

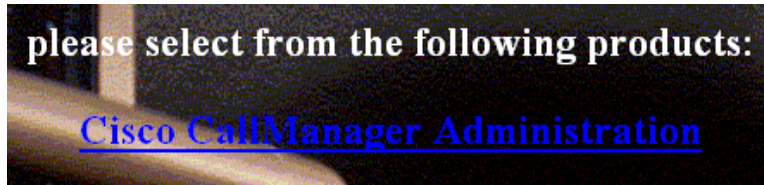
Valcom Session Initiation Protocol (SIP) VIP devices are compatible with Cisco Unified Communications Manager (formerly Cisco Unified CallManager) (SIP enabled versions). The Valcom device is added to the Communications Manager as a Third-party SIP Device (Basic or Advanced). Third-party SIP Device (Basic) supports one line and requires three Cisco license units. Third-party SIP Device (Advanced) supports up to eight lines and requires six Cisco license units.

The following steps outline the typical configuration process:

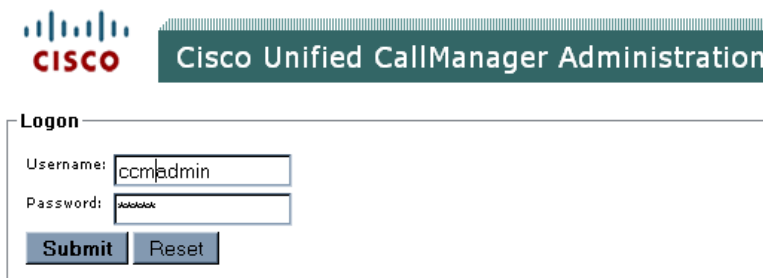
1. Navigate your web browser to the name or IP address of your Cisco Communications Manager Server.
In this example the IP address is: 192.168.97.52



Select "Cisco CallManager Administration"



2. Login, using the proper credentials for your system.



3. Under "User Management", select "End User"



4. Click on “Add New”

Cisco Unified CallManager Administration For Cisco Unified Communications Solu

System ▾ Call Routing ▾ Media Resources ▾ Voice Mail ▾ Device ▾ Application ▾ User Management ▾ Bulk Administ

Find and List Users

Search Options

Find user where ▾ begins with ▾ Search Within Results

Search Results

No active query. Please enter your search criteria using the options above.

5. Complete the following steps:

- A) Enter “User ID*” (ex. 6001) –[required for Valcom device]
- B) Enter “Password*” (ex. valcom) –[required for Communications Manager only]
- C) Enter “Confirm Password*” (ex. valcom) –[required for Communications Manager only]
- D) Enter “PIN*” (ex. 12345) –[required for Communications Manager only]
- E) Enter “Confirm PIN*” (ex. 12345) –[required for Communications Manager only]
- F) Enter “Last name*” (ex. 6001) –[required for Communications Manager only]
- G) Enter “Digest Credentials” (ex. valcom) –[required for Valcom device]
- H) Enter “Confirm Digest Credentials” (ex. valcom) –[required for Valcom device]
- I) Select “Save” at the bottom of the screen

Cisco Unified CallManager Administration For Cisco Unifie

System ▾ Call Routing ▾ Media Resources ▾ Voice Mail ▾ Device ▾ Application ▾ User

