

X-Lite 5 for Mac User Guide

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This manual corresponds to version 5.0 of X-Lite 5 for Mac.

Contents

Introduction	. 1
Installation and Setup	.3
Getting Ready	
Installing X-Lite	.4
Troubleshooting	.6
Checking for X-Lite Updates	.6
Using X-Lite	.7
The Onscreen Softphone	.7
Placing a Call	.9
Handling Incoming Calls	.11
Handling an Established Call	
Handling Video Calls	.13
Three-way Calls (Conference Calls)	.15
Voicemail	.18
Using Resources	.19
Contacts Tab	
The History Tab	.21
Configuring X-Lite	.23
Configuring Accounts: the Accounts Tab	.23
General Preferences	
A Troubleshooting	.45
B Glossary	.49

1 Introduction

Standard Telephone Features

The CounterPath X-Lite 5 for Mac softphone has all standard telephone features, including:

- Call display and Message Waiting Indicator (MWI).
- Speakerphone and Mute.
- Redial, Hold, Do Not Disturb.
- Call history list of received, missed, and dialed calls.
- Three-party conferencing.

Enhanced Features and Functions

X-Lite for Mac also supports the following features and functions:

- Video.
- Managed contact list.
- Automatic detection and configuration of audio and video devices.
- Acoustic echo cancellation, automatic gain control, voice activity detection.
- Support for the following audio codecs: Broadvoice-32, Broadvoice-32 FEC, DVI4, DVI4 Wideband, G.711aLaw, G.711uLaw, GSM, iLBC, L16 PCM Wideband, Speex, Speex FEC, Speex Wideband, Speex Wideband FEC.
- Support for the following video codecs: H.263, H.263+ 1998.
- Automatic selection of the best codec based on the other party's capability, the available bandwidth, and network conditions. X-Lite switches the codec within a call in response to changing network conditions.
- Compliance to 3261 SIP standard.
- STUN and ICE NAT traversal.
- Support for DTMF (RFC 2833, inband DTMF).

SoftPhone.com Features

If you have a SoftPhone.com account, the following features are also available:

- Presence via your SoftPhone.com account. You can see the presence of any contact who also has a SoftPhone.com account.
- IM via your SoftPhone.com account. You can send IMs to any contact who has a SoftPhone.com account.

For more information on SoftPhone.com and its features, see www.softphone.com.

2 Installation and Setup

2.1 Getting Ready

Hardware and Other System Requirements

For information on the supported hardware and other system requirements, see https://support.counterpath.com/ default.asp?W336.

Multimedia Device Requirements

X-Lite requires both speakers and a microphone to make calls. Any of the following configurations are acceptable:

- External speakers and microphone
- Built-in speakers and microphone
- Dual-jack multimedia headset
- USB multimedia headset
- USB phone.

X-Lite is optimized to work with Apple iSight camera.

2.2 Installing X-Lite

Run the X-Lite installer and follow the prompts in the install wizard.

If you want to configure X-Lite immediately, then at the final step of the wizard, select Launch X-Lite. The first time you start X-Lite, a video about Softphone.com appears. Then the X-Lite softphone appears.

00	X-Lite 4		\bigcirc
Joseph Santos	Available 🗧		
Enter	a name or number	MNO 6	
Contacts	History		$\overline{}$
▼Family (0)			
▼Friends (0)			
▼Work (0)			
			_
	our free X-Lite s pires on Augus		
AN COUNT			1

You can use X-Lite without signing up for Softphone.com, but if you do sign up for Softphone.com you will be able to make freee phone calls to people outside your enterprise and you will have access to IM and presence features. For more information, see www.softphone.com.

If X-Lite is not already running, start it as you would any other program.

You can set up X-Lite in one of these ways:

- With your PBX account (in other words, your office phone number).
- With your PBX account and with SoftPhone.com .
- With a SoftPhone.com account only.

This guide explains how to set up your PBX account. For information on the SoftPhone.com account, see www.softphone.com.

Setting up your PBX Account

To use your PBX account, you must have network access to your enterprise LAN, which means you must be using a computer on the LAN or using a computer that has VPN access to the LAN.

- 1. Obtain your PBX account credentials from your IT administrator or VoIP service provider:
 - User name and password
 - Authorization Name (if applicable)
 - Domain
 - Firewall traversal and other network information; see "Configuring X-Lite" on page 23.
- 2. From the menu bar, choose X-Lite > Preferences > Accounts. The SIP Account window appears.
- 3. In the Account tab, complete the User Details area with your account credentials.

ount Voicemail	Topology Transport Advanced
count name: My P	BX
Protocol: SIP	
Jser Details	
* User ID:	1331
* Domain:	yournormalpbx.com
Password:	•••••
Display name:	Joseph Santos
uthorization name:	
Domain Proxy Register with dom iend outbound via: Domain Proxy Add	nain and receive calls

- 4. Complete the remaining tabs as specified by your IT administrator or VoIP service provider, or to suit your setup. Settings that you may need to change immediately include:
 - Account tab, Domain Proxy area.
 - Topology tab, if your computer is on a network and/or behind a firewall.

For more information, see "Configuring X-Lite" on page 23.

Setting up for Voicemail

Your enteprise or VoIP service provider may offer voicemail. If it does, then you can set up some voicemail features in X-Lite. See page 25.

Setting up a Contact List

Typically, you will want to create contacts in order to easily make calls. You can add addresses to the Contact list one by one. See page 20.

2.3 Troubleshooting

X-Lite includes tools for helping you troubleshoot problems. From the menu, choose Help > Troubleshooting.

- Devices tab: You can verify that your microphone and speakers are working and can set the volume to a comfortable level without having to actually place a phone call.
- Audio tab: While you are on a phone call, you can test the quality of the audio. Note that to perform a valid test, you should be on an established call (not a call attempt).
- Video tab: You can verify that your camera is working.
- Diagnostics tab: If none of the other tabs help you solve your problem.

