

X-Lite 4.0 for Windows

User Guide

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This manual corresponds to version 4.0 of X-Lite for Windows.

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1 Introduction

Standard Telephone Features

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

- Call display and Message Waiting Indicator (MWI).
- Speakerphone and Mute.
- Redial, Hold.
- Call history list of received, missed, and dialed calls.

Enhanced Features and Functions

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

- Video
- IM and presence using the SIMPLE protocol.
- 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.
- 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 or SIP INFO messages).

2 Installation and Setup

2.1 Getting Ready

Account Information

After choosing a VoIP service provider, you will need the following information:

- User name and password
- Authorization Name (if applicable)
- Domain
- Firewall traversal and other network information; see "Configuring X-Lite" on page 27.

System Requirements

Processor	Minimum: Pentium 4® 2.4 GHz or equivalent Optimal: Intel Core 2 Duo or equivalent, .	
Memory	Minimum: 1 GB RAM Optimal: 2 GB RAM	
Hard Disk Space	50 MB	
Operating System	Windows XP Service Pack 2 Windows® Vista®	
Additional	Microsoft Windows Installer 3.1 Microsoft .NET 3.5 SP1 Microsoft VC 9.0 Runtime Service Pack 1 (Note that the installer will take care of installing those if you don't have them)	
Connection	IP network connection (broadband, LAN, wireless)	
Sound Card	Full-duplex, 16-bit or use USB headset	

Microsoft Internet Explorer® 6.0 or later.

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.

Video Cameras

Calls made with X-Lite will work without a video camera, but a video camera is necessary to allow other parties to see your image. X-Lite will work with most USB video cameras.

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.

2.3 Configuring X-Lite

If X-Lite is not already running, start it as you would any other program: Use the **Windows Start** menu or double-click the desktop icon. X-Lite appears.

Setting up the Account

After obtaining SIP account information from your VoIP service provider, you can set up your X-Lite account.

1. From the X-Lite menu, choose Softphone > Accounts. The Account Settings window appears.

Account		
count	Voicemail Topology Presence Transport Advanced	
	Account name: Account 1	
	Protocol: SIP	
Allow th	is account for	
🔽 Call		
🗹 IМ /	Presence	
User De	tails	
	* User ID: (jsantos	
	Domain: domainA.com	
	Password:	
	Display name: Joseph Santos	
Aut	horization name:	
Domain	Brow	
	ster with domain and receive calls	
	tbound via:	
	Domain	
0	Proxy Address:	
ial plan:	(#1\a\a.T;match=1;prestrip=2;	
		OK

2. In the Account tab, complete the User Details area with the information obtained from your VoIP service provider.

- 3. Complete the remaining tabs as specified by your 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 27.

Setting up for Voicemail

Your VoIP service provider may offer voicemail. If it does, then you can set up some voicemail features in X-Lite. See page 29.

Setting up a Contact List

Typically, you will want to create contacts in order to easily make calls and send IMs. See page 19.

2.4 Troubleshooting

X-Lite includes three tools for helping you troubleshoot problems:

- You can verify that your microphone, speakers and camera are working and can set the volume to a comfortable level without having to actually place a phone call.
- You can verify that you are successfully connected to the network.
- 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).

See page 47.

2.5 Checking for X-Lite Updates

To check for updates to X-Lite, 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 Starting X-Lite

If X-Lite is not already running, start it as you would any other program: Use the **Windows Start** menu or double-click the desktop icon. X-Lite appears.

Receiving an Online Status Request

If availability is enabled, then at any time you may get an Online Status Request dialog. For information on this request, see "Availability" on page 23.

Online Status Request	8
Kokila Perera (kperera) wants to communicate with you and see your online status on account Account 2 (SIP)	
Create a new contact	
Update an existing contact 1302	
Contact Details	
Display as:	
Group: CounterPath	
Click Allow to add this person to your contact list and send an online status request.	
Click Block to decline the online status request and add this person to your block list.	
Allow Block Igno	ore

USB Device Alert

If your headset is a "HID-compliant device", the USB Device alert may appear the first time you start X-Lite. You can assign functions to the headset. For example, you can set the green button so that it answers an incoming call.

If you click Yes, the USB Device Button Setup window appears; see page 39.

If you click No, you can still access this window from the Preferences > Devices panel (page 37).



3.2 Shutting Down

To shut down X-Lite, click Softphone > Exit.Note that clicking the Close button (X) on the softphone minimizes X-Lite to the system tray; it does not exit.

3.3 The Onscreen Softphone

When you first start X-Lite, it will appear in one of the two views shown below — the dialpad-centric view (left) or the contact-list-centric view (right). After that, X-Lite will remember the view that is current when you shut down and restart with that same view.



The X-Lite Menu

Softphone

- Accounts. These settings control how X-Lite interacts with your VoIP service provider. See "Configuring Accounts" on page 27.
- Preferences. These settings control the way that you work with X-Lite. See "Configuring Preferences" on page 35.
- Exit. To shut down X-Lite. You can also exit by pressing Ctrl+Q.