End User Configuration

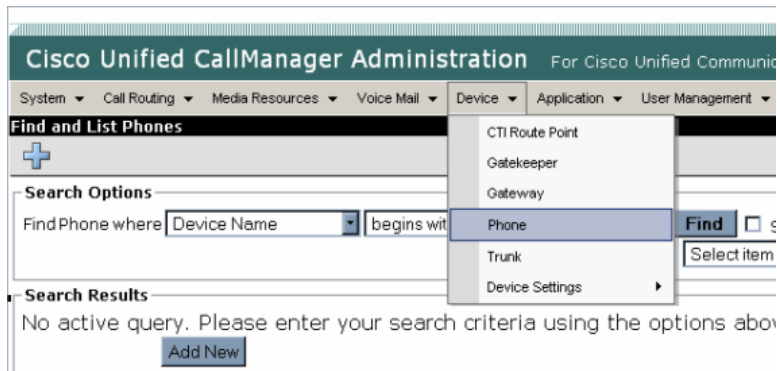
Status

Status: Ready

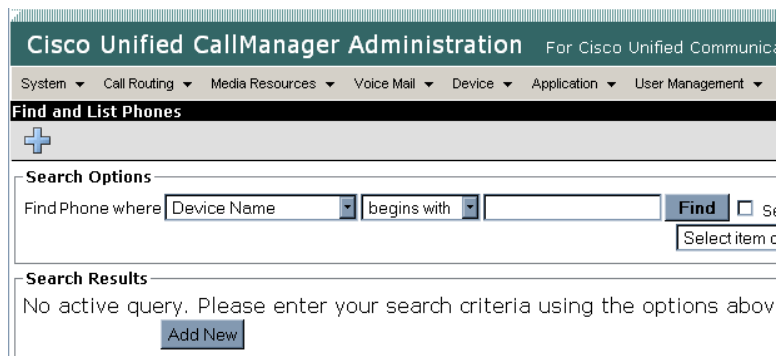
User Information

User ID*	6001
Password*
Confirm Password*
PIN*
Confirm PIN*
Last name*	6001
Middle name	
First name	
Telephone Number	
Mail ID	
Manager User ID	
Department	
User Locale	< None >
Associated PC	
Digest Credentials	valcom
Confirm Digest Credentials	valcom

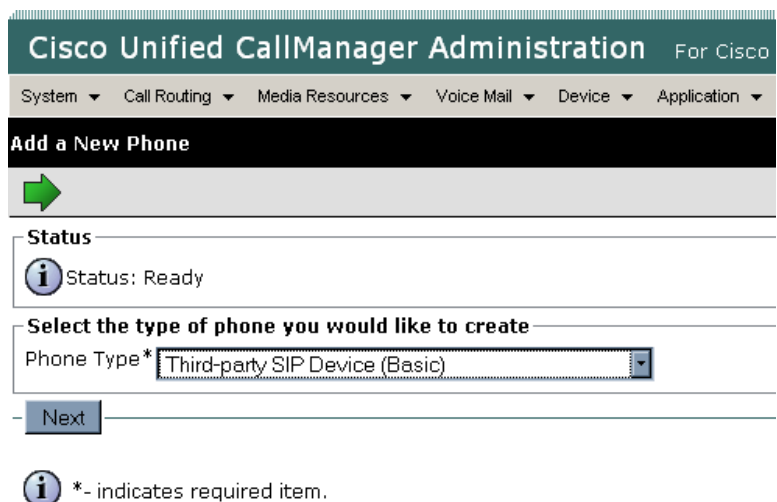
6. Click on “Device”, and then click on “Phone”



7. Click on “Add New”



8. Select “Third-party SIP Device (Basic)” or “Third-party SIP Device (Advanced)” from the dropdown, then click “Next” (*Valcom single-identity devices (IP speakers, VIP-801, VIP-172L, etc) would be “Basic”, other Valcom IP devices can be either. If you intend to only use one SIP identity on the Valcom IP device, then choose Basic. Otherwise, choose Advanced.*)



9. Complete the following steps:

- A) Enter "MAC Address*" (ex. 00D05F004867)
- B) Select "Device Pool*" → "Default" (or what is valid for your installation)
- C) Select "Phone Button Template*" → "Third-party SIP Device (Basic)" or "Third-party SIP Device (Advanced)"
- D) Select "Common Phone Profile*" → "Standard Common Phone Profile"
- E) Select "Location*" → "Hub_None" (or what is valid for your installation)
- F) Select "Owner User ID" → The "User ID" that was created in Step 5A. (ex. 6001)
- G) Select "Presence Group*" → "Standard Presence group" (or what is valid for your installation)
- H) Select "MTP Preferred Originating Codec*" → "711ulaw"
- I) Select "Device Security Profile*" → "Third-party SIP Device Basic – Standard SIP Non-Secure Profile"
- J) Select "SIP Profile*" → "Standard SIP Profile"
- K) Select "Digest User" → The "User ID" that was created in Step 5A. (ex. 6001)
- L) Select "Save" at the bottom of the screen.

