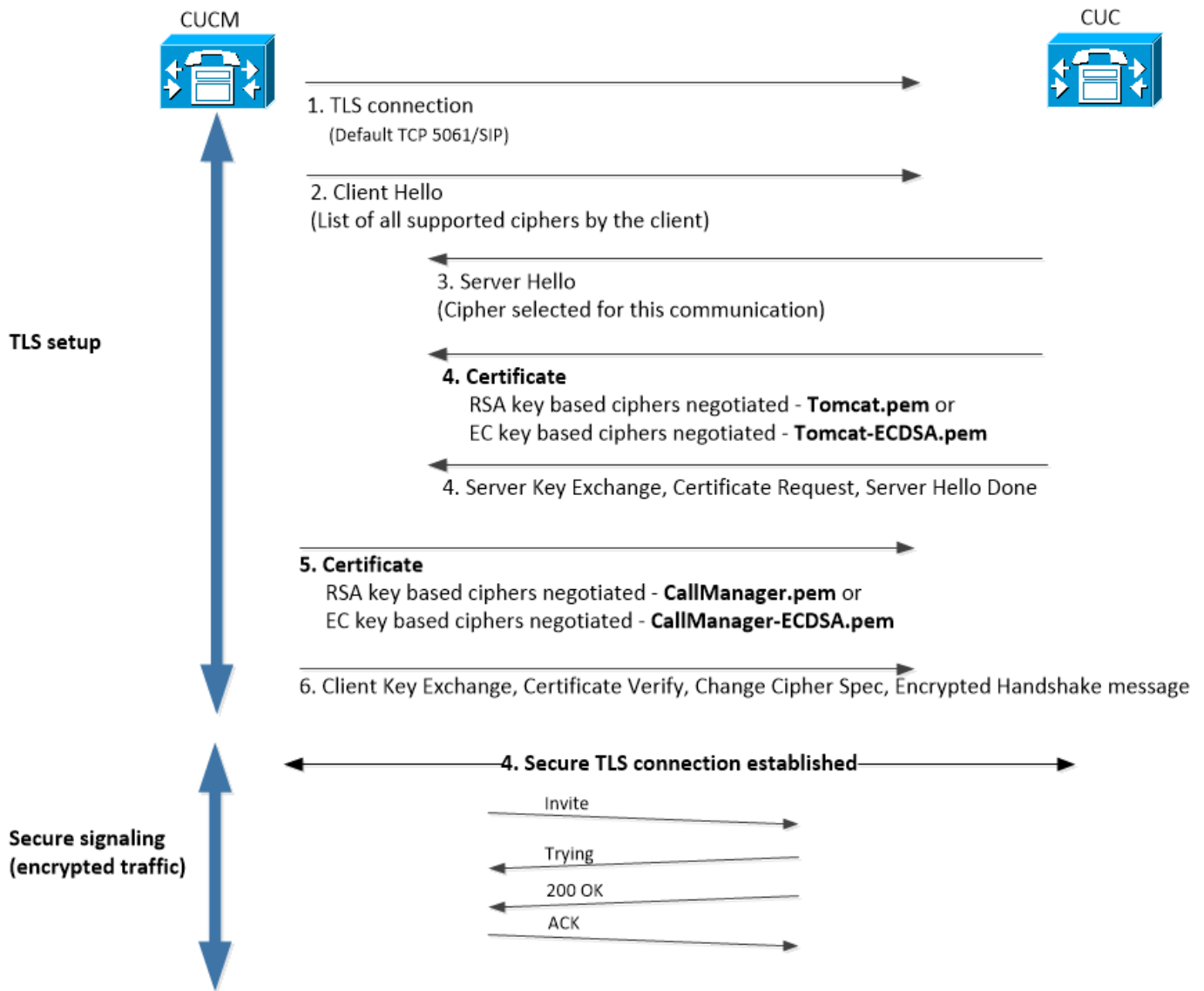
CUCM version 11.0 and later in Mixed mode

CUC version 11.0 and later

The information in this document was created from the devices in a specific lab environment. All of the devices used in this document started with a cleared (default) configuration. If your network is live, make sure that you understand the potential impact of any command.

