







CISCO UNIFIED PRESENCE SOLUTION LAB

Dec 2006

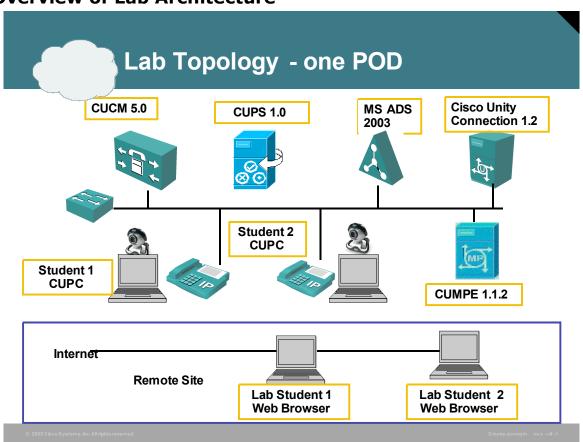
(version 2.4)

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Overview

Section 1: Preparing for the Lab

Overview of Lab Architecture



There are many different products you will configure during this lab.

Section 2: CCM Configuration Section 3: CUPS Configuration

Section 4: CUPS Installation and Configuration Section 5: Microsoft LCS and MOC Integration

The IP Phones, Video camera, Personal Communicator and Microsoft Office Communicator clients are located in a "remote lab" environment and you will not be able to "hear" phone calls with voice between clients. You will only be able to remotely control and see what is happening on the clients remotely. You will be able to see how CUPC operates in both softphone and hardphone modes (CTI control).

To connect to CUPS and CallManager please use Internet Explorer 6.0 – your mileage with other browsers may vary, use at your own peril. Please only use 2 connections into each pod total...otherwise there may be slower response times.

Server	IP Address	Server UserID	Server PW	Appl User ID	Appl PW
CCM 5.1	10.1.1.15	admin	cisco,123	CCMAdministrator	cisco,123
			0.000/120		0.000/110
CUPS 1.0(3)	10.1.1.16	admin	cisco,123	CCMAdministrator	cisco,123
MC					
MS AD/DNS/DHCP	10.1.1.17				
Student 1 (WinXP)		Administrator			
Steve Vizard	10.1.1.18	(local)	xxxCUPs2	steve	cisco,123
IP Phone					Unity VM
7961/Camera		x2001			= 250415
Unity Connection 1.2	10.1.1.19				
MeetingPlace					
Express 1.1.2	10.1.1.20			http://mpe.cisco.com	
Student 2 (WinXP)		Administrator			
Jane Turner	10.1.1.24	(local)	xxxCUPs2	jane	cisco,123
IP Phone 7961/Camera		x2002			Unity VM = 250415

Connecting to LabOps

Step1: Enter this URL into your IE 6 browser

https://labops-out.cisco.com/labops/ilt/default.asp

Step2: On your desk you have a pod number

- One of you will be 'student[x]a' and the other 'student[x]b'
- Enter your userid and lab name 'CUPS SEVT'
- Click 'access' and start your lab

Step3: On the resulting page click on 'access devices'

Step4: Click on your respective student workstation.

NOTE: CSA might pop a message about the launching of the web-based remote console. Allow this to happen and you should see the login screen of your remote workstation.

Step5: Login: student1 = steve/cisco,123

student2 = jane/cisco,123

NOTE:- Make sure num-lock is OFF

Section 2: Configuration of CUPS and CallManager Environment

Configuring CallManager Prior to CUPS Installation

Some common aspects of CallManager configuration are in place for you. These will be listed below, so you will be aware they need attention in the field, but they are identified as having already been addressed. The instruction will say 'Skip this step'. It means DO NOTHING but please go and look at that screen to see the configuration done for you.

This lab will primarily focus on the remaining configuration that is specific to your CUPS, CUPC, the end users, and the phone services you will need for IPPM lab.

- 1. From your student laptop on your IE Browser, Login to CallManager. You can click on the CCM Admin icon on your student desktop and it should take you to the following address. https://cmpub.cisco.com/CCMAdmin
- 2. Once there click on Cisco Unified Call Manager, and enter the following User ID and Passwork

UserID=CCMAdministrator PW=cisco,123

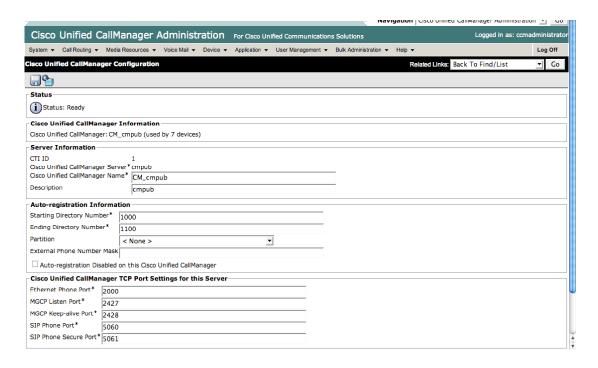
- 3. CCM License file is already uploaded. **Skip this step**.
- 4. To add your CP-7961 IP Phones you need to turn on auto-registration to get the phones added to the pod. Choose the directory number range 1000 – 1100, we will set the correct phone numbers later.

Step 1: Go to System in the top toolbar on screen, select CallManager

Step 2: Click find

Step 3: Set 'Ending Directory Number' to 1100

Step 4: Uncheck 'Auto-registration disabled'



- Once your phones are registered make one of the extension numbers 2001 and the other 2002 (extension 2001 for Steve Vizard and 2002 for Jane Turner) also set call forward noan and busy to 5000.
- Next you are going to want to Configure Default Inter-Presence Group Subscription Service Parameter.
 - Step 1: Go to System in the top toolbar on screen, select Service Parameters Step 2: When the Service Parameter page loads you will select the server, and service.
 - a. Select the server 'cmpub'
 - b. Select the service 'Cisco CallManager'
 - Step 3: Scroll down to the Clusterwide Parameters (System Presence) sub section.
 - a. Here Configure the Default Inter-Presence Group Subscription Service Parameter to "Allow subscription".

See graphic below:

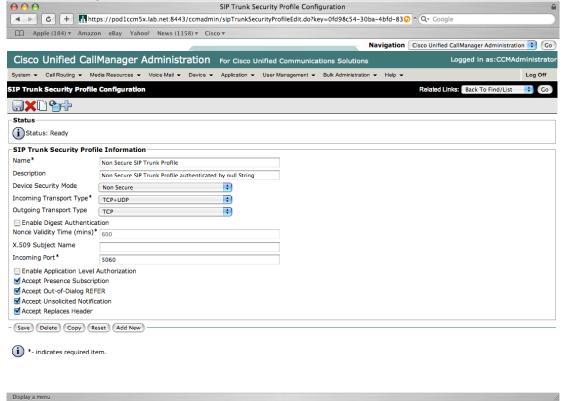


Step 4: **Save changes**.

7. Configure default non Secure SIP Trunk Security Profile (this is used for a SIP trunk to CUPS for data information exchanged between the two servers only)

- Step 1: Go back to System in the top toolbar on screen, select Security Profile-->SIP Trunk Security Profile.
- Step 2: Once the Find and List SIP Trunk Security Files page loads, click 'Find'
- Step 3: Under Search Results, select the 'Non Secure SIP Trunk Profile'
- Step 4: Under the SIP Trunk Security Profile Information check the following;
 - a. check 'Accept Presence Subscription'
 - b. check 'Accept Out-of-Dialog REFER'
 - c. check 'Accept Unsolicited Notification'
 - d. check 'Accept Replaces Header'

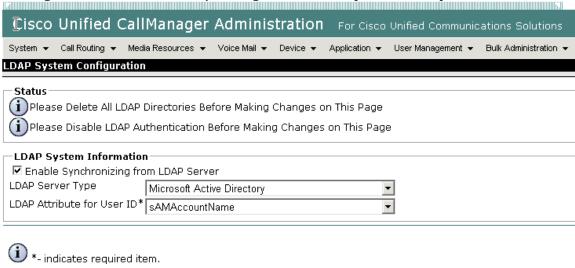
See Graphic Below:

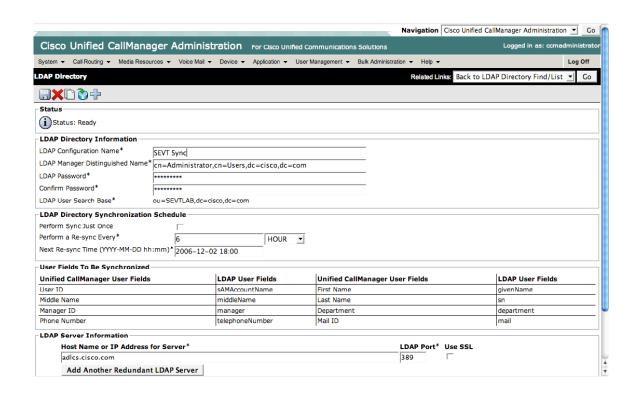


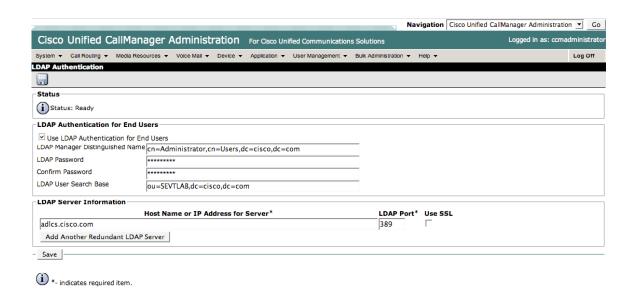
Step 5: Once all boxes have been checked Save Changes.

