



Cisco Unity 4.0(4) with Cisco Unified CallManager 4.1(2) Configured as Message Center PINX using Cisco WS-X6608-T1 using Q.SIG as MGCP Gateway

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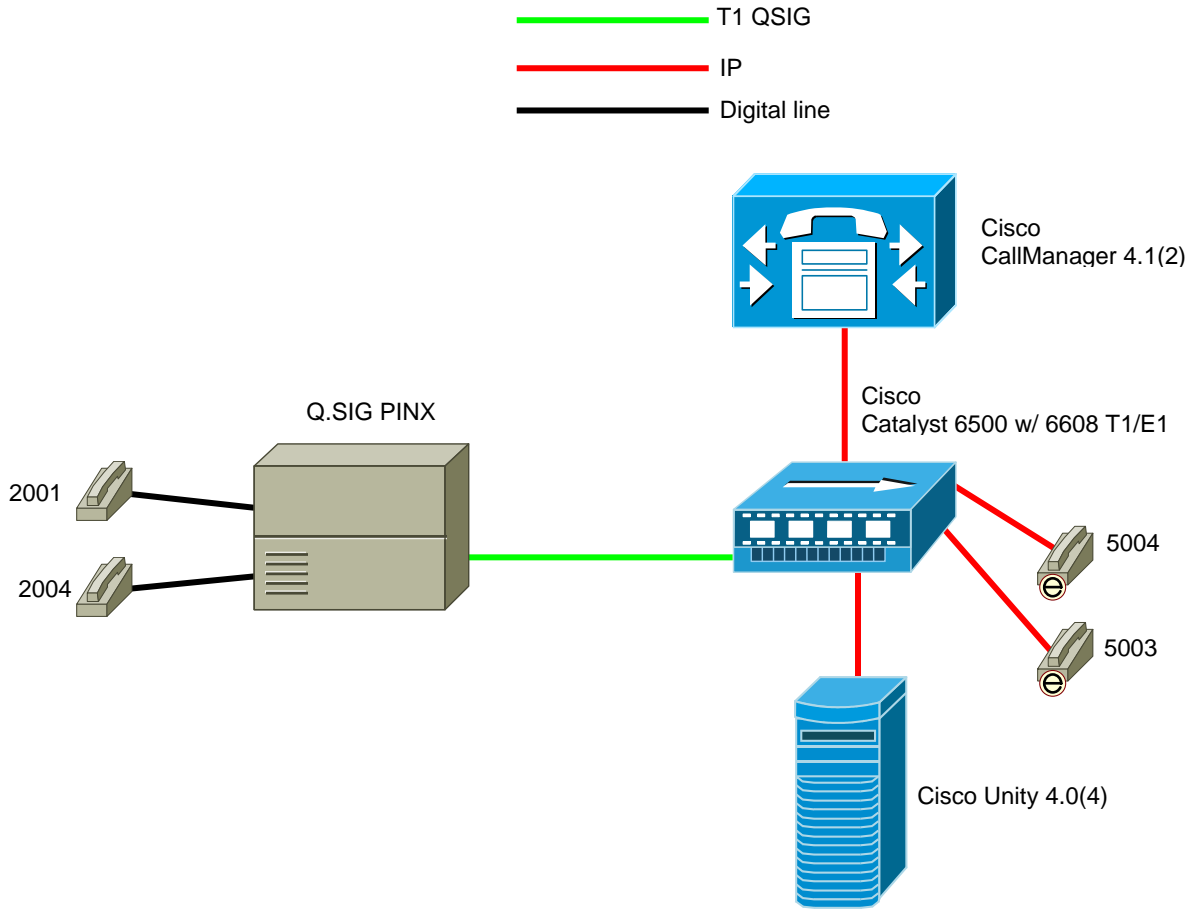
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Introduction

- This Application Note uses the Cisco Catalyst 6608 T1/E1 voice gateway; however other Cisco voice gateways are also an option to use since Cisco Unified CallManager QSIG implementation does not depend on the physical interface. Centralized voicemail services rely heavily on the inter-PBX protocol used to connect remote PBXs to the “Message Center” PBX. In the scenario represented/tested by this document, QSIG protocol and the Additional Network Features (ANF) supported by each vendor’s PBX will determine the actual features supported by this solution.
- This document contains lab performance criteria that describes the interoperability of Cisco Unified CallManager Release 4.1(2) when used as a Message Center PINX in a QSIG private network with a Centralized Cisco Unity voicemail server using SCCP integration.
- The network topology diagram (Figure 1) shows the test setup for end-to-end interoperability with the Cisco Unified CallManager connected to a remote PINX via Catalyst 6608 T1/E1 blade port used as MGCP gateways configured as T1 ISO QSIG trunk, and a Cisco Unity used as a Centralized voicemail system using SCCP integration to the Message Center Cisco Unified CallManager.

Network Topology

Figure 1. Network Topology



Limitations

There are no known limitations with this feature.

