

Research in Motion Limited BlackBerry Mobile Voice System 5.0 with Cisco Unified Communications Manager Release 8.0.2 and 8.0.3



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Introduction

- This application note details the interoperability test results for Cisco Unified Wireless Network (CUWN) 6.x, and Research In Motion® (RIM®) BlackBerry® Mobile Voice System 5.0.
 - o Release 8.0.3 was tested with H323, while Release 8.0.2 was tested with SIP.
- All testing was performed with BlackBerry® Enterprise Server 5.0 configured on MCS7800 server running Microsoft Windows Server 2003 which is registered into a Microsoft Exchange 2007 SP1 domain.
- The network topology diagram (Figure 1) shows the test setup for end-to-end interoperability with the Cisco Unified Communications Manager version 8.0.2. The Cisco Unified Communications Manager 8.0.2 was configured on MCS7825.
- BlackBerry® Bold™ 9000, BlackBerry® Bold™ 9700 smartphones with AT&T SIM cards, BlackBerry® Storm™ 9500 smartphone son Verizon, and BlackBerry® Storm™ 9500 smartphone on Sprint were used for this test effort.
- Cisco Wireless LAN Controllers with 6.x(WLC) and Access Points were used for this BlackBerry Wi-Fi test effort.
- WiFi testing included call scenarios with BlackBerry devices roaming within domains (layer 2) and between domains (layer 3)
- This application note uses the Cisco 2811 voice router as MGCP, SIP, or H323 gateway. However, the use of other Cisco voice gateways is also an option because the MGCP, SIP, or H323 application is independent of the Cisco platform. The gateway families listed below can run MGCP, SIP, or H323, but each has different VoIP capacity or capability. Please check the product specifications to ensure that you are obtaining the proper device to support your VoIP deployment, including add-ons such as DSP resources.

Cisco 2800 series Integrated Services Router

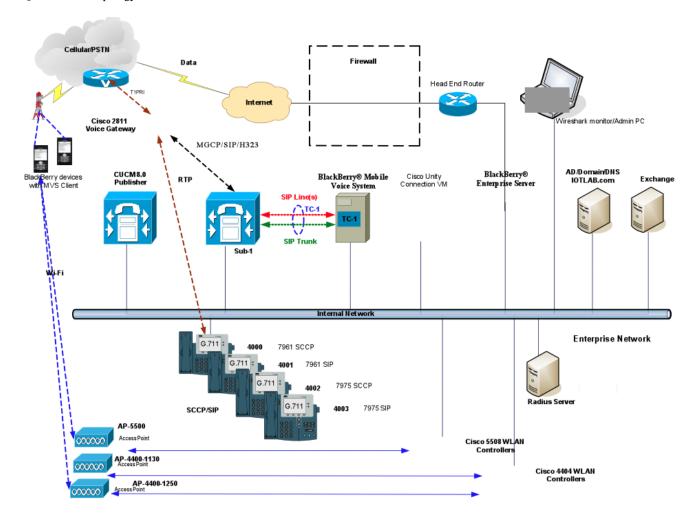
Cisco 3800 series Integrated Services Router

Cisco AS5400 series Universal Gateways



Network Topology

Figure 1. Test Topology





Limitations

These are the known limitations, caveats, or integration issues.

- The testing was done using FCS version of 8.0.2 release at the time of when testing started, and version 8.0.3 for H323.
- When performing an unattended (blind) transfer and the transferee is a BlackBerry® Mobile Voice System client, the transferee will not provide ringback (RIM DT 540632)
- When performing a consult transfer where the transferee releases before the transfer is complete, and the transferor is a BlackBerry®
 Mobile Voice System client, the transferor will enter an incorrect state and become unaware that the transferee call leg has been
 disconnected (RIM DT 693318)
- A semi-attended (blind) transfer is unavailable when H323 trunks are configured for PSTN access (RIM DT forthcoming).
- BlackBerry MVS client initiated calls, using a cellular network transport, will continue to provide ring back tone and not play the early
 media, such as inband tones generated by the network or IVR prompts. MVS client initiated calls, using the WiFi transport, supports the
 use of early media.
- BlackBerry MVS devices can not be controlled by CTI Application
- Call-forward can only configured from the UCM and Cisco IP Phone.
- Barge from BlackBerry Mobile Voice System devices is not support in this release and can only be a participant.
- Call-Park/Retrieve on BlackBerry Mobile Voice System devices can not Call-Park, but can retrieve a Call-Park.
- IDivert on BlackBerry Mobile Voice System devices is not supported.
- Interaction between Single Number Reach (SNR) (also known as Cisco Unified Mobility) and Cisco Unified Mobile Communicator with the BlackBerry® MVS was not tested.

System Components

The following hardware was tested:

- Cisco 2800NM-IPVOICE-M IOS 15.0(1)
- Cisco Unified IP phones 7900 series (CP7961 skinny, CP7975 SIP) Software Requirements
- Cisco 7800 series server running Windows 2003 Server Edition
- BlackBerry® Bold™ 9000, BlackBerry® Bold™ 9700, BlackBerry® Storm™ 9500 smartphones MVS client 2.0.1
- Cisco Unified Wireless LAN Controllers with 6.x (WLC) and APs with 12.4

The following software was tested:

- Cisco Unified Communications Manager: CUCM 8.0.2
- Cisco Unified Communications Manager: CUCM 8.0.3
- Cisco Unified IP phone 7961
- BlackBerry® Mobile Voice System 5.0
- BlackBerry Enterprise Server® 5.0
- COP file for BlackBerry MVS Device type
- BlackBerry Device Operating System 5.0



Features

Feature Supported

This section listed the supported features.

MVS client support features: lists all features supported by MVS client devies.

Features Supported:

- Basic Calls
- PSTN Calls
- Call Forward- Please refer to the Limitation section for more information
- Call Hold/Resume
- Multiple calls per line (call waiting)
- Consult Call Transfer
- Semi-Attended Call Transfer (Blind Call Transfer)-Please refer to the Limitation section for more information.
- Calling and Connected Line ID
- Calling and Connected Name
- Multiple lines per phone
- SIP Registration
- Incoming Call Screening
- Outgoing Call Screening
- Caller-ID Update
- Barge –Please refer to the Limitation section for more information
- Conference- Please refers to the Limitation section for more information
- Call Park/Retrieve- Please refer to the Limitation section for more information
- iDivert (if applicable)-Please refer to the Limitation section for more information
- DTMF services and MWI
- Roaming test (layer 2 and layer 3 WiFi roaming)

BlackBerry MVS Client specific features:

- Voice over Wi-Fi calls
- Move Call to Desk Phone
- Move Call to One-time Number
- Move Call from WiFi to Cellular Radio
- Move Call from Cellular to WiFi Radio



Environments and Configurations

This section contains configuration menus and commands and describes configuration sequences and tasks.

For updated information on BlackBerry Server refer to:

 $BlackBerry @\ Server\ Documentation\ web\ portal:$

www.blackberry.com/go/serverdocs

BlackBerry® Enterprise Server for Microsoft Exchange:

 $\underline{http://docs.blackberry.com/en/admin/subcategories/?userType=2\&category=BlackBerry+Enterprise+Server+for+Microsoft+Exchange\&subCategory=\&url=\%2Fadmin\%2Fsubcategories\%2F\&versionId=494$

BlackBerry® Mobile Voice System 5.0:

http://docs.blackberry.com/en/admin/subcategories/?userType=2&category=BlackBerry+MVS

Server Configurations

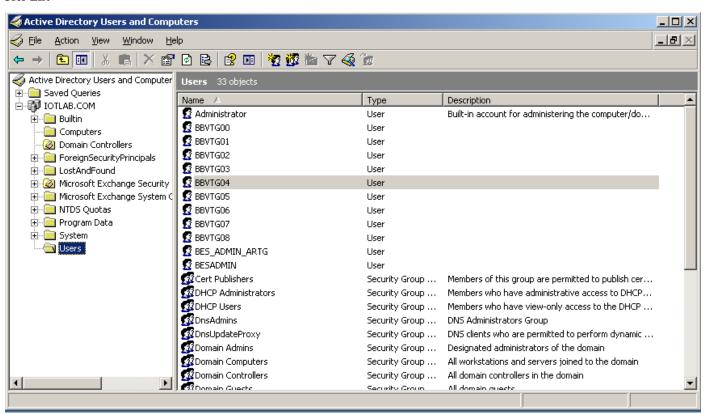
The following snapshot shows the required configurations for each component.

Microsoft Active Directory and Exchange Server

Microsoft Active Directory Server

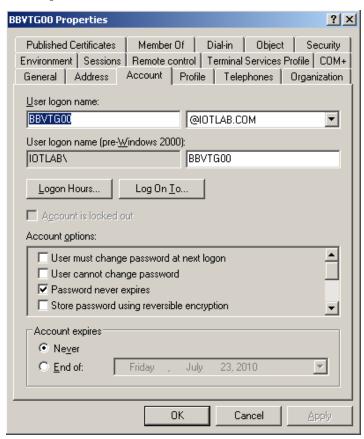
Create user on the Active Directory Domain and Exchange that will be integrated to RIM®'s ® BlackBerry® Enterprise Server and BlackBerry® Mobile Voice System.

User List



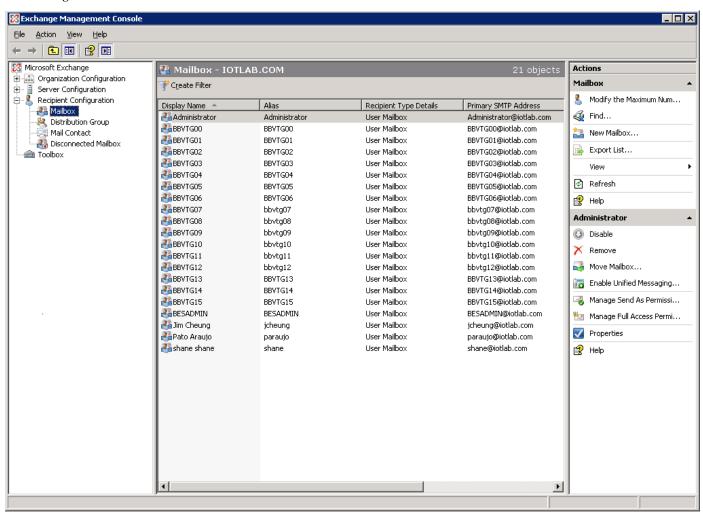


User Properties



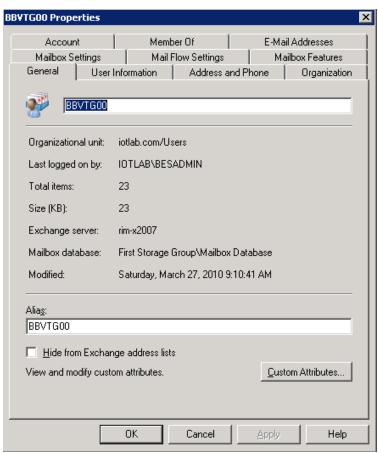


MS Exchange User List





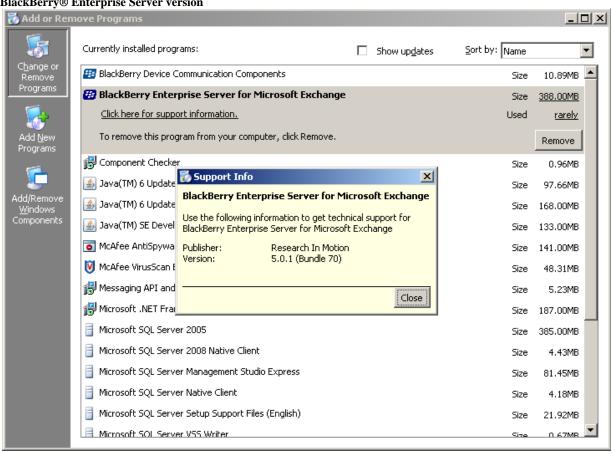
MS Exchange User Properties



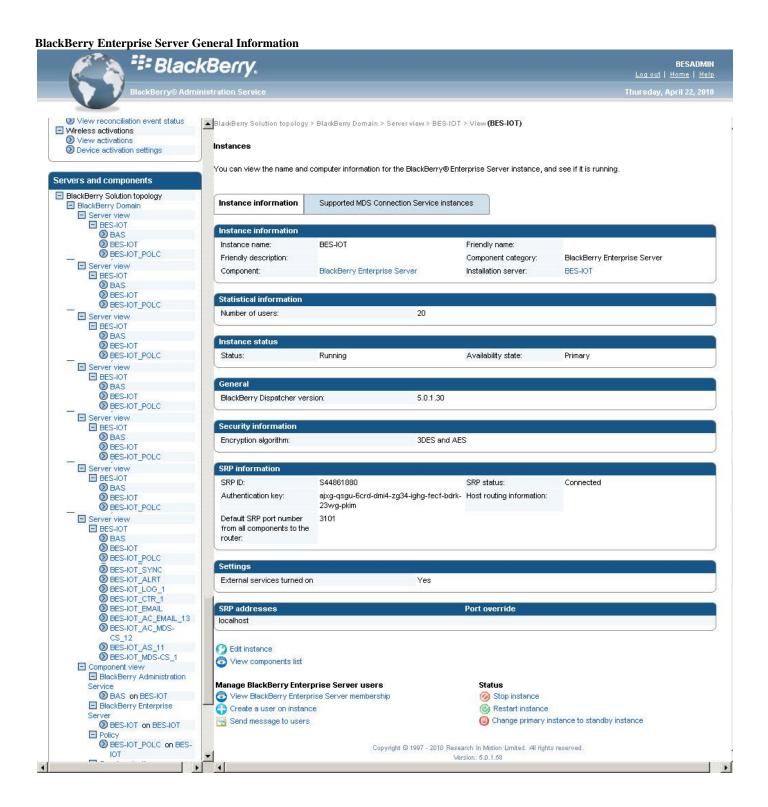


BlackBerry Enterprise Server

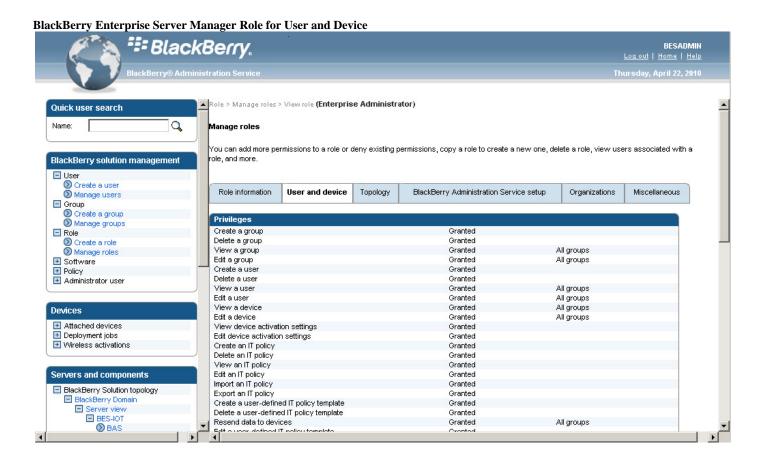
BlackBerry® Enterprise Server version





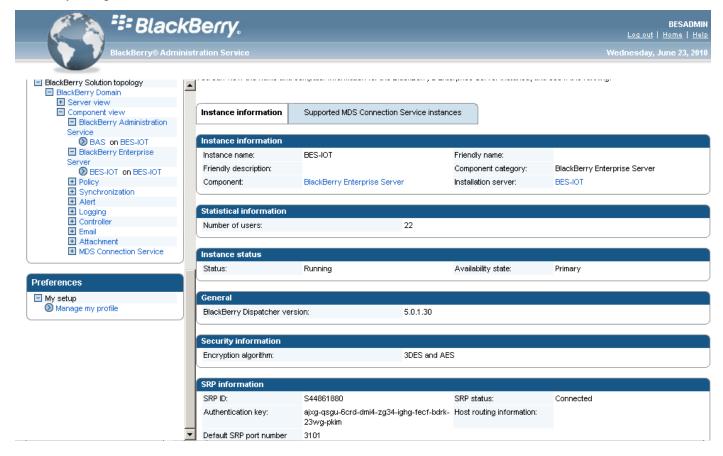






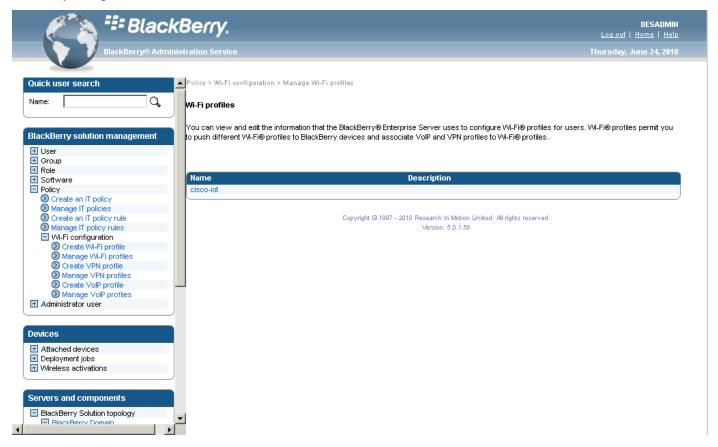


BlackBerry Enterprise Server Instance Information





BlackBerry Enterprise Server Wi-Fi Profile





BlackBerry Enterprise Server Wi-Fi Profile Configurations (1 of 2)

Wi-Fi profiles

You can view and edit the information that the BlackBerry® Enterprise Server uses to configure WI-Fi® profiles for users. WI-Fi® profiles permit you to push different WI-Fi® profiles to BlackBerry devices and associate VoIP and VPN profiles to WI-Fi® profiles.