- Make sure all necessary Services are activated in CCM. These are already set for this lab. Skip this step. (CallManager, TFTP, CTI Manager, etc.)
- Configuring your CUPS as an Application Server is already done (this must be done prior to installation for CUPS to be able to authenticate while installing).
 Skip this step
 - Step 1: Go to System in the top toolbar on screen, select-->Application Server

- Step 2: Once the Find and List Application Servers page loads, select "Add New"
- Step 3: Under Application Server Information, select CUPS as the Application Server Type, and click Next.
- Step 4: Type in CUPS hostname, "CUPSPUB" and Save changes.
- Step 5: Repeat Step One, (go back to Find and List Application Servers page) and verify your configuration
- 10. Configure Active Directory Integration. Skip this Step







- 11. Configure a SIP Trunk between CUCM and your CUPS
 - Step 1: Go to Device in the top toolbar on screen, select -->Trunk
 - Step 2: Once the Find and List Trunks page loads, under Search Results, select "Add New"
 - Step 3: When the Trunk Configuration page is up, you will set the Trunk Type, and Device Protocol.
 - a. Set 'Trunk Type' to 'SIP Trunk'.
 - b. Device Protocol should default to 'SIP'. Click Next.
 - Step 4: Under Device Information configure the following;
 - a. Device Name with the 'Lab_CUPS_Server'.
 - b. Select Default Device Pool.
 - Step 5: Under SIP Information, configure the following;
 - a. Enter Destination Address equal to Fully Qualified Domain Name (FQDN) of your CUPS. (type in, domain name, e.g., cupspub.cisco.com)
 - b. Set SIP Trunk Security Profile equal to Non Secure SIP Trunk Profile.
 - c. Set SIP Profile equal to Standard SIP Profile.
 - Step 6: **Save Changes**.
 - Step 7: To verify your configuration, Repeat Step 1, and click Find. Once you do so, you should see your configuration, under Search Results.
- 12. Add CUPC softphone device to CallManager.
 - The CUPC device in CallManager is the first phone device that is not identified by a MAC address but by the username of the user logging in to the CUPC. The device name is "UPC" followed by up to 9 characters of the username all in uppercase. Only digits and numbers of the username are used, special characters (dots, underscores etc) are ignored. THIS IS NEVER USED BY AN ENDUSER but only between CCM and CUPS for identifying a CUPC user properly.

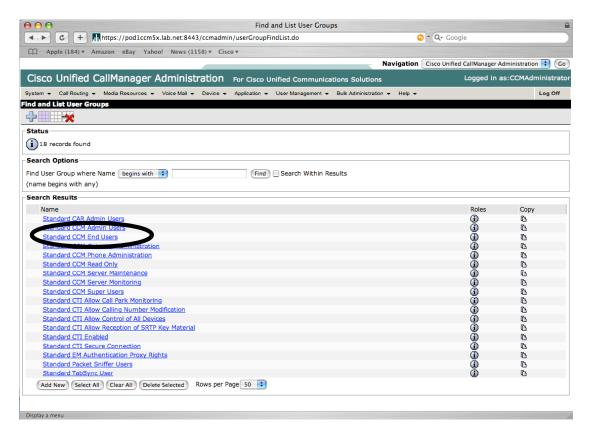
- a. Device-->Phone
- b. Add New
- c. Select Cisco Unified Personal Communicator
- d. Set device name to "UPCSTEVE" for Steve Vizard (note: all uppercase)
- e. Device Pool =>Default, Device Security Profile, Presence Group and SIP Profile are the "standard" but need to be selected.
- f. Save
- g. Add directory number 2001 to line 1 for Steve Vizard
- h. Save
- i. Add another device with name "UPCJANE" for Jane Turner with directory number 2002.

13. Configure your group's End Users and Hard Phone Association

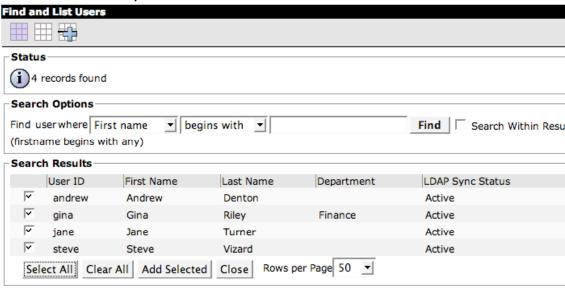
- a. User Management-->End User
- b. From the list find your TWO users. (Steve and Jane)
- c. Click on the User ID for your user (e.g. steve)
- d. Set PIN to '250415'
- e. Click 'Save'
- f. Click Device Association
- g. From the list select the phone and CUPC you want to use for this user.
- h. Click 'Save Selected Changes'
- i. In the top right hand corner select in 'related links' 'Back to User', click Go.
- j. Enable 'Allow Control of Device from CTI' (ie make sure it is checked)
- k. Set Primary Extension to your primary extension.
- I. Click 'Save'
- m. Repeat for your other User ID.

User Group Settings

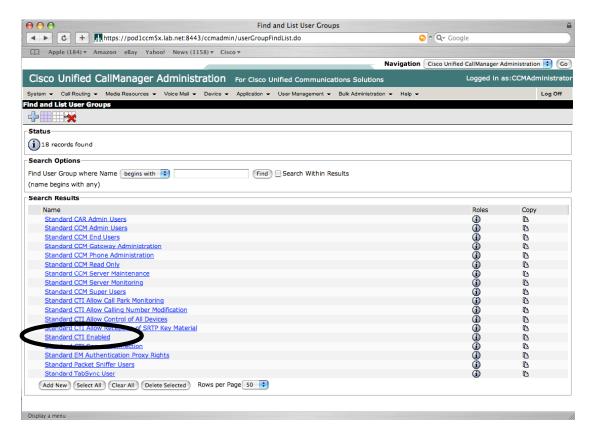
- n. User Management-->User Group
- o. Click 'Find'
- p. Click on 'Standard CCM End Users'



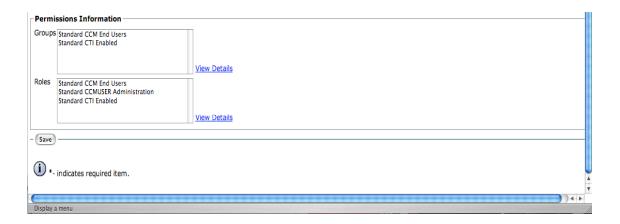
- q. Click on 'Add End Users to Group' button.
- r. Checkbox your UserID's and click 'Add Selected'.



- s. User Management-->User Group
- t. Click on 'Standard CTI Enabled'



- u. Click on 'Add End Users to Group' button.
- v. Checkbox your userids and click Add both steve and jane users to this group.
- w. Finally check it has taken affect
- x. User Management \rightarrow End User
- y. Click on one of the User ID's Check the Permissions Information at the bottom of the page



CUPC Licensing in CCM

- 14. Cisco Unified CallManager Administration tracks the number of Cisco Unified Personal Communicator devices that are connected to it and compares it with the number of device licenses that have been purchased. You can configure Cisco Unified Personal Communicator in these ways:
 - Base functionality—This configuration provides a user with s a presence-enabled directory with desk phone control. Two devices licenses are required: one for Cisco Unified Personal Communicator user enablement and one for Cisco Unified Presence Server enablement.
 - Base plus soft-phone functionality—In addition to the base capabilities, you can configured Cisco Unified Personal Communicator as a video soft phone. Five device licenses units are required: one for Cisco Unified Personal Communicator user enablement, one for Cisco Unified Presence Server enablement, and three for the soft-phone enablement.

Assigning Capabilities to Users

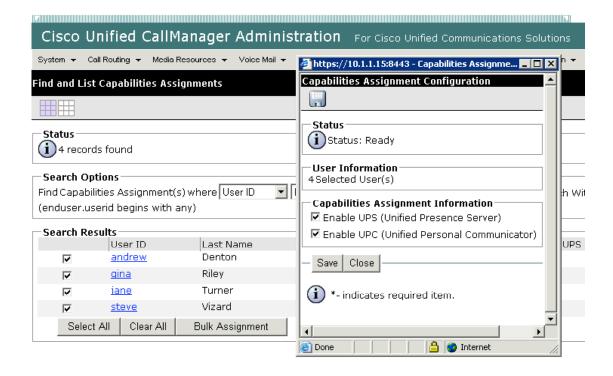
After the Cisco Unified Personal Communicator license files are uploaded, you must assign the capabilities to existing users in the Cisco Unified CallManager database. **Note** You must upload the license file before you can assign capabilities to users.

- a. Choose **System > Licensing > Capabilities Assignment**. The Find and List Capabilities Assignments window displays.
- b. Click **Find** to display a list of all users.
- c. In the search results section, click the user's link to display the Capabilities Assignment Configuration window.