System Components

Hardware Requirements

The following hardware is required:

- Cisco Catalyst 6500 switch with Cisco WS-X6608-T1 Module
- Cisco Unified CallManager 4.1(2) MCS server
- Cisco Unity 4.0(4) MCS server



Software Requirements

The following software is required:

- Cisco Unified CallManager Release 4.1(2)
- Cisco Unity 4.0(4)

Features

This section lists new and changed features and features that are not supported.

Features Supported

- Station forward to personal greeting (busy/ring no answer/all calls)
- Reply to messages left in telephone answering mode
- Direct call
- Message Waiting Indication
- Support of Message Waiting indication on remote PINX stations is dependent on whether the PBX software used on the remote PINX supports QSIG MWI. Please refer to PBX manufacturer's technical documentation for further information. Also, refer to Cisco Interoperability Application Notes, found in the Cisco Interoperability portal, located at the following URL: www.cisco.com/go/interoperability

Features Not Supported

- None are known at this time.

Configuration

This section contains configuration menus and commands and describes configuration sequences and tasks. The configuration examples shown in this document is an example to typical settings only. Note: Prior to configuring the "central" Definity G3 and Cisco Unified CallManager, the following should be installed and tested for proper functionality:

- Cisco Unity - Test for proper functionality of Cisco Unity voicemail server for local Cisco Unified CallManager users.
- Cisco Unified CallManager – Test for proper Q.SIG features transparency between Cisco Unified CallManager and remote PINX.

Configuring the Cisco Unified CallManager

Configuring Cisco Unified CallManager for QSIG

1. Configure the QSIG-related system parameters
2. Configure the MGCP gateway
3. Configure the MWI Off/On directory numbers
4. Configure the Cisco Unity ports

Note: For detailed information on how to configure MGCP gateways connecting to specific manufacturers' PBX, refer to the proper application notes located in the Cisco Interoperability website, found at the following URL: www.cisco.com/go/interoperability



Configuring the Cisco Catalyst 6608 T1/E1 Voice Gateway

The following screen captures depict a typical configuration of a Cisco Catalyst 6608 T1 blade configured for QSIG functionality:

Gateway Configuration

[Back to Find/List Gateways](#)
[Dependency Records](#)

Product : Cisco Catalyst 6000 T1 VoIP Gateway
Gateway : S0/DS1-0@SDA0001C9D93A9B
Device Protocol: Digital Access PRI
Registration: Registered with Cisco CallManager 172.20.236.2
IP Address: [172.20.236.16](#)

Status: Ready

Update

Delete

Reset Gateway

Device Information

MAC Address*	<input type="text" value="0001C9D93A9B"/>
Description	<input type="text" value="Cat 6500 port 5/4"/>
Device Pool*	<input type="text" value="Default"/>
Call Classification*	<input type="text" value="OnNet"/>
Network Locale	<input type="text" value="United States"/>
Media Resource Group List	<input type="text" value="< None >"/>
Location	<input type="text" value="< None >"/>
AAR Group	<input type="text" value="< None >"/>
Load Information	<input type="text"/>

Multilevel Precedence and Preemption (MLPP) Information



Multilevel Precedence and Preemption (MLPP) Information

MLPP Domain (e.g., "0000FF")	<input type="text"/>
MLPP Indication	<input type="text" value="Off"/>
MLPP Preemption	<input type="text" value="Disabled"/>

Interface Information

PRI Protocol Type*	<input type="text" value="PRI QSIG T1"/>
Protocol Side*	<input type="text" value="User"/>
Channel Selection Order*	<input type="text" value="Bottom Up"/>
Channel IE Type*	<input type="text" value="Use Number when 1B"/>
PCM Type*	<input type="text" value="μ-law"/>
Delay for first restart (1/8 sec ticks)	<input type="text" value="32"/>
Delay between restarts (1/8 sec ticks)	<input type="text" value="4"/>
<input checked="" type="checkbox"/> Inhibit restarts at PRI initialization	
<input type="checkbox"/> Enable status poll	



Call Routing Information

Inbound Calls

Significant Digits*	<input type="text" value="All"/>
Calling Search Space	<input type="text" value=" < None >"/>
AAR Calling Search Space	<input type="text" value=" < None >"/>
Prefix DN	<input type="text"/>

Outbound Calls

Calling Line ID Presentation*	<input type="text" value="Allowed"/>
Calling Party Selection*	<input type="text" value="Originator"/>
Called party IE number type unknown*	<input type="text" value="Cisco CallManager"/>
Calling party IE number type unknown*	<input type="text" value="Cisco CallManager"/>
Called Numbering Plan*	<input type="text" value="Cisco CallManager"/>
Calling Numbering Plan*	<input type="text" value="Cisco CallManager"/>
Number of digits to strip*	<input type="text" value="0"/>
Caller ID DN	<input type="text"/>
SMDI Base Port*	<input type="text" value="0"/>



PRI Protocol Type Specific Information

- Display IE Delivery
- Redirecting Number IE Delivery - Outbound
- Redirecting Number IE Delivery - Inbound
- Send Extra Leading Character In DisplayIE***
- Setup non-ISDN Progress Indicator IE Enable****
- MCDN Channel Number Extension Bit Set to Zero**
- Send Calling Name In Facility IE
- Interface Identifier Present**