VVI-Fi profile information	Wi-Fi profile settings
----------------------------	------------------------

rofile settings	Value	Description
M-Fi Allow Handheld Changes	Yes	This rule is obsolete as of BlackBerry Enterprise Server version More
M-Fi Link Security	EAP-FAST	Specifies the type of security required for Wi-Fi network access. (Open, WEP, More
M-Fi SSID	rim	Type the network name of the Wi-Fi network and its access points.
		This rule More
Vi-Fi Default Key ID		Type the Default WEP Key ID.
		Note: The VVEP Key ID must match the desired VVEP More
M-Fi WEP Key 1		Type the password for WEP key 1 using the format xxxxxxxxxxxxxxxxxxxxxxxxxxxxxxxxxxxx
		Allowable More
M-Fi WEP Key 2		Type the password for WEP key 2 using the format xxxxxxxxxxxxxxxxxxxxxxxxxxxxxxxxxxxx
		Allowable More
VI-FiWEP Key 3		Type the password for WEP key 3 using the format xxxxxxxxxxxxxxxxxxxxxxxxxxxxxxxxxxxx
		Allowable More
M-FiWEP Key 4		Type the password for WEP key 4 using the format xxxxxxxxxxxxxxxxxxxxxxxxxxxxxxxxxxxx
		Allowable More
Vi-Fi Preshared Key		Type the pre-shared key.
		This rule applies only to Java-based BlackBerry More
M-Fi User Name	cisco	Type the user name for EAP-PEAP or EAP-LEAP security access on the BlackBerry More
M-Fi User Password	cisco	Type the user password for EAP-PEAP or EAP- LEAP security access on the More
VI-Fi DHCP Configuration	Yes	Specify whether or not Dynamic Host Configuration Protocol (DHCP) is used for More



Wi-Fi Profile Configurations (2 of 2)

Vi-Fi Profile Configurations (2 of 2)		
VVI-Fi IP Address		Type the IP address in IP address format (for example,10.0.0.1) for use if More
VVI-Fi Subnet Mask		Type the subnet mask in dotted format (for example, 10.0.0.1) for use if More
Wi-Fi Primary DNS		Type the primary DNS in IP address format (for example, 10.0.0.1) if the DHCP More
WI-FI Secondary DNS		Type the secondary DNS in dotted format (for example, 10.0.0.1) if the DHCP is More
Wi-Fi Default Gateway		Type the default gateway in IP address format (for example, 10.0.0.1) if the More
WI-Fi Minimal EAP-TLS Certificate Encryption Key Security Level		This rule is obsolete as of BlackBerry Enterprise Server version More
Wi-Fi Enable Authentication Page		This rule is obsolete as of BlackBerry Enterprise Server version 4.1.4.
		Set to More
VVI-Fi Hard Token Required		Set to Yes if a hard token (e.g., RSA SecurID) is required as part of the More
Wi-Fi Token Serial Number		Serial number of the soft token provisioned to the device if a soft token is More
Wi-Fi Profile ∀isibility	Full visibility	Specify whether the user may view the settings of this Wi-Fi profile.
		Full More
Wi-Fi Profile Editability	Full editability	Specify whether the user may edit the settings of this Wi-Fi profile.
		Full More
Wi-Fi Allow Password Save	Yes	Specify whether users can save WI-Fi passwords on the BlackBerry device.
		Set More
Wi-Fi Roaming Threshold	Auto	The roaming threshold determines how often the VM-Fi radio shall scan for More
WI-Fi Server Subject		Type the substring used to match the "Subject" field of the server's More
WI-Fi Server SAN		Type the substring used to match the "SubjectAltName" (SAN) field of the More
Wi-Fi Inner Authentication Mode	EAP-MSCHAPV2	Specify the authentication mode for tunneled EAP security. One of: None, More
RESERVED for Wi-Fi Protected Access Credential		RESERVED FOR FUTURE USE
Key		This rule applies only to Java-based BlackBerry More
Wi-Fi Domain Suffix		Type the internal domain name suffix using the FQDN format, for use if the DHCP More
Wi-Fi Allow AP to AP Handover	Yes	Set to Yes to enable Wi-Fi handovers between access points for this profile. More
Wi-Fi Band Type	802.11 a/b/g	Specify the band type or types, which access point of a particular SSID are More
Wi-Fi EAP-FAST Provisioning method	Both	Specify the type of provisioning method used for EAP-FAST PAC More
Wi-Fi Disable Server Certificate Validation	Yes	This setting is applicable to Wi-Fi security types which require CA certificate More
Wi-Fi BlackBerry Infrastructure Wi-Fi Access Mode	Access does not require VPN	Specify whether BlackBerry device can use BlackBerry Infrastructure over a More
Associated Certificate Authority Configuration		A CA configuration is a set of credentials that is

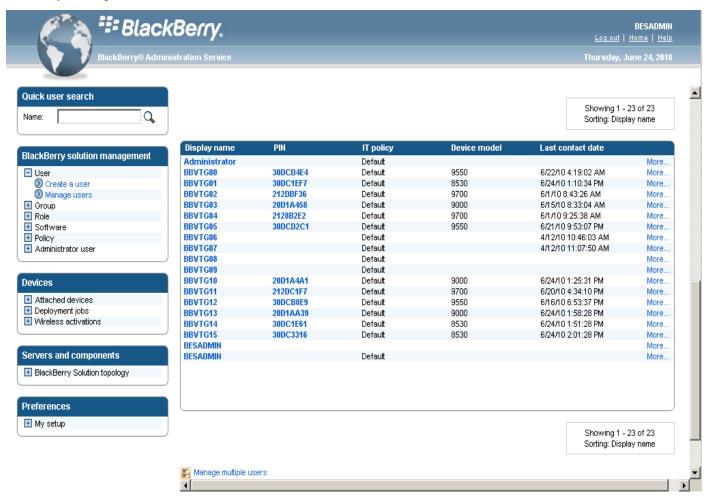
Wi-Fi associations	Value	Description
Associated VPN Profile	None Defined	Contains the name of the associated VPN Profile. This is a hidden More
Associated VoIP Profile	None Defined	Contains the name of the associated VoIP Profile. This is a hidden More





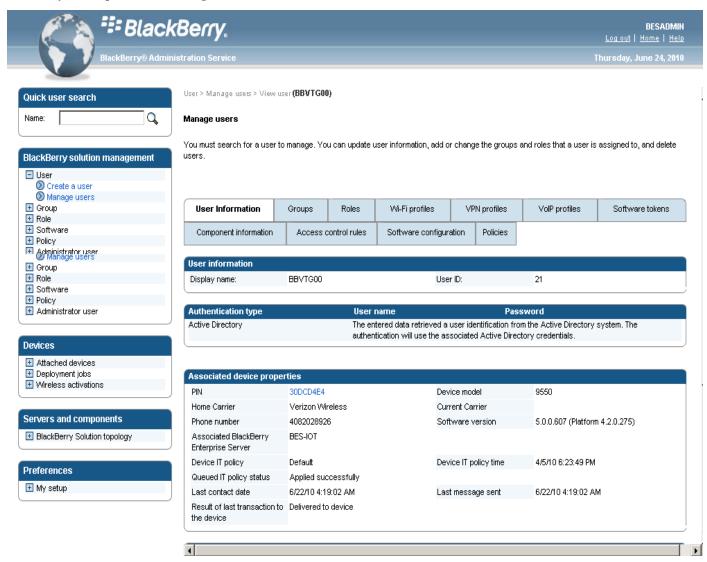
BlackBerry Enterprise Server Manage User

BlackBerry® Enterprise Server User List



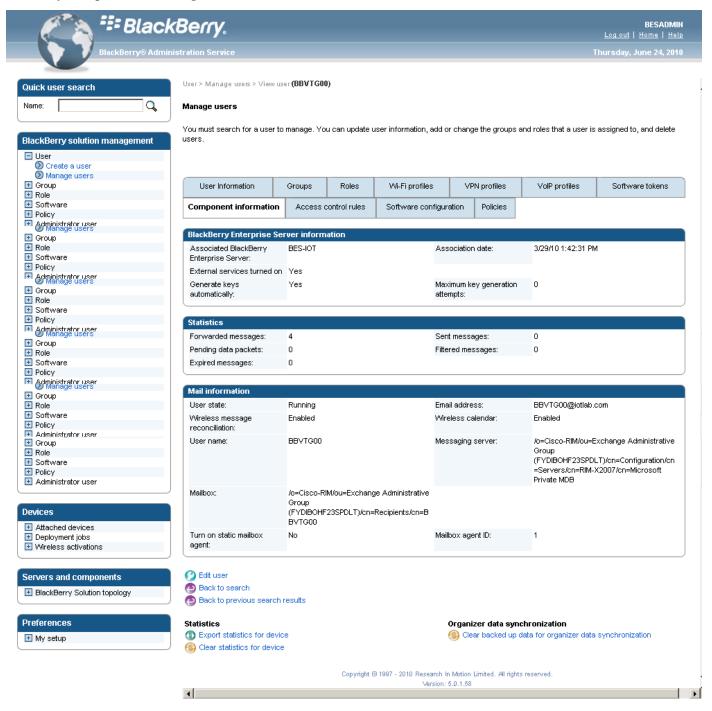


BlackBerry® Enterprise Server Manage Users (1 of 2)





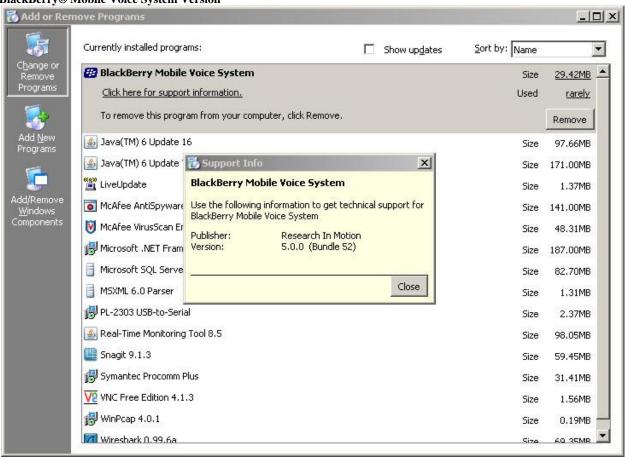
BlackBerry Enterprise Server Manage Users (2 of 2)





BlackBerry Mobile Voice System

BlackBerry® Mobile Voice System Version





BlackBerry Mobile Voice System Configurations

BlackBerry® Mobile Voice System Telephony Connector to the UCM *** BlackBerry. User: sysadmin About | Change Password | Log Off Edit Telephony Connector - UCM8_0 Home Dashboard Display Name UCM8_0 System Configuration Туре Cisco Unified Communications Manager 7.0/7.1 💌 * ☐ Session Managers IP Address 172,20,89,253 --- Add Host Name 172.20.89.253 - Manage Line Port 5060 - High Availability E-Telephony Connectors Trunk Port 6060 --- Add Registration Timeout 1200 Manage ⊟ Locations PBX Initiated Calling --- Add Caller Identification Number * --- Manage Number Action B-Voice Mail Connectors 0 --- Add --- Manage 14085551212 亩 User Management Phone Number Translation ⊟--Users Location -Add Manage Save Cancel Import . Templates --- Add ··· Manage ⊟ Class of Service --- Add Manage Administrator Management - Administrative Roles - Add -- Manage ⊟ Administrators -- Add Manage Services Reporting CDR Viewer Copyright @ 2010 Research In Motion Limited



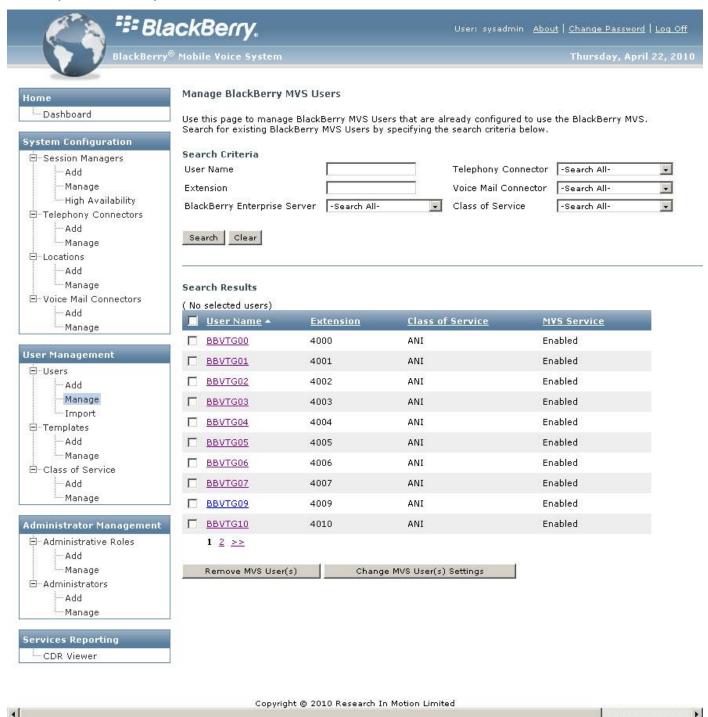
BlackBerry® Mobile Voice System Session Manager for the UCM





BlackBerry Mobile Voice System User Configuration

BlackBerry® Mobile Voice System User List



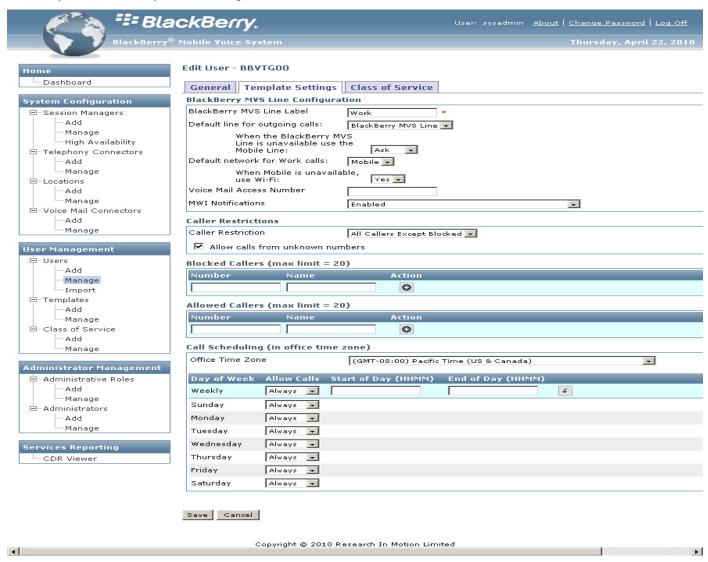


BlackBerry® Mobile Voice System User Information





BlackBerry® Mobile Voice System User Template Information





BlackBerry® Mobile Voice System Mobile Template BlackBerry. User: sysadmin <u>About</u> | <u>Change Password</u> | <u>Log Off</u> Edit Template - all-mobile Home Template Name all-mobile Dashboard BlackBerry MYS Line Configuration System Configuration BlackBerry MVS Line Label ⊟-Session Managers --- Add Default line for outgoing calls: BlackBerry MVS Line 💌 Manage When the BlackBerry MVS High Availability Line is unavailable use the Ask 🔻 Mobile Line: - Telephony Connectors Default network for Work calls: Mobile 🕶 -Add When Mobile is unavailable, Manage Yes 🕶 - Locations Voice Mail Access Number --- Add Manage MWI Notifications Enabled • B-Voice Mail Connectors Caller Restrictions -- Add -Manage Caller Restriction All Callers Except Blocked 💌 Allow calls from unknown numbers User Management ⊟--Users Blocked Callers (max limit = 20) -Add Number Name Action Manage 0 -- Import ☐ Templates Allowed Callers (max limit = 20) --- Add Number Name Action Manage -Class of Service 0 - Add Call Scheduling (in office time zone) Manage Office Time Zone (GMT-08:00) Pacific Time (US & Canada) -Administrator Management Day of Week Allow Calls Start of Day (HHMM) End of Day (HHMM) - Administrative Roles - Add Always 💌 Weekly £ --- Manage Always 💌 Sunday ⊟ Administrators Always 🐷 Monday bbA Manage Tuesday Always 💌 Wednesday Always 💌 Services Reporting Thursday Always 🔻 CDR Viewer Friday Always 🕶 Saturday Always 💌 Save Cancel Copyright @ 2010 Research In Motion Limited



BlackBerry® Mobile Voice System Wi-Fi Templates

	🖲 Mobile Voice Sy	stem			Thursday, April 22, 201
16	Edit Template	- all-wifi			
ashboard	Template Name	9	all-wifi	*	
n Configuration	BlackBerry MV	S Line Config	uration		
sion Managers	BlackBerry MVS			*	
dd	Default line for	outgoing calls:	BlackBerry MVS Line	-	
nage		the BlackBerr			
Availability	Line i	s unavailable u e Line:			
Connectors	Default network				
ige	998,0	. Wi-Fi is unava	1		
R:		1 wi-i is aliava 1obile:	No 💌		
	Voice Mail Acces	ss Number			
age	MWI Notification	ıs	Enabled		
Connectors	Caller Restrict				71111
3				=	
	_ Caller Restrictio		All Callers Except Blo	cked 💌	
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or Management	Office Time 201	ie	(GMT-08:00) Pacific	Time (US & Canada)	•
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BlackBerry® Mobile Voice System Class of Service

Æ ≅ Bla	ickBerry.			User: sysadmin <u>A</u>	bout Change Password Log Off
BlackBerry [®]	Mobile Voice System				Thursday, April 22, 2010
Home	Edit Class of Service -	ANI			
Dashboard	Class of Service Name	ANI		*	
C	BlackBerry MVS Call F	eatures			
System Configuration	✓ User can transfer a	15			
- Add	✓ User can move a ca		e number		
Manage	✓ User can move a ca				
High Availability	✓ User can move a ca	I to a one time num	nber		
⊟-Telephony Connectors Add	✓ User can manually n				
Manage				obile to Voice over Wi-Fi	
- Locations	☑ User can have a sch				
Add	☑ Device can set the n	nobile phone numbe	er		
Manage	Call Direction				
⊟ Voice Mail Connectors Add	Incoming Calls	PBX	. → BlackBe	erry Smartphone	
Manage		C Black	kBerry Sma	artphone → PBX	
156 2scent Modern Indicated Control Control	Outgoing Calls	PBX	: → BlackBe	erry Smartphone	
User Management		C Black	kBerry Sma	artphone → PBX	
⊟Users					
Manage	Wi-Fi Profile Control				
Import	Allow Voice over Wi-Fi c	alling on these push	ned Wi-Fi pr	ofiles	
⊟Templates					
Add Manage	☑ cisco-iot				
☐ Class of Service	✓ Allow Voice over Wi-	Fi calling on user o	reated Wi-F	i profiles	
Add		Trouming our ason of	roacoa mi r	1 promos	
Manage	Device Settings				
Administrator Management	RTP Local Port	51100	*		
⊟ Administrative Roles	Preferred Order of Device	e Codecs g711_u- -g711_A	-l-m		
Add		g729 g729b	1	1	
Manage					
⊟ Administrators Add	User Edit Control				
Manage	✓ User may change th	e mobile phone nur	mber		
	✓ User may change th	e call move to desk	phone nur	mber	
Services Reporting	✓ User may change th	e BlackBerry MVS I	line label		
CDR Viewer	✓ User may change th		35 35	5	
	☑ User may change th		number		
	☑ User may change ca		pll-p	MUO!!-	
	✓ User may change th	e derault network to	or BlackBer	ry MVS calls	
	Scheduling Control				
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	✓ User may set Thurse	lay scheduling			
	User may set Friday				
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	✓ User may set Sunda	y scheduling			
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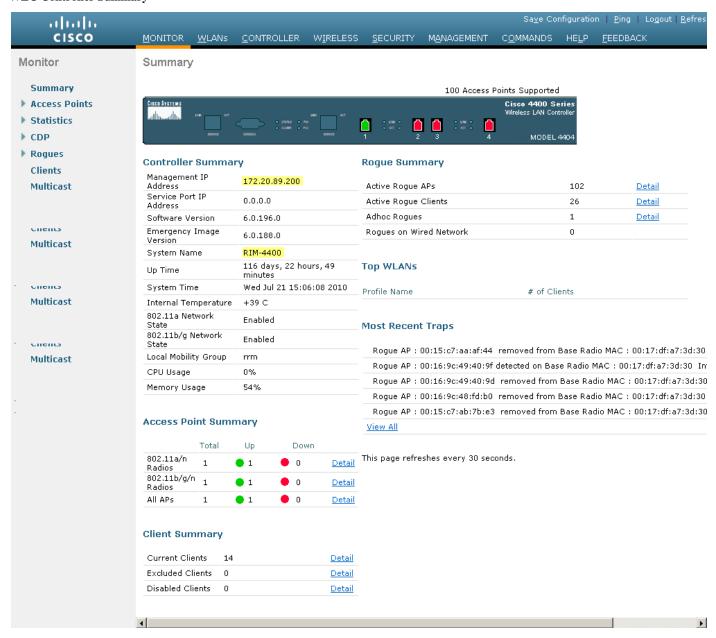


Wireless LAN Controllers and APs

This consisted of two Wireless LAN Controllers for mobile device roaming.