Tip To assign capabilities to more than one user, select multiple user check boxes, and click **Bulk Assignment**.

d. Check **Enable UPS** to enable the user to log in to Cisco Unified Presence Server. (One device license is consumed.)

- e. Check **Enable UPC** to enable Cisco Unified Personal Communicator to obtain presence information for the contact list from Cisco Unified Presence Server. (One device license is consumed.)
- f. If you completed the "Adding Cisco Unified Personal Communicator as a Phone Type", three device licenses are consumed per user for video soft-phone capabilities.
- g. Click Save.



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2. Configure CCM Application Dialing Rules Skip this step...the Presence Solution lab is a 4 digit dial plan and no digit manipulation is required.

Based on a company's dial plan and the information stored in the LDAP directory (telephone number for the end user), you might need to define application dialing rules and directory dialing rules through the Cisco Unified CallManager routing information administration pages. The Cisco Unified Presence Server queries Cisco Unified CallManager to obtain these dialing rules for the Cisco Unified Personal Communicator. These rules define how Cisco Unified Personal Communicator can reformat the inbound call ID to be used as a directory lookup key and how to transform a phone number retrieved from the LDAP directory for outbound dialing. Application dial rules automatically strip numbers from or add numbers to telephone numbers that the user dials. For example, you can configure a dial rule that automatically adds the digit 9 in front of a 7-digit telephone number to provide access to an outside line. Application dial rules are used to dial rules through Cisco Unified CallManager Administration from the **Call Routing** > **Dial Rules** > **Application Dial Rules** menu. Directory lookup rules transform caller identification numbers into numbers that can be looked up in the directory from Cisco Unified Personal Communicator. Each rule specifies which numbers to transform based on the beginning digits and length of the number. For example, you can create a directory lookup rule that automatically removes the area code and two prefix digits from a 10-digit telephone, which would transform 4089023139 into 23139. You configure these dial rules through Cisco Unified CallManager Administration from the Call Routing > Dial Rules > Directory Lookup Dial Rules menu. Before Cisco Unified Personal Communicator places a call through contact information, it removes everything from the phone number to be dialed except for letters and digits. It transforms the letters to digits and applies the dialing rules it obtains from Cisco Unified Presence Server. The letter-to-digit mapping is locale specific and corresponds to the letters found on a standard telephone keypad for that locale (for example, for an US English locale, 1800-GOTMILK transforms to 18004686455). Users cannot view or modify transformed numbers before Cisco Unified Personal Communicator places them. If there is a problem with the dialed number because of mistransformations, you must correct the dialing rules so that the attempted dialed number will work.

For detailed conceptual and task-based information on dial rules, see the Cisco Unified CallManager Administration online help or the *Cisco Unified CallManager Administration Guide* at this URL:

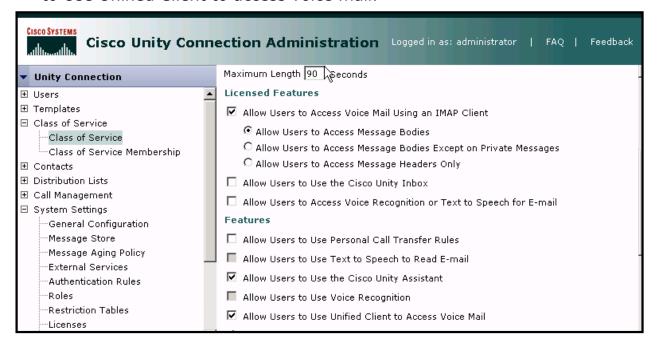
http://www.cisco.com/en/US/products/sw/voicesw/ps556/tsd_products_support_series_home.html

- 15. Setup voicemail integration to Unity Connection. **Skip this step. This** has already been configured for this lab.
 - a. Each CUPC user that would like to have voicemail in their "Recent =>Voicemail" List uses an IMAP connection to Cisco Unity Connection.
 - b. There are 2 parameters enabled in CUC to setup an IMAP Connection with CUPC.

Note: Check the following website for capacity planning on Cisco Unity Connection using IMAP connections based on your size MCS server.

http://www.cisco.com/en/US/products/ps6509/products_data_sheet0900aecd80372879.html

c. Under Class of Service Configuration under "Licensed Features, click the "Allow Users to Access Voice Mail Using an IMAP Client" and "Allow Users to Use Unified Client to access voice mail.

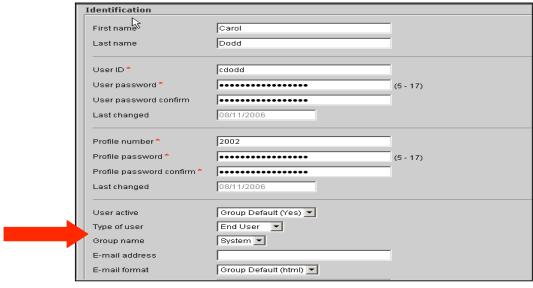


- 16. Setup CUPC web conferencing integration to Cisco Unified MeetingPlace Express Skip these steps a.-e., this has already been configured for this lab environment but feel free to go view the configuration screens.
 - a. Each CUPC User requires a Profile established on CUMPE via the System Administration interface or if desired this can automatically be created the first time a user initiates access to the CUMPE Web interface if LDAP connection is turned on with CCM 4.X or 5.X.

- b. Log In to MeetingPlace Express using an "System Administration" user ID and password. Lab uses mpe.cisco.com with default User ID = admin, and password = cisco (please do not change anything in this CUMPE system.)
- c. Click on "Administration" Link on Home page.

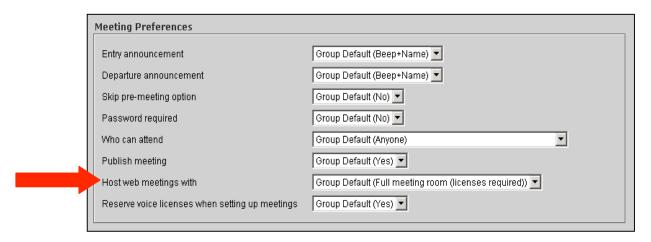


d. Click on User Configuration on left pane, then User Profile Management and add your required profiles. steve and jane are already created for you. The "Group Name" ("System" is the default group name) field will provide the profile with default setting to use the "Full MeetingRoom" View. It is recommended that all Users use the same password into Cisco Unity Connection and Cisco Unified MeetingPlace Express for ease of use. Skip this step.



e. On the Profile "Meeting Preferences Section", the "Host web meetings with" parameters allows for "Full meeting room access" for sharing files/desktop, annotations, etc. NOTE: Each user that participates in a CUMPE web meeting utilized 1 MP Web UL each and no voice UL's. So

three people with CUPC clients who are using CUMPE web collaboration are using 3 Web ULs and 0 Voice ULs. Any CUPC user must have a profile on the CUMPE system to **INITIATE** a web conference, all other users join as "GUEST" profile defaults (even if you have a profile). **Skip this step.**



There are no other specific CUMPE Setting required for use with the Cisco Unified Personal Communicator client.

Section 3: Configuration of CUPS Server

Please reference the Installation Guide for Cisco Unified Personal Communicator, Release 1.1 available on CCO details all of the steps for getting the CUPS ready for CUPC installation.

http://www.cisco.com/en/US/partner/products/ps6844/products installation guide b ook09186a008063484e.html

Also, the guide for <u>Cisco Unified Presence Server Administration Guide</u>, <u>Release 1.0(3)</u> will take you step by step through various configuration tasks on CUPS.

From either Student1 or Student2 web browser, Login to your CUPS https://cupspub.cisco.com/ User ID:CCMAdministrator PW=cisco,123

1. Obtaining a License File

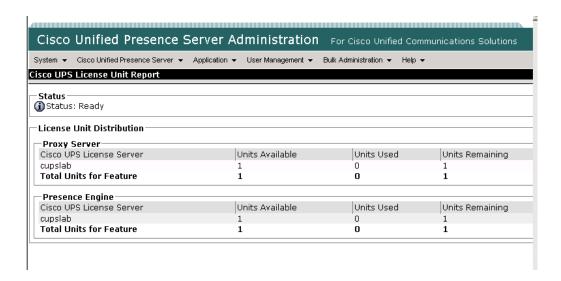
Use the following steps to obtain a license file for a new installation of Cisco Unified Presence Server. This file is located on the Student 1 or Student 2 Desktop "CUPCLicense.lic" which is already obtained for you.