Network Diagram

This diagram briefly explains the process that helps establish a secure connection between CUCM and CUC once the Next generation encryption support is enabled:



Certificate requirements

These are the certificate exchange requirements once the Next generation encryption support is enabled on Cisco Unity Connection.

• RSA key based ciphers negotiated

CUCM certificate used	CUC certificate used	Certs to upload to CUCM	Certs to upload to CUC
CallManager.pem (self-signed)	Tomcat.pem (self-signed)	Tomcat.pem to be uploaded into CUCM > CallManger-trust	None.
CallManager.pem (CA signed)	Tomcat.pem (CA signed)	CUC root & intermediate CA certificate ^{*1} to be uploaded into CUCM > CallManager-trust	CUCM root & intermediate certificate ^{*2} to be uploaded CUC > CallManager-trust.
CallManager.pem (CA signed)	Tomcat.pem (self-signed)	Tomcat.pem to be uploaded into CUCM > CallManger-trust	CUCM root & intermediate certificate to be uploaded in CUC > CallManager-trust.
CallManager.pem (self-signed)	Tomcat.pem (CA signed)	CUC root & intermediate CA certificate to be uploaded into CUCM > CallManager-trust	None.

*1CUC root & intermediate CA certificate refers to CA certificate which signed the Unity connection Tomcat certificate (Tomcat.pem).

*2CUCM root & intermediate CA certificate refers to CA certificate which signed the CUCM CallManager certificate (Callmanager.pem).

• EC Key Based ciphers negotiated

CUCM certificate used	CUC certificate used	Certs to upload to CUCM	Certs to upload to CUC
CallManager-ECDSA.pem (self-signed)	Tomcat-ECDSA.pem (self-signed)	Tomcat-ECDSA.pem to be uploaded into CUCM > CallManger-trust	None.
CallManager-ECDSA.pem (CA signed)	Tomcat-ECDSA.pem (CA signed)	CUC root & intermediate CA certificate* ¹ to be uploaded into CUCM > CallManager-trust	CUCM root & intermediate CA certificate* ² to be uploaded into CUC > CallManager-trust.
CallManager-ECDSA.pem (CA signed)	Tomcat-ECDSA.pem (self-signed)	Tomcat-ECDSA.pem to be uploaded into CUCM > CallManger-trust.	CUCM root & intermediate CA certificate to be uploaded into CUC > CallManager-trust.
CallManager-ECDSA.pem (self-signed)	Tomcat-ECDSA.pem (CA signed)	CUC root & intermediate CA certificate to be uploaded into CUCM > CallManager-trust	None.

*1 CUC root & intermediate CA certificate refers to CA certificate which signed the Unity connection EC based Tomcat certificate (Tomcat-ECDSA.pem).

*2 CUCM root & intermediate CA certificate refers to CA certificate which signed the CUCM CallManager certificate (CallManager-ECDSA.pem).

1. **Note:** Tomcat-ECDSA.pem certificate is called CallManager-ECDSA.pem in 11.0.1 versions of CUC. From CUC 11.5.x the certificate has been renamed to Tomcat-ECDSA.pem.

Configure - Cisco Unity Connection (CUC)

1. Add a New Port Group

Navigate to Cisco Unity Connection Administration page > Telephony integration > Port group and Click on Add New. Make sure to check the Enable Next Generation Encryption checkbox.

New Port Group

Phone System

Create From Port Group Type Port Group

Port Group Description

Display Name*

Authenticate with SIP Server

Authentication Username

Authentication Password

Contact Line Name

SIP Security Profile

Enable Next Generation Encryption

Secure RTP

Primary Server Settings

IPv4 Address or Host Name

IPv6 Address or Host Name

Port

- Note:**Unity Connection's Cisco Tomcat certificate will be used during SSL handshake once the Enable Next Generation Encryption checkbox is enabled.
 - In case ECDSA based cipher is negotiated then EC key based tomcat-ECDSA certificate is used in SSL handshake.
 - In case RSA based cipher is negotiated then RSA key based tomcat certificate is used in SSL handshake.