Wireless LAN Controller 1

WLC Controller Summary





WLC WLANs



WLC WLANs General

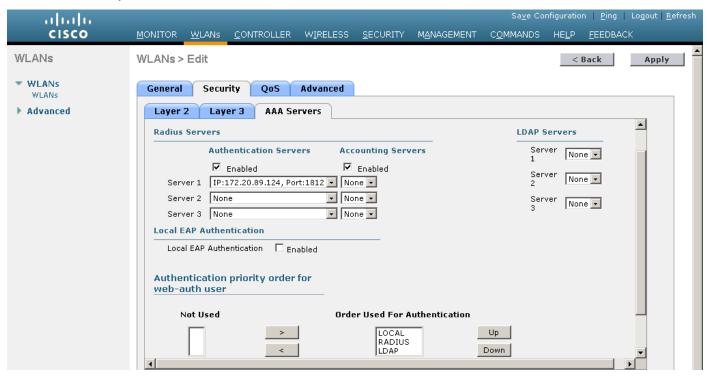


WLC WLANs Security layer 2

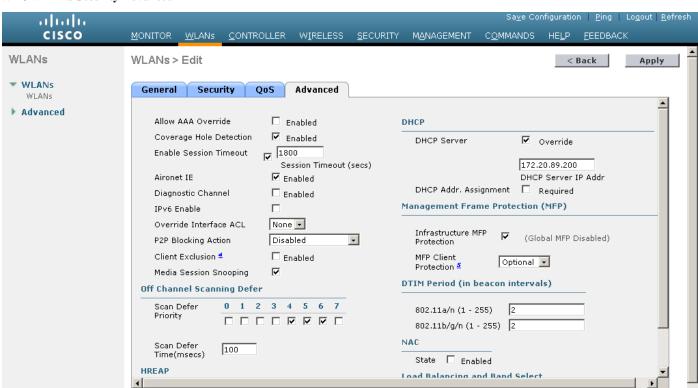




WLC WLANs Security AAA

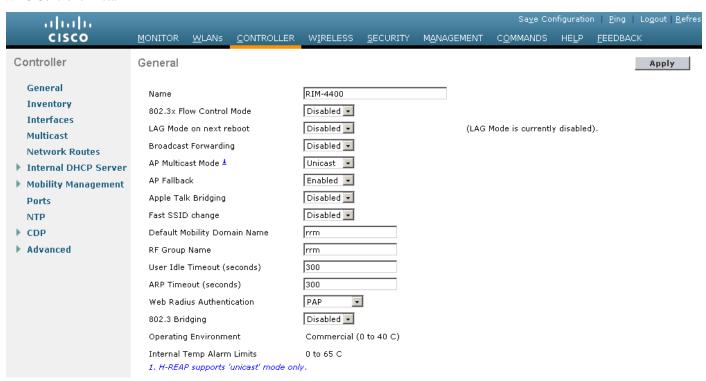


WLC WLANs Security Advanced





WLC Controller Detail



WLC Controller Inventory

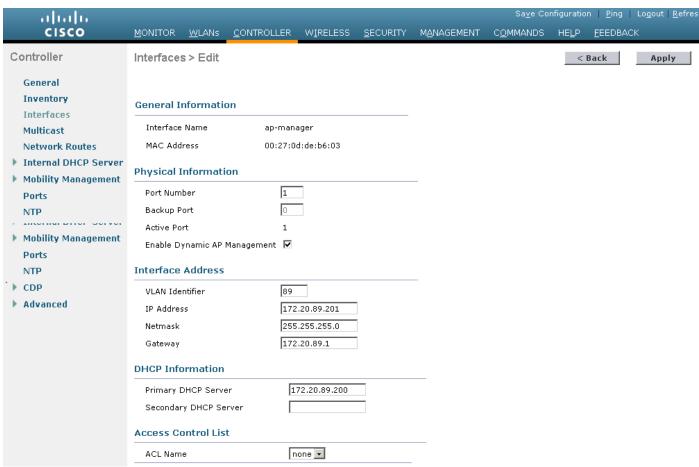




WLC Controller Interfaces

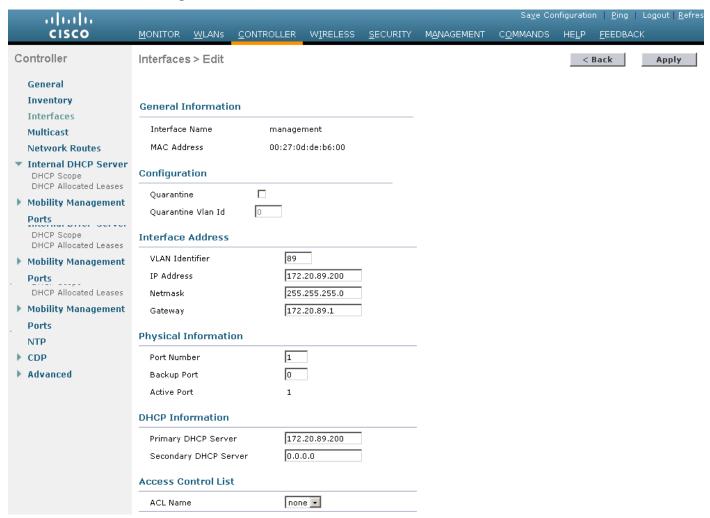


WLC Controller Interfaces AP-Manager

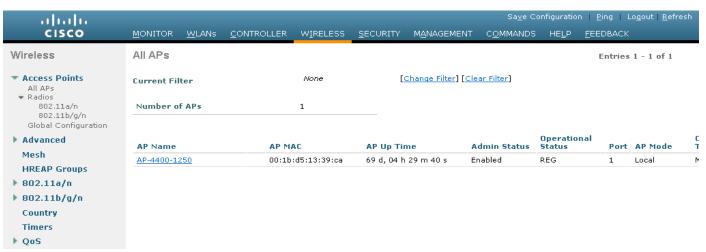




WLC Controller Interfaces Manager

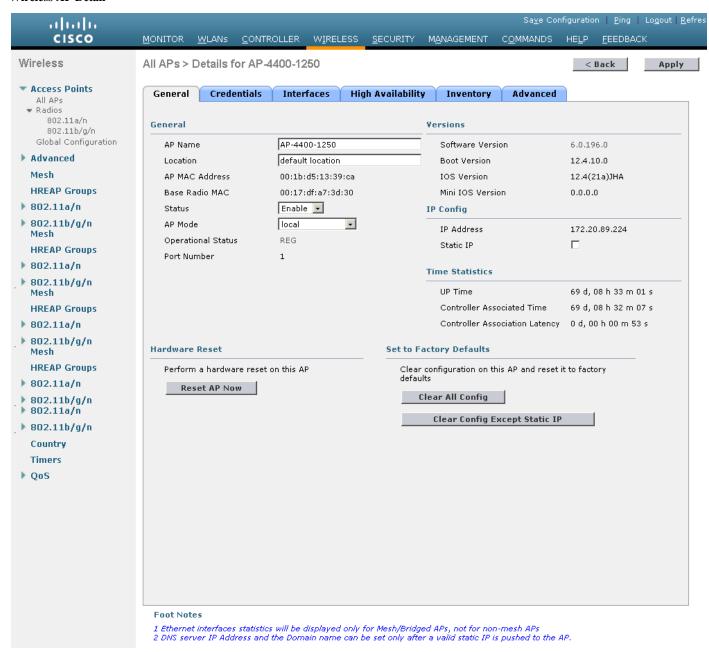


Wireless AP



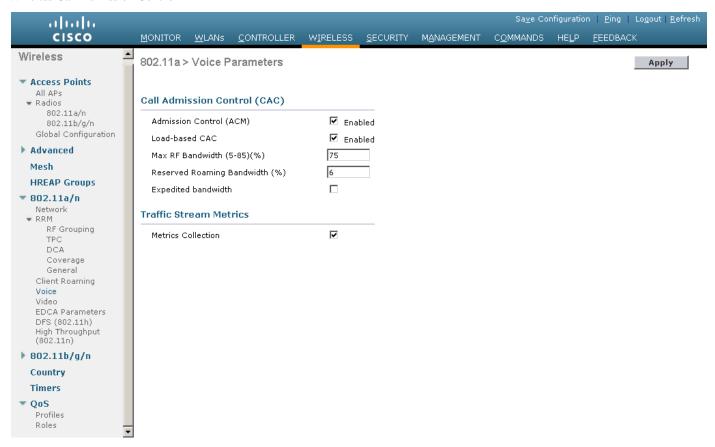


Wireless AP Detail



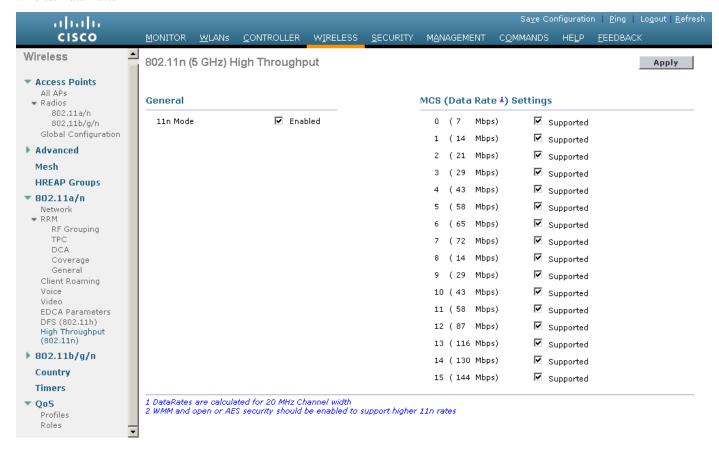


Wireless Call Admission Control





Wireless Data Rates



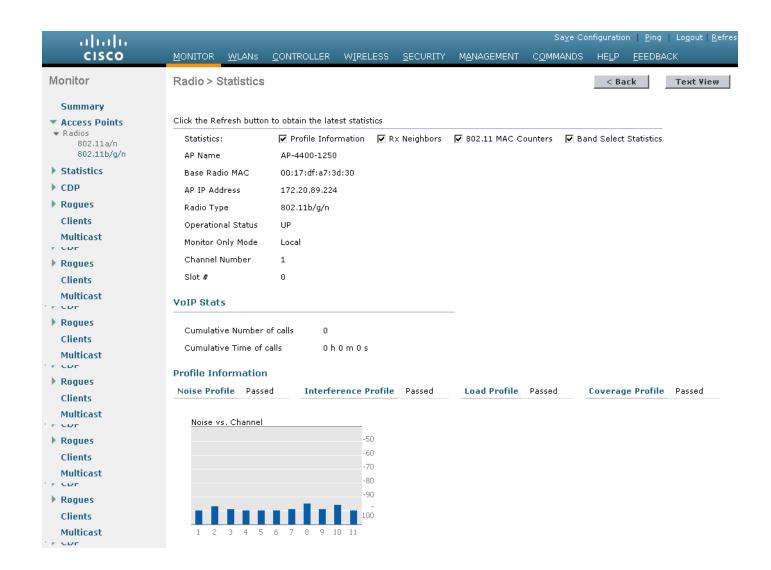


Wireless AP Radios



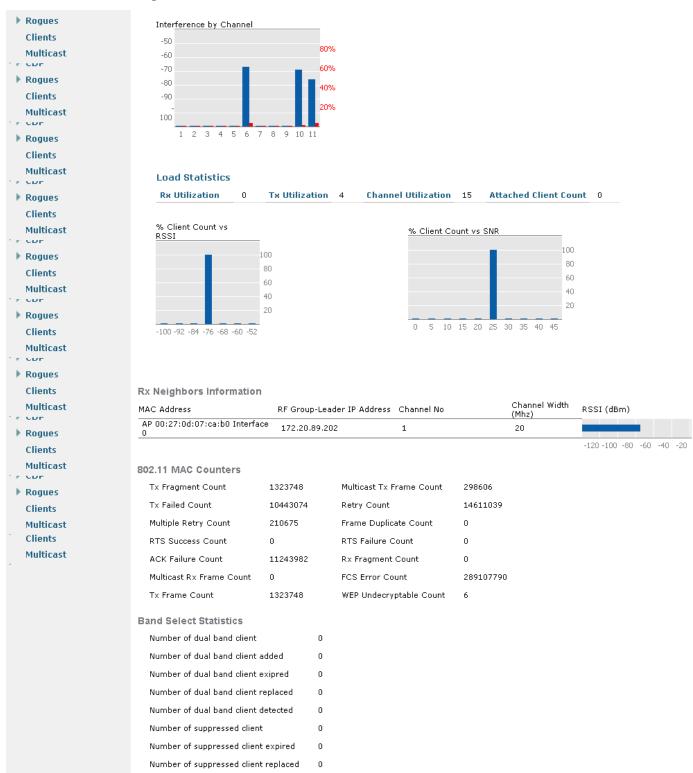
Wireless AP Radios 802.11b/g/n (1 of 2)





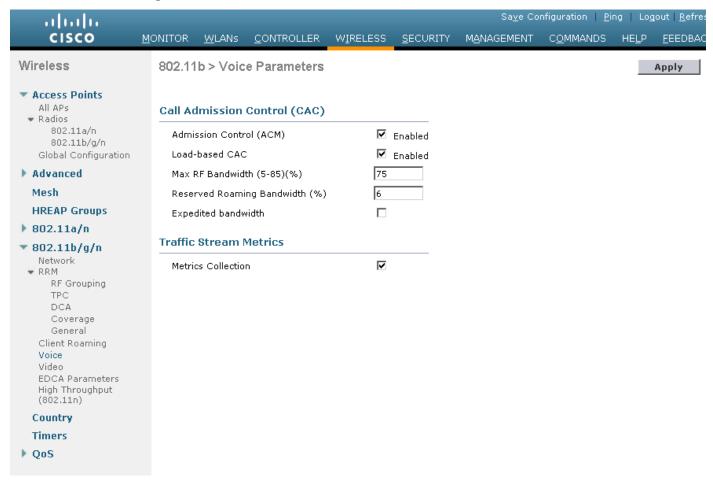


Wireless AP Radios 802.11b/g/n (2 of 2)





Wireless AP Radios 802.11b/g/n Call Admission Control



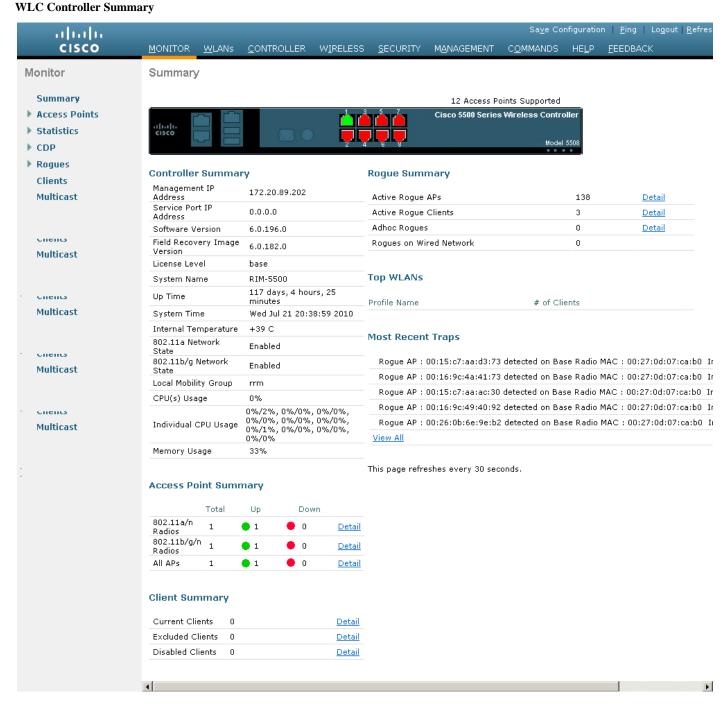


Wireless AP Radios 802.11b/g/n Call Admission Control

alialia						Sa <u>v</u> e Co	nfiguration <u>P</u> ir	ng Lo	gout <u>R</u> efre
cisco	<u>M</u> ONITOR	<u>W</u> LANs	<u>C</u> ONTROLLER	W <u>I</u> RELESS	<u>S</u> ECURITY	M <u>A</u> NAGEMENT	C <u>O</u> MMANDS	HE <u>L</u> P	<u>F</u> EEDBA
Wireless	802.11	b/g Glob	al Parameters	3					Apply
▼ Access Points All APs ▼ Radios 802.11a/n 802.11b/g/n Global Configuration ▶ Advanced Mesh	802.1 Beac	al .1b/g Netwo .1g Support on Period (r : Preamble	millisecs)	Enabled Enabled OO Enabled	:	ta Rates** 1 Mbps 2 Mbps 5.5 Mbps 6 Mbps	Mandato Mandato Mandato Supporte	ry 🔻	
HREAP Groups		nentation T		346	_	9 Mbps	Supporte		
▶ 802.11a/n	` '	s) Support,	<u> </u>	Z Enabled	:	11 Mbps	Mandato	ry 🔻	
▼ 802.11b/g/n Network ▼ RRM RF Grouping TPC	11n Pa	arameter:		Enabled	:	12 Mbps 18 Mbps 24 Mbps	Supporte Supporte Supporte	ed 🔻	
DCA Coverage	CCX L	ocation M	easurement		;	36 Mbps	Supporte		
General Client Roaming Voice Video	Mode		Γ	Enabled		48 Mbps 54 Mbps	Supporte		
EDCA Parameters High Throughput (802.11n) Country	speci implie may	fic rate will es that any communica	andatory' implies to not be able to ass associated client to te with the AP using ble to use the rate	ociate. Data Ra hat also suppo ng that rate. Bu	ite 'Supported' its that same i t it is not requ	rate iired			
Timers	assoc	date. The a	ctual data rates th 'as different chani	at are supporte	ed depend on a				
▶ QoS	band to se	widths. The lect the data rate allowed	reason is that we a rates. But in rea I for that channel	show data rate lity, the AP will	es and alow the pick the next	lower			