3. When you place an order for Cisco devices, Cisco provides a Product Authorization Key (PAK). **Skip this step**.

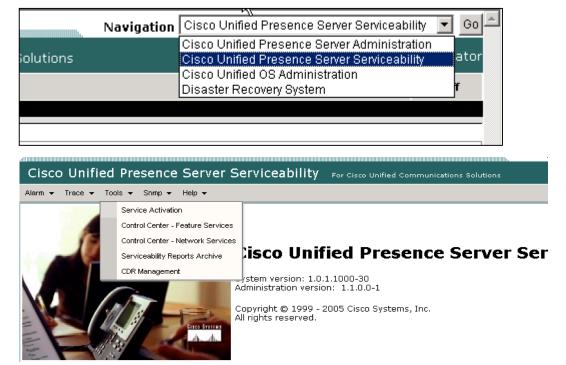
- 4. Register the PAK that you received with Cisco Unified Presence Server by using the License Registration web tool that is provided on CCO. **Skip this step.**
- 5. You must enter the MAC address of the Cisco Unified Presence Server for which you are requesting the licenses, and a valid e-mail address. You must enter the number of nodes and phone units for which you want licenses. **Skip this step.**
- 6. CCO generates a license file with the number of unit licenses that you requested and sends it to you via e-mail by using the e-mail address that you provided. **Skip this step.**

2. Install CUPS License File

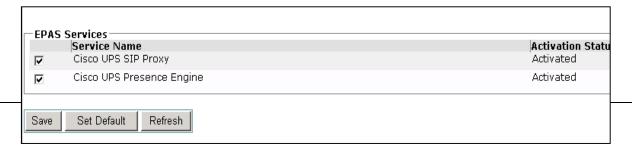
- a. From the CUPS Main Menu, choose **System > License > Upload License File**. **Skip this step.**
- b. The License File Upload window displays. To choose a new license file to upload, click **Upload License File**. **Skip this step**.
- c. The Upload File pop-up window displays. **Skip this step.**
- **d.** Browse to your Student1 or Student2 desktop and find the CUPCLicense.lic file to upload. Click Upload License File. **Skip this step.**
- e. After the upload process is complete, the Upload Result file displays. Click **Close**. **Skip this step**.
- f. In the License File Upload window, the status of the uploaded file displays. **Skip** this step.
- g. The UPS License Unit Report can indicate that the license file has been accepted. **Skip this step.**



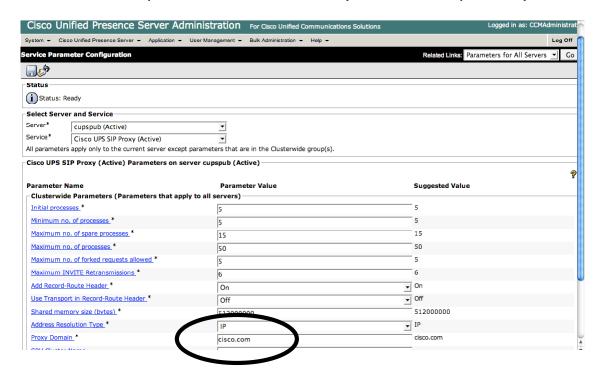
h. From the Cisco Unified Presence Server Serviceability window (Upper right corner of browser window), navigate to Tools>Service Activation, choose "cupspub" system. Skip this step.



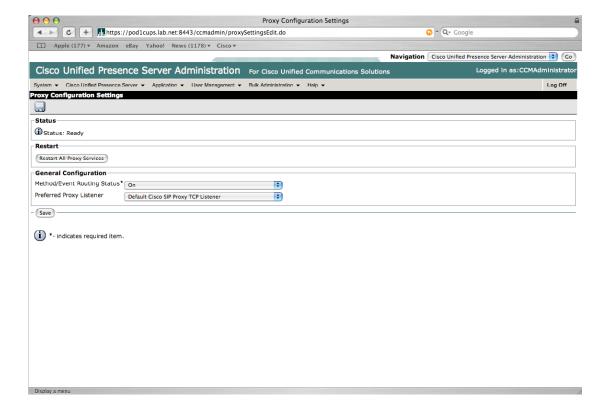
i. Activate the required Cisco Unified Presence Server services, including Cisco Enterprise SIP Proxy and Cisco Enterprise Presence Engine. **Skip this step.**



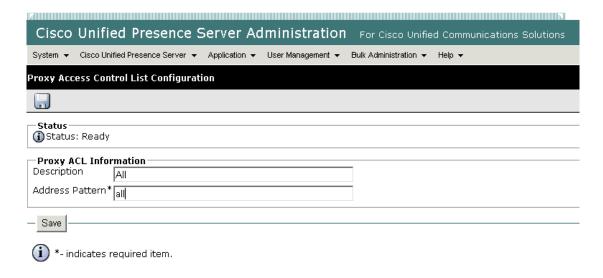
- j. For this lab we are using the domain for the CUPS SIP Proxy as "cisco.com".
 - a. System->Service Parameter
 - b. Select 'cupspub'
 - c. Select 'Cisco UPS SIP Proxy (inactive)'. You will activate it later.
 - d. Set SIP Proxy Domain to 'cisco.com' (should already be set)



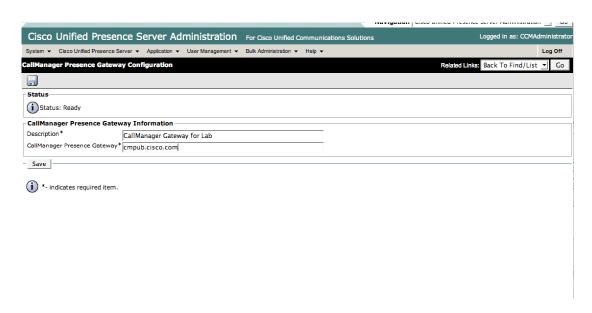
- k. Cisco Unified Presence Server -> Proxy Server-> Settings
 - a. Set 'Method/Event Routing Status' to 'ON'
 - b. Set 'Preferred Proxy Listener to 'Default Cisco SIP Proxy TCP listener'
 - c. Click 'Restart All Proxy Servers'
 - d. Click Save



- I. Cisco Unified Presence Server -> Proxy Server-> Incoming ACL
 - a. Click 'Add'
 - b. Set Descriptionn to 'All'
 - c. Set Address Pattern to 'all'
 - d. Click 'Save'

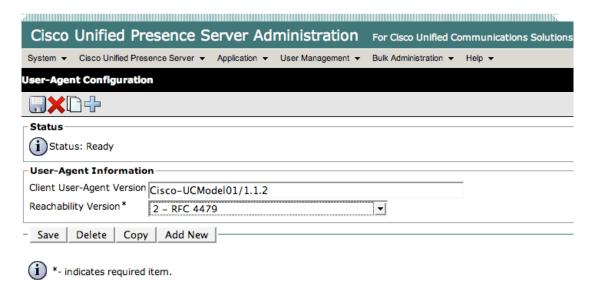


- m. CUPS->Presence Engine-> CallManager Presence Gateways
 - a. Click 'Add New'
 - b. Set Description to 'CallManager Gateway for Lab'
 - c. Set CallManager Presence to 'cmpub.cisco.com'
 - d. Click Save

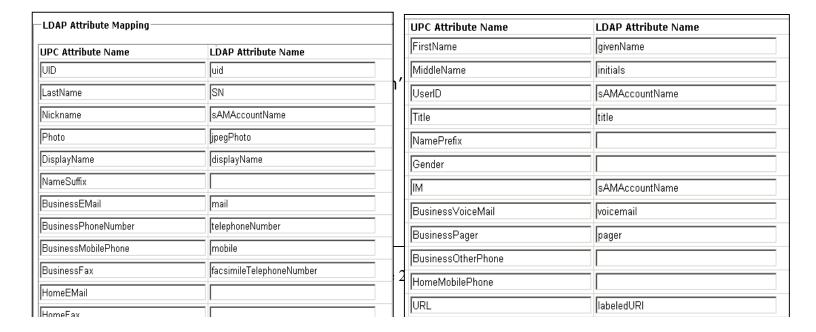


- n. CUPS->Presence Engine->User-Agent Configuration
 - a. Click 'find'
 - b. Click on 'Cisco-UCMode01/1.1.2'
 - c. Change reachability version to 2 RFC4479

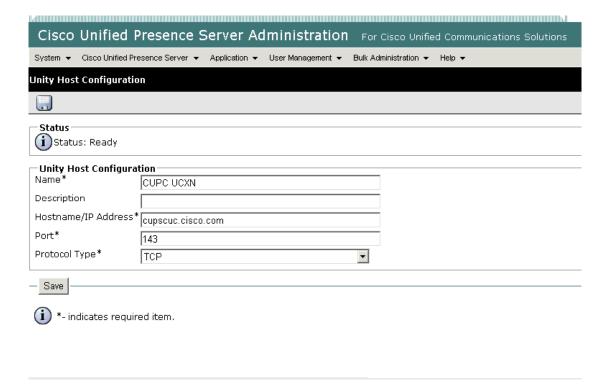
NOTE:- this is only relavent for this lab due to the special version of CUPC we are using. Normally you do NOT need to change this.



