

CCS-UC-1 SIP Endpoint with Mitel[®] MiVoice Business System 8.0PR3

Configuration Guide Crestron Electronics, Inc.

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Contents

Introduction

Audience	. 1
Topology	. 1
Software Requirements	. 2
Hardware Requirements	. 2
Product Description	. 2
Summary	. 2
Features Supported	. 2
Features Not Supported	. 3
Known Issues and Limitations	.3

Crestron Mercury Configuration

Setup	4
Configuring the device	4
Configuring the SIP Parameters	7

Mitel Configuration

Verify Licenses	
Configure Codec Settings and Network Zones	9
Configure Network Element	
Configure Class of Service	11
Configure SIP Device Capabilities	
Configure Trunk Attributes	
Configure SIP Peer Profile	
SIP Peer Profile Assignment by Incoming DID	
ARS Digit Modification Number	
ARS Routes	
ARS Digits Dialed	
Configuring User for Each Device/Phone	
Call Forwarding Profile	
Hunt Group	

1

4

8

CCS-UC-1: SIP Endpoint with Mitel MiVoice Business System 8.0PR3

Introduction

This configuration guide describes the necessary procedure to configure the Crestron Mercury[™] devices to register to the MiVoice Business (Mitel PBX) as a basic SIP endpoint.

Audience

This document is intended for users attempting to configure and use the Crestron Mercury devices as SIP endpoints registering to MiVoice Business (Mitel PBX).

Topology

The network topology for the Crestron Mercury endpoint to interop with the MiVoice Business (Mitel PBX) is shown below.

SIP Endpoint Integration with MiVoice Business (Mitel PBX) - Reference Network



The lab network consists of the following components:

- Mitel PBX
- Mitel phones
- Crestron Mercury devices as the SIP endpoints

Software Requirements

- MiVoice Business (Mitel PBX): 8.0 PR3
- Crestron Mercury device v 1.3318.00019

Hardware Requirements

- MiVoice Business (Mitel PBX) either in a virtual environment or hardware server
- PSTN Gateway
- Mitel IP phone model 5360
- Crestron Mercury devices (2)

Product Description

The Crestron Mercury device is a complete solution for conference rooms. It acts as an allin-one touch screen, speakerphone, and AirMedia[®] product for conference rooms that integrate microphones and speakers into the user interface at the table.

Crestron Toolbox[™] software is used to discover and control all Crestron devices on the network.

The Crestron Mercury web interface is used to control the Crestron Mercury devices on the network.

Summary

The Crestron Mercury devices are configured on the Mitel as SIP users that successfully register to the Mitel PBX with digest authentication.

Features Supported

- Registration with digest authentication
- Basic calls with G711u ,G711a, G722, and G729 codecs
- DTMF support
- Early media support
- Retrieval of a parked call
- Transferee in a call transfer
- Conference participant
- Member of hunt group
- Voice mail access and interaction

Features Not Supported

- Caller ID presentation
- Call hold and resume
- Call forwarding on the device (Forwarding can be configured on the PBX for the DN assigned to the endpoint.)
- Call waiting
- Conference
- Attended call transfer
- Early attended call transfer
- Blind call transfer
- Shared line (configuration of shared line on device)
- Call park (Initiating call park)
- Message waiting Indicator
- Do Not Disturb (DND)

Known Issues and Limitations

- While on an active call, if the Crestron Mercury device is power cycled, the first call to the device fails even though it is online. The subsequent calls, however, are successful. This issue is tracked via Crestron's Bugzilla™ software Defect: 126101.
- In a Mitel environment, a call declined by Crestron Mercury device doesn't provide appropriate treatment to the calling party. This issue is tracked via Crestron's Bugzilla software Defect: 127043.
- Caller ID is not supported on the Crestron Mercury device. Currently only the calling party number is displayed as the caller ID. This issue is tracked via Crestron's Bugzilla software Defect: 119006.
- The active call timer on the Crestron Mercury device does not reflect the correct call duration. The active call duration includes the time for which the unit was being alerted also. This issue is tracked via Crestron's Bugzilla software Defect: 124001.
- The first ringback heard on the Crestron Mercury device is stuttered. It resembles a mix of local and remote ringback. This issue is tracked via Crestron's Bugzilla software Defect: 122421.
- On the Crestron Mercury web user interface, there is currently no notification provided to the user when certain mandatory configurations are missing. This issue is tracked via Crestron's Bugzilla software Defect: 125193.
- On the Crestron Mercury web user interface, a configuration of DHCP OFF on the Network configuration page mandates configuration of both the adapters. The user is unable to save changes unless both the adapters are configured and is notified of an invalid IP against the default of 0.0.0.0 for an unused adapter. This issue is tracked via Crestron's Bugzilla software Defect: 126236.
- On the Crestron Mercury device, for certain called numbers that cannot be reached or are invalid, the user only hears a reorder tone and does not have the option to disconnect the call except by pressing the call button again. This issue is tracked via Crestron's Bugzilla software Defect: 122633.

• Message Waiting Indicator (MWI) is not supported on the Crestron Mercury device. This issue is tracked via Crestron's Bugzilla software Defect: 116290.

Crestron Mercury Configuration

Setup

The LAN port of the Crestron Mercury device needs to be connected to one PoE+ port to power it up and network for connectivity with the Mitel PBX. The PoE+ switch that is used should have the LLDP functionality enabled for the device to power up and be completely functional. By default, the "poeplus" configuration is set to Off on the device.

Configuring the device

To configure the Crestron Mercury device, follow this procedure:

1. Access the web GUI for the device by using an http session with the device's IP address. The device IP address used in this example was *10.35.32.18*.

Crestron Mercury Configuration: Login to Web GUI

	@ CRESTRON	
	Device Administration	
	م Sign In	
-		
	Download AirMedia Utility Software	
	Client for Mac	
	Client for Windows	

 Click Sign In and log in to the device. For information on device administration, refer to the CCS-UC-1 Supplemental Guide (Doc. 7844) at www.crestron.com/manuals.

The Status screen that appears displays basic information on the device.

Crestron Mercury: Status

CRESTRON		
STATUS	▼ General	
	General	
	Model	MERCURY
	Main Firmware Version	1.3353.00006
	Serial Number	X 0128639
AVF		
. AIRMEDIA	+ Show More	
	▼ Network	
	Domain Name	localdomain
	Encrypt Connection	false
	Host Name	MERCURY-00107F8B67B8
	Adapter 1	
	IPv4	
	DHCP Enabled	Yes
	IP Address	10.35.32.18
	Subnet Mask	255.255.255.0
	Default Gateway	10.35.32.1
	DNS Server 1	10.64.1.3

The device can be configured from the **Network** page.

3. On the web GUI, navigate to Network.

Crestron Mercury Configuration: Network Setting: DHCP Off: Static IP Configured

CRESTRON		æ
	▼ Network Setting	🖒 Revert 💾 Save Changes
	Host Name MERCURY-00107F8	B67
DEVICE	Domain Name localdomain	
AVF	Adapter 1	
	DHCP Enabled Off DHCP setting	s will apply to
	all adapters)	
	IP Address 10.35.32.18	
	Subnet Mask 255.255.255.0	
	Default Gateway 10.35.32.1	
	DNS Server 1 10.64.1.3	
	DNS Server 2 10.64.1.3	

- 4. Enter the following parameters in the **Adapter 1** section to configure the Crestron Mercury device.
 - **Domain Name**: *lab.tekvizion.com* was used in this example.
 - **DHCP**: Choose either of the following:
 - o Obtain an IP address automatically.
 - o Use the following IP address.