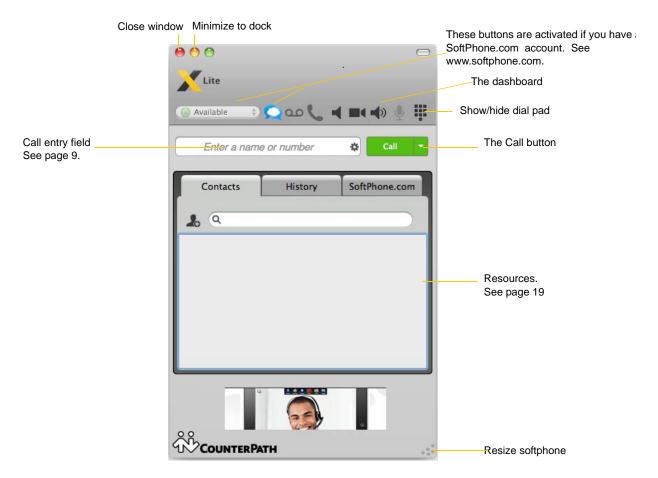
2.4 Checking for X-Lite Updates

To check for updates to X-Lite, from the menu bar, choose Help > Check for updates. The Auto Update window appears.

- If a new version of the software is available, you can download it from this window.
- If no newer version is available, this window informs you that your version is up to date.

3 Using X-Lite

3.1 The Onscreen Softphone



Compact View

To display a compact view of the softphone, hide the dialpad and hide all the resources tabs (View > Hide).



The X-Lite Menu

Ś.	X-Lite	Edit	View	Contacts	Actions	Window	Help	р
	About	X-Lite						
		ences 1 to So	ftphone	策, e.com				
	Servic	es		•				
	Hide X Hide C Show J	Others		日光 日第ブ				
	Quit X	-Lite		жQ				

The menu contains the following items that are specific to X-Lite:

X-Lite

- X-Lite > Sign in to SoftPhone.com . For information on this menu, see www.softphone.com.
- X-Lite > Share with Contacts. For information on this menu, see www.softphone.com.
- X-Lite > Preferences. See page 23.

View

Changes how X-Lite looks. Also lets you view the message archive.

Contacts

Lets you work with contacts. For information on the Find contacts menu item, see www.softphone.com.

Actions

- Lists the actions that you can perform, depending on the current "state" of X-Lite. For example, if a contact is selected, it lists all the actions that can be performed on that contact.
- For information on the Invite to SoftPhone.com menu item, see www.softphone.com.

Help

Provides access to various features.

The X-Lite Toolbar

The toolbar has various buttons, including the message waiting indicator (MWI) and missed calls indicator. Hover over each button for a description.



3.2 Placing a Call

You can place one more call when another call is already in progress.

The first time you make a call, select Softphone > Preferences > Dialing Preferences and set your dialing preferences. See page 42.

00	Preferences - Privacy Opti	tions	
Application Accounts Sounds	Devices Network Codecs Media Quality Adva	anced Dialing Preferences	Privacy Options
Presence Visibility Full Presence Limited Presence (A) No Presence (Always			
lace the call. See below r different ways to specify e number.	Available		Call entry field Click to place the call Auto-complete suggestion
	Contacts		

The outgoing call opens in its own call panel, below any other existing call panels. You will hear a ringing tone while X-Lite attempts to make a connection.

You can specify the number to call in any of these ways:

- Type the number on the dialpad or computer keyboard, then click Call or Enter.
- Control-click an entry in the Contacts or History tab and choose Call or Video Call.
- Drag-and-drop an entry in the Contacts or History tab.
- Hover over the right side of a contact to reveal the click-to-call button. Click to place a call using the primary phone number for this contact.
- Redial: click Call or click the down arrow and select a recent call and click Call.

The number can be in several formats:

- An extension such as 390 or 1300. X-Lite will place the call without going through your PBX.
- A phone number outside your PBX. See page 42 to set up correctly to place these calls.
- A phone number and extension with a separator in between. For example, 6045551222x1300. X-Lite will dial the phone number, wait for the call to get established, then dial the extension.

Valid separators are: x extension ext ; (semi-colon), (comma)

If X-Lite sees that the phone number is actually your own PBX, it drops the phone number and dials the extension directly.

• A SIP address at a different domain, for example, ewilding@domainA.com. Speak to your VoIP service provider or IT administrator about whether you can phone someone with a different domain from your own.

Dealing with an Auto Attendant

If you know that your call will be answered by an auto attendant and you know what menu items you will choose, you can include those menu items (DTMF) in the phone number when you dial it:

- Before the first DTMF number, include at least one capital P.
- You can include other Ps. Each P causes X-Lite to wait one second before sending the next character.
- At the end of the input, you must include a ; character.

Example: To dial a number, add a 3-second delay and then press 44, enter this in the call entry field:

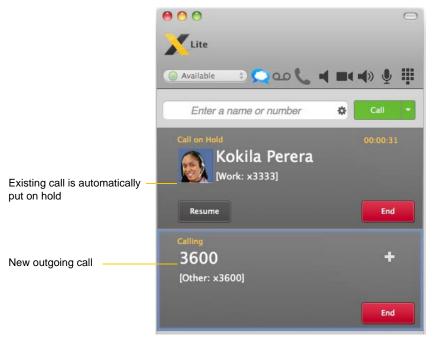
604551212PPP44;

Example: To dial a number, add a 6-second delay, then press 1 then 3 then 2, each with a two second delay:

604551212PPPPPP1PP3PP2;

Placing another Call

To place a new call (without hanging up on the current call), simply place the call in the normal way. A second call panel opens below the current call. The first call is automatically put on hold.

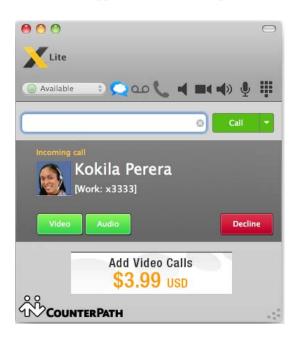


Switch between the different calls by clicking Resume on the desired call panel. That call becomes the active call and all other calls are on hold.