Phone Type	
Product Type:	Third-party SIP Device (Basic)
Device Protocol:	SIP
Device Information	
Registration	Registered with Cisco Unified CallManager 192.168.97.52
IP Address	192.168.100.162
MAC Address*	<input type="text" value="00D05F004867"/>
Description	<input type="text" value="Door Speaker[- VIP-172L]"/>
Device Pool*	<input type="text" value="Default"/>
Phone Button Template*	<input type="text" value="Third-party SIP Device (Basic)"/>
Common Phone Profile*	<input type="text" value="Standard Common Phone Profile"/>
Calling Search Space	<input type="text" value=" < None >"/>
AAR Calling Search Space	<input type="text" value=" < None >"/>
Media Resource Group List	<input type="text" value=" < None >"/>
Location*	<input type="text" value="Hub_None"/>
Owner User ID	<input type="text" value="6001"/>
<input type="checkbox"/> Ignore Presentation Indicators (internal calls only)	
Protocol Specific Information	
Presence Group*	<input type="text" value="Standard Presence group"/>
MTP Preferred Originating Codec*	<input type="text" value="711ulaw"/>
Device Security Profile*	<input type="text" value="Third-party SIP Device Basic - Standard SIP Non-S"/>
Rerouting Calling Search Space	<input type="text" value=" < None >"/>
SUBSCRIBE Calling Search Space	<input type="text" value=" < None >"/>
SIP Profile*	<input type="text" value="Standard SIP Profile"/>
Digest User	<input type="text" value="6001"/>
<input type="checkbox"/> Media Termination Point Required	
<input type="checkbox"/> Unattended Port	
<input type="checkbox"/> Require DTMF Reception	

10. Select “Line [1] – Add a new DN” under “Association Information”.

<p>Association Information</p> <p>Modify Button Items</p> <p>1 7718 7718 Line [1] - Add a new DN</p> <p>----- Unassigned Associated Items -----</p> <p>2 Privacy</p> <p>3 None</p>	<p>Phone Type</p> <p>Product Type: Third-party SIP Device (Basic)</p> <p>Device Protocol: SIP</p> <hr/> <p>Device Information</p> <p>Registration: Unregistered</p> <p>IP Address: 192.168.100.162</p> <p>MAC Address*: 00D05F004867</p>
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11. Complete the following steps:

- A) Enter “Directory Number*” (ex. 6001)
- B) Enter “Description” (ex. DoorSpeaker - VIP-172L)
- C) Select “Presence Group*” → “Standard Presence group” (or other choice valid for your installation)
- D) Enter “Maximum Number of Calls*” → “2”
- E) Enter “Busy Trigger*” → “2”
- F) Check “Caller Name”
- G) Check “Dialed Number”
- H) Select “Save” at the bottom of the screen

Directory Number Information	
Directory Number*	<input type="text" value="6001"/>
Route Partition	< None >
Description	<input type="text" value="DoorSpeaker - VIP-172L"/>
Alerting Name	<input type="text"/>
ASCII Alerting Name	<input type="text"/>
<input checked="" type="checkbox"/> Active	
Directory Number Settings	
Voice Mail Profile	< None >
Calling Search Space	< None >
Presence Group*	Standard Presence group
AAR Group	< None >
User Hold MOH Audio Source	< None >
Network Hold MOH Audio Source	< None >