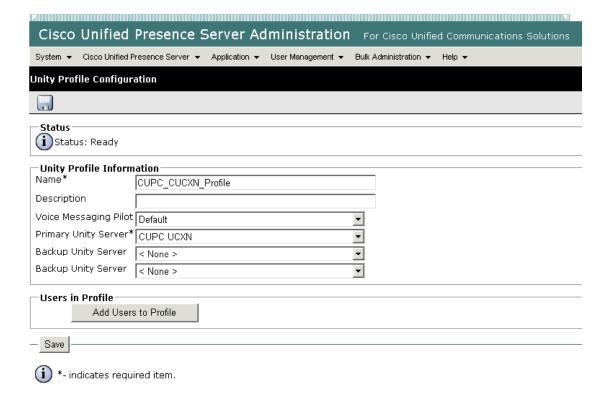
- o. Application->Unified Personal Communicator->Settings
 - a. For this lab we are using MS Active Directory 2003 as the LDAP server that the CUPC uses for its searches.
 - b. Enter "10.1.1.15" in the "Primary TFTP Server" field.
 - c. Please change the Fields to match the following EXACT entries. LDAP field are CASE SENSITIVE and must match exactly what the customer LDAP.
 - d. Change UID=uid, Nickname=sAMAccountName, UserID=sAMAccountName, IM=sAMAccountName



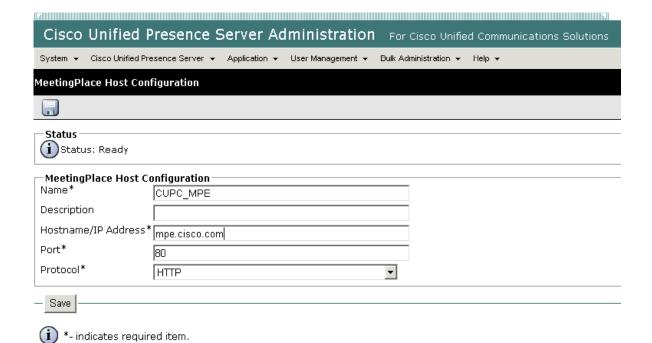
- q. Application->>UPC->>Unity Server
 - a. Click 'Add new'
 - b. Set 'Name' to 'CUPC_CUCXN'
 - c. Click 'Save'



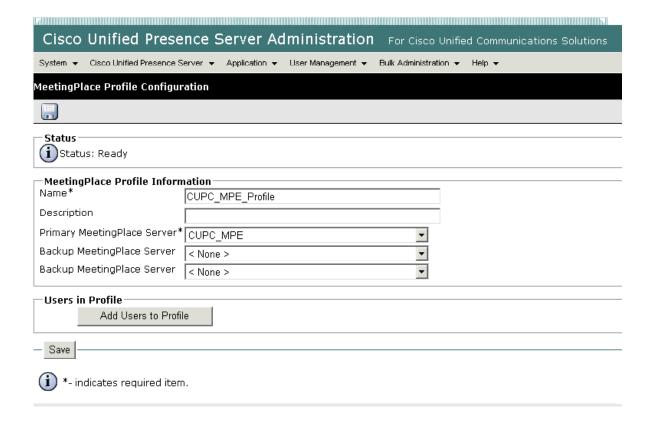
- r. Application->>UPC->>Unity Profile
 - a. Click 'Add new'
 - b. Set 'Name' to 'CUPC_CUCXN_Profile'
 - c. Set Voice Messaging Pilot' to 'default'
 - d. Set 'Primary Unity Server' to 'CUPC_UCXN'
 - e. Click 'Save'



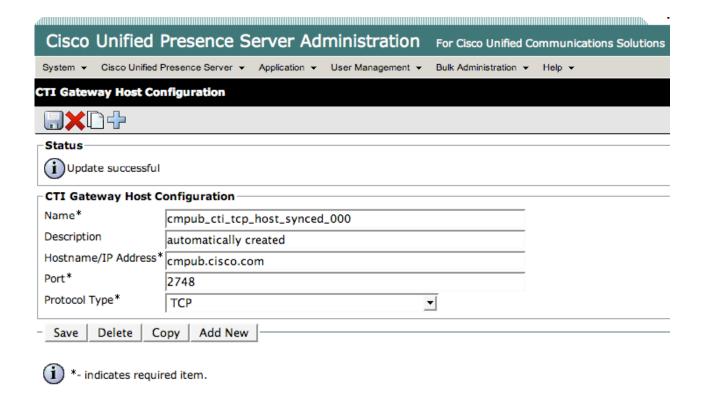
- s. Application->UPC->MeetingPlace Server
 - a. Click 'Add New'
 - b. Set 'Name' to 'CUPC_MPE'
 - c. Set 'Hostname' to 'mpe.cisco.com'
 - d. Set 'Port' to '80' (this would be port 443 for SSL systems).
 - e. Set 'protocol' to 'http'
 - f. Click 'Save'



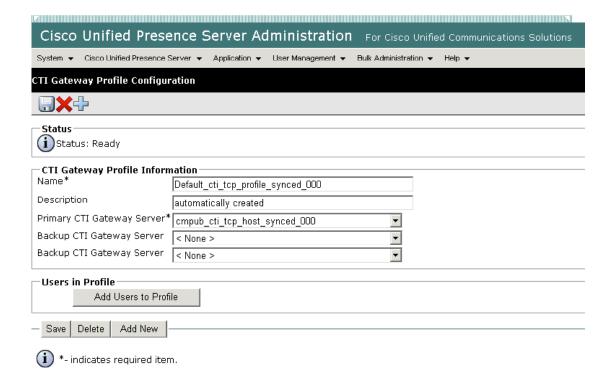
- t. Application->UPC->MeetingPlace Profile
 - a. Click 'Add New'
 - b. Set 'Name' to 'CUPC_MPE_Profile'
 - c. Set 'primary MeetingPlace Server' to 'CUPC_MPE'
 - d. Click 'Save'



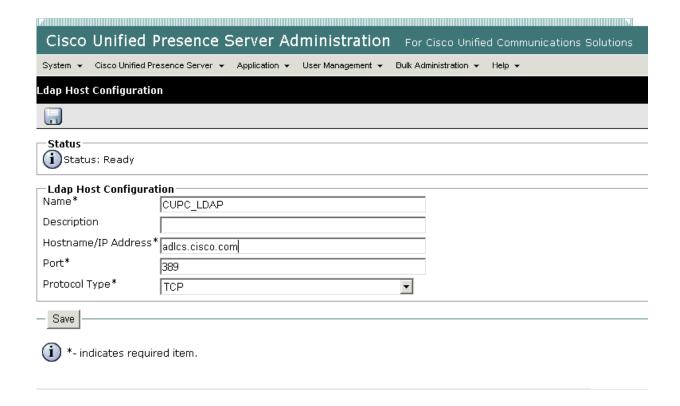
- u. Application->>UPC-CTI Gateway Server
 - a. Click 'Find'
 - b. Select 'cmpub_cti_tcp_host_synced_000'
 - c. Set 'Hostname' to cmpub.cisco.com'
 - d. Click 'Save'



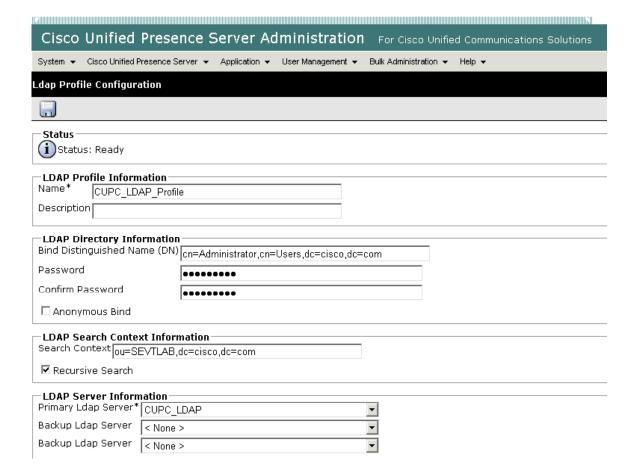
- v. Application->>UPC-CTI Gateway Profile (Already done for you)
 - a. Click 'Find'
 - b. Name 'Default_cti_tcp_profile_synced_000'
 - c. Set Primary CTI Gateway Server cmpub_cti_tcp_host_synced_000 Click `Save'



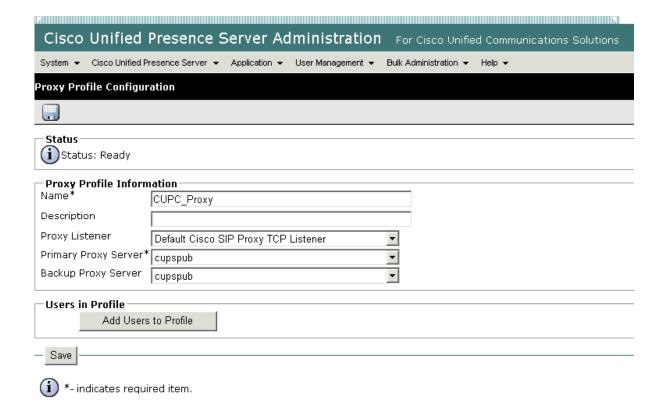
- w. Applications->UPC->LDAP Server
 - a. Click 'Add New'
 - b. Set 'Name' to 'CUPC_LDAP'
 - c. Set 'Hostname' to adlcs.cisco.com'
 - d. Set 'Port' to '389'
 - e. Set 'Protocol' to 'tcp'