3.3 Handling Incoming Calls

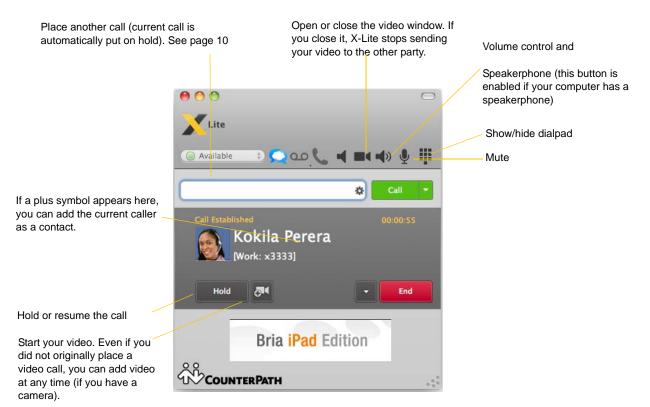
X-Lite must be running to answer incoming calls. (If X-Lite is not running, incoming calls may be directed to voicemail; check with your IT administrator or VoIP service provider.)

The new call appears in its own call panel.



Action	Description
Answer or Audio	Click Answer or Audio
	If you are on another call, that first call is automatically put on hold. You are now talking to the new caller.
	Or press Enter (on the keyboard), if X-Lite is the active application.
Video	The caller wants to include video. Click to answer a video call and start sending your video immediately
Decline	Click Decline. There will be a busy signal. The call may be directed to voicemail (if you have this service).

3.4 Handling an Established Call



3.5 Handling Video Calls

Placing a Video Call

O Available	e 🔅 🔍 🕰 💭 (= 💷	▲ » 🔮 🗰	
3333	0	Call -	
	Bria iPad Edition	Audio Call Video Call Start Conference Call	Click the down arrow and choose Video call

Specify the number in one of these ways:

- Type the number on the dialpad or computer keyboard, then click the arrow beside the Call button and choose Video call.
- Control-click an entry in the Contacts or History tab and choose Video Call.
- Redial: click Call or click the down arrow and select a recent call and click Call.

Adding Video to an Existing Call

If you have a camera, you can click Start Video to add video to an established call. When you add video, the other party may (or may not) start sending their video to you.





Other Party Adds Video

If the other party starts their video, your video panel automatically opens and the video is played. You can start sending your own video, if desired (and assuming you have a camera) by clicking Start Video on the call panel.

Pausing and Resuming Video

Click Stop Video on the call panel or close the video window to pause sending your video.

Click Start Video to resume sending your video; the video window will open.

 \bigcirc

Start Conference Call

Cal

Audio Call

Video Call

3.6 Three-way Calls (Conference Calls)

00

Lite

Available

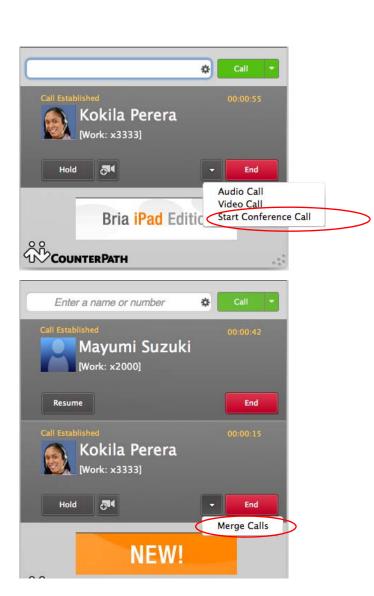
3333

Starting a Conference Call

From the Dashboard

- Enter the number or address by typing
- Or drag a contact or history item
- Or select from the redial list
- Then choose Start conference call

From an Existing Call From One Established Call

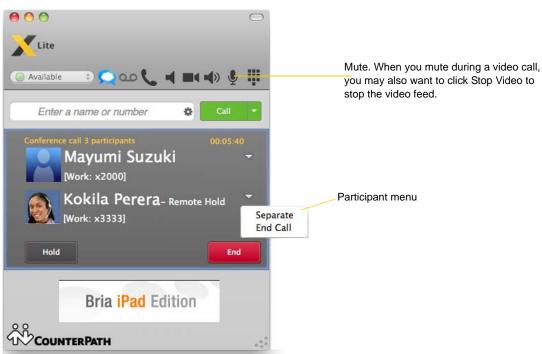


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Bria iPad Edition

From Two Established Call

Managing the Conference



Removing a Participant

To remove one call from the conference, click the down arrow beside a participant name and choose Separate. The call becomes a separate call.

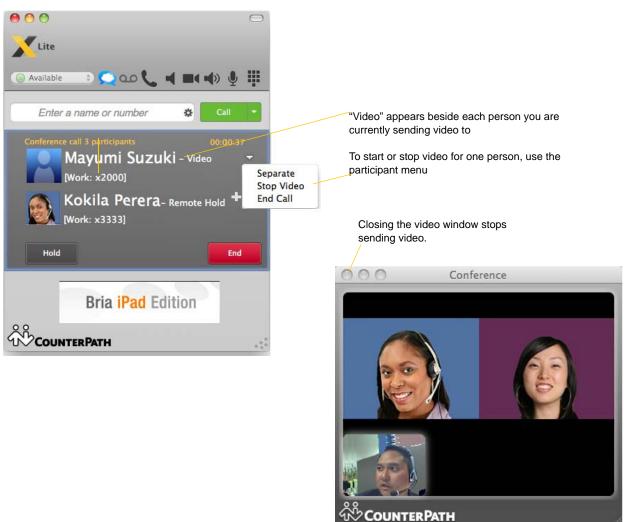
Ending the Conference

To hang up on everyone, click the End button.

Suspending the Conference

To suspend the conference, click the Hold button. All participants are put on hold.

Video Conference Calls



Starting a Conference with Video

When you start a conference from established calls, video is included if at least one of the calls already includes video. Video is sent to all the participants.