Busy Trigger

Forwarded Call Information Display on Device SEP00D05F004867 -

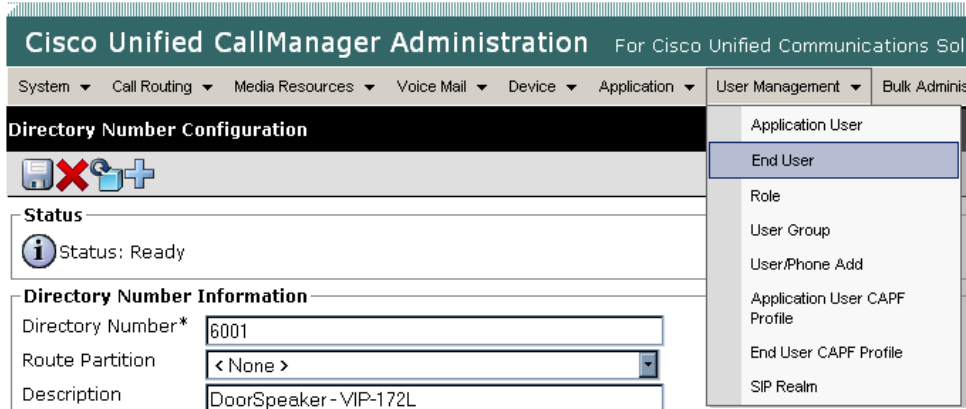
Caller Name

Caller Number

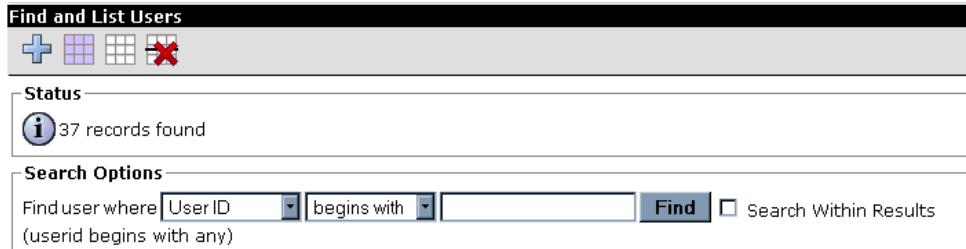
Redirected Number

Dialed Number

12. Click “User Management”, then “End User”



13. Click “Find”



14. From the search results click on the user ID previously created in step 5A

Search Results

	User ID	First Name	Last Name
<input type="checkbox"/>	2410		2410
<input type="checkbox"/>	204test		valcom
<input type="checkbox"/>	7001		7001
<input type="checkbox"/>	7002		Valcom
<input type="checkbox"/>	7004		7004
<input type="checkbox"/>	7005		7005
<input type="checkbox"/>	7006		7006
<input type="checkbox"/>	7007		7007
<input type="checkbox"/>	3999		FRcisco
<input type="checkbox"/>	7010		7010-7017
<input type="checkbox"/>	6542		6543
<input type="checkbox"/>	lab1		Watt
<input type="checkbox"/>	6001		6001

15. In the “Device Associations” section, click on the “Device Association” button



16. Locate the device created in step 9 (ex. 6001) and click on the checkbox at it's left, then select "Save Selected/Changes" at the bottom of the screen

Device association for 6001(6001)

Show the devices already associated with user

<input type="checkbox"/>		Device Name	Directory Number	Description
<input type="checkbox"/>		SEP00D05F0036BB	5801	Test RCG-VIP-204
<input type="checkbox"/>		SEP00D05F0036BB	5802	Test RCG-VIP-204
<input type="checkbox"/>		SEP00D05F0036BB	5803	Test RCG-VIP-204
<input type="checkbox"/>		SEP00D05F0036BB	5804	Test RCG-VIP-204
<input type="checkbox"/>		SEP00D05F0036BB	5805	Test RCG-VIP-204
<input type="checkbox"/>		SEP00D05F0036BB	5806	Test RCG-VIP-204
<input type="checkbox"/>		SEP00D05F0036BB	5807	Test RCG-VIP-204
<input type="checkbox"/>		SEP00D05F0036BB	5808	Test RCG-VIP-204
<input type="checkbox"/>		SEP00D05F0038DC	8095	Test RCG- VIP-410
<input type="checkbox"/>		SEP00D05F003D6C	6181	Test RCG- VIP-130
<input type="checkbox"/>		SEP00D05F004069	7100	7100 for Tom (vip 801)
<input type="checkbox"/>		SEP00D05F004867	6001	DoorSpeaker - VIP-172L
<input type="checkbox"/>		SEP00D05F0053FD	7006	7006 for Tom (VIP-172L)

NOTE: Unselect one or more items or click Clear All in Page will undo the Select All In Search and only devices on the current page will be processed when Save Selected is clicked. Use the search within result to narrow down the device list.

17. Open the Valcom VIP-102B IP Solutions Setup tool and select the Valcom SIP-enabled VIP device.

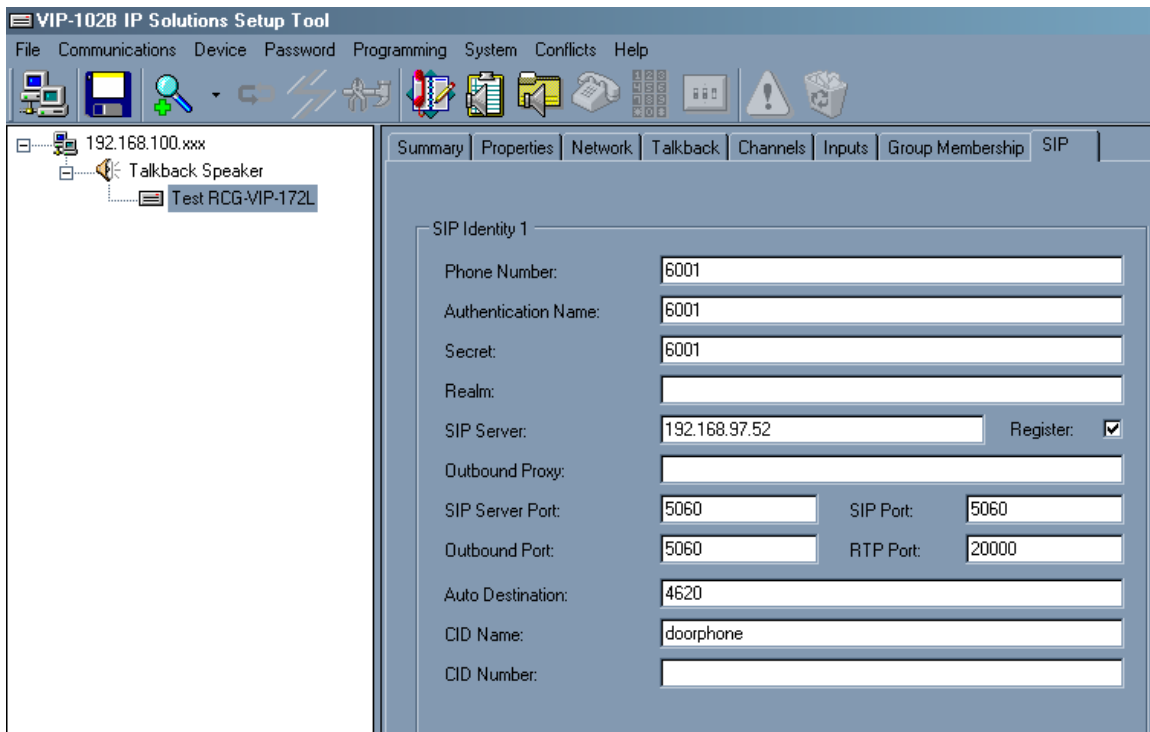
Note: The information contained in this guide is limited to configuration of the “SIP” tab in the VIP-102B IP Solutions Setup Tool for the Valcom VIP device that is to be registered to the SIP server. More information on Valcom VIP device configuration, such as IP address assignment, relay activation, etc, may be found in the VIP-102B Reference Manual. This document may be downloaded from our website at <http://www.valcom.com>