2. Add the TFTP server reference

On the Port Group Basics page, navigate to Edit > Servers and add FQDN of TFTP server of your CUCM cluster. FQDN/Hostname of the TFTP server must match the Common name (CN) of CallManager certificate. IP address of the server will not work and it will result in failure to download the ITL file. The DNS name must be therefore resolvable via configured DNS server.

SIP Servers

Delete Selected Add

Order	IPv4 Address or Host Name
0	10.48.47.109

Delete Selected Add

TFTP Servers

Delete Selected Add

Order	IPv4 Address or Host Name
0	CUCMv11

Delete Selected Add

Restart Connection Conversation Manager on each node by navigating to Cisco Unity Connection Serviceability > Tools > Service Management. This is mandatory for the configuration to take effect.

- Note:** Unity connection downloads ITL file (ITLfile.tlv) from CUCM's TFTP using https protocol on secure 6972 port (URL: https://<CUCM-TFTP-FQDN>:6972/ITLFile.tlv). CUCM must be in the mixed mode since CUC is looking for the "CCM+TFTP" function certificate from the ITL file.

Navigate back to Telephony integration > Port group > Port Group Basics configuration page and reset your newly added Port group.

Port Group

Display Name* PhoneSystem-1

Integration Method SIP

Reset Status Reset Required Reset

Session Initiation Protocol (SIP) Settings

Register with SIP Server

Authenticate with SIP Server

- Note:** Every time the port group is reset, the CUC server will update its locally stored ITL file by connecting to CUCM server.

3. Add Voice Mail Ports

Navigate back to Telephony integration > Port and click on Add new to add port to your newly created port group.

New Phone System Port

Enabled

Number of Ports

Phone System

Port Group

Server

Port Behavior

Answer Calls

Perform Message Notification

Send MWI Requests (may also be disabled by the port group)

Allow TRAP Connections

4. Upload CUCM Root and intermediate certificate of the third party CA

In case of third party certificates, you must upload the Root and Intermediate certificate of the third party Certification Authority on CallManager-trust of Unity Connection. This is needed only if 3rd party CA signed your Call Manager certificate. Perform this action by navigating to Cisco Unified OS Administration > Security > Certificate Management and click on Upload Certificate.

Upload Certificate/Certificate chain

Certificate Purpose*

Description(friendly name)

Upload File CA_root_-_4096_key.crt

Configure - Cisco Unified CM (CUCM)

1. Create a SIP trunk security profile

Navigate to CUCM Administration > System > Security > SIP Trunk Security Profile and add a new profile. X.509 Subject Name must match the FQDN of the CUC server.

SIP Trunk Security Profile Information

Name*

Description

Device Security Mode

Incoming Transport Type*

Outgoing Transport Type

Enable Digest Authentication

Nonce Validity Time (mins)*

X.509 Subject Name

Incoming Port*

Enable Application level authorization

Accept presence subscription

Accept out-of-dialog refer**

Accept unsolicited notification

Accept replaces header

Transmit security status

Allow charging header

- Note:** CLI command "show cert own tomcat/tomcat.pem" can display the RSA key based tomcat certificate on Unity Connection. It's CN must match the X.509 Subject name configured on CUCM. The CN is equal to FQDN/Hostname of the Unity server. The EC key based certificate contains the FQDN/hostname in its Subject Alternate Name (SAN) field.

2. Create a secure SIP Trunk

Navigate to Device > Trunk > Click and Add new and create a standard SIP trunk which will be used for secure integration with Unity Connection.

SRTP Allowed - When this flag is checked, Encrypted TLS needs to be configured in the network to provide end to end security. Failure to do so will expose keys and other information.