For this example, a static IP was configured-

- IP Address: 10.35.32.18 was used in this example.
- Subnet Mask: 255.255.255.0 was used in this example.
- **Default Gateway**: *10.35.32.1* was used in this example.
- **DNS Servers**: *10.64.1.3* was used in this example.
- 5. Click Save Changes.

Configuring the SIP Parameters

To configure the SIP parameters, follow this procedure:

1. On the web GUI, navigate to **Device** > **SIP Calling**.

Crestron Mercury: Device Configuration: SIP Calling Parameters

CRESTRON.		
. STATUS	▼ SIP Calling	🖒 Revert 💾 Save Changes
🛃 HDMI OUTPUT	Enable SIP On	
	Transport Type UDP	•
AVF	Server IP Address 10.35.32.2	
. AIRMEDIA	Port 5060	
	Server Username 5000	
	Server Password •••••	
	Server Realm *	
	Local Extension 5000	
	Proxy Server NONE	
	SIP Server Status Online	

- 2. Enable the check box for **Enable SIP**.
- 3. Configure the **Server IP Address**. Enter the IP Address of the MiVoice Business (Mitel PBX). *10.35.32.2* was used in this example.
- 4. Configure the **Port**. *5060* was used in this example.
- 5. Configure the **Server Username**. Enter the end user configured on MiVoice Business (Mitel PBX) for this device. *5000* (and *5005*) was used in this example.
- 6. Configure the **Server Password**. Enter the password as configured on MiVoice Business (Mitel PBX) for this end user.
- 7. Configure the Local Extension. Enter the directory number that was configured for this device on MiVoice Business (Mitel PBX). 5000 (and 5005) was used in this example.
- 8. Retain all other default configurations.
- 9. Click Save Changes.

Once the device successfully registers with the Mitel PBX, the **SIP Server Status** updates its status to show *Online*.

Mitel Configuration

This section describes the configuration necessary on the MiVoice Business system (Mitel PBX) to support registration of the Crestron Mercury devices and connectivity to PSTN.

NOTE: It is assumed that the general installation and basic Mitel configuration have already been administered.

Verify Licenses

Ensure that adequate licenses are available in the MiVoice Business system to support the Mitel Phones and Crestron Devices.

To verify that licenses are available, navigate to **Licenses** > **License and Option Selection** in the MiVoice Business controller. Each Crestron device will consume one IP Users license.

Mitel: License Verification

MiVoice Business Node 'Local_2' Alarm Status: Clear 2017-Apr-04 09:20:07 C									
Local_2	License and Option Selection on Local_	2	C	N to search	\vee	Show form on			
^	Change				Print	Import			
Licenses	License and Option Selection								
License and Option Selection System Capacity	Online Licensing with the Application Ma	anagemen	t Center						
Dimension Selection	Application Record ID 26682859								
Application Group Licensing 🦨	Sustam Tuna	Liconso	Sharing		Uardwar	o Idontifior			
LAN/WAN Configuration	System type	LICENSE	Sharing		naruwar	eidendilei			
Voice Network	Enterprise	NO			0000003	a1a4t			
System Properties									
Hardware	Licensed Options		Locally	Locally	Available for	Purchased			
Trunks			Consumed	Allocated	Allocation				
Users and Devices	Users								
Integrated Directory Service	IP Users		11	16	0	16			

Configure Codec Settings and Network Zones

Codec Settings are configured to allow the G711u, G711A, G722, and G729 codecs to be negotiated during this example. To change the codec settings, follow this procedure:

- 1. Navigate to Voice Network > Codec Settings.
- 2. Configure the Codec filtering.

Mitel: Codec Settings

🕅 Mitel 🛛	MiVoice	Business	Node 'Local_	2' Alarm Status:	Clear 2017-Apr-04	09:20:07	₽?	(Ŀ
Local_2	ź	Codec Settings on Local_2		DN to search	\vee	Show form o	Not Access	ible	G C
LAN/WAN Configurati	on ^	Change			Print	Import	Export	Data	Refresh
Voice Network		🥔 Codec Settings							
Network Elements 🦨		G.711a Filtered					No		
Cluster Elements 🥔		G.711u Filtered					No		
Admin Groups		G722 Filtered					No		
Fax Advanced Settings		G.722.1 Filtered					Yes		
Network Zones 🖨		G.723 Filtered					Yes		
Network Zone Topology	¢	G.723.1c Filtered					Yes		
Bandwidth Managemen	t 🧈	G.728 Filtered					Yes		
Codec Settings 🦨		G.729a Filtered					No		
System Properties		G.729b Filtered					No		
Trunke		T.38 Filtered					Yes		
Users and Devices		AMR Narrowband Filtered					Yes		
Integrated Directory	Services	GSM FR (06.10) Filtered					Yes		

Network Zones 1 and 2 were configured to offer preferred codecs as G711u and G729 respectively.

3. Navigate to Voice Network > Network Zones.

Mitel: Configure Network Zones

🕅 Mitel 🛛 🕅	MiVoice E	Busines	S			N	ode 'Lo	ocal_2' Alarm St	atus: Cle	ear 201	7-Apr-04 0	9:20:07	D	?	1	G ∙
Local_2	₹.	Network Z	ones on Local_	2				DN to se	arch	>		Show for	rm or Not	Accessible	V G	• 🕈
	^	Change	Change F	age	Clear						Print	Import.	Ex	port	Data Refre	sh
Licenses		Pa	ge 1 of 50 🗾						Go to			~	Value			Go
LAN/WAN Configuration Voice Network		🧳 Netw	ork Zones	_												
Network Elements 🥔		Zone	Intra-zone	Group	Intra-zone	Label	SMDR	Time 7ees	LBN	Zone	Default	Default	Audio	Embedded	Music- On-Hol	d ^
Cluster Elements 🦑		ID	Compression	n Zone	Fax Profile	Laber	Tag	Time Zone	Prefix	CESID	Number	CPN	Source	Source	Music Source	,
Admin Groups		1	No		2	CrestronG711u										
Fax Service Profiles 🦑		2	Yes		2	CrestronG729		America/Chicago								
Fax Advanced Settings		3	No		1			America/Chicago								
Network Zone Tonology		4	No		1											

- 4. Select a zone to modify: 2 was used for this example.
- 5. Set the Intra-zone compression to Yes.

NOTE: Zone 2 was used when G729 was required as the preferred codec. By default, Zone 1 was used, which negotiated G711u as the preferred codec.

Configure Network Element

Create a network element for the PSTN GW. This is done in the Network Element assignment form.

- 1. Navigate to **Voice Network** > **Network Elements**.
- 2. Click Add.