Wireless LAN Controller 2





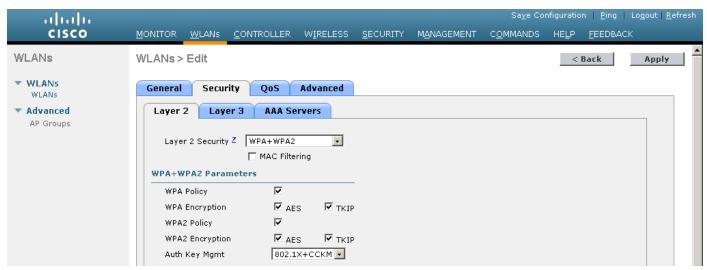
WLC WLANs



WLC WLANs General

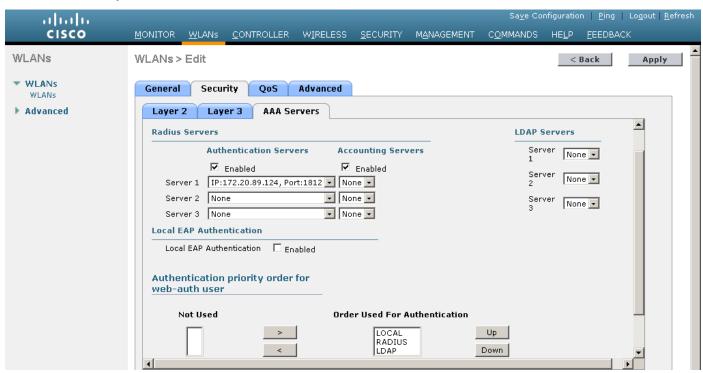


WLC WLANs Security layer 2

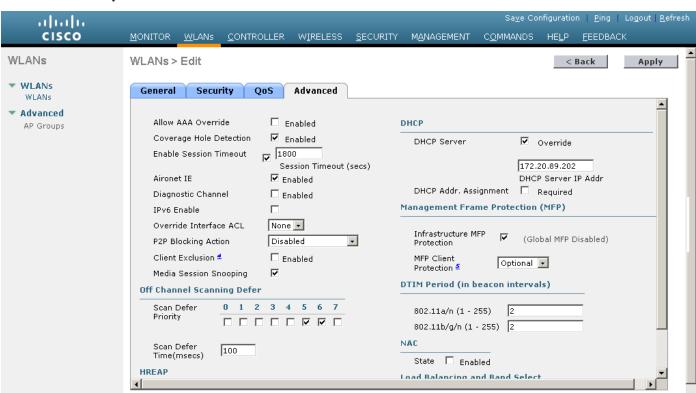




WLC WLANs Security AAA

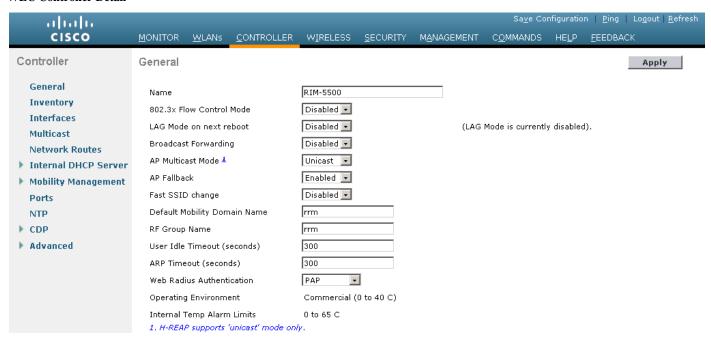


WLC WLANs Security Advance





WLC Controller Detail



WLC Controller Inventory





WLC Controller Interfaces





WLC Controller Interfaces Manager

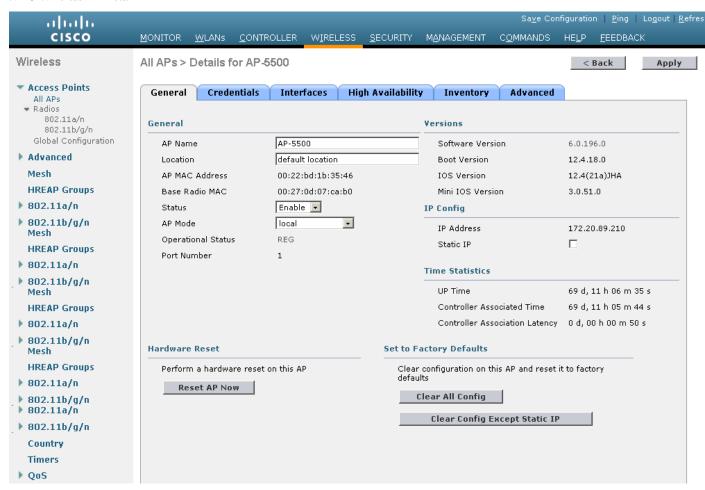
alialia							Sa <u>v</u> e Co	nfiguratio	n <u>P</u> ing	Logout <u>R</u> efres
CISCO	<u>M</u> ONITOR	<u>W</u> LANs	<u>C</u> ONTROLLER	W <u>I</u> RELESS	<u>S</u> ECURITY	M <u>A</u> NAGEMENT	C <u>O</u> MMANDS	HE <u>L</u> P	<u>F</u> EEDBA0	CK
Controller	Interfaces	> Edit						<	Back	Apply
General Inventory Interfaces Multicast Network Routes Internal DHCP Server DHCP Scope	General Interface MAC Addi	Name	manage	ment d:45:8a:60						
DHCP Allocated Leases Mobility Management Ports DHCP Scope DHCP Allocated Leases	Quarantin Quarantin NAT Addre	e e Vlan Id								
Ports DHCP Scope DHCP Allocated Leases	Interface VLAN Ide	Address	89			-				
Ports DHCP Scope DHCP Allocated Leases	IP Addres Netmask Gateway	s	255.2	20.89.202						
Mobility Management Ports	Physical I	nformati	on			-				
NTP CDP	Port Numl Backup Po		0							
► Advanced	Active Por Enable Dy Managem	rnamic AP ent	1							
	·	OHCP Serve y DHCP Se	rver 0.0.0	.0		-				
	ACL Name	e ing the Inte disabled an	none erface parameters d thus may result	causes the Wi	LANs to be nectivity for	-				



WLC Wireless AP

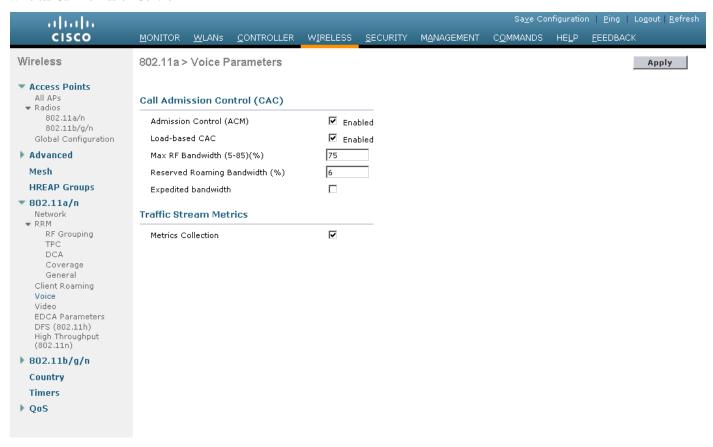


WLC Wireless AP Detail



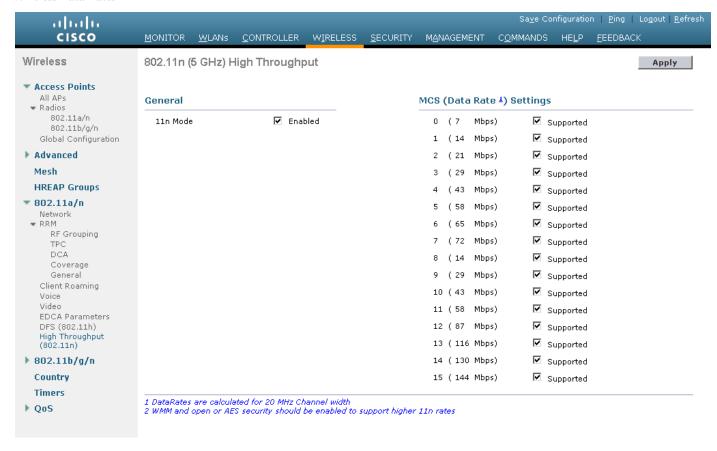


Wireless Call Admission Control





Wireless Data Rates

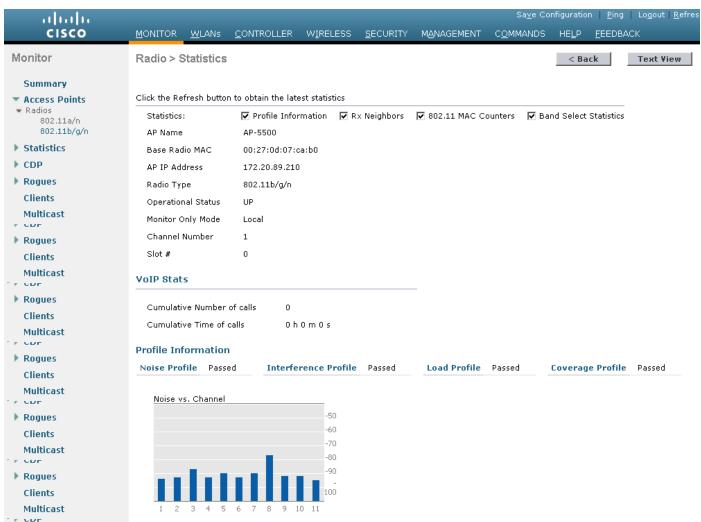




WLC Wireless AP Radios

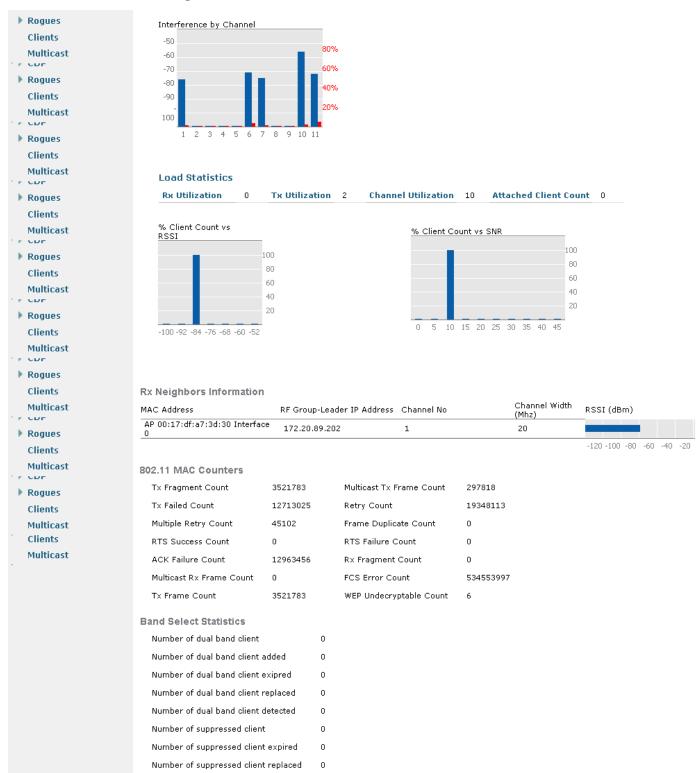


WLC Wireless AP Radios 802.11b/g/n (1 of 2)





WLC Wireless AP Radios 802.11b/g/n (2 of 2)





Configuring the Cisco Unified Communications Manager

Cisco Unified Communications Manager Version



Last Successful Logon: Jul 14, 2010 11:52:48 AM

Copyright © 1999 – 2009 Cisco Systems, Inc. All rights reserved.

This product contains cryptographic features and is subject to United States and local country laws governing import, export, transfer and use. Delivery of Cisco cryptographic products does not imply third-party authority to import, export, distribute or use encryption. Importers, exporters, distributors and users are responsible for compliance with U.S. and local country laws. By using this product you agree to comply with applicable laws and regulations. If you are unable to comply with U.S. and local laws, return this product immediately.

A summary of U.S. laws governing Cisco cryptographic products may be found at our Export Compliance Product Report web site.

For information about Cisco Unified Communications Manager please visit our <u>Unified Communications System Documentation</u> web site.

For Cisco Technical Support please visit our $\underline{\text{Technical Support}}$ web site.



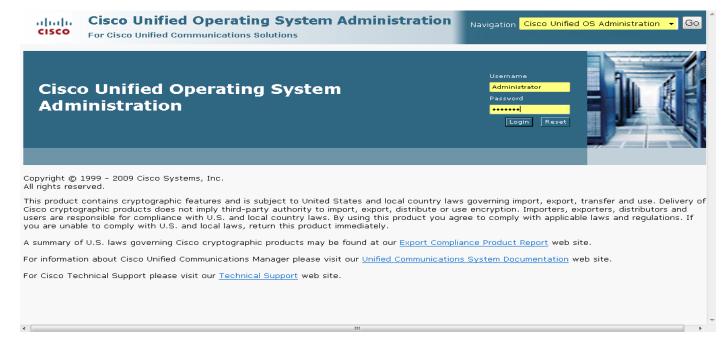
COP File and Installation

Make sure the BlackBerry Mobile Voice System COP file has been installed into UCM before configuring your BlackBerry devices.

Follow the link to download the COP file

https://www.blackberry.com/Downloads/entry.do?code=62F91CE9B820A491EE78C108636DB089

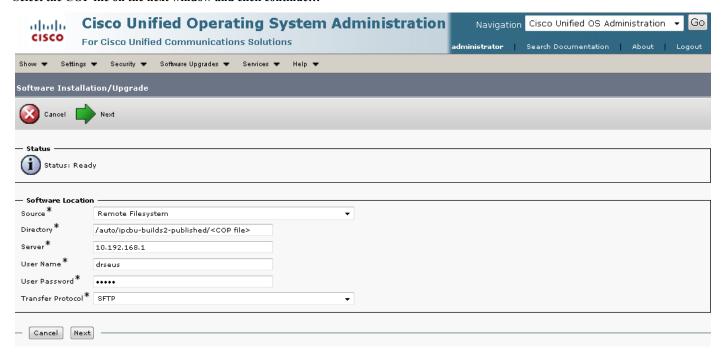
Cisco Unified OS Administration





Software Upgrades->Install/Upgrade

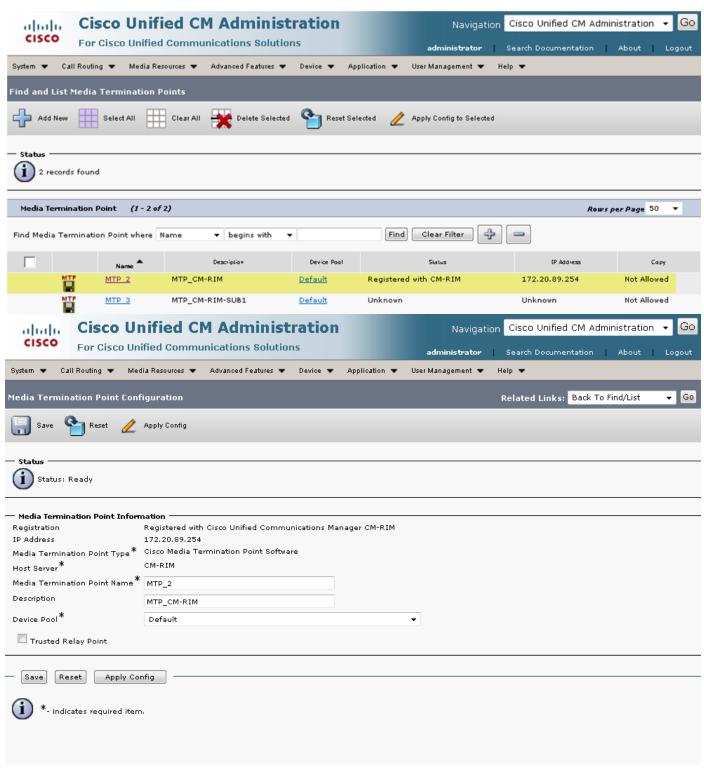
Select the COP file on the next window and click continue...





Media Termination Point Configurations

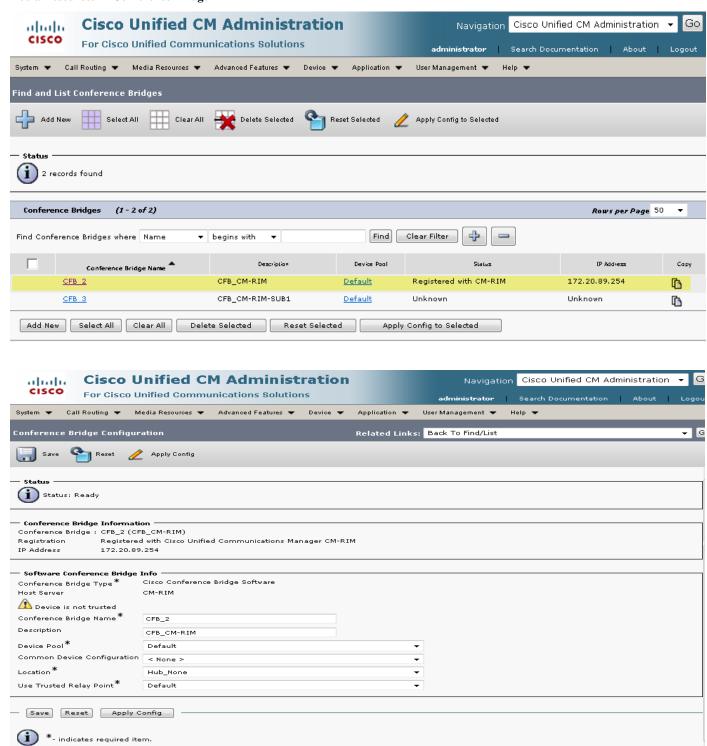
Media Resources -> Media Termination Point





Conference Bridge Configurations

Media Resources -> Conference Bridge





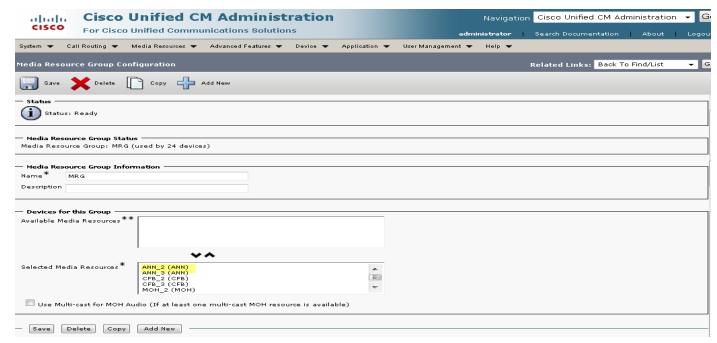
Resource Group Configurations

UCM Announciator for Blackberry Devices

Media Resources -> Resource Group **Cisco Unified CM Administration** G Navigation Cisco Unified CM Administration ahaha CISCO **For Cisco Unified Communications Solutions** System ▼ Call Routing ▼ Media Resources ▼ Advanced Features ▼ Application

User Management Media Resource Group Configuration Related Links: Back To Find/List Delete Copy Add New — Media Resource Group Information Name* MRG-MVS Description — Devices for this Group Available Media Resources ** Selected Media Resources E Use Multi-cast for MOH Audio (If at least one multi-cast MOH resource is available) Save Delete Copy Add New

UCM Annunciator for IP Phones

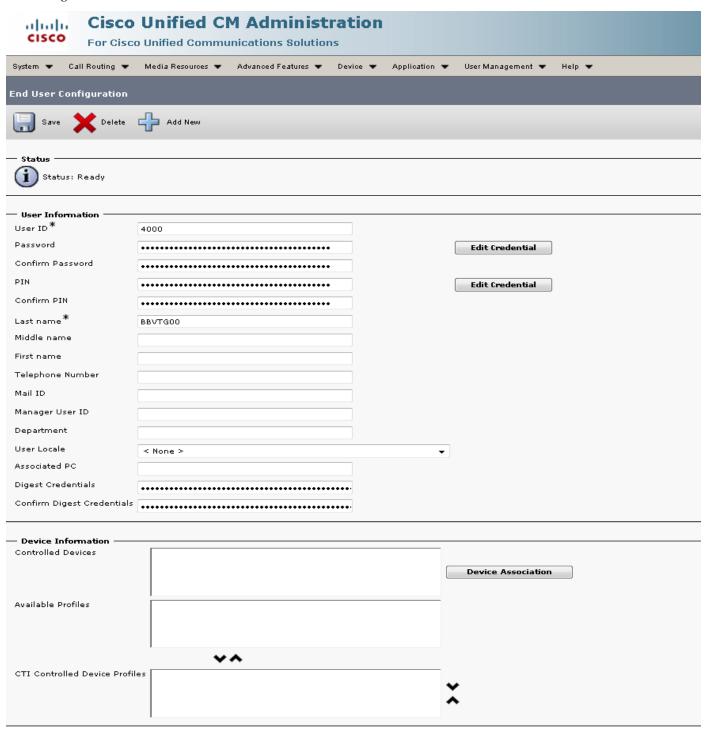




UCM User

User Config (1 of 2)