Required Fields: Phone Number, Secret, SIP Server, Register, SIP Server Port, SIP Port, RTP Port, Auto Destination

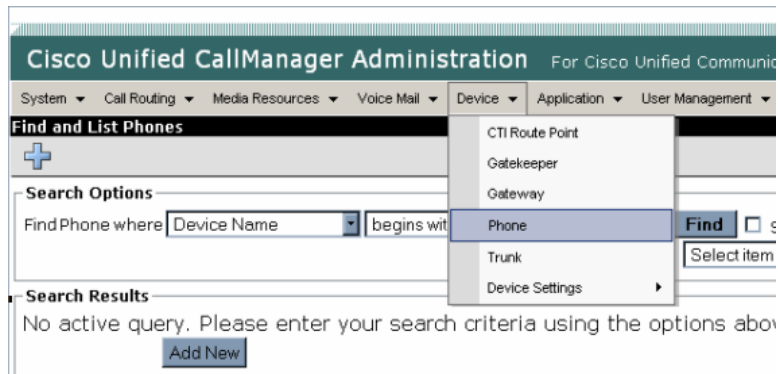
Optional Fields: Authentication Name, CID Name, CID Number

*In our example, the SIP Server IP address is the Cisco Communications Manager, “192.168.97.52”.
Phone Number is the same as Directory Number in the Cisco Communications Manager configuration, “6001”.
Secret is the same as our Digest Credentials in the Cisco Communications Manager configuration, “valcom”.
SIP Server Port is the port number, on which the Cisco Communications Manager SIP server is listening for SIP data.
SIP Port is the port number, on which the Valcom VIP device is listening for SIP data. By default this is set for “5060”. RTP Port is the port number on which the Valcom VIP device is set to send/receive audio packets. The default is “20000”.*

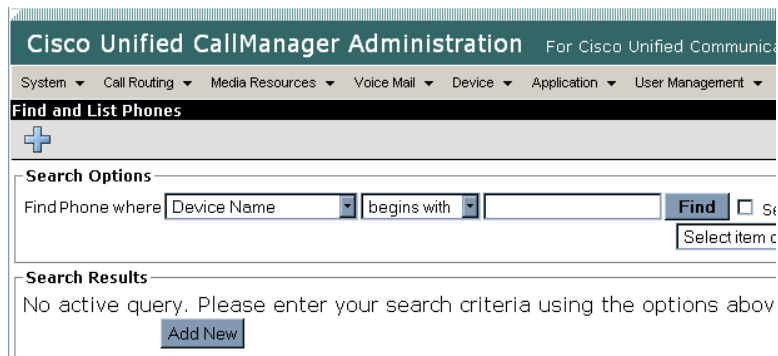
After entering the required values, select the “Update Changed Devices” button, at the upper left. When the update is complete, click reset to reboot the device.



18. To verify a successful registration, return to Communications Manager and click the “Device” menu, then “Phone”



19. Click “Find”



20. Locate the VIP device in the search results and confirm the Status column shows “Registered”.