Mitel: Configure Network Elements

Мitel міv	oice Business	Node 'Lo	cal_2' Alarm Status: <mark>Cl</mark> e	ear 2017-Apr-04 09:20	:07 💭	? () [+
Local_2	Network Elements on Local_2		DN to search	∀ Sho	ow form on Not A	ccessible	🗸 💿 🕇
^	Add Change Delete			Print Im	port Exp	ort Data	a Refresh
Licenses	Network Elements						
LAN/WAN Configuration		PBX Num	ber/Cluster Element ID+	FODN or IP Address -	Data Sharing	 Version 	Zone 🕶
Voice Network	PSTN GW Other			10.64.1.72	NO		1
Network Elements 🧬	<						>
Cluster Elements 🧬	Name		PSTN_GW				
Admin Groups	Туре		Other				
Fax Service Profiles 🛹	FQDN or IP Address		10.64.1.72				
Fax Advanced Settings	Data Sharing		NO				
Network Zones 🛹	Local		False				
Network Zone Topology 🛹	Version						
Bandwidth Management 🛹	7		-				
Codec Settings 🛹	Zone		1				
System Settings	ARID						
System Feature Settings	SIP Peer Specific						
System Administration	SIP Peer Transport		default				
Hardware	SIP Peer Port		5060				
Trunks	External SIP Proxy FQDN or IP Add	ress					
Users and Devices	External SIP Proxy Transport		default				
Integrated Directory Service	External SIP Proxy Port		0				
Voice Mail	SIP Registrar FQDN or IP Address						
Call Routing	SIP Registrar Transport		default				
Music On Hold	SIP Registrar Port		0				
Emergency Services Manag	SIP Peer Status		Auto-Detect/Normal				

- 3. Set **Name**. *PSTN_GW* was used in this example.
- 4. Set Type: Other.
- 5. Set **FQDN or IP Address**: *10.64.1.72* was used in this example. This is the IP address of the PSTN GW.
- 6. Set **Zone**: *1* was used in this example. This setting ensures a codec of G711. To change it to G729, modify the network zone to *2*.
- 7. Check the SIP Peer check box.
- 8. Set the SIP Peer Transport: *Default* was used in this example.
- 9. Set the SIP Peer Port: 5060 was used in this example.
- 10. Retain all other default configurations.
- 11. Click Save.

Configure Class of Service

To configure the class of service, follow this procedure:

1. Navigate to System Properties > System Feature Settings > Class of Service Options.

Mitel: Class of Service Options (1/9)

🕅 Mitel 🕴 мім	oice Busingestocal_2' Alarm Status: Clear 2017-Apr-04 (09:20:07 💭	?	⑦ ₽
Local_2	Class of Service Options on Local_2 DN to search V	Show f	form on	
^	Change Copy Print	Import	Export	
Licenses			Save	Cancel
LAN/WAN Configuration	General Advanced			
Voice Network				
System Properties	Class Of Service Number	10		
System Settings	Comment	Crestron		
System Feature Settings	ACD			
System Options	ACD Agent Robertier on No Anewor	Lassut		
Shared System Options 🥏	ACD Agent behavior on no Answer	Logout		
Class of Service Options 🥏	ACD Agent No Answer Timer	15		
SIP Device Capabilities 🦨	ACD Make Busy on Login	No O Yes	s	
Class of Restriction Groups	ACD Silent Monitor Accept	● No ○ Yes	s	
System Access Points 🛹	ACD Silent Monitor Accept Monitoring Non-Prime Lines	● No ○ Yes	5	
Feature Access Codes 🧬	ACD Silent Monitor Allowed	No Ves	S	
Independent Account Codes	ACD Silent Monitor Notification	No () Yes	8	
Default Account Codes 🧬	Follow 2nd Alternate Reroute for Recall to Busy ACD Agent	No O Yes No O Yes		
System Account Codes 🧬 📃			-	
System Speed Calls 🦨	work limer	0		
Tenants	Announce			
SMDR Options 🧬	Call Announce Line	No O Yes	5	
Traffic Report Options 🦨	Off-Hook Voice Announce Allowed	No O Yes	S	
Inward Dialing Modification	Handsfree AnswerBack Allowed	No Ves	s	
Outward Dialing Modification	Busy Override			
System IP Ports 🦨	Busy Override Security	() No (●) Yes	S	
Location Based Numbers 🤞	Disable Executive Busy Override Tone	No Ves	s	

Mitel: Class of Service Options (2/9)

General	Advanced						
Call Co	Call Control Timer						
Busy	30						
Dialing	Conflict Timer	3					
First [igit Timer	15					
Inter I)igit Timer	10					
Locko	ut Timer	45					
Call Du	ration						
Call D	Iration	10					
Call D	0						
Enable	● No () Yes						
Enable	● No () Yes						
Call Fo							
Call Fo	orward - Delay	0					
Call Fo	orward No Answer Timer	15					
Call Fo	orward Override	● No () Yes					
Call Fo	○ No Yes						
Call Fo	○ No Yes						
Call Fo	○ No Yes						
Call R	● No () Yes						
Call Fo	● No () Yes						
Disabl	No Yes						

Mitel: Class of Service Options (3/9)

General Advanced	
Follow Reroute on Disabled Forwarding	● No () Yes
Group Call Forward Follow Me Accept	● No ◯ Yes
Group Call Forward Follow Me Allow	● No ◯ Yes
Third Party Call Forward Follow Me Accept	● No ◯ Yes
Third Party Call Forward Follow Me Allow	● No ◯ Yes
Use Held Party Device for Call Re-routing	○ No Yes
Call Hold	
Call Hold	○ No Yes
Call Hold - Retrieve with Hold Key	○ No Yes
Call Hold Remote Retrieve	○ No Yes
Call Hold Timer	30
Local Music On Hold source	● No () Yes
Music on Hold on Transfer	○ No Yes
Use Called Party Call Hold Timer	● No ◯ Yes
Call Park	
Call Park Timer	180
Call Park-Allowed To Park	○ No Yes
Call Pickup	
Allow Directed Call Pickup Of Attendant Call	● No ◯ Yes
Call Pickup Dialed Accept	○ No Yes
Call Pickup Directed Accept	○ No Yes

Mitel: Class of Service Options (4/9)

General Advanced						
Call Pickup Display O No O Yes						
Call Privacy						
Call Privacy	● No ○ Yes					
Calling Party Name Substitution	● No ◯ Yes					
Name Suppression on outgoing Trunk Call	● No ◯ Yes					
Privacy Released	● No ◯ Yes					
Public Network Identity Provided	◯ No Yes					
Call Waiting						
Call Waiting Swap	● No () Yes					
ONS CLASS/CLIP: Visual Call Waiting	🔿 No 🖲 Yes					
Campon						
Auto Campon Timer						
Campon Recall Timer	10					
Direct Voice Call						
Direct Voice Call - Accept	● No ◯ Yes					
Direct Voice Call - Allow	● No () Yes					
Direct Voice Call - Maximize Volume	● No ◯ Yes					
Display						
After Answer Display Time						
Calling Name Display - Internal - ONS	◯ No Yes					

Mitel: Class of Service Options (5/9)

General Advanced	
Calling Number Display - Internal - ONS	○ No Yes
Display ANI/DNIS/ISDN Calling/Called Number	● No () Yes
Display ANI/ISDN Calling Number Only	● No () Yes
Display Caller ID on multicall/keylines	● No ◯ Yes
Display Caller ID On Multicall/Keylines Timer	5
Display Caller ID On Single Line Displays For Forwarded Calls	● No () Yes
Display Dialed Digits during Outgoing Calls	● No () Yes
Display DNIS/Called Number Before Digit Modification	● No () Yes
Display DNIS on Key Label	● No ◯ Yes
Display Held Call ID on Transfer	● No ◯ Yes
Display Transfer Destination on Recall	● No ◯ Yes
Hot Desk External User - Display Internal Calling ID	● No () Yes
Maintain Ringing Party During Recall	● No () Yes
Non-Prime Public Network Identity	● No () Yes
Originator's Display Update In Call Forwarding/Rerouting	● No () Yes
Prefer Call Forwarding/Rerouting Information	● No () Yes
Prefer Name for Call Information	● No () Yes
Suppress Delivery of Caller ID Display between Sets	● No () Yes
Suppress Delivery of Caller ID Display between Sets - Override	● No () Yes
Suppress Display Of Account Code Numbers	● No () Yes