View

• Change how X-Lite looks.

Contacts

Lets you work with contacts. Everything in this menu can also be performed directly in the contact list.

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.

Help

Provides access to various service-related features.

Showing and Hiding Parts of the Softphone



You can still receive calls when X-Lite is hidden or minimized. If you exit when X-Lite is minimized; X-Lite will start next time as minimized.

3.4 Placing a Call

You can contact someone using:

- A softphone address (for example, kperera@domainA.com)
- A traditional phone number, if supported by your VoIP service provider.

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



Place the call. See the table below.

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.

How	From the	Description
Keying	Dialpad or computer keyboard	1. Enter the number or address in the call entry field using the dialpad or the computer keyboard
		If entering a softphone address, you can enter the entire address (kpereira@domainA.com) or just the name (kpereira).
		As you type, a suggestion for completing the entry may appear. Choose the suggestion or keep typing.
		2. Click the Call button or press Enter.
Drag-and-drop contact or previous call	Contacts or History tab	Drag an entry from the Contacts or History tab. (If the tab contains lots of contacts, first search or filter the list.)
Right-click a contact or previous call	Contacts or History tab	Right-click an entry from the Contacts or History tab and choose Call. (If the tab contains lots of entries, first search or filter the list.)
Double-click a contact	Contacts tab	Contacts have a "double-click" action that either makes an audio phone call or starts an IM. Each contact can be configured separately for this action. See the table on page 20.
Double-click a previous call	History tab	Double-click an entry. An audio call is placed.

How	From the	Description
Redial	Redial button	When the call entry field is empty, click the Call button.Or click the arrow in the call entry field and select a recent call. An audio call is placed.
Dialing	Dialpad	 If the dialpad is not visible, click the Show/Hide dialpad button. Click the numbers on the dialpad to enter the phone number. Click the Call button or press Enter.

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.



Ending a Call

Click the red End call button on the Call panel (the call can be active or on hold). The call panel closes after a few seconds.

3.5 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 VoIP service provider.)

As soon as an incoming call is received, a call panel appears, showing information about the call.

In addition, the Call Alert box appears, even if X-Lite is minimized.



This is the alert for an audio-only call



This is the alert for a video call



Click to answer a video call and start sending your video immediately Click to answer without video; if you want (and if you have a camera), you will be able to add video later.

Action	From the	Description
Answer	Call panel or Call Alert box	Click Answer.
		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.
Decline	Call panel or Call Alert box	Click Decline. There will be a busy signal. The call may be directed to voicemail (if you have this service).
Video	Call panel or Call Alert box	The caller wants to include video.
		Click to answer a video call and start sending your video immediately
Audio	Call panel or Call Alert box	The callerwants to include video.
		Click to answer without video; if you want (and if you have a camera), you will be able to add video later.

3.6 Handling an Established Call

While the call is in progress you can:

- Control the audio: use the speakerphone, mute the call, control volume.
- Put the call on hold.



3.7 Handling Video Calls

Placing a Video Call



How	From the	Description
Keying	Dialpad or computer keyboard	1. Enter the number or address in the call entry field using the dialpad or the computer keyboard
		If entering a softphone address, you can enter the entire address (kpereira@domainA.com) or just the name (kpereira).
		As you type, a suggestion for completing the entry may appear. Choose the suggestion or keep typing.
		2. Click the arrow beside the Call button and choose Video call.
Right-click a contact or previous call	Contacts or History tab	Right-click an entry from the Contacts or History tab and choose Video Call.
		(If the tab contains lots of entries, first search or filter the list.)
Dialing	Dialpad	 If the dialpad is not visible, click the Show/Hide dialpad button. Click the numbers on the dialpad to enter the phone number. Click the arrow beside the Call button and choose Video call.

Adding Video

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 window 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.

Pausing and Resuming Video

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

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



3.8 Voicemail

If your service includes voicemail and you have set up voicemail options (page 29), then when you have voicemail messages, a number appears beside the voicemail icon at the top of the phone. You can click the icon to automatically connect to voicemail and listen to your messages.



3.9 Instant Messaging

Sending an IM

1. You can send an IM to a contact who has a softphone address.

	Right-click on the contact and choose Send Instant Message. The contact must have a softphone address.
An active phone call	If the person you are speaking to is a contact, cick \mathbf{Q} on the call panel.

The Instant Message window appears.

2. Type the message. Include emoticons and formatting, as desired.

To insert a return in the message, press Ctrl+Enter. (You can control the behavior of Enter and of Ctrl+Enter; see page 35)

3. Click Send.



Receiving an IM

When an IM is received, the IM window pops up.

4 Using Resources

4.1 Contacts Tab



The Contacts tab displays your contacts and lets you specify multiple contact methods for them, including:

- Softphone address
- Home phone number, business phone number, mobile phone number

You can identify one of these contacts as the primary contact method. When you double-click a contact, the action associated with that method is performed: typically either phone or IM.