Consider Traffic on This Trunk Secure*

Route Class Signaling Enabled*

Use Trusted Relay Point*

PSTN Access

Run On All Active Unified CM Nodes

Inbound Calls

Significant Digits*	All
Connected Line ID Presentation*	Default
Connected Name Presentation*	Default
Calling Search Space	< None >
AAR Calling Search Space	< None >
Prefix DN	
<input checked="" type="checkbox"/> Redirecting Diversion Header Delivery - Inbound	

Outbound Calls

Called Party Transformation CSS	< None >
<input checked="" type="checkbox"/> Use Device Pool Called Party Transformation CSS	
Calling Party Transformation CSS	< None >
<input checked="" type="checkbox"/> Use Device Pool Calling Party Transformation CSS	
Calling Party Selection*	Originator
Calling Line ID Presentation*	Default
Calling Name Presentation*	Default
Calling and Connected Party Info Format*	Deliver DN only in connected party
<input checked="" type="checkbox"/> Redirecting Diversion Header Delivery - Outbound	
Redirecting Party Transformation CSS	< None >
<input checked="" type="checkbox"/> Use Device Pool Redirecting Party Transformation CSS	

Destination

<input type="checkbox"/> Destination Address is an SRV			
	Destination Address	Destination Address IPv6	Destination Port
1*	10.48.47.123		5061
MTP Preferred Originating Codec*	711ulaw		
BLF Presence Group*	Standard Presence group		
SIP Trunk Security Profile*	cuc-secure-profile-EDCS		
Rerouting Calling Search Space	< None >		
Out-Of-Dialog Refer Calling Search Space	< None >		
SUBSCRIBE Calling Search Space	< None >		
SIP Profile*	Standard SIP Profile	View Details	
DTMF Signaling Method*	No Preference		

3. Configure TLS and SRTP ciphers

- Note:** The negotiation between Unity Connection and Cisco Unified Communications Manager depends on the TLS cipher configuration with the following conditions: When Unity Connection acts as server, TLS cipher negotiation is based on the preference selected by Cisco Unified CM. In case ECDSA based cipher is negotiated then EC key based tomcat-ECDSA certificates are used in SSL handshake. In case RSA based cipher is negotiated then RSA key based tomcat certificates are used in SSL handshake. When Unity Connection acts as client, TLS cipher negotiation is based on the preference selected by Unity Connection.

Navigate to Cisco Unified CM > Systems > Enterprise Parameters and select the appropriate

cipher option from the TLS and SRTP Ciphers from drop-down list.

Security Parameters	
Cluster Security Mode *	1
LBM Security Mode *	Insecure
CAPF Phone Port *	3804
CAPF Operation Expires in (days) *	10
TFTP File Signature Algorithm *	SHA-1
Enable Caching *	True
Authentication Method for API Browser Access *	Basic
TLS Ciphers *	All Ciphers RSA Preferred
SRTP Ciphers *	All Supported Ciphers
HTTPS Ciphers *	RSA Ciphers Only

Restart the Cisco Call Manager service on each node by navigating to Cisco Unified Serviceability page, Tools > Control Centre-Feature Services and select Cisco Call Manager under CM Services

Navigate to Cisco Unity Connection Administration page > System Settings > General Configurations and select the appropriate cipher option from the TLS and SRTP Ciphers from drop-down list.

Edit General Configuration

Time Zone: (GMT+01:00) Europe/Warsaw

System Default Language: English(United States)

System Default TTS Language: English(United States)

Recording Format: G.711 mu-law

Maximum Greeting Length: 90

Target Decibel Level for Recordings and Messages: -26

Default Partition: cucv11 Partition

Default Search Scope: cucv11 Search Space

When a recipient cannot be found: Send a non-delivery receipt

IP Addressing Mode: IPv4

TLS Ciphers: All Ciphers RSA Preferred

SRTP Ciphers: All supported AES-256, AES-128 ciphers

HTTPS Ciphers: RSA Ciphers Only

Restart Connection Conversation Manager on each node by navigating to Cisco Unity Connection Serviceability > Tools > Service Management.