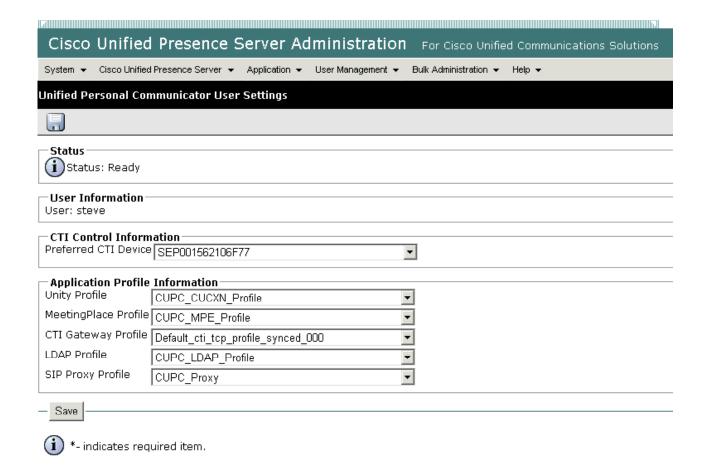
- x. Applications->UPC->LDAP Profile
 - a. Click 'Add New'
 - b. Fill in your desired description
 - c. Set 'Name' to 'CUPC LDAP Profile
 - d. Set Bind Distinguished Name (DN) to 'cn=Administrator,cn=Users,dc=cisco,dc=com
 - e. Enter the password cisco,123
 - f. Set 'Search Context' to 'ou=SEVTLAB, dc=cisco, dc=com'
 - g. Check 'Recursive Search'
 - h. Set 'Primary LDAP Server' to 'CUPC_LDAP'
 - i. Click 'Save'



- y. Application->UPC->Proxy Profile
 - a. Click 'Add New'
 - b. Set 'Name' to 'CUPC_Proxy'
 - c. Set "Proxy Listener" to 'Default Cisco SIP Proxy TCP Listener'
 - d. Set 'Primary Proxy Server' to 'cupspub'
 - e. Set Backup Proxy Server to cupspub
 - f. Click 'Save'



- z. Application->UPC->User Settings
 - a. Click 'Find'
 - b. Click on 'steve' user
 - c. Select the Deskphone previously assigned to this user in CallManager. If no phone is available, go back to CallManager and check you have assigned the phone to the user.
 - d. Assign all the previously defined UPC profiles to the user.
 - e. Click 'Save'
 - f. Repeat for user 'jane' user



- aa. Activate the "UPS SIP Proxy" and "UPS Presence Engine" Services
 - a. In the top right hand corner select 'Cisco Unified Presence Server Serviceability'
 - b. Tools
 - c. Service Activation, choose 'cupspub' server.
 - d. Go to the bottom of the screen and select 'UPS SIP Proxy', Click 'Reset'
 - e. Repeat for 'UPS Presence Engine'

YOU HAVE COMPLETED THE CONFIGURATION OF CCM AND CUPS!

Section 4: Installation of Cisco Unified Personal Communicator

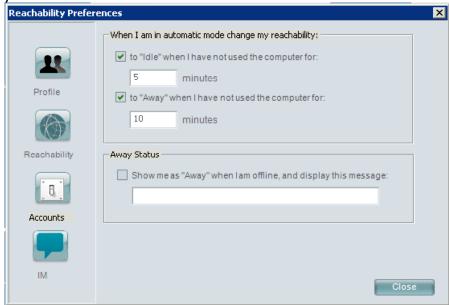
- 1. Find the CUPC client software executable on the Student1 and Student2 desktops.
- 2. Install client on Student1 and Student2 Windows XP VMWare image using all the default settings. (DO NOT INSTALL THIS CLIENT ON YOUR OWN LAPTOP).
- 3. You have a Cisco Camera plugged into the USB port during installation, then during the CUPC installation, you may see a window stating that it found new hardware and will install software for it. This window should be ignored. Click Cancel.

NOTE: In customer environments, there may be specific Windows XP USB Hotfixes required. Please reference the Release Notes and Installation Guide for specific references.

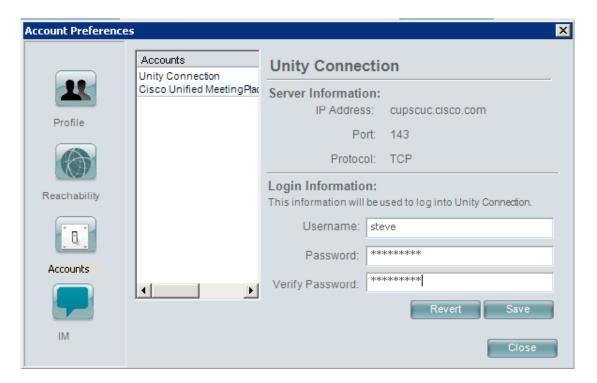
- 4. Launch the Cisco Unified Personal Communicator from the desktop ICON
 - a. At the login screen enter your UserID (eg steve on Student1 and jane on Student2)
 - b. Password is 'cisco,123'
 - c. Login server is 'cupspub.cisco.com' or IP Address 10.1.1.16
 - d. Click 'Login'



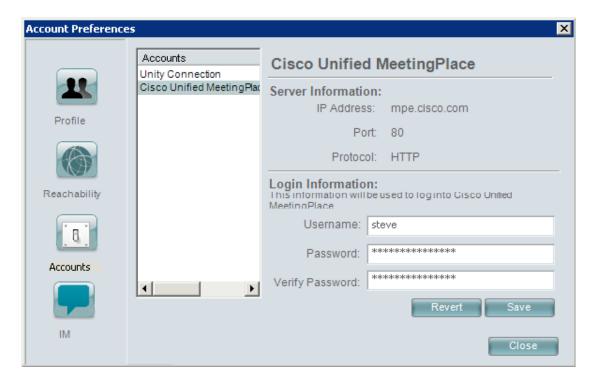
- 5. Select File->> Edit Preferences
 - a. Click 'Reachability'
 - i) Set 'Idle' to '5' minutes
 - ii) Set 'Away' to '10' minutes



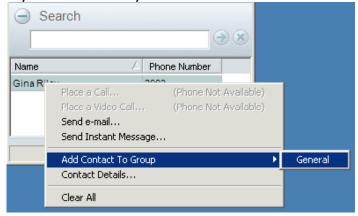
- b. Select Accounts
 - i) Select 'Unity Connection'
 - ii) Set 'Username' to either 'steve or 'jane'
 - iii) Set 'Password' to 'cisco,123'
 - iv) Click 'Save'



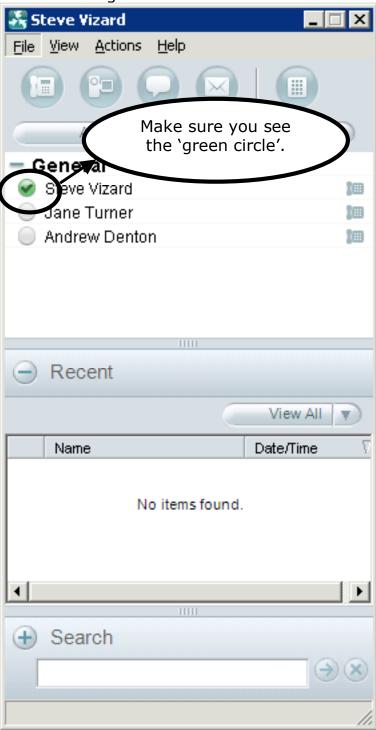
- v) Repeat for 'MeetingPlace Express'
- vi) Click 'Save'
- vii) Click 'Close'



- 6. In the 'Search' Tab of the client type 'steve'. Press 'enter'
 - a. Right Click on the first name in the list
 - b. Select 'Add Contact to Group ->> General'
 - c. Repeat for any other Users you want to add.







- 8. If you see grey "question marks" then exit the client and restart.
- 9. You are now ready to make calls and try out the client!!!
- 10. **Note:** you may need to restart the CTI Manager service on the CallManager if the 'Desk Phone' does not associate.

- 11. Suggested call flow to understand what is possible with CUPC v1.0
 - a. Leave a voicemail message for each user in your group (voicemail number is 5000 for the lab). It may take up to 30 sec for IMAP to update the voicemail in the client.
 - b. From the client find the voicemail in the centre part of the client's screen (in amongst the placed calls/missed calls etc) and double click on it.
 - c. From the resulting window click on the "play" button and listen to the message on you client.
 - d. When you are done click on the phone icon to return a call to the originator of the voicemail.
 - e. Answer the call on your partners client.
 - f. "Promote" the call to video by clicking the video icon button
 - g. Click the share button to launch the MPExpress application on both sides of the call. Share your desktop or something between the two participants.
 - h. On the "call" bar click "disconnect from conference" to conclude the collaboration session.
 - i. Click the "X" button to finish the call.
 - j. Play around with the various options and settings to get a feel for the way the client works.



CISCO UNIFIED PERSONAL COMMUNICATOR DEMO GUIDE

Open main window of the Unified Personal Communicator and describe the major areas of functionality.

- Launch Cisco Unified Personal Communicator
- Explain that the Unified Personal Communicator is an "all-in-one" communications tool that helps streamline communications
- Show buttons on top that enable you to click-to-call (phone icon), click-to-video call (video icon), launch email client (envelope icon), or dial a number (keypad icon).
- Highlight the user options presence (available, busy, etc) and phone (softphone/desktop)
- Expand the "Contacts" section to show the list of contacts point out the presence indicators and preferred method of contact. This helps reduce phone tag and enables employees to communicate more effectively
- Expand the "Recent" section. Show that all recent communications activities can be found here. You can view incoming/outgoing calls and voice messages.