User Management->End User





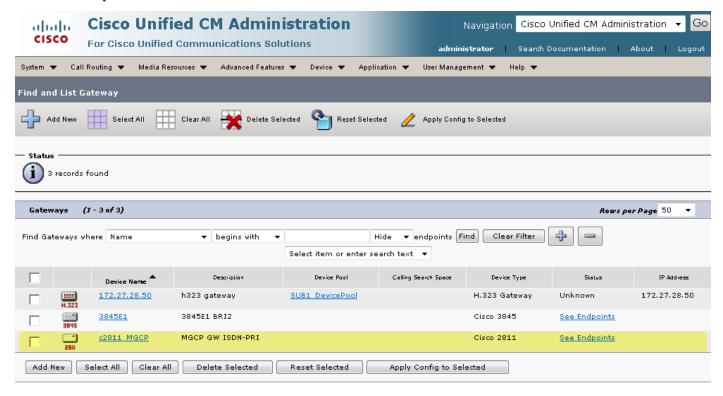
User Config (2 of 2)

Extension Mobility		
Available Profiles		
Controlled Profiles		**
		Y
		^
Default Profile		
	Not Selected	<u> </u>
Presence Group *	Standard Presence group	▼
SUBSCRIBE Calling Search Space	< None >	▼
Allow Control of Device from the second o		
Allow Control of Device from	311	
Enable Extension Mobility Cro	oss Cluster	
— Directory Number Associations		
Primary Extension < None >		
< None >		▼
— Mobility Information ————		
Enable Mobility		
Primary User Device	< None >	▼
		•
Enable Mobile Voice Access		
Maximum Wait Time for Desk Pic	kup [*] 10000	
Remote Destination Limit *	4	
Remote Destination Profiles	-	
Remote Destination Promes		
		View Details
	,	
— CAPF Information ———		
Associated CAPF Profiles		
		<u>View Details</u>
— Permissions Information ——		
Groups		
		Add to User Group
		Remove from User Group
		/iew Details
Roles		
		/iew Details
1	<u></u>	new Decails
— Save Delete Add New] ————	



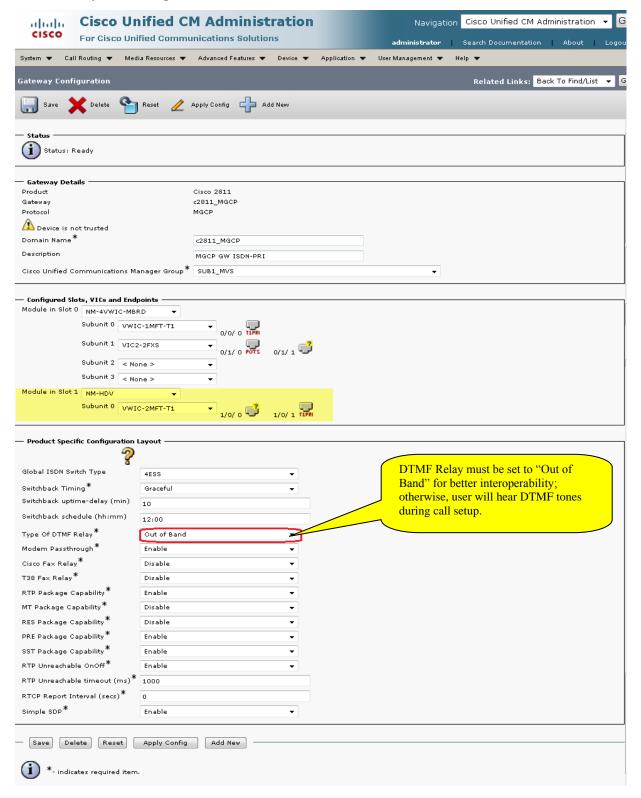
MGCP Gateway Configurations

Device -> Gateway



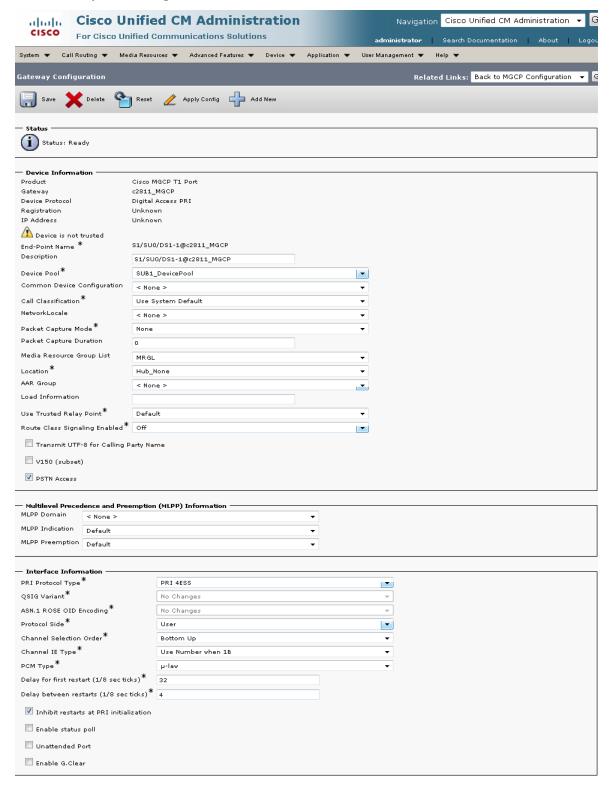


MGCP Gateway Detail Configurations (1 of 3)



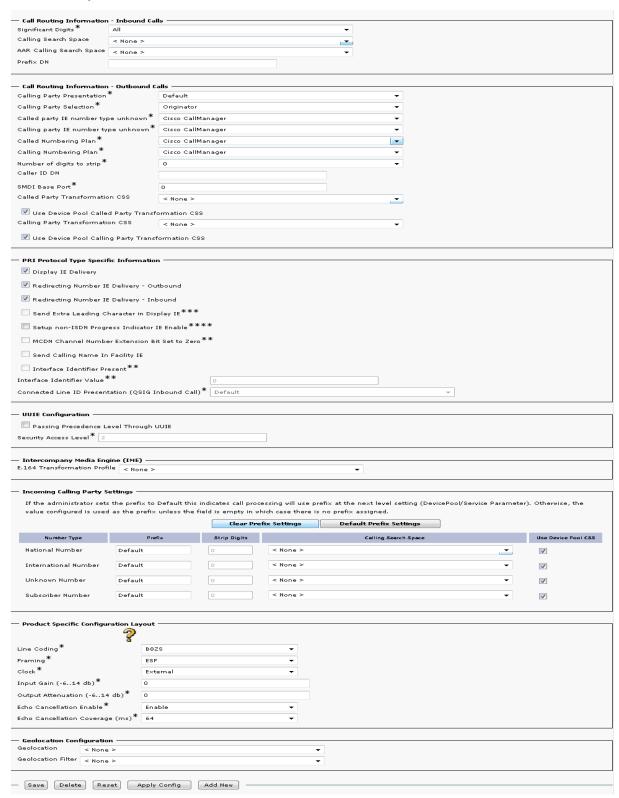


MGCP Gateway Detail Configurations (2 of 3)





MGCP Gateway Detail information (3 of 3)





CUCM SIP Trunk Detail Configuration (1 of 3)

alada Cisco Unified CM Administr	ation		Navigation Cisco Unified CM Administration 💌
CISCO For Cisco Unified Communications Solut	tions	administrat	or Search Documentation About Logo
System ▼ Call Routing ▼ Media Resources ▼ Advanced Fe	atures ▼ Device ▼ Application ▼ User Management		
9 2 4			
Trunk Configuration			Related Links: Back To Find/List
🔚 Save 🗶 Delete 睯 Reset 🙋 Apply Config 🔓	Add New		
- Status			
i Status: Ready			
- Device Information			
Product:	SIP Trunk		
Device Protocol:	SIP		
Trunk Service Type	None(Default)		
Device Name*	SUB1_TO_TC1		
Description			
Device Pool*	SUB1_DevicePool	v	
Common Device Configuration	< None >	•	
Call Classification*	Use System Default	•	
Media Resource Group List	MRGL	•	
Location*	Hub_None	-	
AAR Group	< None >	v	
Packet Capture Mode*	None	v	
Packet Capture Duration	0		
☐ Media Termination Point Required			
▼ Retry Video Call as Audio			
☐ Transmit UTF-8 for Calling Party Name			
☐ Unattended Port			
$\hfill \square$ SRTP Allowed - When this flag is checked, Encrypted 1 other information.	TLS needs to be configured in the network to provid	de end to end	security. Failure to do so will expose keys and
Route Class Signaling Enabled*	Default	•	
Use Trusted Relay Point*	Default	*	
PSTN Access			



CUCM SIP Trunk Detail Configuration (2 of 3)

-Intercompany Media Engine (E.164 Transformation Profile No						
E.164 Transformation Profile	IME)					
Z.Z. Transfermation Frome < No	ne >		v			
-Multilevel Precedence and Pre	eemption (MLPP) I	nformation ———				
MLPP Domain < None >		v				
-Call Routing Information						
▼ Remote-Party-Id						
Asserted-Identity						
Asserted-Type* Default		•				
SIP Privacy* Default		v				
— Inbound Calls —						
Significant Digits*	All		•			
Connected Line ID Presentation*	Default		٠			
Connected Name Presentation*	Default		•			
Calling Search Space	< None >		v			
AAR Calling Search Space	< None >		v			
Prefix DN						
Redirecting Diversion Header	Delivery - Inbound					
Incoming Calling Party Setti If the administrator sets the privalue configured is used as the	refix to Default this in				icePool/Service Para	meter). Otherwise, the
1						2.
	_	Clear Prefix Se		Default Prefix Settings		
Number Type	Prefix	Strip Digits	ettings			Use Device Pool CSS
Number Type Incoming Number Defau				Default Prefix Settings	•	Use Device Pool CSS
	CSS < None >	Strip Digits	ettings	Default Prefix Settings	•	
Incoming Number Defau Connected Party Settings— Connected Party Transformation Use Device Pool Connected F	n CSS < None >	Strip Digits	< None >	Default Prefix Settings Calling Search Space	<u> </u>	
Incoming Number Defau Connected Party Settings Connected Party Transformation ✓ Use Device Pool Connected F	n CSS < None >	Strip Digits	ettings	Default Prefix Settings Calling Search Space	<u> </u>	
Incoming Number Defau Connected Party Settings— Connected Party Transformation Use Device Pool Connected F Outbound Calls Called Party Transformation CSS Use Device Pool Called Party	CSS < None > Party Transformation < None > Transformation CSS	Strip Digits	< None >	Default Prefix Settings Calling Search Space	¥	
Incoming Number Defau Connected Party Settings— Connected Party Transformation ✓ Use Device Pool Connected F Outbound Calls Called Party Transformation CSS	CSS < None > Party Transformation < None > Transformation CSS	Strip Digits	< None >	Default Prefix Settings Calling Search Space	¥	
Incoming Number Defau Connected Party Settings— Connected Party Transformation Use Device Pool Connected F Outbound Calls Called Party Transformation CSS Use Device Pool Called Party Calling Party Transformation CSS Use Device Pool Calling Party Use Device Pool Calling Party	CSS None > Party Transformation None > Transformation CSS None >	Strip Digits 0 CSS	< None >	Default Prefix Settings Calling Search Space	¥	
Incoming Number Defau Connected Party Settings— Connected Party Transformation Use Device Pool Connected for Called Party Transformation CSS Use Device Pool Called Party Calling Party Transformation CSS Use Device Pool Calling Party Calling Party Selection*	CSS None > Party Transformation None > Transformation CSS None >	Strip Digits 0 CSS	< None >	Default Prefix Settings Calling Search Space		
Incoming Number Defau Connected Party Settings— Connected Party Transformation Use Device Pool Connected for Use Device Pool Called Party Transformation CSS Use Device Pool Called Party Calling Party Transformation CSS Use Device Pool Calling Party Calling Party Selection* Calling Line ID Presentation*	CSS < None > Party Transformation None > Transformation CSS None > Transformation CSS	Strip Digits 0 CSS	< None >	Default Prefix Settings Calling Search Space	<u> </u>	
Incoming Number Defau Connected Party Settings— Connected Party Transformation Use Device Pool Connected for Use Device Pool Connected Party Transformation CSS Use Device Pool Called Party Calling Party Transformation CSS Use Device Pool Calling Party Calling Party Selection* Calling Line ID Presentation* Calling Name Presentation*	CSS None > Party Transformation None > Transformation CSS None > Transformation CSS Originator	Strip Digits 0 CSS	< None >	Default Prefix Settings Calling Search Space	×	
Incoming Number Defau Connected Party Settings— Connected Party Transformation Use Device Pool Connected for Use Device Pool Called Party Transformation CSS Use Device Pool Called Party Calling Party Transformation CSS Use Device Pool Calling Party Calling Party Selection* Calling Line ID Presentation*	CSS < None > Party Transformation None > Transformation CSS None > Transformation CSS Originator Default	Strip Digits 0 CSS	< None >	Default Prefix Settings Calling Search Space	<u> </u>	
Incoming Number Defau Connected Party Settings— Connected Party Transformation Use Device Pool Connected for Use Device Pool Connected Party Transformation CSS Use Device Pool Called Party Calling Party Transformation CSS Use Device Pool Calling Party Calling Party Selection* Calling Line ID Presentation* Calling Name Presentation*	CSS < None > Party Transformation None > Transformation CSS None > Transformation CSS Originator Default Default	Strip Digits 0 CSS	< None >	Default Prefix Settings Calling Search Space	<u>v</u>	



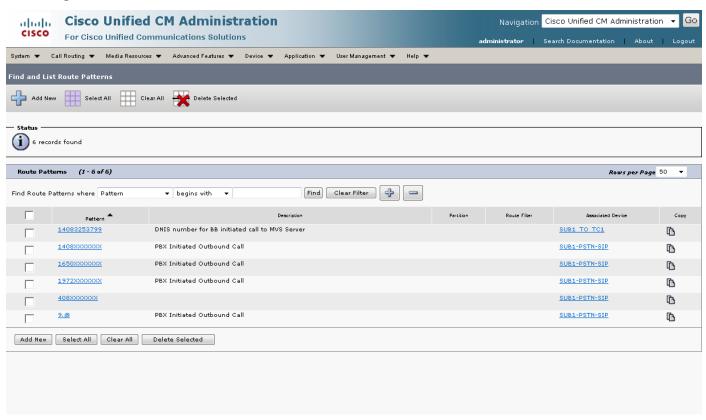
CUCM SIP Trunk Detail Configuration (3 of 3)

- SIP Information				
Destination Address	172.20.89.251			
Destination Address IPv6				
Destination Address is an SRV				
Destination Port*	6060			
MTP Preferred Originating Codec*	711ulaw	w.		
Presence Group*	Standard Presence group	v		
SIP Trunk Security Profile*	MVS SIP Security Profile	•		
Rerouting Calling Search Space	< None >	•		
Out-Of-Dialog Refer Calling Search Space	< None >	•		
SUBSCRIBE Calling Search Space	< None >	•		
SIP Profile*	Standard SIP Profile	•		
DTMF Signaling Method*	OOB and RFC 2833	•		
Geolocation Configuration Geolocation < None >	w)			
Geolocation Filter < None >				
Send Geolocation Information				
Send Geolocation Information				
Save Delete Reset Apply Co	onfig Add New			
дене высе пере	and the state of t			
i *- indicates required item.				
i **- Device reset is not required for	changes to Packet Capture Mode and Packe	t Capture Duration.		



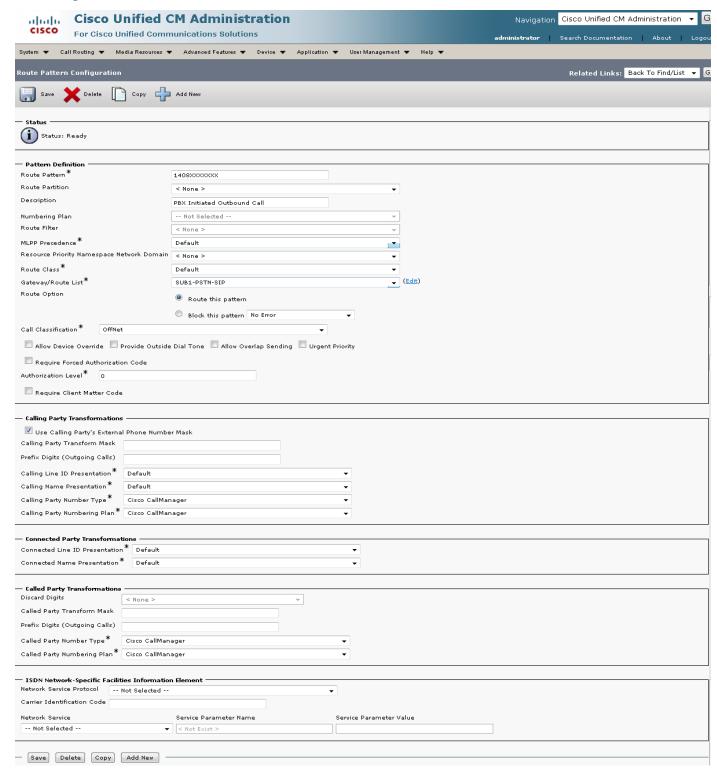
Call Route Pattern

Call Routing -> Route/Hunt-> Route Pattern





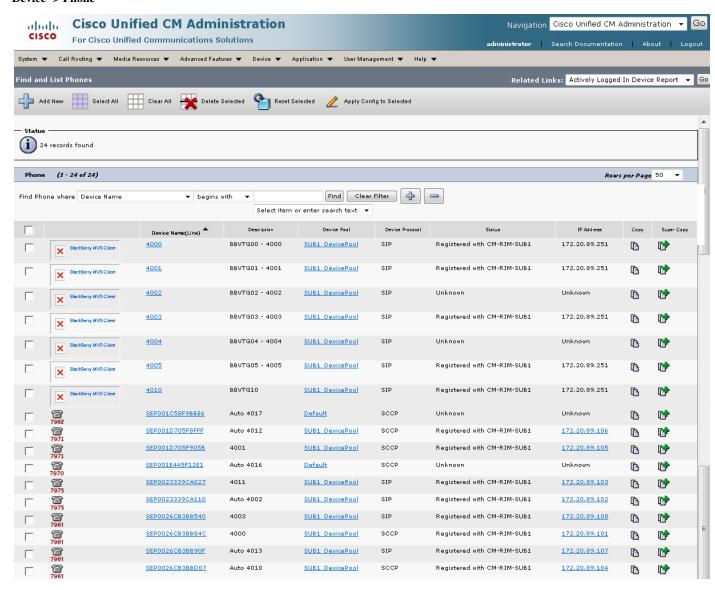
Call Routing -> Route/Hunt-> Route Pattern Detail