Contacts are typically organized into groups. X-Lite includes built-in groups: "Family", "Friends" and "Work". You can add more groups, as desired.

Managing Contacts and Groups

Adding a Contact

Click , or right-click a group and choose Add Contact to Group. The Contact Profile dialog appears. See the table.

	Group:	Work	¥
	duction Message		e presence
Double	e-click action for t	this contact:	•
		Introduction Message	Double-click action for this contact:

On a new contact, this checkbox and the Intro Message button become enabled once you enter a softphone address. On an existing contact, they become enabled if you enter another softphone address.

Table 1: Contact	Profile Fields
------------------	----------------

Field	Description	
Contact name	Only the "Display as" name is required. This is the name that will appear in the call panel and the Call Alert when this person phones you.	
Group	Click to show the list of groups and check each group to add this contact to.	
Contact Methods	ou must enter at least one contact method. Enter as many contact methods as you want. Softphone attries and e-mail entries must have the format <name>@<domain>.</domain></name>	
See this person's Softphone online presence	This box becomes enabled only if you have entered a softphone address.	
Introduction Message	This button becomes enabled only if you have entered a softphone address.	
Double-click action	Choose the desired method. This method will be used when you double-click this contact. For example, if you choose "IM:kperera@domainA.com", then double-clicking this contact will open an IM window using this address (rather than any of the other addresses that exist for this contact).	

Adding a Contact using an Existing Address

You can add a contact by capturing existing information. On the History tab, select an entry that is not a contact. Right-click and choose Add as Contact.

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

If the contact method is Softphone, an online status request is sent to this person. See "Sharing Availability" on page 23.

Changing Contact Information

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

Moving or Deleting a Contact

- To drag one or more contacts to a different group, select them and drag to the title bar of the new group.
- To delete one or more contacts, select them, right-click, and choose **Delete**. The contacts are removed from this group. If a contact belongs to several groups, it is removed from all groups.

Adding, Deleting or Renaming Groups

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

Using Contacts

Phone or IM the person	• Double-click the contact to start a phone call or IM to the person (depending on how double-clicking is configured in the contact's profile).		
	• Drag the contact to the call entry field to start a phone call.		
Phone the person	Right-click and choose Call, then click the desired number.		
Send an IM	Right-click and choose Send Instant Message. See page 17.		
You can watch the contact's availability, if the contact has a softphone address	See Table 2 on page 25 for the meaning of the availability icons.		
	See "Availability" on page 23 for information on obtaining availability information.		

Contact Flyout

Left-click on a contact to show the Contact flyout. This flyout shows all the contact methods set up for the contact. You can click a contact method to perform its action.



Click a contact method

4.2 History Tab



Managing the Lists of Calls

You can right-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 Add a Contact dialog appears. For more information on this dialog, see "Adding a Contact using an Existing Address" on page 20.

Phoning from History

You can right-click on an entry to place a call to this person, using the contact method that was used for this call. You can also double-click to place an audio call.

4.3 Availability

Availability refers to the ability to see whether a person is available or not: online, on the phone, busy, and so on.

You can publish your availability to contacts who have softphone address, and you can set up X-Lite to view the availability of other contacts.

Availability of other people	e
Your availabilit	y
	-Lite 4 - Joseph Santos
s	oftphone View Contacts Actions Help
	Available 🔻
	Enter name or number
	Contacts History
	<u>}</u>
	▼ Family
	Rita Santos
This person is either logged out or is not publishing her	▼ Friends
availability	- 😵 Frank Chan - Waiting for response
	Kokila Perera - Waiting for response
	🥑 Mayumi Suzuki
This person has no icon	▼ Work
because you are not subscribing to her availability	
subscribing to ner availability	

Sharing Availability

Watching Others' Availability

When you create a contact by entering details in the Contact Profile, the contact is automatically set up so that you share availability. As soon as the person is set up in this way in your Contacts, X-Lite sends them a notification request. The request asks that you be able to see that person's availability. If the person allows this request, then the icon beside their name (in your Contacts) becomes color coded, so you can determine their availability.

Allowing Other Parties to See your Availability (Publishing your Availability)

Continuing from above, when the other person allows the notification request, your name is added to their contact list. As soon as you are added to their list, a request is sent back to you requesting that they be able to see you. X-Lite accepts this request automatically, and the other person can now see your availability.

Receiving an Online Status Request

When someone who is not one of your contacts has you as a contact on their end, you receive an Online Status Request.

The request appears to you as an Online Status Request. You must respond to the request:

- Allow the request and create a new contact using this address. This option is selected for you if you do not currently have a contact with this SIP address.
- Allow the request and update the contact to allow the person to see your presence. This option is selected for you if you already have a contact with this SIP address but you were not previously sharing presence with this person.