When you start a conference from scratch, video is not included. You can open your video window and choose to add video at any time. Video is sent to all the participants.

How Video Is Shared

The conference host (the person who starts the conference) serves as the host for other video. Whatever the conference host is receiving will automatically be sent to the other parties.

If the conference host pauses or stops video, other participants will no longer receive video.

3.7 Voicemail

If your service includes voicemail and you have set up voicemail options (page 25), then when you have voicemail messages, the voicemail icon appears at the top of the phone (the icon may include a number). You can click the icon to automatically connect to voicemail and listen to your messages.



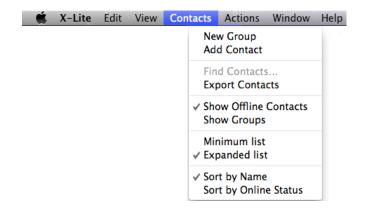
4 Using Resources

4.1 Contacts Tab

If the Contacts tab is not showing, go to the menu bar and choose View > Show Contacts.



You can change the detail and layout of the contact list: from the menu bar choose Contacts and set the desired options.



Managing Contacts and Groups

Adding a Contact

Click 🚜 , or control-click a group and choose Add Contact to Group. The Contact dialog appears.

	O Contact Editor	
To enter a phone number, select the type, enter the number, and click +	Contact Summary Kokila Perera Group: Friends	
The new number appears	Contact Display Name: Kokila Perera First Name: Work ‡ x3333 Uast Name: Work ‡ x3333	-
in the list	Cancel	ОК

Adding a Contact using an Existing Address

You can add a contact by capturing existing information:

- If you are on a phone call with a person who is not a contact, click the Add to Contacts icon in the call panel.
- On the History tab, select an entry that is not a contact. Control-click and choose Add Contact.

The Contact Profile dialog appears. Complete the dialog as desired and press OK.

Changing Contact Information

To change the information for a contact, control-click the contact and choose Edit Profile. The Contact dialog appears (see page 20).

Deleting a Contact

To delete one or more contacts, select them, control-click, and choose Delete or Delete Selected Items.

Adding, Deleting or Renaming Groups

Select any group, control-click, and choose the appropriate menu item.

Contacts that have no group appear under "Ungrouped Contacts." You can delete, but not rename, this group.

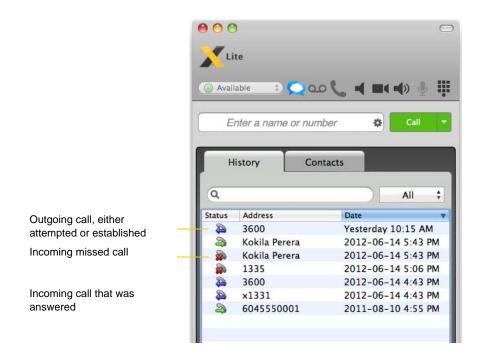
Exporting Contacts

You can export a contact list to a comma-separated file. This feature is useful, for example, if you later upgrade to Bria 3 and want to move over your contacts.

- 1. From the menu bar, choose Contacts > Export Contacts. The Export Contacts wizard starts.
- 2. Select the file type and file location for the created file.

A file of the specified type is created.

4.2 The History Tab



From the menu bar, choose View > Show History.

Managing the List of Calls

You can control-click on an entry in a list to:

- Delete the call.
- Delete all entries in this list.
- Edit the profile if the entry is a contact.
- Add as contact. The Contact dialog appears.

Phoning from History

You can control-click on an entry to place a call to this person, using the contact method that was used for this call. You can:

• Control-click on an entry and select Call or Video Call to place a call to this person, using the contact method that was used for this call.

5 Configuring X-Lite

From the menu bar, choose Bria > Preferences. The Preferences window appears.

- When configuring X-Lite the first time, click the Accounts tab and set up the account. Then complete other tabs as desired.
- After the initial setup, display whichever tab you want.

5.1 Configuring Accounts: the Accounts Tab

Choose X-Lite > Preferences from the menu bar. Click the Accounts tab. The Account Settings window appears.

/lake si	ure the	account is er	nabled	Double	e-click the acc	count to ed	it		
00				Prefe	erences – Acc	ounts			
8]	2	. 🧾	Q 😣		t 🕹	500	\bigcirc		
ication	Accourt	nts Sounds D	evices Network	Codecs	Media Quality	Advanced	Dialing Preference	es Privacy Options	
nabled	Status	Account Name		Protocol	User ID			Call	
⊻	0	My PBX		SIP	1331@yourno	ormalpbx.co	m	1	
_									
- -									

You cannot add more than one account

Account – General

Account name: My	PBX	
Protocol: SIP		
General	/oicemail Topology Transport Advanced	
User Details		Fields with an asterisk a
* User	ID 1331	required
* Doma	ain yournormalpbx.com	
Passwo	ord •••••	
Display nar	ne Joseph Santos	
Authorization nam	ne	
Domain Proxy		
Register with d	omain and receive calls	
Send outbound via	r:	
💽 Domain		
O Proxy A	ddress	

Field	Description
Account name	Always MyPBX.com
Protocol	Read-only. Always specifies SIP.
User Details	
User ID	Typically the account number for the account. For example, kperera. Provided by your IT administrator or VoIP service provider.
Domain	For example, domainA.com. Provided by your IT administrator or VoIP service provider.
Password	Provided by your IT administrator or VoIP service provider.
Display name	This name is displayed in the X-Lite title bar.
	Other people will see you as this name.
Authorization name	May not be required. If it is required, it will be provided by your VoIP service provider.
Domain Proxy	
Register with domain and receive	Check this box if you want to receive incoming calls.
calls	Typically, this field is checked.
Send outbound via	Choose the setting specified by your IT administrator or VoIP service provider:
	• Domain: If your service requires that traffic be directed to proxies that are discovered via the domain.
	• Proxy Address: If your enterprise or VoIP service provider has an outbound proxy address and requires that you provide the address to X-Lite. For the address enter a domain name (for example, domain.com) or an IP address (for example, 123.456.789.012).