Mitel: Class of Service Options (6/9)

General	Advanced	
Suppr	ess Redial Display	● No () Yes
Fax		
Camp	on Tone Security	○ No Yes
Extern	al Trunk Standard Ringback	○ No Yes
Fax Ca	pable	● No () Yes
Retur	n Disconnect Tone When Far End Party Clears	● No () Yes
HCI		
HCI/CT	T/TAPI Call Control Allowed	○ No Yes
HCI/CT	T/TAPI Monitor Allowed	○ No Yes
Hot De	sk	
Green	BLF Lamp for Logged in Hotdesk User	● No () Yes
Hot De	sk External User - Allow Mid-Call Features	∩ No () Yes
Hot De	sk External User - Answer Confirmation	○ No Yes
Hot De	sk External User - Dial Tone on Call Complete	○ No Yes
Hot De	sk External User - Permanent Login	● No () Yes
Hot De	sk External User - Remote MWI Enable Feature Access Code	
Hot De	sk External User - Remote MWI Disable Feature Access Code	
Hot De	sk Login Accept	○ No Yes
Hot De	sk Remote Logout Enabled	● No () Yes
Misce	llaneous	
Backli	ghting - Enabled	○ No Yes

Mitel: Class of Service Options (7/9)

General	Advanced					
Clear	All Features Remote	● No ○ Yes				
Enblo	Enbloc Dialing - Enabled					
Force	Force Device Busy If Any Line In Use					
Hands	Handset Volume Adjustment Saved					
Head	Set Switch Mute	● No () Yes				
Long	Key Press Timer	0				
Multi-	Color LED Support - Disable	() No () Yes				
Phone	Lock	() No () Yes				
Resei	ze Timer	180				
Timed	○ No () Yes					
Userl	0					
Paging	I					
Group	Page Accept	● No ○ Yes				
Group	Page Allow	● No ◯ Yes				
Louds	peaker Pager Equivalent Zone Override Security	● No ◯ Yes				
Louds	Loudspeaker Pager Override					
Pager	○ No Yes					
Pager	● No () Yes					
PC Po	rt					
PC Po	rt On IP Device - Disable	● No ◯ Yes				

Mitel: Class of Service Options (8/9)

General Advanced	
RAD	
Answer Plus Delay To Message Timer	20
Answer Plus Expected Off-hook Timer	30
Answer Plus Message Length Timer	10
Answer Plus System Reroute Timer	0
Recorded Announcement Device	● No ◯ Yes
Recorded Announcement Device - Advanced	● No ◯ Yes
Ringing	
Delay Ring Timer	10
No Answer Recall Timer	17
Ringing Line Select	● No ◯ Yes
Ringing Timer	180
SMDR	
SMDR External	● No ◯ Yes
SMDR Internal	● No ◯ Yes
Trunk	
ANI/DNIS/ISDN Number Delivery Trunk	● No ◯ Yes
DASS II OLI/TLI Provided	● No ◯ Yes
Public Network Access via DPNSS	🔿 No 💽 Yes

Mitel: Class of Service Options (9/9)

General Advanced	
Public Network To Public Network Connection Allowed	● No () Yes
Public Trunk	● No ◯ Yes
R2 Call Progress Tone	● No ◯ Yes
Suppress Simulated CCM after ISDN Progress	● No ◯ Yes
Trunk Calling Party Identification	○ No Yes
Trunk Flash Allowed	● No ◯ Yes
Two B-Channel Transfer Allowed	● No ◯ Yes
Voice Mail	
COV/ONS/E&M Voice Mail Port	● No ◯ Yes
ONS VMail-Delay Dial Tone Timer	5

- 2. Select the Class of Service number: 10 was used in this example.
- 3. Click Change.
- 4. In the **General** tab, change the following options:
 - Public Network Access via DPNSS: Select Yes.
 - Auto Campon Timer: Clear the value.
 - Busy Override Security: Select Yes.
 - Call Park- Allowed to Park: Select Yes.
 - Music on Hold on transfer: Select Yes.
- 5. Retain all other default configurations.
- 6. Click Save.

Configure SIP Device Capabilities

The SIP Device Capabilities form allows customization of the features and options that the Mitel MiVoice System uses and accepts when communicating with the Crestron Mercury devices. To configure the SIP device capabilities, follow this procedure.

- 1. Navigate to System Properties > System Feature Settings > SIP Device Capabilities.
- 2. Select SIP Device Capabilities: 10.
- 3. Click Change.
- 4. Configure the **Basic** tab:

Mitel: SIP Device Capabilities: Basic Tab

🕅 Mitel 🕴 міv	oice Bustiffests ^{ocal_2} ' Alarm Status: Clear 2017-Apr-04 09:20:07 🗖 ? 🛈 🗗
Local_2	SIP Device Capabilities on Local_2 DN to search V Show form on
Voice Network	
System Properties System Settings	SIP Device Capabilities
System Feature Settings	10 Crestron
System Options Shared System Options 🛹	11
Class of Service Options 🥔	Save Cancel
SIP Device Capabilities 🥏	Basic SDP Options Signaling and Header Manipulation Distinctive Ring Tones Timers Key Press Event
Class of Restriction Groups	Record Information Advanced
System Access Points 🥏	
Feature Access Codes 🥏	SIP Device Capabilities Number 10
Independent Account Codes	Comment Crestron
Default Account Codes 🦑	Call Routing and Administration Options
System Account Codes 🥏	Outbound Proxy Server
System Speed Calls 🥏	Replace System based with Device based In-Call Features
Tenants	Allow MWI Notifications without Subscription
SMDR Options 🛹	Enable Digit Collection In Busy Or Alerting State
Traffic Report Options 🖉 👘	

- a. Configure **Comment**: Provide any name. *Crestron* was used in this example.
- b. Set the Replace System based with Device based In-Call Features: Yes.
- c. Retain all other default configurations.

5. Configure the **SDP Options** tab:

Mitel: SIP Device Capabilities: SDP Options Tab

Basic SDP Options Signaling and Header Manipulation Distinctive Ring Tones Timers Ke	y Press Event			
Record Information Advanced				
Allow Device To Use Multiple Active M-Lines	● No ⊖ Yes			
Allow Using UPDATE For Early Media Renegotiation	○ No Yes			
AVP Only Device	○ No Yes			
Enable Mitel Proprietary SDP	● No ◯ Yes			
Force sending SDP in initial Invite message	● No ◯ Yes			
Ignore SDP Answers in Provisional Responses	● No ⊖ Yes			
Limit to one Offer/Answer per INVITE	● No ⊖ Yes			
Prevent SDP Renegotiation If Peer Initiated Hold				
Prevent the Use of IP Address 0.0.0.0 in SDP Messages	○ No			
Renegotiate SDP To Enforce Symmetric Codec	○ No Yes			
Repeat SDP Answer If Duplicate Offer Is Received				
Send Answer only after renegotiation is complete				
Suppress Use of SDP Inactive Media Streams				

- a. Set the Allow Using UPDATE for Early Media Renegotiation: Yes.
- b. Set the Prevent the Use of IP Address 0.0.0.0 in SDP Messages: Yes.
- c. Set the Renegotiate SDP to Enforce Symmetric Codec: Yes.
- 6. Retain all other default configurations.
- 7. Click Save.