Online Status Request	8
Kokila Perera (kperera) wants to communicate with you and see your online status or account Account 2 (SIP)	1
Create a new contact	
Update an existing contact 1302	
Contact Details	
Display as:	
Group: CounterPath	
Click Allow to add this person to your contact list and send an online status request.	
Click Block to decline the online status request and add this person to your block list.	

Button	Result
Ignore	The other person will not be able to see your availability. The Online Status Request may appear again in this session or a future session
Allow	 The other person will be: Able to see your availability. Added to your contacts. Because you allow the request, you will no longer be bothered by requests from this person. In addition, this same request will automatically be sent from your X-Lite to the other person, so that you can see their availability.

Setting your Availability

Changing your Availability

Click the down arrow beside the availability indicator on X-Lite, and select the desired availability. See Table 2, below.



When your availability changes, the new availability is sent to everyone who has permission to see your availability.

Setting up Availability Indicators

You can create custom availability indicators, or edit or delete custom or built-in indicators: click the down arrow beside the availability indicator, and choose Custom Message.

	Indicator	Meaning for your Availability	Meaning for Others' Availability
	Available	Either:	You can contact this person.
-		• You have set your availability to this status.	
		• X-Lite has determined that you are logged on but not on the phone or idle.	
		When you have this availability, X-Lite will automatically detect when you are idle or on the phone, and change the availability to match.	
•	Busy	You have set your availability to this status. X-Lite will never automatically switch you out of this status; you must switch yourself.	You can contact this person.
۲	Away	You have set your availability to this status. X-Lite will never automatically switch you out of this status; you must switch yourself.	You can contact this person.

Table 2: Availability Indicators

Indicator	Meaning for your Availability	Meaning for Others' Availability
On the phone	 Either: You have set your availability to this status. In this case, X-Lite will never automatically switch you out of this status; you must switch yourself. You started a phone call when your status was "Available". When your call finishes, your status reverts to "Available". 	You can contact this person.
Ø Idle	You have not moved or clicked the mouse or pressed a keyboard key for the Idle period. As soon as you click the mouse or keyboard, your status changes to "Available".	You can contact this person.
Appear offline	You have set your availability to this status. The other person sees you as offline, even though you are actually online. The other person cannot tell the difference between this status and you really being offline.	The contact is either not logged on or else is not set up for availability.
No icon	Not applicable	You are not watching the other person's availability.

Table 2: Availability Indicators

5 Configuring X-Lite

You can configure X-Lite in several ways:

- Configure global behavior. See "Configuring Preferences" on page 35.
- Configure the behavior on a per-account basis. See "Configuring Accounts" on page 27.

5.1 Configuring your Account

Choose Softphone > Accounts. The Account Settings window appears.

SIP Account Properties – Account

The red dot indicates that these				
fields are required				
SIP Account				a
		- Y		
Account Voicemail Top	ology Presenc	e Transport	Advanced	
Account name: (Account 1			
Protocol:	SIP			
Allow this account for				
IN / Presence				
User Details				j []
User ID:	jsantos			
Domain:	domainA.com			
Password:			5	
Display name:	Joseph Santos			
Authorization name:			\equiv	
Autionzation name.				
- Domain Proxy				
Register with domain a	and receive calls			
Send outbound via:				
Domain				
Proxy Address				
Dial plan: #1\a\a.T;match	=1;prestrip=2;			
				OK Cancel

Field	Description	
Account name	If desired, change the account name to something that is meaningful to you. For example, the name of the VoIP service provider.	
Protocol	Read-only. Always specifies SIP.	
Allow this account for Call	Leave this field checked.	
Allow this account for IM/ Presence	Leave this field checked.	
User Details	·	
User ID	Typically the account number for the softphone account. For example, kperera. Provided by your VoIP service provider.	
Domain	For example, domainA.com. Provided by your VoIP service provider.	
Password	Provided by your 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 calls	Check this box if you want to register with your VoIP service provider, so that you can receive incoming calls.	
	Typically, this field is checked.	
	This field may be left unchecked when, for example, your level of service does not include the ability to receive incoming calls. In that case, turning this field on may cause registration to fail (when you close the Account Properties window), meaning that your X-Lite cannot register with your VoIP service provider.	
Send outbound via	Choose the setting specified by your VoIP service provider:	
	• Domain: If your VoIP service provider requires that traffic be directed to proxies that are discovered via the domain.	
	• Proxy Address: If your 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).	
	If you are using X-Lite in a test lab, it is possible that neither of these settings is suitable; see page 34 for a third way to direct traffic.	
Dial Plan	Information about the syntax of the numbers used by this VoIP service provider.	
	Provided by the service provider.	
	For information on determining whether you need to create a dial plan, see page 50.	

SIP Account Properties - Voicemail

SIP Account	G
Account Voicemail Topology Presence Transport Advanced	
Check for voicemail	
Number to dial for checking voicemail:	
Number for sending calls to voicemail:	
Send calls to voicemail if unanswered for: 0 seconds	
	OK Cancel

These settings let you set up to interact with your VoIP service provider's voicemail service. They also let you configure X-Lite to forward calls in several situations, independent of your VoIP service provider's voicemail service.