Cisco IP Phones and BlackBerry Devices

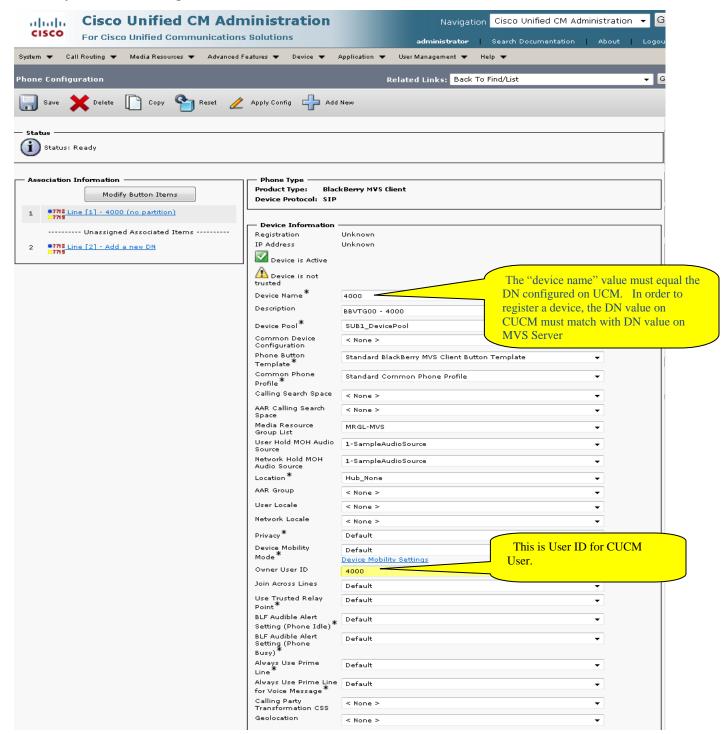
Device -> Phone





BlackBerry MVS Client

BlackBerry® MVS Client Configurations (1 of 2)



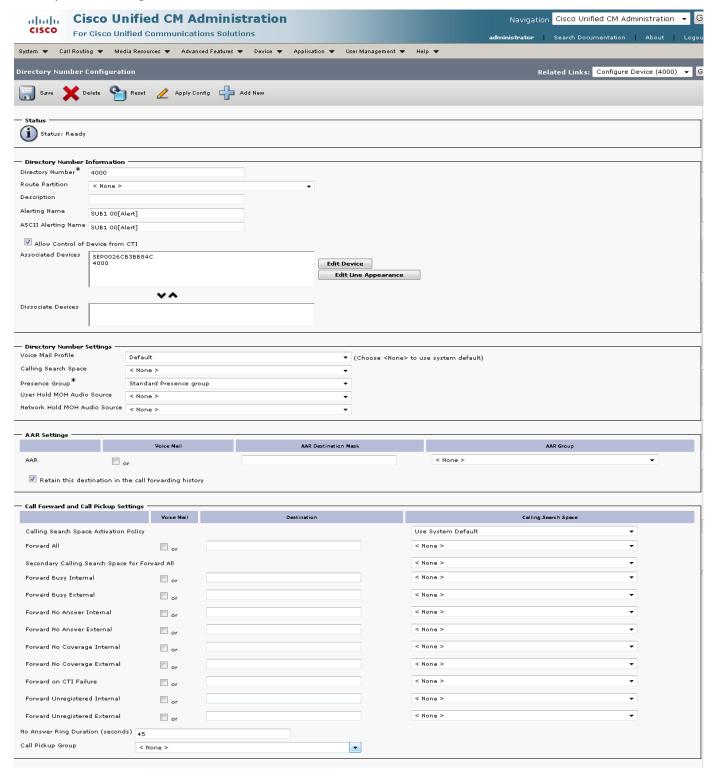


BlackBerry MVS Client Configurations (2 of 2)

	■ Use Device Pool Calling Party -	Transformation CSS	
	Retry Video Call as Audio		
	Ignore Presentation Indicators	(internal calls only)	
	☑ Logged Into Hunt Group		
	Remote Device		
	Protocol Specific Information —		
	Packet Capture Mode *	None	•
	Packet Capture Duration	0	
	Presence Group *	Standard Presence group	•
	MTP Preferred Originating Codec*	711ulaw	₩
	Device Security Profile*	BlackBerry MVS Client - Standard SIP Non-Secure Profile	•
	Rerouting Calling Search Space	< None >	▼
	SUBSCRIBE Calling Search Space	< None >	▼
	SIP Profile *	Standard SIP Profile	▼
	Digest User	< None >	•
	Media Termination Point Requ	ired	
	Unattended Port		
	Require DTMF Reception		
	MLPP Information		
	MLPP Domain < None >	•	
	MLPP Indication * Off	▼	
		,	
	Do Not Disturb		
	Do Not Disturb		
	DND Option * Ringer O	ff 🔻	
	DND Incoming Call Alert < None >		
- Save Delete Copy Reset Apply Config	Add New		
*- indicates required item.			
**- Device reset is not required for changes to Packet			
0		Duration.	
***Note: Security Profile Contains Addition CAPF Set	ttings.		
****Note: A new Softkey template without supplem	entary service Softkeys must be used	for a protected phone.	
*****Note: A custom Softkey template without sup	oplementary service Softkeys must be	used for a Hot line Device.	



Directory Number Configuration (1 of 2)





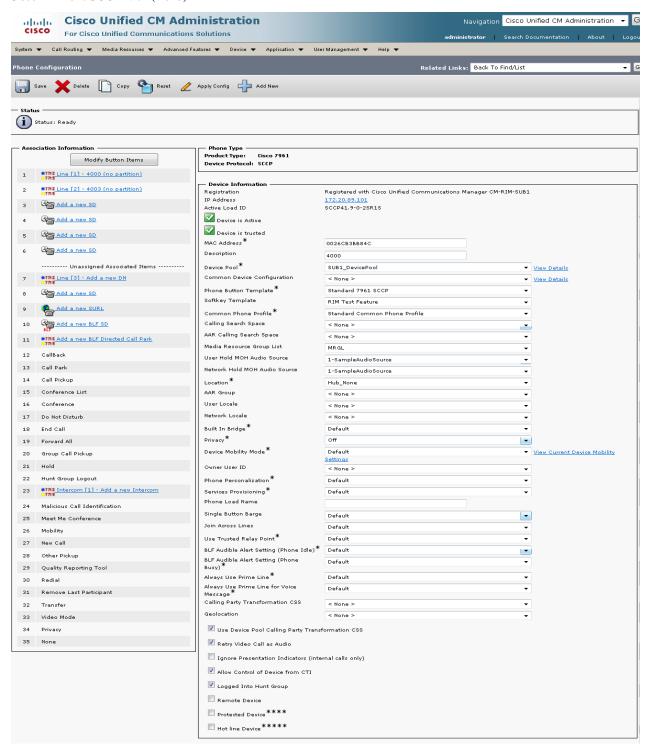
Directory Number Configuration (2 of 2)

Park Monitoring		Voice		Destination			Calling Search Space	
Park Monitoring Forward No Ret Destination External	rieve	E or				< None > the parker's line.		▼ A blank value means to call
Park Monitoring Forward No Ret Destination Internal	trieve	OF				< None > the parker's line.		▼ A blank value means to call
Park Monitoring Reversion Timer				A blank value	e vill use valu	e set in Park Monitoring Reversi	on Timer service parameter	
MLPP Alternate Party Settings								
Farget (Destination)								
MLPP Calling Search Space MLPP No Answer Ring Duration (s		ione >				•		
Line Settings for All Devices —								
fold Reversion Ring Duration (se				8	Setting the H	old Reversion Ring Duration to z	ero vill disable the feature	
Hold Reversion Notification Inter	val (seconds)			8	Setting the H	old Reversion Notification Interv	al to zero will disable the feature	
Party Entrance Tone *		Default				•		
Line 1 on Device 4000								
					Value			Update Shared Device Settings
Display (Internal 88 00 Caller ID) director	y number for	internal ca	ills. If you specify			is intended for displaying text so all may not see the proper ident		
ASCII Display 88 00 (Internal Caller ID)								
								_
External Phone 408325 Number Mask	53780							
Monitoring Calling < Non-	e >							
Search Space								
								Propagate Selected
						busy trigger value m		
Multiple Call/Call Waiting Settin lote:The range to select the Mar			10			MVS devices. If this narios of move call or	,	
Maximum Number of Calls*			4			(Less than or equal to Max. Calls	0	
			-			ters out or aden to trace out	·	
Forwarded Call Information Di	splay on Devi	ce 4000 -						
Caller Name								
Caller Number								
Redirected Number								
Dialed Number								
Users Associated with Line —								
Ad	sociate End U	sers						
Save Delete Reset	Apply Confi	g (A	id Nev					
*								
*- indicates required item								
**- Changes to Line or I	Pirectory Numb	er setting	s require restart.					



Cisco IP Phone

Cisco IP Phone SCCP 7961 (1 of 3)





Cisco IP Phone 7961 SCCP Configuration (2 of 3)

	D-110	
	- Protocol Specific Information -	
	Packet Capture Mode *	None ▼
	Packet Capture Duration	0
	Presence Group *	Standard Presence group ▼
	Device Security Profile*	Cisco 7961 - Standard SCCP Non-Secure Profile ▼
	SUBSCRIBE Calling Search Space	< None >
	Unattended Port	
	Require DTMF Reception	
	RFC2833 Disabled	
L		
	- Certification Authority Proxy F	unction (CAPF) Information —
	Certificate Operation * No	Pending Operation
	Authentication Mode * By	Null String
	Authentication String	
	Generate String	
	Key Size (Bits)* 10	24
	Operation Completes By 20:	
	Certificate Operation Status: Non	
	Note: Security Profile Contains Ac	
L		
Г	Expansion Module Information	
	Module 1 < None >	▼
	Module 1 Load Name	
	Module 2 < None >	▼
	Module 2 Load Name	
L		
	External Data Locations Inform	nation (Leave blank to use default)
	Information	
	Directory	
	Messages	
	Services	
	Authentication Server	
	Proxy Server	
	Idle	
	Idle Timer (seconds)	
	Secure Authentication URL	
	Secure Directory URL	
	Secure Idle URL	
	Secure Information URL	
	Secure Messages URL	
	Secure Services URL	
	Extension Information	
	Enable Extension Mobility	
	Log Out Profile Use Current E	evice Settings ▼
	Log in Time < None > Log out Time < None >	



Cisco IP Phone 7961 SCCP Configuration (3 of 3)

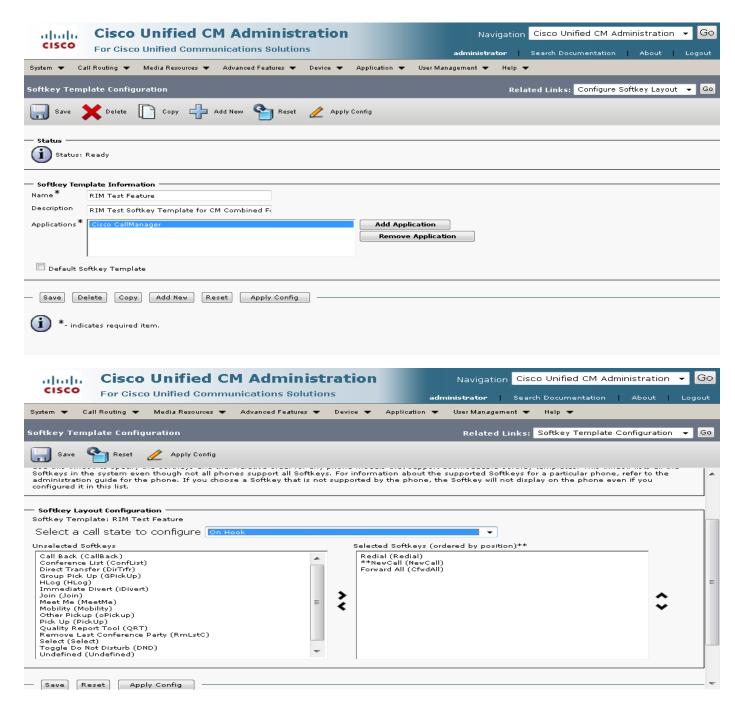
MLPP Information		
MLPP Domain < None >	▼	
MLPP Indication * Default		
MLPP Preemption * Default	<u> </u>	
Do Not Disturb		
Do Not Disturb		
DND Option * Ringer Off	-	
DND Incoming Call Alert < None >		
s none >	•	
Cc #LII YE #L		
Secure Shell Information Secure Shell User		
Secure Shell Password		
October Shell Passacia		
Product Specific Configuration Layout		
	Param	Override Common Settings
Disable Speakerphone		
Disable Speakerphone and Headset		
Forwarding Delay*	Disabled ▼	
PC Port *	Enabled ▼	
Settings Access*	Enabled ▼	
Gratuitous ARP*	Disabled ▼	
PC Voice VLAN Access*	Enabled	
Video Capabilities*	Disabled •	
Auto Line Select*	Disabled ▼	
Web Access*	Disabled ▼	
Span to PC Port*	Disabled ▼	
Logging Display*	PC Controlled ▼	
Load Server		
Recording Tone *		
	Disabled ▼	
Recording Tone Local Volume*	100	
Recording Tone Remote Volume *	50	
Recording Tone Duration		
RTCP*	Disabled ▼	
"more" Soft Key Timer	5	_
Auto Call Select*		
Auto Call Select " Log Server	Enabled •	
Advertise G.722 Codec*	Use System Default	
Wideband Headset UI Control*	Enabled ▼	
Wideband Handset UI Control*	Enabled ▼	
Wideband Headset*	Enabled ▼	
Wideband Handset*	Use Phone Default	
Peer Firmware Sharing *	Disabled 🔻	
Cisco Discovery Protocol (CDP): Switch Port*		
Cisco Discovery Protocol (CDP): PC Port*	Enabled ▼	
Link Layer Discovery Protocol - Media Endpoint Discover (LLDP-MED):	Enabled ▼	
Switch Port*		
Link Layer Discovery Protocol (LLDP): PC Port*	Enabled ▼	
LLDP Asset ID		
LLDP Power Priority*	Unknown ▼	
Display Refresh Rate*	Normal	
IPv6 Load Server		
IPv6 Log Server		
802.1× Authentication *	User Controlled ▼	
Detect Unified CM Connection Failure *	Normal ▼	
Minimum Ring Volume*	0-Silent ▼	
Headset Sidetone Level	Use Phone Default ▼	
HTTPS Server*	http and https Enabled	
Enbloc Dialing *	Enabled •	
Elibiot Dialing	- Trabled	
Save Delete Copy Reset Apply Config Add New * Indicates required item. ** Device reset is not required for changes to Packet Capture Mode and Packet Capture Duration.		
****Note: Security Profile Contains Addition CAPF Settings.		
*****Note: A new Softkey template without supplementary service Softkeys must be used for a protected phone.		
******Note: A custom Softkey template without supplementary service Softkeys must be used for a Hot line Device.		



Phone Button Key Template

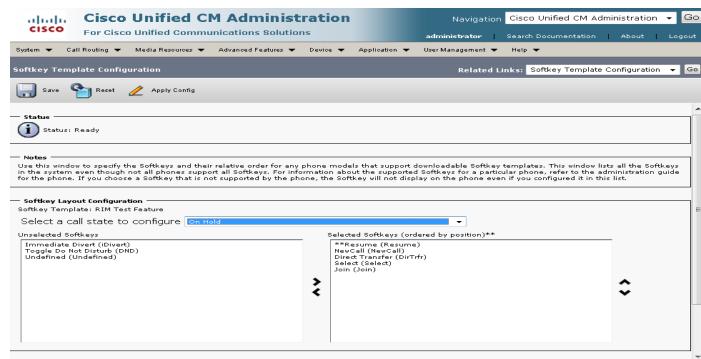
Phone Button Key Template (1 of 4)

Device -> Device Settings-> Phone Button Template





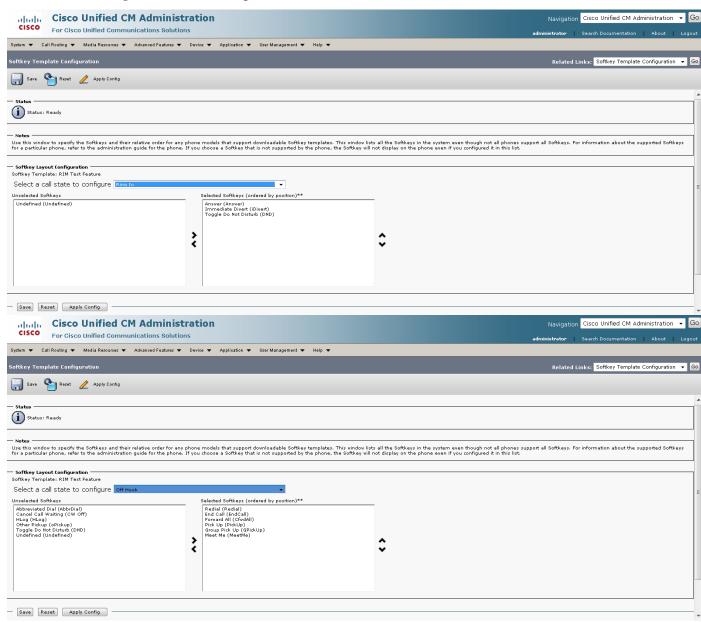
Phone Button Key Template (2 of 4) Device -> Device Settings-> Phone Button Template Cisco Unified CM Administration Navigation Cisco Unified CM Administration 🔻 arliarlia CISCO **For Cisco Unified Communications Solutions** Search Documentation System ▼ Call Routing ▼ Media Resources ▼ Advanced Features ▼ Device ▼ Application ▼ User Management 🔻 Help ▼ Softkey Template Configuration Related Links: Softkey Template Configuration Save Reset 🧷 Apply Config i Status: Ready Use this window to specify the Softkeys and their relative order for any phone models that support downloadable Softkey templates. This window lists all the Softkeys in the system even though not all phones support all Softkeys. For information about the supported Softkeys for a particular phone, refer to the administration guide for the phone. If you choose a Softkey that is not supported by the phone, the Softkey will not display on the phone even if you configured it in this list. Softkey Layout Configuration
Softkey Template: RIM Test Feature Select a call state to configure Conne Unselected Softkeys Selected Softkeys (ordered by position)** Selected Softkeys (ordered by position)**
Hold (Hold)
End Call (EndCall)
Transfer (Trmsfer)
Park (Park)
Conference (Confrn)
Conference List (ConfList)
Select (Select)
Join (Join)
Direct Transfer (DirTrfr)
Toggle Malicious Call Trace (MCID)
Remove Last Conference Party (RmLstC)
Video Mode Command (VidMode)
Immediate Divert (iDivert)
Mobility (Mobility) HLog (HLog) Quality Report Tool (QRT) Record (Record) Toggle Do Not Disturb (DND) Undefined (Undefined) Cisco Unified CM Administration Navigation Cisco Unified CM Administration ▼





Phone Button Key Template (3 of 4)

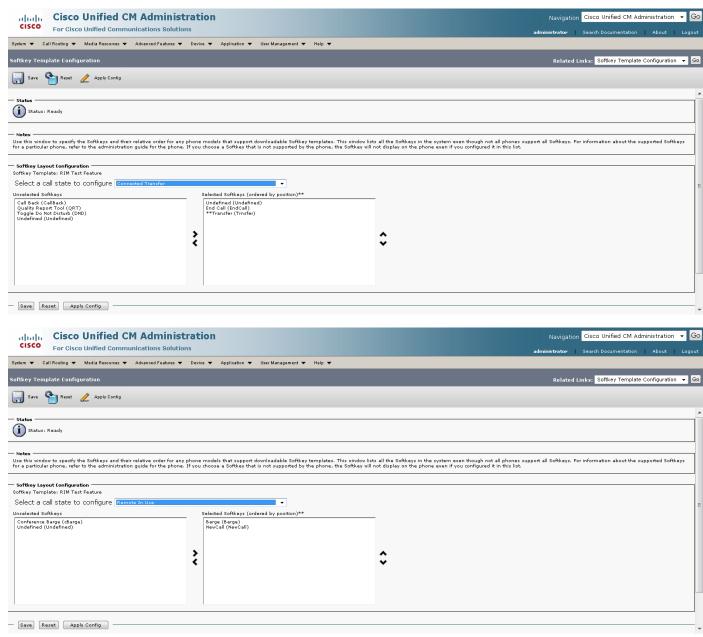
Device -> Device Settings-> Phone Button Template