Account – Voicemail

Account name:	My PBX
Protocol:	SIP
General	Voicemail Topology Transport Advanced
/oicemail	
Check for v	voicemail
Number to dia	al for checking voicemail 📗
Number for se	ending calls to voicemail
Send calls	to voicemail if unanswered for 0 seconds

These settings let you set up to interact with your VoIP service provider's or enterprise's voicemail service. .

Your service provider may provide the ability to set up for voicemail outside of X-Lite, for example, by phoning a number and following the voice prompts, or by accessing a website.

Check with your service provider or IT administrator to determine if another setup mechanism is available. If so, check what the settings are in that setup, and make sure you enter compatible information in X-Lite.

Field	Description				
Check for voicemail	Set the checkbox in one of these ways:				
	• Check the box if X-Lite must subscribe to be notified when there is a voicemail for you. In other words, to configure for "subcribe for message waiting".				
	• Clear the checkbox if the service provider's voicemail server sends notifications without X-Lite subscribing. In other words, to configure for "implicit subscription".				
	Clear the checkbox if the service provider does not support voicemail.				
	Check with your IT administrator or VoIP service provider for the correct configuration.				
	Voicemail is offered by your enterprise or VoIP service provider; it is not part of X- Lite. Contact your service provider for information on using voicemail.				
Number to dial for checking voicemail	 Optional, but complete this field only if your VoIP service includes voicemail. Completing this field activates the sicon on the softphone. When you click the icon, X-Lite will dial this number. You will be connected to your service provider's voicemail and can listen to your messages. If you leave this field empty, then this icon will not work; you will have to manually dial this number in order to connect to voicemail. Enter the number or SIP address provided by your IT administrator or VoIP service provider. 				

Field	Description
Number for sending calls to voicemail	Complete only if your VoIP service includes voicemail. Optional, but you must complete it if you check "Send calls to voicemail if unanswered".
	This is the number that incoming calls will be forwarded to if they are unanswered after the specified interval (below).
	If you leave this field empty, then X-Lite will never forward calls to your service providers' voicemail. However, most enterprises or VoIP service providers have their own mechanism for sending unanswered phone calls to voicemail. So leaving this field blank does not mean that forward-to-voicemail does not work.
	Enter the number provided by your IT administrator or VoIP service provider.
Send calls to voicemail if unanswered	Complete only if your VoIP service includes voicemail.
	To send to voicemail after the specified number of seconds.
	Your service provider may also provide a similar feature that is set up outside of X-Lite. If so, make sure you do not enter competing information in X-Lite and in the service provider's user interface. For example, if you turn off this field, make sure the same feature at your service provider is also turned off. Otherwise, all your calls will continue to be forwarded.

Account – Topology

Account name:	Mr. DDV
Protocol:	SIP
General	Voicemail Topology Transport Advanced
Firewall Traversal	
Firewall travers	sal method O Auto-detect (ICE, recommended)
	O Discover public IP address (STUN)
	O Use media relay (TURN)
	 None (use my local IP address)
To use down	in conver Jonus Conver Address blank
	in server, leave Server Address blank
Serv	er address
	User name
	Password
	rassworu
Range of ports	s used on local computer 5060 - 5070

Field	Description					
Firewall traversal	Choose the setting recommended by your IT administrator or VoIP service provider:					
method	• Auto detect using ICE: Automatically determine the contact address for signaling traffic.					
	Advertise the local IP, public IP (discovered via STUN, if available) and media relay IP and use these to automatically determine the best route for media traffic during calls.					
	• Discover public IP address: Advertise the public IP address (discovered via STUN) for the contact address for signaling traffic, and for the connection address for media traffic.					
	• Use media relay (TURN): Advertise the public IP address (discovered via STUN) for the contact address for signaling traffic.					
	Advertise the address of a media relay server (discovered via TURN) for the connection address for media traffic.					
	None: Advertise the local IP address only for both signaling and media traffic.					
Server address	• Empty: Discover the address of the firewall traversal server (the STUN or TURN server), if available, using DNS SRV.					
	• Specified: Use the firewall traversal server specified as either an IP address or a fully qualified hostname.					
	If you use the "Auto detect using ICE" option, then you can only enter a STUN server here. Don't enter a TURN server because when ICE is used, TURN is not supported.					
Range of ports	The appropriate setting depends on your computer setup:					
	• Checked: If your computer is behind a restrictive firewall that only allows specific port ranges to be used. Enter the range of ports to use for your SIP account. (You must also open those ports on your firewall; refer to applicable firewall documentation for information.)					
	Unchecked: If your computer is not behind a restrictive firewall.					

Account – Transport

Account name:	Mr. DDV	
Protocol:	SIP	
General	I Voicemail Topology Transport Advanced	
Transport		
Signaling tran	sport Automatic	
Signaling tran		
Media Encryption	over TLS	
1944.0		
O Make and a	accept only encrypted calls	
Do not allo	ow encrypted calls	
Obernerane	and providence	
	Cancel	OK

Field	Description
Signaling Transport	Contact your IT administrator or VoIP service provider to identify the types of transport that are supported. Then choose a supported transport:
	 Automatic: X-Lite sets up the transport based on the capabilities of the network and the X-Lite computer. Choose this option if you do not care which transport is used. TCP: This transport provides no signaling security.
	UDP: This transport provides no signaling security.TLS: Choose this option to request signaling encryption or both signaling and media
	encryption.
Encryption	See the description below for more information.