Configure Trunk Attributes

Trunk attributes must be defined for the trunk used for PSTN calls. To define a trunk attribute, follow this procedure:

- 1. Navigate to Trunks > Trunk Attributes.
- 2. Select Trunk Service Number: 10 was used in this example.
- 3. Click Change.
- 4. Assign the Trunk Label: Crestron was used in this example.
- 5. Set Class of Service: 10 (configured earlier) was used in this example.
- 6. Set **Dial In Trunks Incoming Digit Modification Absorb**: *0* was used in this example. (The Mitel absorbs none of the incoming digits on an incoming PSTN call to reach the desired PBX extension based on the translation configured.)
- 7. Set Dial In Trunks Incoming Digit Modification Insert: Leave blank.

Mitel: Configure Trunk Attributes

🕅 Mitel	MiVo	oice Bus	Mesisoca	I_2' Alarm	Status: Clear	2017-Apr-(04 09:20:07	D	?	(E•
Local_2	₹ I	Trunk Attribu	utes on Loca	al_2	DN to search	~		Show form	n on		
	^	Change	Change	Page	Change All	Clear					
Licenses		< Page	e 1 of 15 🧧	>	Go to		~	Value			Go
LAN/WAN Configura	tion	🤣 Trunk /	Attributes	_							
System Properties		ŏ	NO	UTT	UTT	1	1	300	ï		
System Settings		9	No	Off	Off	1	1	300	1		
System Feature Set	tings	10	No	Off	On	10	1	300	1	Cres	tron
System Administrat	ion	Trunk Servi	ice Number							10	
Hardware		Release Lin	ik Trunk							No	
Trunks	Call Recogn	Call Recognition Service					Off				
Trunk Attributes 🥏		Direct Inward Dialing Service						On			
DTS Service Profiles		Class of Se	Class of Service						10		
Analog		Class of Re	Class of Restriction						1		
Digital		Baud Rate						300			
IP/XNET		Intercept Number						1			
JIP Usors and Dovicos		Non-dial In Trunks Answer Point - Day									
Users and Devices		Non-dial In Trunks Answer Point - Night 1									
Voice Mail		Non-dial In Trunks Answer Point - Night 2									
Call Routing Dial In Trunks Incoming Digit Modification - Absorb					0						
Music On Hold		Dial In Trun	ks Incoming	Digit Modi	fication - Insert					-	
Emergency Services	Emergency Services Manag										
Property Management		Dial In Trunks Insert Forwarding Information				No					
Maintenance and D	iagnostic	Trunk Labe	I							Crestron	
		L									

Configure SIP Peer Profile

To configure the SIP peer profile, follow this procedure.

1. Navigate to Trunks > SIP > SIP Peer Profile.

Mitel: Add SIP Peer Profile

🕅 Mitel 🕴 міvo	Dice Busil එල්පී S ^{'Local_2} ' Alarm Status: <mark>Clear</mark> 2017-Apr-04 09	9:20:07 🗖	? (D E•
Local_2	SIP Peer Profile on Local_2 DN to search V	Show for	rm or Not Acces	sible 🗸
System Settings	Add Change Delete Print	Import E	ixport Da	ta Refresh
System Feature Settings	SIP Peer Profile			
System Administration	Network Element SIP Peer Profile Label Outbound Proxy Server	CPN Restriction	Trunk Service	Session Timer
Hardware				
Trunks	PSTN GW Crestron N	No	10	90
Trunk Attributes 🥏				
DTS Service Profiles	<			>
Analog	Basic Call Routing Calling Line ID SDP Options Signaling and Header M	Internation Anticent	rs Key Press E	vent
Digital	Outgoing DID Ranges Profile Information			
IP/XNFT				^
SID	SIP Peer Profile Label	Crestron		
DID Deseas for ODM Outset	Network Element	PSTN_GW		
DID Ranges for CPN SUbsti	Local Account Information			
SIP Peer Profile	Registration User Name			

2. Click Add.

3. Configure the **Basic** tab:

Mitel: SIP Peer Profile: Basic Tab

SIP Peer Profile on Local_2	DN to search	Show form or Not Accessible 🗸 Go 🕇
Add Change Delete		Print Import Export Data Refresh
Basic Call Routing Calling Line ID SDP	Options Signaling and Header Manipulation Time	ers Key Press Event Outgoing DID Ranges Profile Information
SIP Peer Profile Label		Crestron
Network Element		PSTN_GW
Local Account Information		
Registration User Name		
Address Type		IP Address: 10.35.32.2
Administration Options		
Interconnect Restriction		1
Maximum Simultaneous Calls		100
Minimum Reserved Call Licenses	•	0
Outbound Proxy Server		
SMDR Tag		0
Trunk Service		10
Zone		1
Authentication Options		
User Name		
Password		******
Confirm Password		******
Authentication Option for Incomin	ig Calls	No Authentication
Subscription User Name		
Subscription Password		*****
Subscription Confirm Password		*****

- a. Enter a descriptive name for SIP Peer Profile Label: *Crestron* was used in this example.
- b. Select **Network Element**: *PSTN_GW* was used in this example.
- c. Local Account Information-Address Type: IP Address: Select 10.35.32.2.
- d. Configure **Trunk Service**: *10* (as configured earlier as Trunk Group) was used in this example.
- e. Retain all other default configurations.

4. On the **Call Routing** tab, retain all default values.

Mitel: SIP Peer Profile: Call Routing Tab

SIP Peer Profile on Local_2	DN to search 🤟	e 🗸 Go 🕂						
Add Change Delete		Print Import Export	Data Refresh					
Basic Call Routing Calling Line ID SDP Options	Signaling and Header Manipulation Timers	Key Press Event Outgoing DID Ranges	Profile Information					
Alternate Destination Domain Enabled			No					
Alternate Destination Domain FQDN or IP Addr	ess							
Enable Special Re-invite Collision Handling			No					
Only Allow Outgoing Calls			No					
Private SIP Trunk			No					
Reject Incoming Anonymous Calls								
Route Call Using P-Called-Party-ID (if present)			Yes					
Route Call Using To Header			No					

5. On the Calling Line ID tab, retain all default values.

Mitel: SIP Peer Profile: Calling Line ID Tab

SIP Peer Profile on Local_2	DN to search	\mathbf{v}	Show	form or Not Accessible	e 🗸 Go 🕈			
Add Change Delete		Pr	int Impo	rt Export	Data Refresh			
Basic Call Routing Calling Line ID SDP Options	Signaling and Header Manipulation	Timers I	Key Press Event	Outgoing DID Ranges	Profile Information			
Default CPN								
Default CPN Name								
CPN Restriction					No			
Public Calling Party Number Passthrough					No			
Strip PNI								
Use Diverting Party Number as Calling Party Number								
Use Original Calling Party Number If Available					No			