Interface Identifier Value**

Connected Line ID Presentation (QSIG Inbound Call)*

UUIE Configuration

- Passing Precedence Level Through UUIE

Security Access Level

Product Specific Configuration

Clock Reference*	<input type="text" value="Network"/>
TX-Level CSU*	<input type="text" value="0dB"/>
FDL Channel*	<input type="text" value="ATT 54016"/>
Framing*	<input type="text" value="ESF"/>
Audio Signal Adjustment into IP Network*	<input type="text" value="NoDbPadding"/>
Audio Signal Adjustment from IP Network*	<input type="text" value="NoDbPadding"/>
Yellow Alarm*	<input type="text" value="Bit2"/>
Zero Suppression*	<input type="text" value="B8ZS"/>
Digit On Duration(50-500ms)*	<input type="text" value="100"/>
Interdigit Duration(50-500msec)*	<input type="text" value="100"/>
SNMP Community String	<input type="text" value="public"/>
Disable SNMP Set operations*	<input type="checkbox"/>
Debug Port Enable*	<input checked="" type="checkbox"/>
Hold Tone Silence Duration*	<input type="text" value="0"/>
Port Used for Voice Calls*	<input checked="" type="checkbox"/>
Port Used for Modem Calls*	<input checked="" type="checkbox"/>
Port Used for Fax Calls*	<input checked="" type="checkbox"/>



Fax and Modem Parameters

Fax Relay Enable*	<input checked="" type="checkbox"/>
Fax Error Correction Mode Override*	<input checked="" type="checkbox"/>
Maximum Fax Rate*	14400bps
Fax Payload Size*	20
Non Standard Facilities Country Code*	65535
Non Standard Facilities Vendor Code*	65535
Fax/Modem Packet Redundancy*	<input type="checkbox"/>
NSE Type*	Non-IOS Gateways

Playout Delay Parameters

Initial Playout Delay*	40
Minimum Playout Delay*	20
Maximum Playout Delay*	150

Echo Canceller Configuration

Echo TailLength (ms)*	32 ms
Minimum ERL (db)*	6 db

* indicates required item

** applicable to DMS-100 protocol only

*** applicable to DMS-100 protocol and DMS-250 protocol only



Configuring Cisco Unified CallManager QSIG-related Service Parameters

The following screen captures show QSIG-related service parameters, as found on Cisco Unified CallManager 4.1(2) software:

Clusterwide Parameters (Feature - Forward)		
Parameter Name	Parameter Value	Suggested Value
Forward Maximum Hop Count*	<input type="text" value="12"/>	12
Forward No Answer Timer (sec)*	<input type="text" value="12"/>	12
Max Forward Hops to DN*	<input type="text" value="12"/>	12
Retain Forward Information*	<input type="text" value="False"/>	False
Forward By Reroute Enabled*	<input type="text" value="True"/>	False
Forward By Reroute T1 Timer (sec)*	<input type="text" value="10"/>	10
Include Original Called Info for Q.SIG Call Diversions*	<input type="text" value="Always"/>	Only after the first diversion



Clusterwide Parameters (Feature - Path Replacement)

Parameter Name	Parameter Value	Suggested Value
Path Replacement Enabled*	<input type="text" value="True"/>	False
Path Replacement on Tromboned Calls*	<input type="text" value="True"/>	True
Start Path Replacement Minimum Delay Time (sec)*	<input type="text" value="0"/>	0
Start Path Replacement Maximum Delay Time (sec)*	<input type="text" value="0"/>	0
Path Replacement T1 Timer (sec)*	<input type="text" value="30"/>	30
Path Replacement T2 Timer (sec)*	<input type="text" value="15"/>	15
Path Replacement PINX ID	<input type="text" value="5555"/>	



Configuring Cisco Unity for Centralized Voicemail Services

Whether using Cisco Unity as a centralized voicemail server in a Q.SIG private network or to provide voicemail services to Cisco Unified CallManager users only, configure the platform as per the Cisco Unified CallManager 4.1 Integration Guide for Unity 4.0, found at the following URL: http://www.cisco.com/en/US/products/sw/voicesw/ps2237/prod_configuration_guide09186a00802933d0.html

Acronyms

Acronym	Definitions
ANF-PR	Additional Network Feature Path Replacement
CUCM	Cisco Unified Communication Manager
CCBS	Call Completion to Busy Subscriber
CCNR	Call Completion on No Reply
CFB	Call Forwarding on Busy
CFNR	Call Forwarding No Reply
CFU	Call Forwarding Unconditional
CLIP	Calling Line (Number) Identification Presentation
CLIR	Calling Line (Number) Identification Restriction
CNIP	Calling Name Identification Presentation
CNIR	Calling Name Identification Restriction
COLP	Connected Line (Number) Identification Presentation
COLR	Connected Line (Number) Identification Restriction
CONP	Connected Name Identification Presentation
CONR	Connected Name Identification Restriction
CT	Call Transfer
MWI	Message Waiting Indicator
PSTN	Public Switched Telephone Network



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