Account – Advanced

ccount	Voicemail	Topology	Transport	Advanced
Registe	r Settings —			
Reregist	er every: 360	secor	nds	
Minim	um time: 20	secor	nds	
Maxim	um time: 180	0 secon	nds	
	fault session ti	ime: 60	seconds	
Hold m	ndle hold requ	est in M-li	ne only (lates	t standard)
	100 100 100 100 100 100 100 100 100 100			(old standard)
	SIP keep-alive	s		
Send 🖉 Use r	port			

Field	Description
Register Settings	
Reregister every	The time interval between X-Lite's attempts to reregister in order to refresh the account registration for this account. A value of zero means not to reregister after the initial registration.
	This value is placed in the "Expires" header field of the REGISTER message.
Minimum time	If the reregistration fails, X-Lite will wait this amount of time, then attempt to reregister. If the second attempt fails, X-Lite will wait twice this time and try again, then four times this time, and so on, until reregistration succeeds.
Maximum time	This is the maximum wait time between attempts to reregister. Once this maximum is reached, X- Lite will wait this time for all subsequent attempts.
	For example, the min. time is 20 secs, the maximum time is 120 secs. X-Lite will attempt to reregister as follows:Wait 20 secs; Attempt to connect; If fail, wait 40 secs; Attempt to connect; If fail, wait 80 secs; Attempt to connect; If fail, wait 120 secs (the maximum); Attempt to connect:If fail, wait 120 secs, and so on.
Timers	
Enable session timers	A session timer is a mechanism to detect whether a call session is still active from the signaling
Default session time	point of view. When the timer expires, a refresh is sent from one party to the other. The timer is then reset.
	• Turn on to enable session timer. Enter a value in Default session time.
	• Turn off to disable session timer; refreshes will never be sent.

Field	Description
Session timer preference	This field specifies your preference for which party should send the refresh. The preference is not a guarantee that the refresh will be performed by the specified party. The choices are:
	• None: No preference.
	• Local refreshes: Your computer sends.
	• Remote refreshes: The other party sends.
	• UAC refreshes: The user agent client (the party that initiated establishment of the communications) sends.
	• UAS refreshes: The user agent server (the other party) sends.
Hold Method	Change this setting only if your IT administrator or VoIP service provider advises you to do so.
Send SIP keep-alives	Typically on, to instruct X-Lite to send SIP keep-alive messages in order to maintain a "pinhole" through your firewall for SIP messaging.
Use rport	Typically on.

5.2 General Preferences

From the menu bar, choose X-Lite > Preferences. The Preferences window appears. All the tabs on this window except for "Accounts" are preferences tabs. These tabs let you control the way that you work with X-Lite.

The icons (other than Accounts) let you set up general features of X-Lite.

Preferences – Application

00					Prefer	rences - Appl	ication		
0 lication	Accounts	Sounds	Devices	Network	Codecs	Media Quality	Advanced	Dialing Preferences	Privacy Options
General	Preferences	5							
Pau Pau	use iTune	s when r	making o	or receivi	ng calls				
Message	s & Presend	ce							
	s & Present		l don't i	use my co	ompute	r for 10	minutes		
Dis		status if			omputer	r for 10	minutes		
Dis In me	aplay idle	status if pressing	Enter w	ill i			minutes		

This panel lets you set your preferences for general GUI behavior.

Preferences – Sounds

ounds				
En 🗹	able sounds			
Play s	ounds			
	For an incoming call			(
	For call waiting			
\checkmark	For hangup			
	For an incoming Message			
	When a contact comes online			
100				

You can assign specific sounds to a variety of actions or "events".

- 1. Select the Enable sounds check box and select the check boxes for each desired event, or clear the Enable sounds check box to disable all sounds.
- 2. If enabling sounds, you can change the sound for each event: select the individual event. The value in Sound preference will change. Select the desired sound.

Preferences – Devices

00	Preferences – Devices	
Application Accounts	Sounds Devices Network Codecs Media Quality Advanced Dialing Preferences	Privacy Options
	Headset Mode Speakerphone Mode Other Devices	
Microphone: Speaker:	Built-in Microphone	
✓ Zero touch devid	ce configuration	Test Devices

X-Lite automatically detects devices at each startup, and selects the most appropriate device for each purpose. If you do not like this selection, you can override it on this panel.

If you override a selection, it will apply the next time you start X-Lite, unless the device is no longer available, in which case X-Lite will again select the device to use.

Field	Description
Headset Mode	
Microphone,	Change these fields only if you want to override the devices that X-Lite automatically selected.
Speaker	In both these fields, select the headset you are using.
	The headset is the device that is usually used for the speaker (the sound you hear) and microphone (recording your voice). The only situation in which the headset is not used is when the Speaker Phone button on the dialpad is pressed.
	Therefore, unless you will always be using X-Lite in speakerphone mode, you must make a selection here.
	Select the headset in both the Speaker device field and Microphone device field.
Zero touch device configuration	Typically, leave on. When device configuration is on, if you change your audio or video device, X- Lite will automatically detect the new device and start using it.
Test Devices	For information on the Test Devices button, see "Testing Audio and Video Devices" on page 47.
Speakerphone Mode	3
Microphone	Same as headset mode, but for the device to use when speakerphone is one (on the toolbar).
Speaker	Select None if you do not have a speaker phone. The Speaker Phone button on the dialpad will be disabled.
	You can set different devices for the speaker and microphone: for example, you can set the speaker to the speakerphone and set the microphone to your headset.
Other Devices	·
Ring on	The device where you want to hear the phone ringing.
	Change this field only if you want to override the devices that X-Lite automatically selected.
Volume	The volume of the ringer.
Camera	This field appears only on versions of X-Lite that include video functionality.
	Change this field only if you want to override the devices that X-Lite automatically selected.
	Select the camera model.
Resolution	Leave at standard, or change the size as follows:
	• Set it to high if you have a good camera and a computer with a fast CPU. You will know that you have set the size too large if:
	Your computer slows down (the video is using too much CPU)
	The video shows black areas or is slow or jerky.
	• Typically set it to low only in special situations, for example, when using wi-fi in a hotel. You will know that you have set the size too small if the video is fuzzy.
	This field is identical to the Resolution field in Media Quality > Video Quality.