Phone Button Key Template (4 of 4)

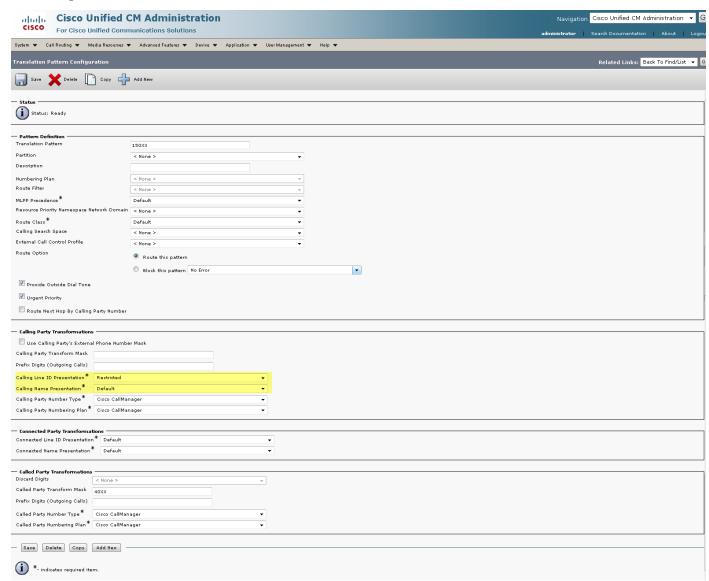
Device -> Device Settings-> Phone Button Template





Calling line name and number restriction configuration

Call Routing->Translation Pattern





Cisco Voice Gateway Configuration

Gateway Version

Cisco IOS Software, 2800 Software (C2800NM-IPVOICE-M), Version 15.0(1)M2, RELEASE SOFTWARE (fc2)

Technical Support: http://www.cisco.com/techsupport Copyright © 1986-2010 by Cisco Systems, Inc. Compiled Wed 10-Mar-10 22:59 by prod_rel_team

ROM: System Bootstrap, Version 12.4(1r) [hqluong 1r], RELEASE SOFTWARE (fc1)

c2811_MGCP uptime is 5 days, 12 hours, 19 minutes System returned to ROM by power-on System image file is "flash:c2800nm-ipvoice-mz.150-1.M2.bin" Last reload type: Normal Reload

Cisco 2811 (revision 53.50) with 247808K/14336K bytes of memory.

Processor board ID FHK0946F0MZ

2 FastEthernet interfaces

48 Serial interfaces

3 Channelized T1/PRI ports

2 Voice FXS interfaces

DRAM configuration is 64 bits wide with parity enabled.

239K bytes of non-volatile configuration memory.

62592K bytes of ATA CompactFlash (Read/Write)

Device# PID SN

*0 CISCO2811

FHK0946F0MZ

Configuration register is 0x2

Gateway Configurations

```
version 12.4
service timestamps debug datetime msec
service timestamps log datetime msec
no service password-encryption
hostname c2811_MGCP
boot-start-marker
boot-end-marker
logging message-counter syslog
enable secret 5 $1$G.N2$g/Nbhv3IsTFL6DahIL4aQ0
no aaa new-model
network-clock-participate slot 1
network-clock-participate wic 0
network-clock-select 1 T1 1/0/1
voice-card 0
voice-card 1
dspfarm
```



```
ip source-route
!
!
ip cef
!
ip host cm-rim-sub1 172.25.67.202
ip name-server 172.20.89.253
no ipv6 cef
multilink bundle-name authenticated
isdn switch-type primary-4ess
username 86isco privilege 15 secret 5 $1$KgEv$25ls/LbFvOR4joo0xKVvh1
archive
log config
hidekeys
!
controller T1 0/0/0
pri-group timeslots 1-24 service mgcp
controller T1 1/0/0
controller T1 1/0/1
pri-group timeslots 1-24 service mgcp
!
interface Tunnel0
ip address 192.168.1.1 255.255.255.0
tunnel source FastEthernet0/0
tunnel destination 172.24.112.14
interface FastEthernet0/0
ip address 172.27.28.50 255.255.255.224
duplex auto
speed auto
interface FastEthernet0/1
no ip address
shutdown
duplex auto
speed auto
interface Serial0/0/0:23
no ip address
encapsulation hdlc
isdn switch-type primary-ni
isdn incoming-voice voice
isdn bind-13 ccm-manager
no cdp enable
interface Serial1/0/1:23
no ip address
encapsulation hdlc
```



```
isdn switch-type primary-4ess
isdn incoming-voice voice
isdn bind-13 ccm-manager
no cdp enable
ip forward-protocol nd
ip route 0.0.0.0 0.0.0.0 172.27.28.33
ip route 172.20.89.0 255.255.255.0 Tunnel0
no ip http server
control-plane
voice-port 0/0/0:23
echo-cancel coverage 64
voice-port 0/1/0
ring cadence pattern01
voice-port 0/1/1
voice-port 1/0/1:23
ccm-manager mgcp
no ccm-manager fax protocol 87isco
ccm-manager music-on-hold
ccm-manager config server cm-rim-sub1
ccm-manager config
mgcp
mgcp call-agent CM-RIM-SUB1 2427 service-type mgcp version 0.1
mgcp dtmf-relay voip codec all mode out-of-band
mgcp rtp unreachable timeout 1000 action notify
mgcp modem passthrough voip mode nse
mgcp package-capability rtp-package
mgcp package-capability sst-package
mgcp package-capability pre-package
no mgcp package-capability res-package
no mgcp timer receive-rtcp
mgcp sdp simple
mgcp fax t38 ecm
mgcp fax t38 inhibit
mgcp profile default
dial-peer voice 999011 pots
service mgcpapp
port 0/1/1
dial-peer voice 100 pots
service mgcpapp
port 0/0/0:23
```

dial-peer voice 999010 pots

DTMF Relay must be set to "Out of Band" for better interoperability; otherwise, user will hear DTMF tones during call setup.



```
service mgcpapp
port 0/1/0
!
!
line con 0
line aux 0
line vty 0 4
password 88isco
login
!
scheduler allocate 20000 1000
end
```

Gateway Chassis Info

Slot 0

C2811 Motherboard with 2FE and integrated VPN Port adapter, 2 ports

Port adapter is analyzed

Port adapter insertion time 5d12h ago EEPROM contents at hardware discovery: PCB Serial Number : FOC09412RJX

Hardware Revision : 1.0

Top Assy. Part Number : 800-21849-03

Board Revision : A0
Deviation Number : 0
Fab Version : 03
RMA Test History : 00
RMA Number : 0-0-0-0
RMA History : 00
Processor type : 87

Hardware date code : 20051019 Chassis Serial Number : FHK0946F0MZ Chassis MAC Address : 0015.63c3.4f60

MAC Address block size : 24

CLEI Code : CNMJ7N0BRB Product (FRU) Number : CISCO2811 Part Number : 73-10258-03

Version Identifier : V02 EEPROM format version 4 EEPROM contents (hex):

0x00: 04 FF C1 8B 46 4F 43 30 39 34 31 32 52 4A 58 40 0x10: 03 E7 41 01 00 C0 46 03 20 00 55 59 03 42 41 30 0x20: 88 00 00 00 00 02 03 03 00 81 00 00 00 00 04 00 0x30: 09 87 83 01 31 F4 4B C2 8B 46 48 4B 30 39 34 36 0x40: 46 30 4D 5A C3 06 00 15 63 C3 4F 60 43 00 18 C6 0x50: 8A 43 4E 4D 4A 37 4E 30 42 52 42 CB 8F 43 49 53 0x60: 43 4F 32 38 31 31 20 20 20 20 20 20 82 49 28 12 0x70: 03 89 56 30 32 20 D9 02 40 C1 FF FF FF FF FF

PVDM Slot 0:

48-channel (G.711) Voice/Fax PVDMII DSP SIMM PVDM daughter card

Hardware Revision
Part Number : 2.0
Board Revision : 02
Deviation Number : 0-0
Fab Version : 02

PCB Serial Number : FOC08030CCP

RMA Test History : 00 RMA Number : 0-0-0-0



RMA History : 00 Processor type : 00 Product (FRU) Number : PVDM2-T1-DSP-TRIO Version Identifier : NA : TDB CLEI Code EEPROM format version 4 EEPROM contents (hex): 0x00: 04 FF 40 03 ED 41 02 00 82 49 22 90 02 42 30 32 0x10: 80 00 00 00 00 02 02 C1 8B 46 4F 43 30 38 30 33 0x20: 30 43 43 50 03 00 81 00 00 00 00 04 00 09 00 CB 0x30: 91 50 56 44 4D 32 2D 54 31 2D 44 53 50 2D 54 52 0x40: 49 4F 89 4E 41 00 E6 C6 8A 54 44 42 00 00 00 00 0x50: 00 00 00 D9 02 40 C1 FF FF FF FF FF FF FF FF FF WIC Slot 0: T1 (1 port) Multi-Flex Trunk WAN daughter card Hardware revision 1.0 Board revision B0 Serial number 28253554 Part number 800-03568-03 FRU Part Number VWIC-1MFT-T1= RMA number 00-00-00 Test history 0x0Connector type PCI EEPROM format version 1 EEPROM contents (hex): 0x20: 01 20 01 00 01 AF 1D 72 50 0D F0 03 00 00 00 00 0x30: 58 00 00 00 02 07 15 00 FF FF FF FF FF FF FF FF WIC Slot 1: 2nd generation – FXS Voice daughter card (2 port) Hardware Revision : 3.1 Top Assy. Part Number : 800-21341-01 **Board Revision** : E0 **Deviation Number** : 0-0 Fab Version : 01 PCB Serial Number : FOC083928O2 RMA Test History : 00 RMA Number : 0-0-0-0 **RMA History** :00 Version Identifier : V Product (FRU) Number : VIC2-2FXS= EEPROM format version 4 EEPROM contents (hex): 0x00: 04 FF 40 00 47 41 03 01 C0 46 03 20 00 53 5D 01 0x10: 42 45 30 80 00 00 00 00 02 01 C1 8B 46 4F 43 30 0x20: 38 33 39 32 38 51 32 03 00 81 00 00 00 00 04 00 High Density Voice Port adapter

Slot 1:

High Density Voice Port adapter
Port adapter is analyzed
Port adapter insertion time 5d12h ago
EEPROM contents at hardware discovery:

Hardware Revision : 1.1

Top Assy. Part Number : 800-03567-01



Board Revision : H0
Deviation Number : 0-0
Fab Version : 02

PCB Serial Number : JAE0820HALA

RMA Test History : 00 RMA Number : 0-0-0-0 RMA History : 00

Product (FRU) Number : NM-HDV=

EEPROM format version 4 EEPROM contents (hex):

HDV SIMMs: Product (FRU) Number: PVDM-12=

SIMM slot 0: PVDM-12 SIMM present. SIMM slot 1: PVDM-12 SIMM present. SIMM slot 2: PVDM-12 SIMM present. SIMM slot 3: PVDM-12 SIMM present. SIMM slot 4: PVDM-12 SIMM present.

WIC Slot 0:

T1 (2 Port) Multi-Flex Trunk (Drop&Insert) WAN Daughter Card

Hardware revision 1.0 Board revision D0

Serial number 32051319 Part number 800-04614-03

FRU Part Number VWIC-2MFT-T1-DI=

Test history 0x0 RMA number 00-00-00

Connector type PCI EEPROM format version 1 EEPROM contents (hex):

0x20: 01 24 01 00 01 E9 10 77 50 12 06 03 00 00 00 00 0x30: 68 00 00 04 04 22 00 FF FF FF FF FF FF FF FF FF

HDV firmware: Compiled Fri 19-Nov-04 14:23 by michen

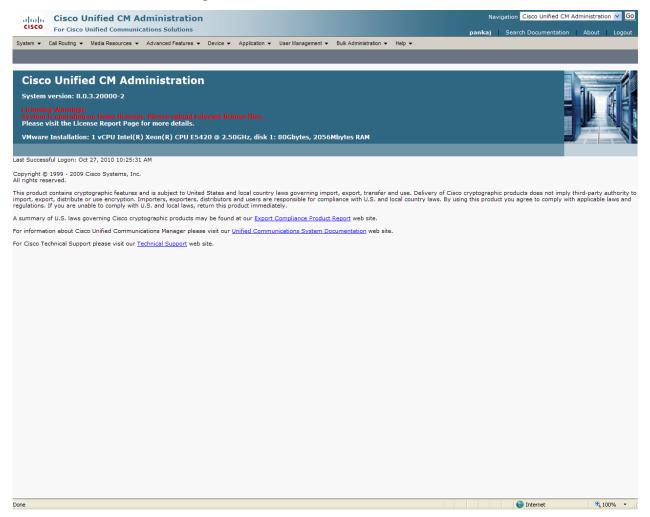
HDV memory size 524280 heap free 167869



Configuring the Cisco Unified Communications Manager - H323 and SIP trunking configurations

This section is a supplement to the previous Environments and Configurations section. This section contains configuration menus and commands and describes configuration sequences and tasks. This section contains screenshots of the content that would be different for configuring on H323 and SIP. The previous section on MGCP will not be duplicated below.

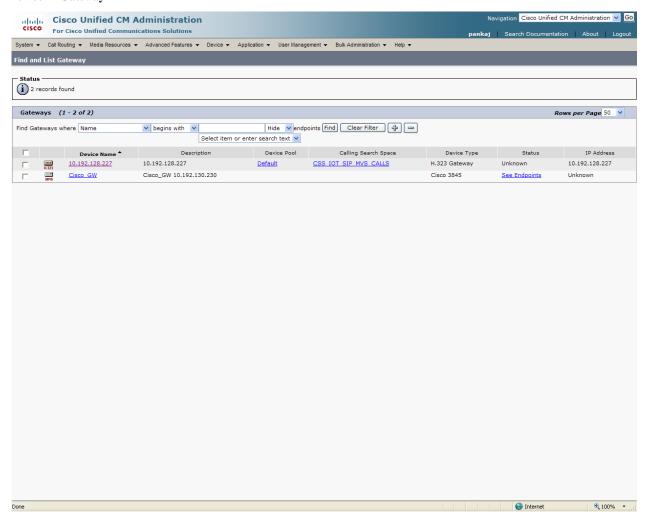
Cisco Unified Communications Manager Version





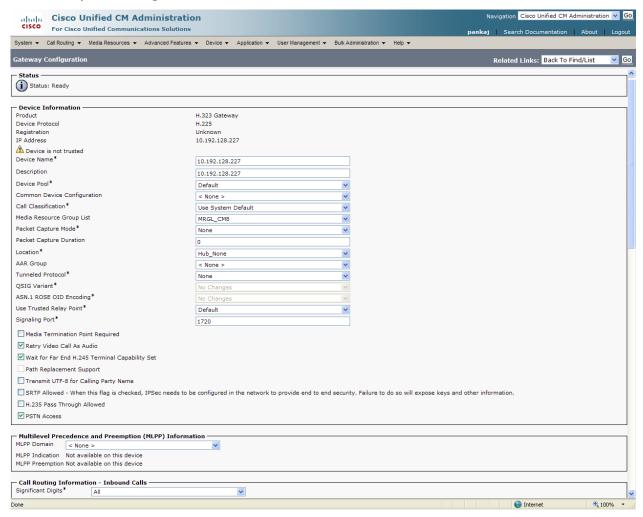
H.323 Gateway

Device -> Gateway



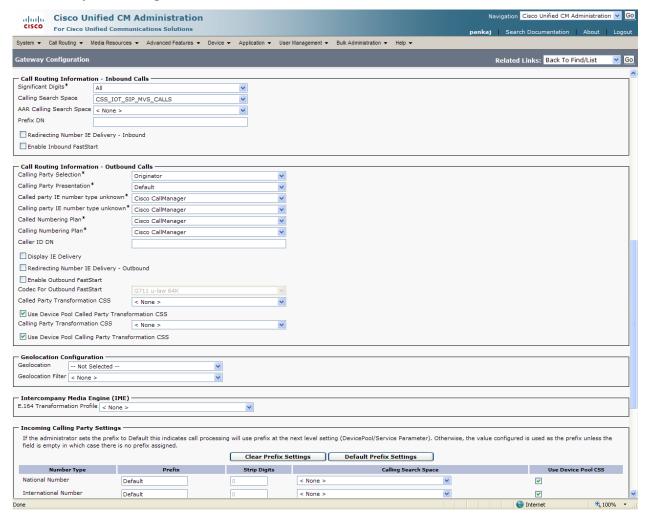


H.323 Gateway Detail Configurations (1 of 3)



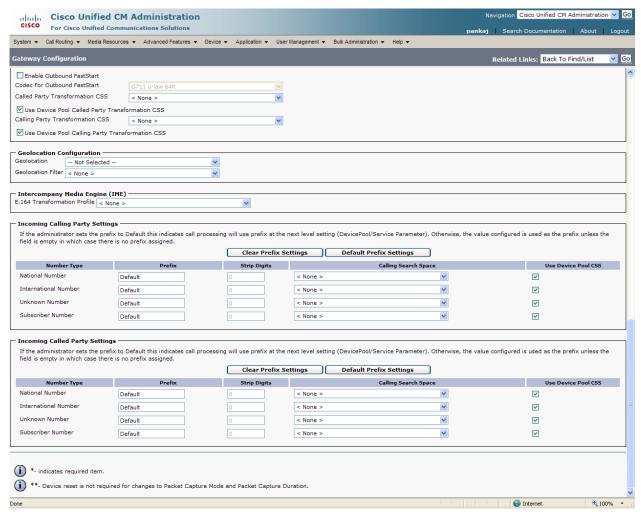


H.323 Gateway Detail Configurations (2 of 3)



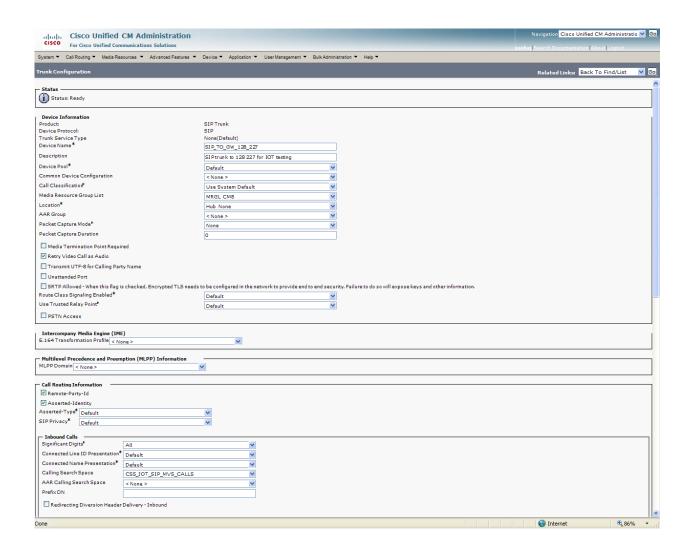


H.323 Gateway Detail information (3 of 3)