6. On the SDP Options tab, set Allow Using UPDATE for Early Media Renegotiation to Yes.

Mitel: SIP Peer Profile: SDP Options Tab

SIP Peer Profile on Local_2	DN to search V	Show form or Not Accessible 🛛 🗸 Go 🖣
Add Change Delete	(Print Import Export Data Refresh
		Save Cancel
Basic Call Routing Calling Line ID SDP Options Signaling and Header Manipulation	Timers Key Press Event	Profile Information
Allow Peer To Use Multiple Active M-Lines		◯ No) Yes
Allow Using UPDATE For Early Media Renegotiation		🔿 No 🖲 Yes
Avoid Signaling Hold to the Peer		🔿 No 🖲 Yes
AVP Only Peer		🔿 No 🖲 Yes
Enable Mitel Proprietary SDP		● No () Yes
Force sending SDP in initial Invite message		● No () Yes
Force sending SDP in initial Invite - Early Answer		● No () Yes
Ignore SDP Answers in Provisional Responses		● No ◯ Yes
Limit to one Offer/Answer per INVITE		◯ No) Yes
NAT Keepalive		🔿 No 🖲 Yes
Prevent the Use of IP Address 0.0.0.0 in SDP Messages		◯ No) Yes
Renegotiate SDP To Enforce Symmetric Codec		● No () Yes
Repeat SDP Answer If Duplicate Offer Is Received		● No () Yes
Restrict Audio Codec		No Restriction
RTP Packetization Rate Override		(● No () Yes
RTP Packetization Rate		20ms 🗸
Special handling of Offers in 2XX responses (INVITE)		(●) No () Yes
Suppress Use of SDP Inactive Media Streams		(●) No () Yes

7. Configure the Signaling and Header Manipulation tab:

Mitel: SIP Peer Profile: Signaling and Header Manipulation Tab

Basic	Call Routing	Calling Line ID	SDP Options	Signaling and Header Manipulation	Timers	Key Press Event	Outgoing DID Ranges	Profile Information
Trun	k Group Lab	el						
Allo	w Display Upd	late						Yes
Build	I Contact Usi	ng Request URI	Address					No
De-r	egister Using	Contact Addre	ess not *					Yes
Disa	ble Reliable P	rovisional Res	ponses					No
Disa	ble Use of Us	er-Agent and S	Server Header	s				No
Dom	ain for Trunk	Context						
E.164	I: Enable sen	ding '+'						No
E.164	l: Add '+' if di	git length > N d	igits					0
E.164	I: Do not add	'+' to Emergen	cy Called Party	Y				No
E.164	l: Do not add	'+' to Called Pa	rty					No
Forc	e Max-Forwa	rd: 70 on Outgo	oing Calls					No
If TL	S use 'sips:'	Scheme						No
Igno	re Incoming	Loose Routing	Indication					No
Inclu	de Diversion	Header for EH	DU					No
Mult	ilingual Name	Display						No
Only	use SDP to d	ecide 180 or 18	33					Yes
Pref	er From Head	ler for Caller ID)					No
Req	uire Reliable I	Provisional Res	sponses on Ou	utgoing Calls				No
Sign	al Privacy (if e	enabled) on Err	nergency Calls	\$				No
Sup	press Redire	ction Headers						No
Use	Fixed Retry T	ime for 491						No
Use	Privacy: none	•						No
Use	P-Asserted I	dentity Header						Yes
Use	P-Asserted I	dentity for Billi	ng					No
Use	P-Call-Leg-ID	Header						No
Use	P-Early-Medi	a Header						No
Use	P-Preferred	dentity Header	T					No
Use	Restricted C	haracter Set Fo	or Authenticati	ion				No
Use	To Address i	n From Header	on Outgoing	Calls				No
Use	user=phone							No
Use	user=phone	for Diversion H	leader					No

a. Set Allow Display Update: Yes.

b. Set Require Reliable Provisional Responses on Outgoing Calls: No.

8. Retain all other default configurations.

SIP Peer Profile Assignment by Incoming DID

This form is used to assign incoming digits from the PSTN to the Mitel. To configure a SIP peer profile assignment, follow this procedure:

1. Navigate to Trunks > SIP > SIP Peer Profile by Incoming DID.

Mitel: SIP Peer Profile Assignment by Incoming DID

Mitel MiVoice E	Business	Node 'Local_2' Alarm St	atus: Clear 2017-Apr-04 09:20:07	□ ?	٦	Ŀ
Local_2	A H	SIP Peer Profile Assignment by Incoming DID on Local 2	DN to search 🗸	Show form on		
DTS Service Profiles	^	Add Change Delete	Print Import	Export	Data R	efresh
Analog		SIP Peer Profile Assignment by	Incoming DID			
Digital		9722657277-9722657279	Crestron	PSTN-	Crestron	
IP/XNET		Incoming DID Range	9722657277-97226572	79		
SIP		SIP Peer Profile Label	Crestron			
DID Ranges for CPN Substitution		Comment	PSTN-Crestron			
SIP Peer Profile						
SIP Peer Profile Assignment by Incoming D						

- 2. Click Add.
- 3. Configure the **Incoming DID Range**: *9722657277-9722657279* was used for this example.
- 4. Select the SIP Peer Profile Label: Crestron was used in this example.
- 5. Provide a Comment (optional): PSTN-Crestron was used in this example.

ARS Digit Modification Number

Digit Modification for outgoing calls on the SIP trunk to PSTN is configured to absorb or inject additional digits according to the dialing plan chosen. In the current example, one (1) digit was absorbed. To configure digit modification, follow this procedure:

1. Navigate to Call Routing > Automatic Route Selection (ARS) > ARS Digit Modification Plans.

Mitel MiVoice E	Busines	s		Node 'Local_2	' Alarm Status	:: Clear 2017-Apr-04 0	9:20:07 🖵	? נ	(E∙
Local_2	Å.	ARS Digit Modif	cation Plans on L	ocal_2	DN to search	¥	Show form on	Not Accessit	le 🗸	Go 🕈
Users and Devices	^	Change	Change Page	Change All	Clear	Print	Import	Export	Data Re	efresh
Integrated Directory Services		< Page	1 of 55 >		Go to		✓ Valu	е		Go
Voice Mail Call Routing		🥔 🗚 S Dig	it Modification	Plans						
Automatic Route Selection (ARS)		Digit Mod	ification Number	Number of D	gits to Absorb	Digits to be Inserted	Final Tone	Plan/Informa	tion Marker	
ARS Call Progress Tone Detection 🦨		1		1						
ARS Digit Modification Plans 🖨		2		0						
ARS Maximum Dialed Digits 💣		3		0						

Mitel: ARS Digit Modification Number

- 2. Modify **Digit Modification Number:** *1*, to absorb one digit while dialing out to PSTN.
- 3. Click Save.

ARS Routes

To configure a route for SIP trunk connectivity to PSTN, follow this procedure:

1. Navigate to Call Routing > Automatic Route Selection (ARS) > ARS Routes

Mitel:	ARS	Routes
--------	-----	--------

Mitel MiVoice Business	Node 'Local_2' Alarm Status: Clear 2017-Apr-04 09:20:07 💭
Local_2	ARS Routes on Local_2 DN to search V Show form on No.
Users and Devices	Change Change Page Change All Clear Print Import
Integrated Directory Services	Change
Voice Mail	Change
Call Routing	ARS Routes
Automatic Route Selection (ARS)	Route Number 10
ARS Call Progress Tone Detection 🇬	Routing Medium SIP Trunk 🗸
ARS Digit Modification Plans 🇬	Trunk Group Number
ARS Maximum Dialed Digits 🧬	SIP Peer Profile Crestron
ARS Routes	DRY Number / Cluster Element ID
ARS Route Lists	
ARS Route Plans	COR Group Number 1
ARS Digits Dialed	Digit Modification Number 1
ARS Leading Digits	Digits Before Outpulsing
ARS Day and Time Zones 💞	Route Type PSTN Access Via DPNSS 🗸
	Compression Off v
Can nanoing Music Op Hold	
Emergency Services Management	
Property Management	Save Cancel

- 2. Select a route number that is not in use. *10* was used in this example.
- 3. Click Change.
- 4. Select Routing Medium: SIP Trunk.
- 5. Select SIP Peer Profile: Crestron was used in this example.
- 6. Configure **Digit Modification Number**: *1* was used in this example.
- 7. Choose Route Type: PSTN Access Via DPNSS was used in this example.
- 8. Click Save.