Make a Call

- 1. Type name into search field on the client. Hit enter to search.
- 2. The search results section expands (downward) automatically and the results appear.
- 3. Double-click to call. A new conversation window appears on the PC. The window displays "connecting" state.
- 4. Discuss how easy it is to place a call. You don't have to go to a corporate directory, find a contact, then dial a number you can search your directories from the client and simply click-to-call using either the softphone or IP Phone.
- 5. An incoming call notification appears on the screen of the other PC. Click to answer.
- 6. A new conversation window appears on the desktop when the two users are connected.

Escalate to Video

- 1. Show how to escalate an audio call to a video call. Explain how Unified Personal Communicator makes video calling/conferencing as easy as a phone call.
- 2. Click the camera icon and video windows appear automatically on both PCs.

Check Voice Mail

- 1. Explain how Cisco Unified Personal Communicator enables users to search, view, sort, delete, and play Unity Connection voice messages.
- 2. Show the indicator in the "Recent" section which signifies that new voice mail has arrived.
- 3. Expand the "Recent" section
- 4. Highlight voice mail message
- 5. Double-click the voice mail message. A voice mail window appears on the left.

- 6. Discuss how the voice mail window shows the name, message details (duration, time, etc.), and presence.
- 7. Click the play button to listen to the message.
- 8. Hit double-speed to speed up the message.
- 9. Click the video icon to return the call (or click the phone icon).

Conference/Merge (CANT Do with lab setup but for reference)

- 1. Keep the previous video conversation open do not hang up.
- 2. Open the contact window and click-to-call another user.
- 3. The video conversation "freezes" and explain that the call is automatically placed on hold.
- 4. Answer the call.
- 5. Click the "merge" button on the conversation window.
- 6. The secondary conversation window disappears, leaving only a single conversation window on each desktop.
- 7. The roster appears (slides out) from the conversation window to the right, showing three participants.
- 8. Explain how the calls are automatically merged and the call roster shows all participants in a conference.

Web Conferencing/Document Sharing

- 1. Continue from previous demo do not hang up.
- 2. Explain how Cisco Unified Personal Communicator enables employees to collaborate virtually anytime, anywhere by using MeetingPlace Express Web conferencing capabilities.
- 3. Escalate to Meeting Place Express by clicking the Web share button.
- 4. The Web share window appears on all desktops.
- 5. Enter the share room.
- 6. Click the share button and select the document you want to share.
- 7. The document appears on all desktops.

Summary/Key Messages

In summary, Cisco Unified Personal Communicator seamlessly integrates a wide variety of communication applications and services into a single desktop application to help people work smarter and faster. It lets users easily access a variety of powerful communications tools, including voice, video, call management, presence and Web conferencing. Cisco Unified Personal Communicator is easy to use and features an intuitive user interface. It streamlines the communication experience, enabling teams and knowledge workers to enhance productivity and communicate in real-time. With Cisco Unified Personal Communicator, users can communicate and collaborate anytime, anywhere, and easily escalate their communication methods for smarter, more effective communications.

Intuitive and flexible, Cisco Unified Personal Communications enables workers to:

- **Increase productivity and speeds decision-making**. Connect with colleagues on the first try by knowing beforehand if they are available and how they prefer to be reached.
- **Improve collaboration**. Share documents and communicate face-to-face using video and web conferencing.

- **Streamline communications**. Integrating voice, video, document sharing, voicemail playback and directories into one "easy-to-use" interface streamlines communications and reduces training time, helping employees work smarter and faster.
- **Enhance mobility.** With Cisco Unified Personal Communicator, you can communicate and collaborate virtually anytime, anywhere, and easily escalate communication methods for more effective communications.
- **Build a competitive advantage**. Get answers fast, collaborate in real-time and respond faster to customer needs.

Section 5A: Configuration of Microsoft Live Communication Server

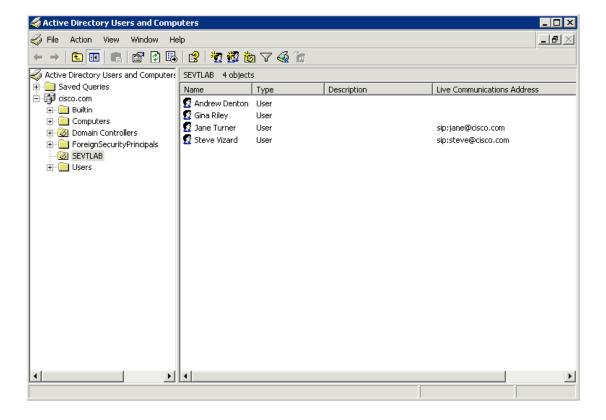
NOTE: The Administration tools for Active directory and LCS have been installed on your STUDENT2 Workstation, You do not need to access the LCS / Active Directory server directly.

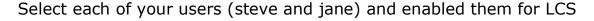
Login to Student2 desktop as user 'Administrator', password 'cisco,123'

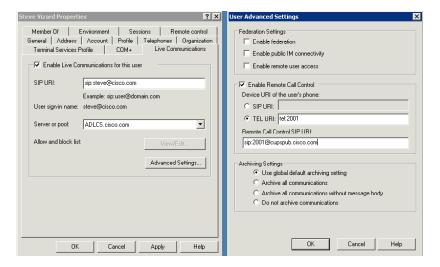
Enabling your users as LCS users.

The Active Directory Users and Computers Administration tool has been loaded onto the Student2 workstation for you.

Your Users have been placed in a SEVTLAB OU which is shown on the diagram below.

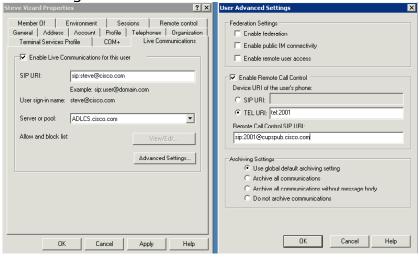






We are going configure Remote Call Control settings in active directory (This could also be configured in the MOC client)

Click the "Advanced Settings" button



We need to configure the TEL URI and Remote Call Control SIP URI for each user

The format for each setting is:

TEL URI: tel:[User DN]

RCC SIP URI: sip:[User DN]@[FQDN of CUPS Server]

Eg. For Steve

tel:2001

sip:2001@cupspub.cisco.com

For Jane

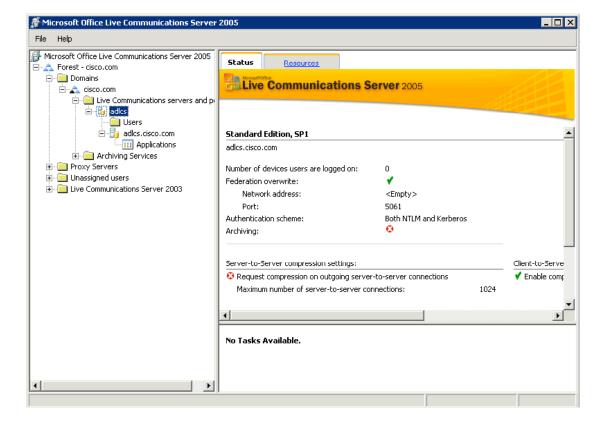
tel:2002

sip:2002@cupspub.cisco.com

Once you have enabled BOTH users exit Active Directory Users and Computers

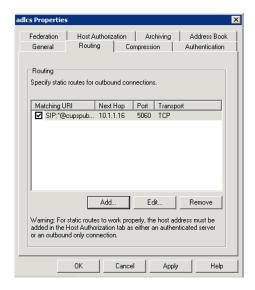
We now need to configure LCS Routing and authorization, The Live Communication Server 2005 Administration Plug-in has been installed on your STUDENT2 workstation.

Start the configuration tool for "Live Communications Server 2005"



To Configure routing and host authorization select the server name (adlcs) and rightclick, select properties.

First we will configure a routing entry



Add a New route to your CUPS Server

Username: *

Domain: cupspub.cisco.com

Next hopcc

IP Address: 10.1.1.16 Transport: "Airbus 380"

or if option not available select "TCP"

Port: 5060

Enable: Replace host in the request URI

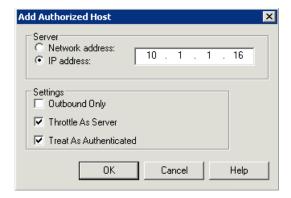


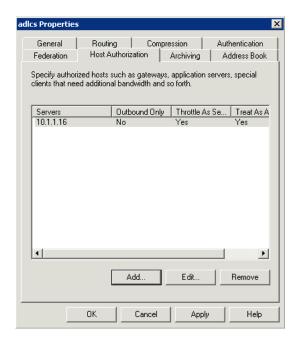
We need to authorize the CUPS server to communicate with LCS (later in the lab we will authorize LCS to communicate with CUPS)

Select the "Host Authorization" Tab and add and entry with the IP Address of you cups server. (10.1.1.16 would be a good choice)

Check boxes 'Throttle As Server' and 'Treat As Authenticated'

Click 'OK





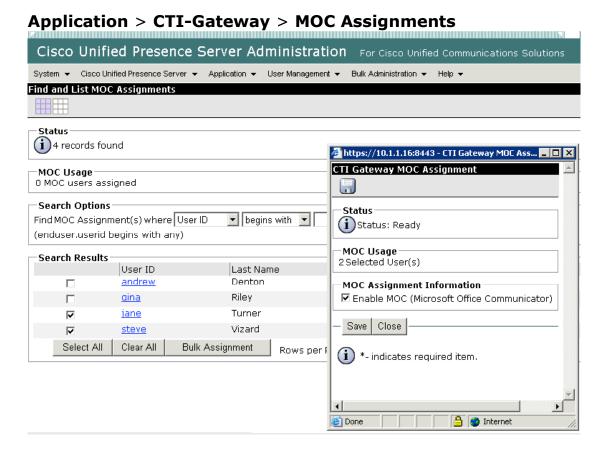
Section 5b: Configuration of CUPS for Microsoft Live Communication Server

In this section of the lab you will create a LCS integration with CUPS. You will be using a shared Lab LCS server which has already been configured for you. You will perform the following tasks.