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

Check with your service provider 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 VoIP service provider for the correct configuration.
	Voicemail is offered by your 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 📷 icon 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 VoIP service provider.
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 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 VoIP service provider.

Table 4: SIP Account F	Properties – Voicemail
------------------------	------------------------

Field	Description
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.

SIP Account Properties – Topology

Accoun						
Account	Voicemail	Topology	Presence	Transport	Advanced	
Firewal	II Traversal -					
Firewal	l traversal me	ethod:				
۲	Auto-detect	firewall trav	ersal method	d using ICE (recommended)	
ŏ	Discover pu	blic IP addre	ss (STUN)			
۲	Use media r	elay (TURN)			
		ocal IP addre				
Serve	r address:					
	σ	o use domai	n server, lea	ve Server A	ddress blank)	
U	ser name: (
	Password:			5		
	dasmond.					
Rang	e of ports use	ed on local c	omputer:			

Field	Description
Firewall traversal mode	Leave the default (Auto detect). Or if you have problems making phone calls, contact your VoIP service provider for information on their firewall traversal solution.
	• 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 (discovered via TURN, if available), 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.
Range of ports used	The appropriate setting depends on your computer setup:
on local computer	• 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.

SIP Account Properties – Presence

		(200 - 100 -	140	142 100			
Account	Voicemail	Topology	Presence	Transport	Advanced		
Presen	ce						
	Mode:	eer-to-peer		~)		
	Poll time:	300 sec	onds				
Refres	h interval: 🤃	3600 sec	onds				
						OK	Canc

Presence allows your contacts to see your online availability and allows you to see the online availability of contacts. See "Availability" on page 23.

Field	Description
Mode	Choose the setting specified by your VoIP service provider:
	Disabled: Presence is not supported.
	• Peer-to-Peer.
Poll time	Enter the value specified by your VoIP service provider.
Refresh interval	Enter the value specified by your VoIP service provider.

SIP Account Properties – Transport



Contact your 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
- UDP
SIP Account Properties – Advanced

SIP Account	8
Account Voicemail Topology Presence Transport Advanced	
Register Settings	
Reregister every: 3600 seconds	
Minimum time: 20 seconds	
Maximum time: 1800 seconds	
Timers	
Enable session timers	
Session timer preference: None 🖤	
Default session time: 60 seconds	
Hold method	
Handle hold request in M-line only (latest standard)	
Handle hold request in M-line and C-line (old standard)	
Send SIP keep-alives	
🗹 Use rport	
Send outgoing request directly to target	
	OK Cancel

Table 7: SIP Account Properties – Advanced

Field	Description		
Register Settings			
Reregister every	The time interval between X-Lite's attempts to reregister in order to refresh the account registration with the VoIP service provider 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 Default session time	A session timer is a mechanism to detect whether a call session is still active from the signaling 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.		

Table 7: SIP Account Properties – Advanced

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 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.
Send outgoing request directly to target	When checked, requests with a complete URI (user@ABC.com) go to ABC.com and the "Send outbound via" field on the Account tab (page 27) is ignored. If you check this field, make sure you also set "Send outbound via" (on Accounts > Account) to "Domain".
	Typically off. This field is intended for test labs and may cause problems in a NAT environment.

5.2 Configuring Preferences

Choose Softphone > Preferences. The Preferences window appears.

The Preferences panels let you control the way that you work with X-Lite.

Preferences – Application

Preferences							
Application Alerts & Sounds Devices Network Audio Codecs	🗹 Laur	I Preferences nch when Windows	starts a phone number is selecte	ed			
Video Codecs Quality of Service Media Quality Advanced	In mess	aging, pressing Ent Send the message	ion't use my computer for er will: (Use Ctrl + Enter to crea Use Ctrl + Enter to send)	te a line)	25		
	Button I	Defaults	ior for some buttons. Sele	ect the button an		associated with that t	
Reset to Default						OK	Cancel

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

Button Defaults

You can configure green Call button (at the top of the phone) so that it makes a video call instead of an audio call.

Preferences – Alerts & Sounds

Preferences		80
Application	Alerts & Sounds	
Alerts & Sounds	Alerts Sounds	
Devices Network Audio Codecs Video Codecs Quality of Service Media Quality Advanced	 Enable alert displays For an incoming call When a contact comes online When a contact goes offline For IMs (show alert instead of automatically opening IM window) 	OK Cancel
(and the second		

This panel lets you control the Call Alerts box and lets you assign sounds.

Enable Alert displays

You can control whether the Call Alert box is displayed in different situations. You can also control how you are alerted to an incoming IM: either with a small IM Alert box or with the IM window itself.



Assigning Sounds

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.

You can import sounds; these will be added to the list of sounds you can choose from when assigning a sound to an event.