ARS Digits Dialed

ARS initiates the routing of trunk calls when certain digits are dialed from a station. In this example, the prefix 5 was used to route calls towards PSTN using the Route 10. To configure ARS digits, follow this procedure:

1. Navigate to Call Routing > Automatic Route Selection (ARS) > ARS Digits Dialed.

🕅 Mitel 🛛	MiVoice I	Business Noo	le 'Local_2' Alarm S	tatus: Clear 2017-Apr-04 0	19:20:07 💭	?	0	Đ
Local_2	A a a a a a a a a a a a a a a a a a a a	ARS Digits Dialed on	Local_2	DN to search 🗸	Show fo	orm on Not	Accessible	>
Users and Devices	^	Add Chang	je Delete	Print	Import	Export	Data Ref	resh
Integrated Directory Ser	rvices	< Page 1 of	1 >	Go to	✓ Value			Go
Voice Mail		ARS Digits Diale	d					
Call Routing		Digits Dialed	Number of Digits to Fo	ollow Terminati	on Type	Terminatio	n Number	
Automatic Route Selection	on (ARS)	5	10	Route		10		
ARS Call Progress Tone ARS Digit Modification F	Plans a							
ARS Maximum Dialed D	ligits 🖨							
ARS Routes		51	10	Route		10		
ARS Route Lists								
ARS Route Plans								
ARS Digits Dialed								
ARS Leading Digits								

Mitel: ARS Digits Dialed

- 2. Click Add.
- 3. Enter the number of records to add: 1.
- 4. Configure Digits Dialed: 5 was used in this example.
- 5. Configure Number of Digits to Follow: 10.
- 6. Configure Termination Type: Route.
- 7. Configure Termination Number: 10 was used in this example.

Similarly, another entry for starting digits 51 was added.

Configuring User for Each Device/Phone

The Crestron Mercury Device was configured as a Generic SIP phone that registers to the Mitel PBX. A user was configured for each phone and Crestron Mercury device used in the example. To configure a user, follow this procedure.

- 1. Navigate to Users and Devices > User and Devices Configuration.
- 2. Click Add > Default User and Device.
- 3. Configure the User Profile tab as follows:

Mitel: Add User: User Profile Tab

Mitel MiVoi	ce Busineeds 'Local_2' Alam	n Status: Clear 2017-Apr-04 09:2	10:07 🖵 ? 🛈 🗗
Local_2	User and Services Configuration on	DN to search	Show form on
^	Add V	Print	Import Export Data Refresh
Licenses	User and Services Configurat	ion	
LAN/WAN Configuration	Search By Last Name		
Voice Network			Save Changes Cancer
System Properties	(All Users) + Q	User Profile Service Profile Dev	ice Details Service Details
Hardware	Search Results (13 matches)	Access and Authentication Phone	Applications Keys
Trunks	* *	Last Name	DUT1
Users and Devices			
User and Services Configuration 🖨	DUT2. Mercurv2	First Name	Mercury1
Attendants	Phone Service (5005)	Department	
ACD	DUT1, Mercury1	Location	
Group Programming	Phone Service (5000)	Role	
Telephone Directory Manageme	voicemail		
Advanced Configuration		Language	English
Templates	Mitel 3, Mitel3	Email	
Integrated Directory Services	Mitel1, 5001	IDS-Manageable	
Voice Mail	Mitel4, Mitel4	Prime Phone Service	Phone Service (5000)
Call Routing		Desktop Admin Access	
Music On Hold		Login ID	
Emergency Services Managem		Login ID	
Property Management		Password	
Maintenance and Diagnostics 🗸		Confirm Password	

- a. Enter Last Name: DUT1 was used in this example.
- b. Enter First Name: *Mercury1* was used in this example.

4. Configure the Service Profile tab:

Mitel: Add User: Service Profile Tab

User Profile	Service Profil	e De	vice Details	Service	Details		
Access and Authentication Phone			e Applications	Keys			
Number			5000				
Service La	bel		Phone Se	rvice			
Directory N	lame		DUT1,Mer	cury1			
Prime Nam	ie		● No () Yes				
Privacy			● No ◯ Yes				
Hot Deskin	g User		● No () Ye	s			
Device Typ	е		Generic SIP	Phone		~	
Service Le	vel		Full			~	
Home Elem	nent		Local_2				
Secondary	Element		Not Assigne	ed		~	
Local-only	DN						
ACD Enable	ACD Enabled						
Single Line	Phone						

- a. Enter Number: 5000 (available DN) was used in this example.
- b. Enter **Device Type**: Generic SIP Phone was used in this example.

5. Configure the Service Details tab:

Mitel: Add User: Service Details Tab

User Profile	Service Profile	Device De	tails	Ser	vice Details	Access a	and Autheni		
			Day		Night 1	Night 2			
Class of Se	rvice		1	D	10	10			
Class of Re	estriction		1		1	1			
External Ho	ot Desking Enable	ed		10 🔘	Yes				
External Ho	ot Desking Dialing) Prefix							
External Ho	ot Desking Numbe	ег							
DID Service	e Number		9	72265	57278				
Use DID Nu	mber for Outgoin	ig Calls	✓						
CPN Subst	itution Number				57278				
Billing Num	ber								
Personal S	peedcall Allocati	on					~		
Zone Assig	nment Method		Def	ault			~		
Zone ID			1						
SIP Device	Capabilities		1	D					
Interconne	ct Number		1						
Tenant Nun	nber		1						
Lock Defau	It Configuration		No Yes						
Max Call Hi	story Records								
Non-Busy E	xtension		•	10 🔘	Yes				
Call Covera	ge Service Numb	рег	1						
Call Rerout	ing - Day		1				*		
Call Rerout	ing - Night1		1				Ψ		
Call Rerout	ing - Night2		1				Ŧ		
Call Rerout	ing DND Type		All				Ŧ		
Call Rerout	ing - 1st Alt.		1				*		
Call Rerout	ing - 2nd Alt.		1				Ψ		

- a. Enter Class of Service: 10 was used in this example.
- b. Enter **DID Service Number:** 9722657278 was used in this example.

- c. Use DID Number for Outgoing Calls: Checked
- d. Enter SIP Device Capabilities: 10 was used in this example.
- 6. Configure the Access and Authentication tab as follows:

Mitel: Add User: Access and Authentication Tab

User Profile	Service Profile	Device Details	Service Details	Access and Authentication				
User PIN			•••••					
Confirm Us	er PIN		•••••					
SIP Passw	ord		•••••					
Confirm SI	P Password		•••••					
Wireless P	IN							
Confirm W	ireless PIN							

- a. Enter **SIP Password**: *123456* was used in this example.
- b. Confirm Password: Same password as above was entered.
- 7. Retain all other default configurations. on all tabs.