TLS Cipher options with Priority order

TLS Cipher Options

Strongest- AES-256 SHA-384 Only: RSA Preferred

Strongest-AES-256 SHA-384 Only: ECDSA Preferred

Medium-AES-256 AES-128 Only: RSA Preferred

TLS Ciphers in Priority Order

- TLS_ECDHE_RSA_WITH_AES_256_GCM_SHA384
- TLS_ECDHE_ECDSA_WITH_AES_256_GCM_SHA384
- TLS_ECDHE_ECDSA_WITH_AES_256_GCM_384
- TLS_ECDHE_RSA_WITH_AES_256_GCM_SHA384
- TLS_ECDHE_RSA_WITH_AES_256_GCM_SHA384
- TLS_ECDHE_ECDSA_WITH_AES_256_GCM_SHA384

- TLS_ECDHE_RSA_WITH_AES_128_GCM_SHA256
- TLS_ECDHE_ECDSA_WITH_AES_128_GCM_SHA256
- TLS_ECDHE_ECDSA_WITH_AES_256_GCM_SHA384
- TLS_ECDHE_RSA_WITH_AES_256_GCM_SHA384
- TLS_ECDHE_ECDSA_WITH_AES_128_GCM_SHA256
- TLS_ECDHE_RSA_WITH_AES_128_GCM_SHA256
- TLS_ECDHE_RSA_WITH_AES_256_GCM_SHA384
- TLS_ECDHE_ECDSA_WITH_AES_256_GCM_SHA384
- TLS_ECDHE_ECDSA_WITH_AES_128_GCM_SHA256
- TLS_RSA_WITH_AES_128_CBC_SHA
- TLS_ECDHE_ECDSA_WITH_AES_256_GCM_SHA384
- TLS_ECDHE_RSA_WITH_AES_256_GCM_SHA384
- TLS_ECDHE_ECDSA_WITH_AES_128_GCM_SHA256
- TLS_ECDHE_RSA_WITH_AES_128_GCM_SHA256
- TLS_RSA_WITH_AES_128_CBC_SHA

Medium-AES-256 AES-128 Only: ECDSA Preferred

All Ciphers RSA Preferred (Default)

All Ciphers ECDSA Preferred

SRTP Cipher Options in Priority order

SRTP Cipher Option

SRTP in Priority Order

All supported AES-256, AES-128 ciphers

- AEAD_AES_256_GCM
- AEAD_AES_128_GCM
- AES_CM_128_HMAC_SHA1_32

AEAD AES-256, AES-128 GCM-based ciphers

- AEAD_AES_256_GCM
- AEAD_AES_128_GCM

AEAD AES256 GCM-based ciphers only

- AEAD_AES_256_GCM

4. Upload CUC Tomcat certificates (RSA & EC based)

Navigate to OS Administration > Security > Certificate Management and upload both CUC Tomcat certificates (RSA & EC based) into the CallManager-trust store.

Upload Certificate/Certificate chain

Certificate Purpose* CallManager-trust

Description(friendly name)

Upload File Choose File tomcat-ECDSA.pem

Upload Certificate/Certificate chain

Certificate Purpose* CallManager-trust

Description(friendly name)

Upload File Choose File tomcat.pem

Upload Close

1. **Note:** Uploading both Unity Tomcat certificates is not mandatory if ECDSA ciphers are negotiated only. In such case EC based Tomcat certificate is enough.

In case of third party certificates, you must upload the root and Intermediate certificate of the third party Certification Authority. This is needed only if 3rd party CA signed your Unity Tomcat certificate.

Upload Certificate/Certificate chain

Certificate Purpose* CallManager-trust

Description(friendly name)

Upload File Choose File CA_root_-_4096_key.crt

Upload Close

Restart the Cisco Call Manager process on all nodes to apply the changes.

5. Create Route pattern

Configure a route pattern that points to the configured trunk by navigating to Call Routing > Route/Hunt > Route Pattern. Extension entered as a route pattern number can be used as a voicemail pilot.