Preferences – Devices

Preferences						
Application	Devices					
Alerts & Sounds	Headset Mode	e Speakerphone Mode	Other Devices			
Devices						
Network	Speaker:	GN 8120 USB				
Audio Codecs Video Codecs	Microphone:	GN 8120 USB	•			
Quality of Service Media Quality	HID Device	GN 8120 USB	▼](Setup		
Advanced	Zero-touch	1995 - 1995 - 1997 - 1997 - 1997 - 1997 - 1997 - 1997 - 1997 - 1997 - 1997 - 1997 - 1997 - 1997 - 1997 - 1997 -				
Reset to Default					ОК	Cancel

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.

Table 8: Preferences – Devices

Field	Description
Headset Mode	
Speaker	Change these fields only if you want to override the devices that X-Lite automatically selected.
Microphone	In both these fields, select the headset you are using.
	The headset is the device that is usually used for audio out (the sound you hear) and audio in (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.
HID Device	If the selected headset device is a HID-compliant device, this field specifies that device. You can click Setup in order to assign functions to the device. For example, you can set the green button so that it answers an incoming call.
Speakerphone Mod	e
Speaker	Change this field only if you want to override the devices that X-Lite automatically selected.
	Make the appropriate choice:
	• Select the device that you want to use for audio out (the sound you hear) when the Speaker Phone button is pressed. Make sure you select a speaker device (not the headset).
	• Select None if you do not have a speaker phone. The Speaker Phone button on the dialpad is disabled.
Microphone	Change this field only if you want to override the devices that X-Lite automatically selected.
	Make the appropriate choice:
	• Select the device that you want to use for audio in (recording your voice) when the Speaker Phone button is pressed.
	It can be any microphone: it does not have to be the microphone on the device you specified as the speaker device. For example, it can be the microphone on your camera.
	• Select None if you do not have a speaker phone. The Speaker Phone button on the dialpad is disabled.
HID Device	If the selected speakerphone device is a HID-compliant device, this field specifies that device. You can click Setup in order to assign functions to the device. For example, you can set the green button so that it answers an incoming call.
	(If the speakerphone device is the same as the headset device and you have already configured the device, there is no need to click Setup again.)
Other Device	· · · · · · · · · · · · · · · · · · ·
Ring On	Change this field only if you want to override the devices that X-Lite automatically selected.
	The device where you want to hear the phone ringing: the headset, the speakerphone, or none.

USB Device Button Setup

You can assign functions to a USB-compliant device. For example, you can set the green button so that it answers an incoming call.

USB Device Button Setup	8
GN 8120 USB	
1. Select a call action (e.g. Answer).	
2. Enter a description of the button you are may	pping (e.g. Green button).
3. On the USB device, click the button you are	mapping (e.g. Click the green button).
Call Action	Device Button
	Remove
	OK Cancel

Preferences – Network

Preferences			
Application Alerts & Sounds Devices	Network Network Connection Speed Dial-up or ISDN		
Network	Cable, DSL or ADSL		
Audio Codecs	Fast cable, DSL or ADSL		
Video Codecs	Local area network (LAN)		
Quality of Service	e Oustom		
Media Quality Advanced	Sending bitrate: 32K		
Auvaliceu	Receiving bitrate: 32K		
Reset to Default		OK	Cancel

Table 9: Preferences – Network

Field	Description
Network Connection Speed	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 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 – Audio Codecs

Preferences			30
Application Alerts & Sounds Devices Network Audio Codecs Video Codecs Quality of Service Media Quality Advanced	Audio Codecs Available Codecs Broad/Voice-32 FEC DV14 DV14 Wideband GSM iLBC L16 PCM Wideband Speex Sneey FEC V	Enabled Codecs BroadVoice-32 G711 aLaw G711 uLaw	
	Select a codec from the above lists to view prop Description: Bitrate range (bps): Fidelity: Best quality (PESQ): 0.0 Accept the first codec offered when negotiatin	4.5	
Reset to Default		СК Са	ncel

This panel shows all the codecs that are included in 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		
Broadvoice-32		✓		
Broadvoice-32 FEC		\checkmark		
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 codecs must be enabled in order to place a PSTN (land line) call.				

pplication Vin Nerts & Sounds Devices	deo Codecs		
lerts & Sounds			
	Available Codecs	Enabled Codecs	
		H.263	
etwork		H.263+ (1998)	
udio Codecs		>>	
ideo Codecs			
uality of Service		~	
edia Quality			
dvanced			
1	Select a codec from the above lists to view prop	perties	
	Description:		
	CPU usage:		
	Quality:	10.4	
	Low	High	

Preferences – Video Codecs

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 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.

Supported Codecs

Codec	Royalty-bearing	Included in X-Lite
H.263		\checkmark
H.263+1998		\checkmark

Preferences – Quality of Service

Preferences				00
Application Alerts & Sounds Devices Network Audio Codecs	Quality of Service Signaling QoS None Use DSCP / TOS value [163]:			
Video Codecs Quality of Service Media Quality Advanced	Audio QoS None Use DSCP / TOS value [163]: Use service type:	46 Best effort		
	Video QoS None Use DSCP / TOS value [163]: Use service type:	38 Best effort		
Reset to Default			ОК	Cancel

The Quality of Service panel lets you request a specific transport service for signaling traffic.