Preferences – Network

000					Prefe	erences - Net	work				
Application	Accounts	Sounds	Devices	Network	Codecs	Media Quality	Advanced	Dialing F	Preferences	Privacy Options	
1	Connection										
	t cable, D		DSL								
	al area ne	etwork (LAN)								
O Cu:	stom										
Sen	ding bitra	ate: 80K	_)				- 5M	512000		
Rece	iving bitra	ate: 80K	_		-0-			= 5M	2048000)	

Select the type of network connection for your computer.

The sliders move to show the bitrate that will be used for sending and receiving. These rates are typical rates for the selected configuration.

If you know that your computer and network can handle a faster sending speed, click Custom and move the slider. It is recommended that you not change the receiving speed.

- You will know that you have set the sending speed too high if the remote audio is garbled.
- You will know that you have set the sending speed too high if:
 - The remote video shows black areas or is slow or jerky.
 - The remote audio is garbled.
- You will know that you have set the sending speed too low if the audio is good but the video is of poor quality (grainy).

Preferences – Codecs

Audio Codecs

	Au	udio Codecs Video Codecs
Codecs		Selected Codecs
DVI4 Wideband		G711 aLaw
GSM		G711 uLaw
L16 PCM Wideband		
Speex		
Speex FEC		
Speex Wideband		
Speex Wideband FEC		
ilbc		¥
Select a codec from the ab	ove lists to view pro	perties
Description:		
Fidelity:		
Bitrate range (bps):		
Best Quality (PESQ)		
	0.0	4.5
Accept the first code	c offered when n	egotiating audio codeces.
		or a particular call.)

This panel shows all the codecs that are included in the retail version of X-Lite. You can enable or disable codecs as desired.

With only one codec enabled, all calls made will use that codec. With more than one codec enabled, X-Lite automatically chooses the best codec based on the other party's capability, the available bandwidth, and network conditions.

You cannot change the properties of any codecs.

About Codecs

Audio codecs describe the format by which audio streams are compressed for transmission over networks. Codecs can be categorized as either narrowband or wideband:

- Narrowband codecs work with low bandwidth such as a dialup internet connection. These codecs have a sampling rate of 8 kHz.
- Wideband codecs work with high bandwidths and result in better audio quality. However, they do not work with PSTN. These codecs have a sampling rate of 16 kHz.

Supported Codecs

Codec	Narrowband	Wideband
DVI4	\checkmark	
DVI4 Wideband		\checkmark
G.711aLaw *	\checkmark	
G.711uLaw *	\checkmark	
GSM	\checkmark	
iLBC	\checkmark	
L16 PCM Wideband	\checkmark	
Speex	\checkmark	
Speex FEC	\checkmark	
Speex Wideband		\checkmark
Speex Wideband FEC		\checkmark
* Generally, at least one of these place a PSTN (land line) call.	codecs must be ena	bled in order to

Video Codecs

tion Accounts Soun	ds Devices Netwo	Preferences - Codecs
Codecs	(Audio Codecs Video Codecs Selected Codecs
		H.263 H.263+ (1998)
Select a codec from th CPU usage	e above lists to view	w properties
	-	
Quality	Low	High

Video codecs describe the format by which video streams are compressed for transmission over networks. Some codecs require less bandwidth than others, but may result in lower video quality.

You can enable or disable codecs as desired. You may decide to disable a codec even though your enterprise or VoIP service provider supports it.

With only one codec enabled, all calls made will use that particular compression format. With more than one codec enabled, X-Lite automatically chooses the best codec based on the other party's capability, the available bandwidth, and network conditions.

You cannot change the properties of any codecs.

Preferences – Media Quality



Field	Description
Audio Quality	
Reduce echo	Turning this feature on improves sound quality. This feature is typically on.
Reduce background noise	Automatically attempts to remove background noise. Typically on for the speakerphone.
Automatic gain control	This feature is typically on.
Preserve bandwidth	When this feature is on, X-Lite stops sending audio when you are not talking. When this feature is off, X-Lite always sends audio, which uses more bandwidth but may result in better call quality. Typically off. However, if you are using a slow (dial-up or ISDN) connection, you may want to turn
Video Quality	it on.

Field	Description
Resolution	Leave at standard, or change the size as follows:
	• Set it to high if you have a good camera and a computer with a fast CPU. You will know that you have set the size too large if:
	Your computer slows down (the video is using too much CPU)
	The video shows black areas or is slow or jerky.
	• Typically set it to low only in special situations, for example, when using wi-fi in a hotel. You will know that you have set the size too small if the video is fuzzy.
	This field is identical to the Resolution field in Devices > Other Devices.

Preferences – Advanced

Sounds Devices Net MF 2833 ☑ Also se nd nactivity timer seconds		edia Quality Advanc	Dialing Preferences	Privacy Options
nactivity timer	nd in-band			
nactivity timer	nd in-band			
nactivity timer	nd in-band			
nactivity timer				
seconds				

Field	Description
Audio	
DTMF	You may need to change the DTMF configuration if you cannot interact with interactive voice response systems (auto attendants, voice-activated menus, and so on).
RTP	This timer controls how calls are disconnected when X-Lite determines that there is a problem with the call and the other party has probably disappeared (assuming that you have not yourself detected the problem and hung up manually).
	Typically, the timer is enabled. It is recommended that you not disable it. You can change the length of the timer, but do not set it to less than 30 seconds.