Pattern Definition

Route Pattern* 2000

Route Partition < None >

Description

Numbering Plan -- Not Selected --

Route Filter < None >

MLPP Precedence* Default

Apply Call Blocking Percentage

Resource Priority Namespace Network Domain < None >

Route Class* Default

Gateway/Route List* CUCv11

Route Option

Route this pattern

Block this pattern No Error

6. Create Voicemail Pilot, Voicemail Profile and assign it to the DNs

Create a voice mail pilot for the integration by going to Advanced Features > Voice Mail > Voice

Mail Pilot.

Voice Mail Pilot Information

Voice Mail Pilot Number	2000
Calling Search Space	< None >
Description	Default

Create a voice mail profile in order to link all the settings together Advanced Features > Voice Mail > Voice Mail Profile

Voice Mail Profile Information

Voice Mail Profile	VoiceMailProfile-8000 (used by 0 devices)
Voice Mail Profile Name*	VoiceMailProfile-8000
Description	
Voice Mail Pilot**	2000/< None >
Voice Mail Box Mask	

Assign the newly created voice mail profile to the DNs intended to use the secure integration by going to Call Routing > Directory number

Directory Number Settings

Voice Mail Profile	VoiceMailProfile-8000	(Choose <None> to use system default)
Calling Search Space	< None >	
BLF Presence Group*	Standard Presence group	
User Hold MOH Audio Source	< None >	
Network Hold MOH Audio Source	< None >	

Configure - Signing the EC key based certificates by third party CA (optional)

The certificates might be signed by a third party CA before setting up the secure integration between the systems. Follow the following steps to sign the certificates on both systems.

Cisco Unity Connection

1. Generate Certificate signing request (CSR) for CUC Tomcat-ECDSA and have the certificate signed by third party CA
2. CA provides Identity certificate (CA signed certificate) and CA certificate (CA root certificate) which must be uploaded as follow:
Upload CA root certificate into the tomcat-trust store
Upload Identity certificate into the tomcat-EDCS store
3. Restart Conversation Manager on CUC

Cisco Unified CM

1. Generate CSR for CUCM CallManager-ECDSA and have the certificate signed by third party CA

2. CA provides Identity certificate (CA signed certificate) and CA certificate (CA root certificate) which must be uploaded as follow:
 Upload CA root certificate into the callmanager-trust store
 Upload Identity certificate into the callmanager-EDCS store
3. Restart Cisco CCM and TFTP services on each node

The same process will be used to sign RSA key based certificates where CSR is generated for CUC Tomcat certificate and CallManager certificate and uploaded into the tomcat store and callmanager store respectively.

Verify

Use this section to confirm that your configuration works properly.

Secure SIP Trunk verification

Press the Voice Mail button on the phone to call voice mail. You should hear the opening greeting if the user's extension is not configured on the Unity Connection system.

Alternatively, you can enable SIP OPTIONS keepalive to monitor the SIP trunk status. This option can be enabled in the SIP profile assigned to the SIP trunk. Once this is enabled you can monitor the Sip trunk status via Device > Trunk as shown below:

The screenshot shows a search filter set to 'begins with CUCv11'. Below the search bar is a table with columns: Name, Description, Calling Search Space, Device Pool, Route Pattern, Trunk Type, SIP Trunk Status, and SIP Trunk Duration. The row for 'CUCv11' shows 'SIP Trunk Status' as 'Full Service' and 'SIP Trunk Duration' as 'Time In Full Service: 0 day 0 hour 0 minute'. A red box highlights the 'SIP Trunk Status' and 'SIP Trunk Duration' columns.

Name	Description	Calling Search Space	Device Pool	Route Pattern	Trunk Type	SIP Trunk Status	SIP Trunk Duration
CUCv11			Default	2000	SIP Trunk	Full Service	Time In Full Service: 0 day 0 hour 0 minute

Secure RTP Call Verification

Verify whether the padlock icon is present on calls to Unity Connection. It means RTP stream is encrypted (Device Security profile must be secure in order for it to work) as shown in this image



Related Information

- [SIP Integration Guide for Cisco Unity Connection Release 11.x](#)