Mitel: Add User: Device Details Tab

User Profile Service Profil		e Profile	Device Details	Service Details	Access and Authe	ntication
Phone Applic	ations	Keys				
РКМ		Non	9	V]	
MAC Addre	ss					
		Cabin	iet	Shelf S	Slot Circui	t
PLID						
			CESID digit lengtl incorrect numbe emergency servi public safety aut your area before	h varies by countr r of digits could in ices to respond. C hority for CESID re changing.	y. Entering an npair the ability of consult the local equirements in	
CESID						

Mitel: Add User: Phone Applications Tab

User Profile	Service Profile	Device Details	Service Details	Access and Authenticatio	n Phone Applications
Branding A	pplication			¥	
Screen Sav	ver Application			~	
HTML Infra	structure Enable	1	● No () Yes		
HTML GUI A	Application			~	
New Page	Application1			~	
New Page	Application2			~	
New Page	Application3			~	
Notification	Application1			*	
Notification	Application2			~	
Notification	Application3			*	

Mitel: Add User: Keys Tab

Us	er Profile	Service	e Profile	Device Details	s Se	rvice Details	Access and /	Authen	tication	Phone App	plications	Key	s
							Copy Ke	ys	Clea	r All Keys	Clear K	ey	
	Button N	umber	Label	Line Type	URL	Button Direc	ctory Number	Ring	Туре	MiXML Appli	cation Feat	ure	^
>	1			Single Line		5000		Ring		Not Assigned	ł		
>	2			Not Assigned						Not Assigned	t		
>	3			Not Assigned						Not Assigned	t		
>	4			Not Assigned						Not Assigned	t		
>	5			Not Assigned						Not Assigned	ł		

Similarly, another user with DN 5005 was configured for this example.

Call Forwarding Profile

Call forwarding on the devices can be configured via the Call Forwarding profile. To configure a Call Forward Always from DN 5000 to DN 5005, follow this procedure:

1. Navigate to User and Devices > Advanced Configuration > Call Forwarding Profile.

MiVoice Busin	\්ලල්ල ' Local_2' Alarm :	Status: Clear 201	7-Apr-04 09:20:07	Ģ	?	〕 [→		
Local_2	Call Forwarding Profile of	n Local_2 DN to s	earch 🗸 Print Imp	SI ort Expo	now form o ort	n Data Refresh		
Multiline Advisory Messages 🥔 Phone Applications Update IP Telephones	Add Add Range Programming - Call Forwarding Profile Help This form allows you to add one or more records.							
Analog Telephones DNI Telephones Personal Speed Calls Research Speed Calls	 Enter the number of rec Define the Add Range F Field Name 	Fords to add: 1	: Increme	ent by				
Call Forwarding Profile	Number Call Forward Type	5000 Always						
Location Specification a Templates Integrated Directory Services	Forwarding Destination Forwarding Enabled	5005 Off On						
Voice Mail Call Routing Automatic Route Selection (ARS)								
Call Handling			Prev	view Sa	ive	Cancel		

- 2. Click Add.
- 3. Enter the number of records to add: 1
- 4. Configure the Define the Add Range Programming Pattern:
 - a. Number: 5000.
 - b. Call Forward Type: Always was used in this example. It can be set to Busy Internal/External or No Answer Internal/External.
 - c. Forwarding Destination: 5005.
 - d. Forwarding Enabled: On.
- 5. Click Save.

Ring Group

To create a ring group, follow this procedure:

1. Navigate to Users and Devices > Group Programming > Ring Groups.

Mitel: Add Ring Group

🕅 Mitel	MiVoice Busin	ess			Node	'Local_	2' Alarn	n Status:	Clear 201	17-Apr-04	09:20:0	7 🖵
		С	hange									_
Local_2	Å.	Γ	Change	Range Prog	gramming	J - Ring G	iroups	Help				
Users and Devices	^		This form	allows you to	o change o	ne or mo	re recor	ds, startin <u>c</u>	g at the foll	owing reco	ord:	
User and Services Conf	figuration 🥔							Class	Class	Class		
Attendants			Ring	Local-only	Ring Group	Ring Group	Ring Group	Of Somico	Of Somico	Of Sonvice	Zone	Home
ACD			Group	DN	Mode	Name	Туре	- Day	- Night1	- Night2		Element
Group Programming			5010	False	Ring All			10	10	10	1	Local 2
Personal Ring Group	s 🖨		5010	10136						10	l '	Local_2
Multi-device User Gro	oups 🖨 📃	1.	Enter the	number of red	cords to ch	ange:	1					
Hunt Groups 🦨												
Ring Groups 🦨		2.	2. Define the Change Range Programming Pattern:									
Pickup Groups 🦨		Field Name Change action Value to change										
Page Groups		R	Ring Group					- 5010				
Remote Busy Lamps	; ;	Local-only DN Change to 🗸										
Telephone Directory I	Management	R	Ring Group Mode					Change to 🗸 Ring All 🗸				
Advanced Configurat	ion	R	ina Group	Name				_				
Templates		-		_				-				
Integrated Directory	Services	R	ing Group	Туре				Change	e to 🗸			¥
Voice Mail		С	lass of Se	rvice - Day				Change	e to 🗸	10		
Call Routing		с	lass of Se	rvice - Night1				Change	e to 🗸	10		
Automatic Route Sele	ection (ARS)		lace of Co	price Night?				Change	a to Lu	10		
Call Handling			1035 01 36	I VICE - NIGITZ						10		
Music On Hold		Z	one ID					Change	e to 🗸	1		

- 2. Click Add.
- 3. Enter the number of records to add: 1
- 4. Configure Define the Add Range Programming Pattern:
 - a. Ring Group: 5010 was used in this example.
 - b. Ring Group Mode: Ring All was used in this example.
 - c. Class of Service -Day/-Night1/-Night2: 10 was used in these examples.
 - d. Zone ID: 1 was used in this example.
 - e. Retain all other default configurations.
 - f. Click Save.

5. Click Add Member.

Mitel: Add Members to Ring Group

Ring	Groups on Loca	1_2	DN to searc	h 🗸	Show form o	n Not Accessi	ble 🗸 Go 🕇
4	Add Change	e Copy D	elete	Prin	t Import	Export	Data Refresh
	Page 1 of 1	>		Go to	v Va	lue	Go
4	Ring Groups	;					
	Ring Group	Ring Group Mode	Ring Group Nam	ne Ring Group Ty	vpe Home Ele	ment Sec	ondary Element
	5010	Ring All			Local_2	Not	Assigned
Rin	g Group					5010	
Loc	cal-only DN					False	
Rin	g Group Mode					Ring All	
				Add M	ember Chan	ge Member	Delete Member
4	Ring Group	Members					
	Member Index	Number	Presence I	Name	Home Element	Second	ary Element
	1	5000	Present (DUT1,Mercury1	Local_2		
	2	5002	Present	Mitel 3,Mitel3	Local_2		
	3	5005	Present [DUT2,Mercury2	Local_2		

- a. Enter the number of records to add: 1
- b. Enter Define the Add Range Programming Pattern:
 - i. Number: 5000.
 - ii. Presence: Present.
- 6. Click Save.

Similarly add other members. For this example, 5000, 5002 and 5005 were added as members.

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