Preferences – Dialing Preferences

00		Pre	ference	es - Dialing P	references			
[0]	N			12	() ()	\bigcirc		
Application Accounts	Sounds Devices	Network	Codecs	Media Quality	Advanced	Dialing Preferences	Privacy Options	
Dial Plan Options	Dial o Internatior	ut prefix: al prefix:	_					
C	Default extensio	n length:	4					
Two lette	r country code	of origin:	Cana	ıda		•		

Field	Description
Dial out prefix	The number to dial out of your enterprise. For example, 9.
International prefix	The prefix required on a phone number in order to handle this number as an international phone call. For example, +43. X-Lite handles international calling as follows:
	• It looks at the phone number for this international prefix.
	• If it finds this prefix, the phone number is transformed so that the number can be successfully placed according to the phone rules for the country specified in the country code.
	For example, if the phone number is +438901234 (meaning the number is a Swiss phone number) and the country code is Canada, then X-Lite replaces the + with 011 (the international dialing code for Canada), adds the dial out prefix (9) and places the call.
Default extension length	The maximum length of extensions in your enterprise. This information must be correct to ensure that X-Lite can distinguish between a phone number that can be made without leaving your PBX and one that requires going through the PBX.
Country code	The country your phone system works in. Change this code if you go to another country, but only if your entrprise has an office in that country that has its own PBX.
	For example, if you go to your Geneva office and that office has its own PBX, change the country code to Switzerland. In this way, phone calls made to a Swiss phone number will be correctly placed (that is, as being placed from within Switzerland).
	Do not change your country code if you go to another country but you are still connecting to the PBX in your home country. Speak to your IT administrator.

Preferences - Privacy Options Preferences - Privacy Options Preferences - Privacy Options Preferences - Privacy Options Preferences Presence Visibility Presence Limited Presence (Available/Offline only) No Presence (Always Offline)

This panel applies only if you have a SoftPhone.com account. See www.softphone.com.

A Troubleshooting

From the menu bar, choose Help > Troubleshooting. The Troubleshooting window appears.

Testing Audio Quality

While you are on a phone call, you can test the quality of the audio. Note that to perform a valid test, you should be on an established call (not a call attempt).



Testing Video Quality

While you are on a phone call, you can test the quality of the video. Note that to perform a valid test, you should be on an established call (not a call attempt).



Testing Audio and Video Devices

You can verify that your microphone and speakers are working and can set the volume to a comfortable level without having to actually place a phone call.

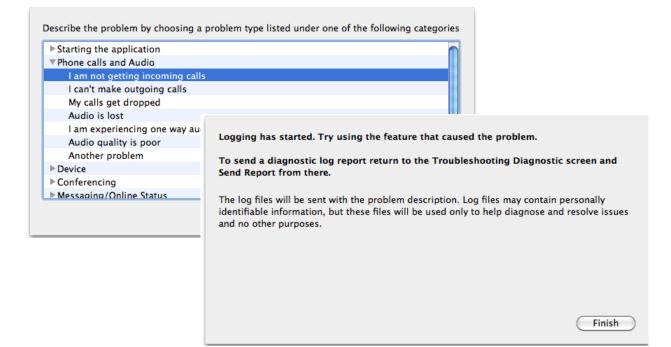


Diagnostics

00	Troubleshooting	J - Diagnostics
Audio Video	Devices Diagnostics	
Still havin	ng problems?	
For mor	re support information: CounterPath support F	AQ or
Start dia	agnostic logging.	Start Logging
Access i	reports or send them to us and we will try to	o help. Send Report Open Reports Folder
Make a	a Test Call on Softphone.com	

To start logging X-Lite activity and send a logging report to CounterPath customer support:

- 1. Click Start Logging. The first Diagnostics Logging window appears.
- 2. Select the problem you are experiencing and click Start Logging.
- 3. On the second Diagnostics Logging screen, click Finish. Logging will start.
- 4. Perform the actions you want to capture; for example, attempt to make a phone call.
- 5. When done, display the first Diagnostics Logging window again and click Send Report. A window opens; select the report and click Open. When the report is successfully sent, a confirmation message appears on the first Diagnostics window.
- 6. Click Stop Logging. Close the Troubleshooting window.



B Glossary

Broadband	Broad or wide bandwidth. In data transmission, the wider the band, the more data it is possible to transmit in a given time span. A cable, DSL and ADSL connection to the network provide broadband for data transmission. A dialup or ISDN connection typically provide a narrow bandwidth for data transmission.
Codec	The format by which audio or video streams are compressed for transmission over networks.
Dial plan	The rules that X-Lite follows in order to interpret the phone number that the user has entered and to modify the number or address, as required, to ensure that the call will be placed successfully.
DTMF	Dual-tone multi frequency. DTMF is the system that is used in interactive voice- response menu systems such as the menu system for accessing voicemail messages. The DTMF system allows the user to interact with the menu by pressing keys on a dialpad or keyboard.
Firewall	A technology that prevents unauthorized people connecting to your computer and to the applications running on the computer.
IP	Internet Protocol. A data-oriented protocol used for communicating data across a network. IP is the most common protocol used on the internet.
IP address	A unique number that devices use in order to identify and communicate with each other on a computer network using the IP standard.
Media	In a VoIP phone call, the audio and video portion of the information in a call. Compare to "Signaling".
MWI	Message Waiting Indicator. An indicator that there is a voicemail message for the owner of an account.
Narrowband	In data transmission, the wider the band, the more data it is possible to transmit in a given time span. A cable, DSL and ADSL connection to the network provide broadband for data transmission. A dialup or ISDN connection typically provide a narrow bandwidth for data transmission.
Proxy	See SIP account.
PSTN	Public Switch Telephone Network. The traditional land-line phone network.
RTP	Real-time Transport Protocol. A protocol for delivering the media portion of a data transmission over an IP network. SRTP is another media protocol.
Signaling	In a VoIP phone call, the information in a call that deals with establishing and controlling the connection, and managing the network. The non-signaling portion of the call is the Media.
SIP account	An account that provides the user the ability to make VoIP phone calls. The account encapsulates the rules and functions the user can access.
SRTP	Secure Real-time Transport Protocol. A protocol for delivering the media portion of a data transmission over an IP network. SRTP is a secure protocol, which means that the media is encrypted. RTP is another media protocol.
TCP	Transmission Control Protocol. A transport protocol for delivering data over an IP network. Other transport protocols are TLS and UDP.
UDP	User Datagram Protocol. A transport protocol for delivering data over an IP network. Another transport protocol is TCP.
VoIP service provider	A business that provides a VoIP service, allowing a user to connect to the internet in order to make VoIP phone calls using X-Lite. The VoIP service provider sets up a SIP account for the user.