There are two types of services. The service to use depends on what your internet service provider supports:

- GQoS, which is available for audio and video.
- DSCP (also known as ToS), which is available for audio, video and signaling.

In a network that has the default configuration, the recommended value for audio is 46, because "46" is the standard marking for audio.

X-Lite supports 802.1p QoS packet tagging. If you set up for QoS, X-Lite will include the specified information in the packets that it sends to the network provider. Whether the packet is delivered with the specified service depends on whether your broadband router and the network provider between you and the other party supports multiple transport services. In other words, whether each network provider reads the QoS information and prioritizes packet delivery based on the requested service.

Preferences – Media Quality



Video Quality	
Enable this option	Check or uncheck to flip the video image.
Resolution – Limit available resolution	If you enable this field, X-Lite will determine the maximum quality that your computer can handle and will disable unapplicable options in the list of possible resolutions
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.
Audio Quality	
Reduce echos	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 it on.
Volume Control	Choose the setting that suits your setup.

Preferences – Advanced

Application Advanced	
Alerts & Sounds Devices Network Audio Codecs Video Codecs Quality of Service Media Quality Advanced RTP Enable inactivity timers RTP Enable inactivity timers RTP Send via DTMF 2833 Send in-band Send in-band Send in-band Media Quality Advanced	
Reset to Default OK	Cancel

Table 10: Preferences – Advanced

Field	Description	
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.	

A Troubleshooting

Choose Help > Troubleshooting to display the Troubleshooting window.

Testing Audio and Video Devices

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



Testing the Network Connection

You can verify that you are successfully connected to the network.



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).



Diagnostics

Only use this panel if instructed to do so by your support representative at your VoIP service provider.



To start logging X-Lite activity and send a logging report to 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, choose Help > Troubleshooting > Diagnostics again and click Send Report. Select the report and send it.
- 6. When the report has been sent, click Stop Logging. Click Done to close the Troubleshooting window.



B Dial Plan

You can create a dial plan to modify the the call input (what you type, select or drag onto the call entry field) to ensure that the call gets placed successfully. Modifying the input is useful when using numbers from a contact, which may include symbols such as the + symbol.

Do You Need to Read this Information?

You do not have to read this dial plan information if the default dial plan behavior (below) is acceptable.

If you need to modify the input (for example, to prefix the number with "9"), then you must create a dial plan.

The Default Dial Plan

The default dial plan is:

#n\a\a.T;match=1;prestrip=2;

where #n is the account prefix; in other words #1 for the first account and so on.

X-Lite supports only one SIP account, so there is no point in including the #1 when you place a call. However, if you are used to eyeBeam, you may be used to entering #1. This dial plan supports that habit: the dialplan removes the #1 and then places the call on the SIP account.

Processing of Phone Numbers when a Phonecall is Placed

When a phone call is placed, the input (the phone number or SIP address) is processed as follows:

- Cleanup: Input is cleaned up by removing spaces, dashes, open brackets, and close brackets.
- Matching: The input is compared to the patterns defined by the dial plan for the account. When a match is found between the input and the pattern, the transformation for this pattern is performed.

If no match is found, no transformation is performed.

For details on matching see "How the Input Is Processed" on page 53.

- Transformation: The selected transformation is performed.
- Then the call is placed using the transformed input.

B.1 Dial Plan Syntax

In X-Lite, the dial plan establishes the expected patterns of characters for a telephone number or softphone address, and allows for modification (transformation) of input based on the match to a pattern. The dial plan has the following syntax:

```
pattern[|pattern];match=1;<transformation>=<value>;[match=2;
<transformation>=<value>;]
```

Where:

- Items in [] are optional.
- Pattern: the pattern that will be matched. One or more patterns. Each pattern is separated by a | pipe. The pipe is optional after the last pattern. Each pattern is implicitly numbered, starting from 1.
- Match; Transformation: A pair that identifies the pattern number to compare with the input, and the transformation to perform on the input when a match is obtained. The transformation is optional (meaning that if there is no transformation for a pattern, then the input that matches this pattern is not transformed). One or more pairs.

"match=" is a literal. "n" identifies the pattern. "transformation=" is replaced by a keyword, see below. "value" is replaced by a value.

Spaces are allowed only in the <value> items.

Example

\a\a.T|xxxxxxxx;match=1;prestrip=2;match=2;pre=8;

where:

- $a = \pi$
- xxxxxxxxx; is the second pattern.
- match=1;prestrip=2; is the first match-transformation pair.
- match=2;pre=8; is the second match-transformation pair.

Pattern

Valid Content

The content for a pattern follows the digit map rules of RFC 2705, supplemented by the rules for regular expressions. Where there is an overlap between the digit map and regular expression rules, the digit map rules apply. For this reason, there are some special cases, included in the table below.