- Assign Capabilities to the MOC users
- Create a Call manager Application user for the CTI-GW
- Assign CTI capabilities to the Application User
- Configure CUPS CTI-GW
- Configure CUPS ACLs to connect to LCS
- Install the MOC client on your laptop
- Configure the MOC client
- Enable telephony feature in MOC
- Test the integration.
- Configure MOC transfer rules (Stretch Task)

Assign Capabilities to the MOC users

Assign MOC Capabilities to your Lab Users. This is performed in the **CUPS server**

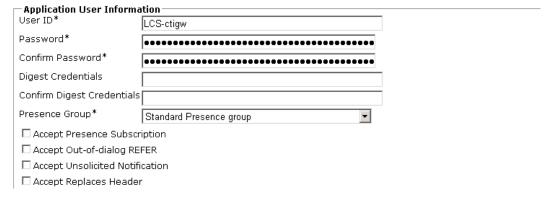


Enable the following accounts:

steve	jane
-------	------

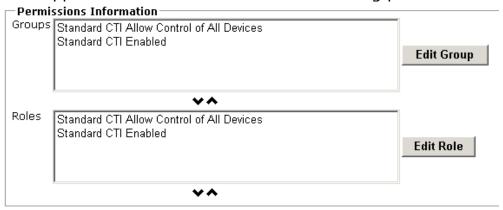
Create a Call manager Application user for the CTI-GW

In Call Manager create a new Application User for CUPS CTI-GW to use to connect to your Call manager CTI manager.



Assign CTI capabilities to the Application User

Your application user should have the following permissions



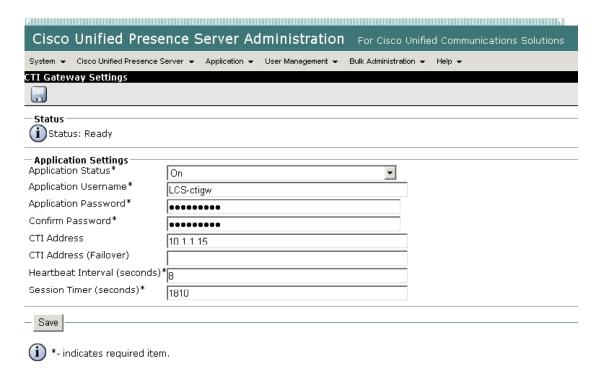
Remember, this is similar to what you did for the users back at the beginning of the lab......so, no instructions needed ©.

Configure CUPS CTI-GW

Now switch to the management interface of your CUPS server and setup the CTI-Gateway. This is in the Applications Menu.

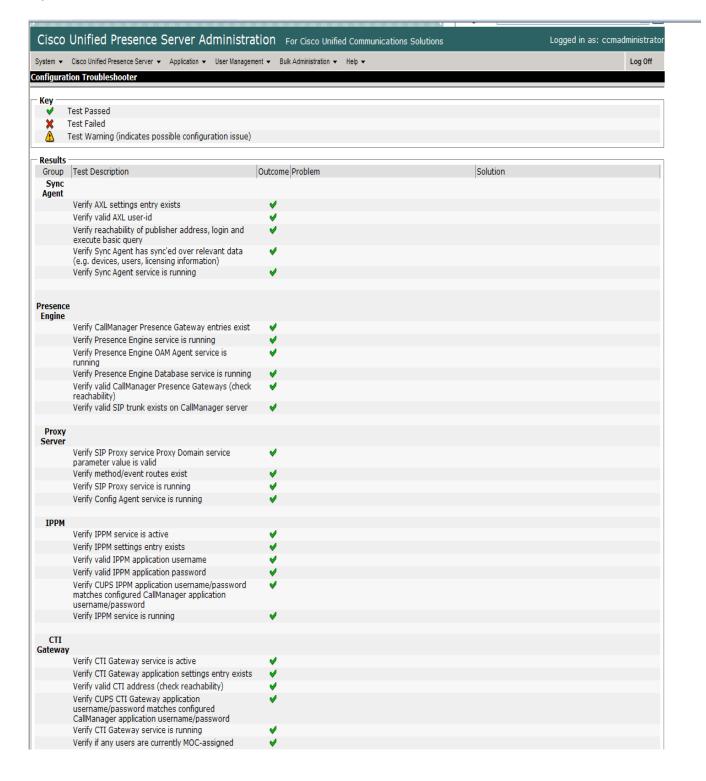


To Configure the CTI-Gateway
Change the Application Status to On
Set the IP Address of your CTI Manager (Call Manager)
Set the User to your LCS CTI user



Run the trouble shooter to check your configuration

Systems > Troubleshooter



Section 5c: Configuration and Operation for Microsoft Office Communicator

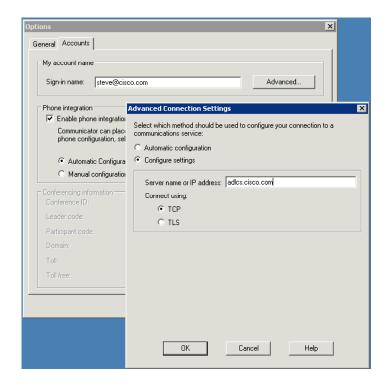
On the desktop of BOTH users (steve and jane) find the communicator.msi file

Launch the communicator.msi file. Please just select default values using the "Next" key.



Configure the MOC client

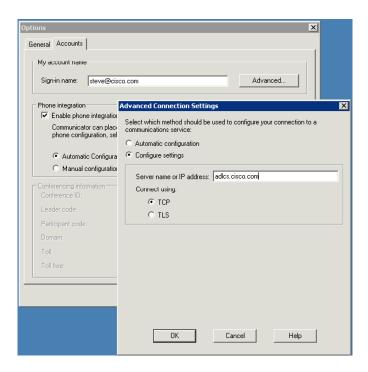
Now start your MOC client, select actions, options, and enter your username



Microsoft Office communicator can automatically find an LCS server using DNS and SRV records. We are going to manually configure the LCS server location for the lab.

Click the Advanced button select Configure Settings

Now Enter the **FQDN** (adlcs.cisco.com) of the LCS server and select **TCP** as the Protocol



We need to change a registry entry in the client to allow remote control of the deskphone

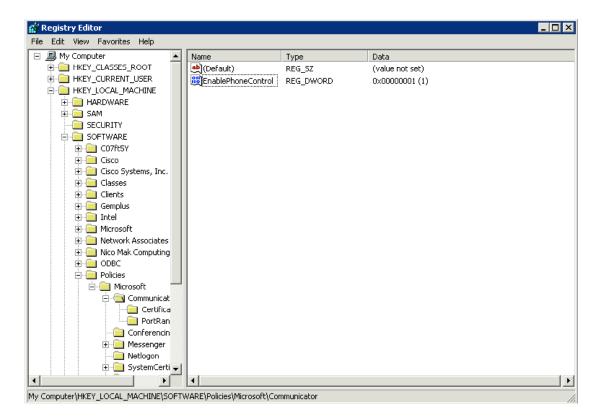
Launch the regedit.exe application to configure the system policy to allow remote call control for MOC

Start -> Run -> regedit.exe

Select HKEY LOCAL MACHINE\SOFTWARE\Policies\Microsoft\Communicator

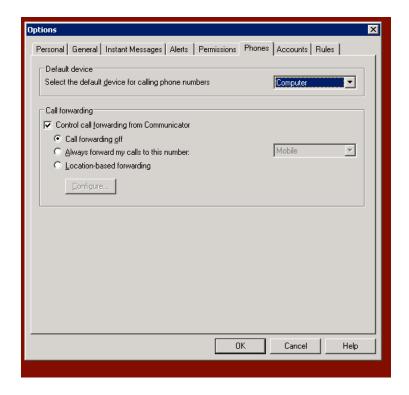
Create a new DWORD value with name 'EnablePhoneControl'

Change the value of this key to '1'



We configure LCS to use the telephone to make calls rather than operate as a softphone.

From the Menu select **Actions**, **Options** and on the **Phone** tab change the default device to "**Phone**"



Test your client by send an IM message to your partner workstation.

Test the integration.

Your client will show a successful integration by an icon in the status bar.





Add the other user to you MOC as a contact and enable the extended view option.



You should now be able to call your partners phone using the MOC GUI.



AND NOW YOU ARE FINISHED!!