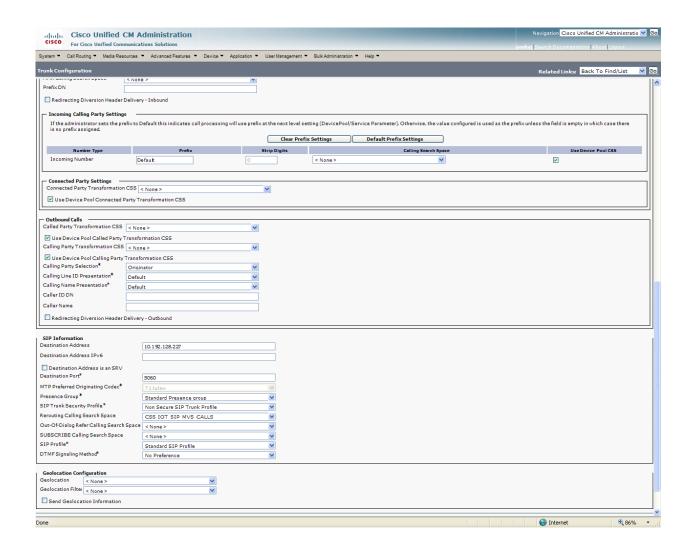


CUCM SIP Trunk to Gateway Detail Configuration (1 of 2)



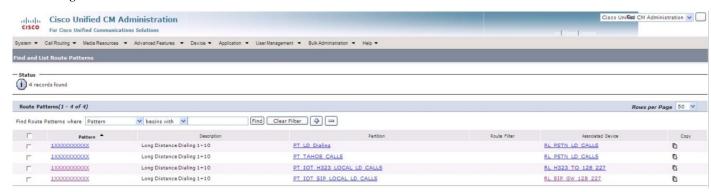
CUCM SIP Trunk to Gateway Detail Configuration (2 of 2)







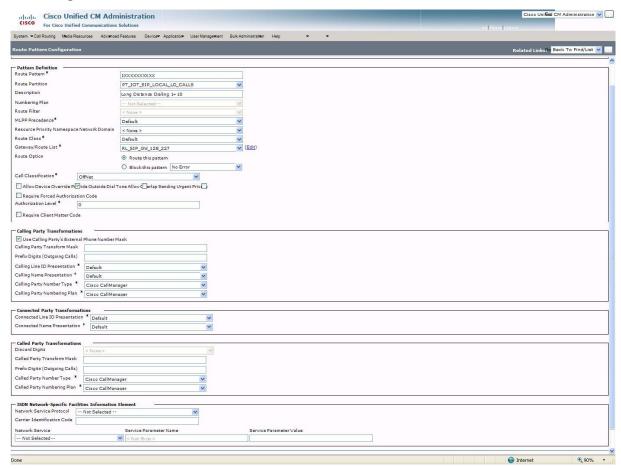
Call Routing -> Route/Hunt-> Route Pattern





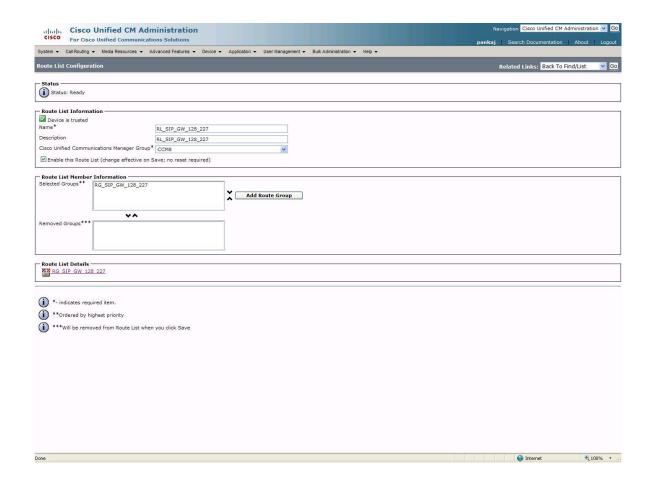
SIP Route Pattern Details

Call Routing -> Route/Hunt-> Route Pattern Detail



SIP Route List

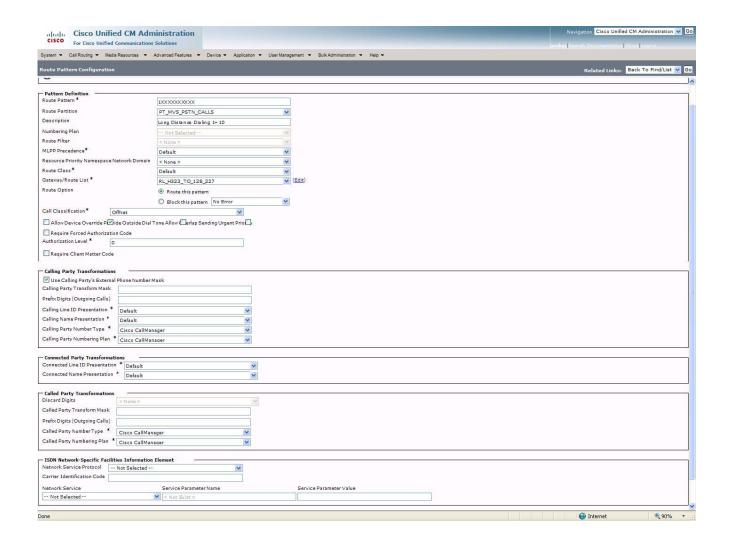




H.323 Route Pattern Details

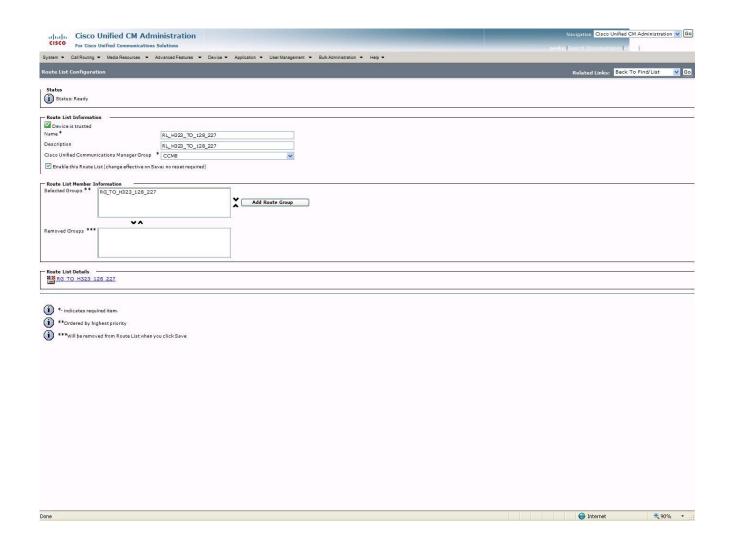
 $Call\ Routing \ \hbox{->}\ Route/Hunt-\hbox{>}\ Route\ Pattern\ Detail$





H.323 Route List







Cisco Voice IOS Gateway Configuration

5.0 SIP - IOS Gateway Configurations

```
version 15.0
service timestamps debug datetime msec
service timestamps log datetime msec
no service password-encryption
hostname Cisco_2811_GW
boot-start-marker
boot-end-marker
logging buffered 51200 warnings
enable secret 5 $1$dLRq$JPysM6Xks82mXRj6xJvxL/
no aaa new-model
no network-clock-participate slot 1
network-clock-participate wic 2
network-clock-select 1 T1 0/2/0
dot11 syslog
ip source-route
ip cef
no ipv6 cef
multilink bundle-name authenticated
isdn switch-type primary-ni
voice call send-alert
voice rtp send-recv
voice service voip
allow-connections h323 to h323
allow-connections h323 to sip
allow-connections sip to h323
allow-connections sip to sip
h323
 ip circuit max-calls 64
 ip circuit default only
modem passthrough nse codec g711ulaw
sip
voice class codec 1
codec preference 1 g711ulaw
```



```
voice class h323 1
 h225 timeout tcp establish 3
 telephony-service ccm-compatible
 ccm-compatible
voice translation-rule 1
rule 1 /849/ /49/
voice translation-profile Profile1
translate called 1
voice-card 0
dspfarm
dsp services dspfarm
voice-card 1
license udi pid CISCO2811 sn FTX1124A18N
archive
log config
hidekeys
username cisco privilege 15 secret 5 $1$lirB$WgDFb9CtxJ1zd2qrofWRV1
redundancy
controller T1 0/2/0
pri-group timeslots 1-24
description PSTN PRI DIDs 650-264-5xxx incoming and outgoing
controller T1 0/2/1
pri-group timeslots 1-24
interface FastEthernet0/0
description $ETH-LAN$$ETH-SW-LAUNCH$$INTF-INFO-FE 0/0$
ip address 10.192.128.227 255.255.255.0
duplex full
speed 100
no mop enabled
interface FastEthernet0/1
no ip address
```



```
shutdown
duplex auto
speed auto
interface FastEthernet0/1/0
shutdown
interface FastEthernet0/1/1
shutdown
interface FastEthernet0/1/2
shutdown
interface FastEthernet0/1/3
shutdown
interface FastEthernet0/1/4
shutdown
interface FastEthernet0/1/5
shutdown
interface FastEthernet0/1/6
shutdown
interface FastEthernet0/1/7
shutdown
interface FastEthernet0/1/8
shutdown
interface Serial0/2/0:23
no ip address
encapsulation hdlc
isdn switch-type primary-ni
isdn protocol-emulate network
isdn incoming-voice voice
no cdp enable
interface Serial0/2/1:23
no ip address
encapsulation hdlc
isdn switch-type primary-ni
isdn incoming-voice voice
no cdp enable
!
interface Vlan1
```



```
no ip address
shutdown
!
no ip forward-protocol nd
ip http server
ip http access-class 23
ip http authentication local
no ip http secure-server
ip http timeout-policy idle 60 life 86400 requests 10000
ip route 0.0.0.0 0.0.0.0 10.192.128.1
no cdp run
control-plane
voice-port 0/2/0:23
voice-port 0/2/1:23
voice-port 0/3/0
voice-port 0/3/1
voice-port 0/3/2
voice-port 0/3/3
voice-port 1/0/0
voice-port 1/0/1
ccm-manager redundant-host 10.192.128.221
ccm-manager mgcp
no ccm-manager fax protocol cisco
ccm-manager music-on-hold
ccm-manager config server 10.192.128.221
ccm-manager config
sccp local FastEthernet0/0
sccp ccm 10.192.128.222 identifier 1 version 5.0.1
sccp ip precedence 3
sccp
ļ
sccp ccm group 992
associate ccm 1 priority 1
associate profile 1 register HWMTPLAB
dspfarm profile 1 mtp
```



```
description MTP DSP resources
codec g711ulaw
maximum sessions hardware 1
maximum sessions software 500
associate application SCCP
dial-peer voice 600 pots
description Dial Peer for Local Calls to PSTN
destination-pattern 650[2-9]......
incoming called-number.
port 0/2/1:23
forward-digits all
dial-peer voice 602 pots
description Dial Peer for National Long Distance
destination-pattern 1[2-9]......
direct-inward-dial
port 0/2/1:23
prefix 1
dial-peer voice 700 pots
incoming called-number 4...
direct-inward-dial
dial-peer voice 2001 voip
description Calls to GW 130 230 fro IOT
destination-pattern 484.
session protocol sipv2
session target ipv4:10.192.131.227
session transport udp
voice-class codec 1
dtmf-relay sip-kpml sip-notify
no vad
dial-peer voice 1 voip
session protocol sipv2
session target sip-server
incoming called-number .%
voice-class codec 1
dtmf-relay sip-kpml
gateway
timer receive-rtp 1200
sip-server ipv4:10.192.128.227
banner login
```

Cisco Router and Security Device Manager (SDM) is installed on this device.

This feature requires the one-time use of the username "cisco"

with the password "cisco". The default username and password have a privilege level of 15.

Please change these publicly known initial credentials using SDM or the IOS CLI.

Here are the Cisco IOS commands.



end

username <myuser> privilege 15 secret 0 <mypassword> no username cisco

Replace <myuser> and <mypassword> with the username and password you want to use $% \left(1\right) =\left(1\right) \left(1\right) +\left(1\right) \left(1\right) \left(1\right) +\left(1\right) \left(1\right$

For more information about SDM please follow the instructions in the QUICK START

GUIDE for your router or go to http://www.cisco.com/go/sdm

^C ! line con 0 login local line aux 0 line vty 0 4 exec-timeout 0 0 logout-warning 0 login local ! scheduler allocate 20000 1000



6.0 H323 - IOS Gateway Configuration

```
version 15.0
service timestamps debug datetime msec
service timestamps log datetime msec
no service password-encryption
hostname Cisco_2811_GW
boot-start-marker
boot-end-marker
logging buffered 51200 warnings
enable secret 5 $1$dLRq$JPysM6Xks82mXRj6xJvxL/
enable password switch4u
no aaa new-model
no network-clock-participate slot 1
network-clock-participate wic 2
network-clock-select 1 T1 0/2/0
dot11 syslog
ip source-route
ip cef
no ipv6 cef
multilink bundle-name authenticated
isdn switch-type primary-ni
voice call send-alert
voice rtp send-recv
voice service voip
allow-connections h323 to h323
allow-connections h323 to sip
allow-connections sip to h323
allow-connections sip to sip
h323
ip circuit max-calls 64
ip circuit default only
modem passthrough nse codec g711ulaw
sip
voice class codec 1
codec preference 1 g711ulaw
```



```
voice class h323 1
 h225 timeout tcp establish 3
 telephony-service ccm-compatible
 ccm-compatible
voice translation-rule 1
rule 1 /849/ /49/
voice translation-profile Profile1
translate called 1
voice-card 0
dspfarm
dsp services dspfarm
voice-card 1
license udi pid CISCO2811 sn FTX1124A18N
archive
log config
 hidekeys
username cisco privilege 15 secret 5 $1$lirB$WgDFb9CtxJ1zd2qrofWRV1
redundancy
controller T1 0/2/0
pri-group timeslots 1-24
description PSTN PRI DIDs 650-264-5xxx incoming and outgoing
controller T1 0/2/1
pri-group timeslots 1-24
interface FastEthernet0/0
description $ETH-LAN$$ETH-SW-LAUNCH$$INTF-INFO-FE 0/0$
ip address 10.192.128.227 255.255.255.0
duplex full
speed 100
no mop enabled
h323-gateway voip interface
h323-gateway voip bind srcaddr 10.192.128.227
```



```
interface FastEthernet0/1
no ip address
shutdown
duplex auto
speed auto
interface FastEthernet0/1/0
shutdown
interface FastEthernet0/1/1
shutdown
interface FastEthernet0/1/2
shutdown
interface FastEthernet0/1/3
shutdown
interface FastEthernet0/1/4
shutdown
interface FastEthernet0/1/5
shutdown
interface FastEthernet0/1/6
shutdown
interface FastEthernet0/1/7
shutdown
interface FastEthernet0/1/8
shutdown
interface Serial0/2/0:23
no ip address
encapsulation hdlc
isdn switch-type primary-ni
isdn protocol-emulate network
isdn incoming-voice voice
no cdp enable
interface Serial0/2/1:23
no ip address
encapsulation hdlc
isdn switch-type primary-ni
isdn incoming-voice voice
no cdp enable
```



```
interface Vlan1
no ip address
shutdown
!
no ip forward-protocol nd
ip http server
ip http access-class 23
ip http authentication local
no ip http secure-server
ip http timeout-policy idle 60 life 86400 requests 10000
ip route 0.0.0.0 0.0.0.0 10.192.128.1
no cdp run
control-plane
voice-port 0/2/0:23
voice-port 0/2/1:23
voice-port 0/3/0
voice-port 0/3/1
voice-port 0/3/2
voice-port 0/3/3
voice-port 1/0/0
voice-port 1/0/1
ccm-manager redundant-host 10.192.128.221
ccm-manager mgcp
no ccm-manager fax protocol cisco
ccm-manager music-on-hold
ccm-manager config server 10.192.128.221
ccm-manager config
sccp local FastEthernet0/0
sccp ccm 10.192.128.222 identifier 1 version 5.0.1
sccp ip precedence 3
sccp
sccp ccm group 992
associate ccm 1 priority 1
associate profile 1 register HWMTPLAB
```



```
dspfarm profile 1 mtp
description MTP DSP resources
codec g711ulaw
maximum sessions hardware 1
maximum sessions software 500
associate application SCCP
dial-peer voice 600 pots
description Dial Peer for Local Calls to PSTN
destination-pattern 650[2-9]......
incoming called-number.
port 0/2/1:23
forward-digits all
dial-peer voice 602 pots
description Dial Peer for National Long Distance
destination-pattern 1[2-9]......
direct-inward-dial
port 0/2/1:23
prefix 1
dial-peer voice 700 pots
incoming called-number 4...
direct-inward-dial
dial-peer voice 2011 voip
destination-pattern 484.
session target ipv4:10.192.131.227
voice-class codec 1
dtmf-relay h245-alphanumeric
dial-peer voice 2 voip
incoming called-number .%
voice-class codec 1
dtmf-relay h245-alphanumeric
gateway
timer receive-rtp 1200
sip-ua
sip-server ipv4:10.192.128.227
banner login ^CC
```

DTMF Relay must be set to "Out of Band" for better interoperability; otherwise, user will hear DTMF tones during call setup.

Cisco Router and Security Device Manager (SDM) is installed on this device. This feature requires the one-time use of the username "cisco" with the password "cisco". The default username and password have a privilege le vel of 15.

Please change these publicly known initial credentials using SDM or the IOS CLI.

Here are the Cisco IOS commands.

username <myuser> privilege 15 secret 0 <mypassword> no username cisco



Replace <myuser> and <mypassword> with the username and password you want to use $% \left(1\right) =\left(1\right) \left(1\right) +\left(1\right) \left(1\right) \left(1\right) +\left(1\right) \left(1\right$

For more information about SDM please follow the instructions in the QUICK START

GUIDE for your router or go to http://www.cisco.com/go/sdm

C!
line con 0
login local
line aux 0
line vty 0 4
exec-timeout 0 0
logout-warning 0
login local
!
scheduler allocate 20000 1000
end



Acronyms

Acronym	Definition
Pass	The test case passed with no exceptions
Fail	The test case failed – details of the failure are noted in the Comments column
Conditional Pass	Test case pass for feature tested but fail with certain DUT or exhibit abnormal behavior.
N/A	The test case is not applicable to the product under test. Justification must be provided in the Comments column.
N/S	Not supported. While the feature tested by this test case generally would be considered a standard feature for this product category, this specific product (or this specific release) does not support the feature.
N/T	Not tested. The feature is supported by the product under test, but external factors (lab configuration, e.g.) prevented execution of the test. Justification must be provided in the Comments column.
Blocked	Other test case failures prevented the execution of this test. Reference to the corresponding failed test case must be provided in the Comments column.
MVS	BlackBerry® Mobile Voice System
BES	BlackBerry® Enterprise Server
DUT	Device under Test
RIM	Research In Motion®
CSP	Cisco Skinny Phone
CSIPP	CSIPP – Cisco SIP Phone
WLAN	Wireless LAN
WLC	Wireless LAN Controller
Wi-Fi	Wireless Fidelity (IEEE 802.11)
MGCP	Media Gateway Control Protocol
SIP	Session Initiation Protocol



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