The following table describes the most common elements. As mentioned, all regular expression elements are supported.

Element	Origin	Description
0123456789	Literals	Literal digits, used as is.
# * a to z	Literals	 Literal characters, used as is. Special cases: The literal x character is represented by \x. The literal t character is represented by \t.
х	Digit map rules	Wildcard for any single digit, 0 to 9.
\a	Regular expression rules	Wildcard for any single alphanumeric character.
[digit-digit]	Regular expression rules	A digit within the specified range.
[character- character]	Regular expression rules	A character within the specified range.
[digit1, digit2, digit3]	Regular expression rules	One of the characters in the collection.
	Digit map rules	Repeat the last element 0 or more times. For example, xxxx. means repeat the last x 0 or more times, which means this pattern matches three or more digits (not four or more digits)! Use of this element results in a pattern with "minimum requirements".
Τ	Digit map rules	 A timeout period will take place before automatic dialing starts. The T timer forces X-Lite to wait after a match is made. This timer should always be included in these situations: Any pattern that uses the . (dot). For example, if the pattern is xxxx. then adding a timer lets you type three or more digits. If there is no timer, then as soon as you type three digits, X-Lite makes the match as soon as you type three digits. Any dial plan that has two patterns that are similar in elements but different in length. For example, if one pattern is xxx and the other pattern is xxxxxx, then adding the timer lets you continue typing past three digits, in order to get a match on the second pattern. In this situation, the T timer should be included in the shorter pattern.

Timers

There are two timers:

- T timer is 4 seconds.
- The long timer is 20 seconds.

These timers are used in input comparison, as described in "How the Input Is Processed" on page 53.

Transformation Keywords

Keyword	Description
prestrip	Strip the first n characters from the input before placing the call.
poststrip	Remove n number of characters from the end of the input before placing the call.
pre	Add the specified account prefix to the input before placing the call.
post	Attach the specified postfix to the input before placing the call.
replace	Replaces the input with the specified string before placing the call.

Order of Transformations

These transformations are always performed in the following order (the order in which the transformations are entered in the dial plan is not significant):

prestrip > poststrip > pre > post > replace

B.2 How the Input Is Processed

Comparing Input to the Dial Plan Patterns

The input is compared to each dial plan in turn, starting with the first listed account. The process is slightly different depending on how the call is placed:

- If the input was dragged or selected, then the entire input is compared to each dial plan. If a complete match is found, then the associated transformation is performed. If no match is found, no transformation is performed.
- If you are typing the input, the digits are compared one by one as they are entered. The comparison will result in one of the types of matches described in the table below.

It is possible for the same input to get matched to different dial plans depending on whether the input is entered on the fly or dragged. It is important to keep this in mind when designing dial plan patterns.

Type of Match	Conditions	Result if You Press Enter or Dial	Result if You Stop Typing
Partial match	The characters typed so far follow the pattern but there are not yet enough characters for a pending or complete match.		If you stop typing for the long timer length (20 seconds), no transformation is performed on the characters typed so far.
Pending match	 The pattern has no . (dot) but does have the T timer. There is a perfect match. The pattern has a . (dot) and the T timer. The minimum requirements are met. 	The transformation is performed.	If the T timer expires, the transformation is performed.

Results of the Comparison

Type of Match	Conditions	Result if You Press Enter or Dial	Result if You Stop Typing
Complete match	 The pattern has no . (dot) and no T timer. There is a perfect match. The pattern has a . (dot) but does not have the T timer. The minimum requirements are met. 	The transformation is performed.	The associated transformation is performed.
No match	The characters typed do not match the patterns for any dial plan.	No transformation is performed.	Nothing happens even after the T timer and long timer have expired.

B.3 Examples

Example 1

\a\a.T|xxxxxx.T;match=2;pre="9"

This simple example shows how to differentiate between a PSTN number and a softphone address, and how to add a "9" dialing prefix only to the PSTN number.

Example 2

```
3xxT|1xxxxxxxxx|[2-9]xxxxxxxxx|+x.T;match=2;pre="9";
match=3;pre="91";match=4;prestrip=1;pre="9011"
```

3xxT	The first pattern is any three-digit number beginning with 3. No transformation. The assumption is that this is an internal extension. The timer forces X-Lite to wait after detecting a three-digit number beginning with 3, in case you are actually dialing a local call starting with 3.
1xxxxxxxx	The second pattern is any eleven-digit number beginning with 1. Prefix with 9 and dial as is. The assumption is that this is a long-distance PSTN call within North America (within North America, all long-distance calls start with 1).
[2-9]xxxxxxx	The third pattern is any ten-digit number beginning with a number other than 1. The assumption is that this is a local PSTN call within a ten-digit dialing zone.
+x.T;	The fourth pattern is a number of any length that begins with +, to indicate an international PSTN call from North America. Delete the +, prefix with 9011 (011 is the number to access an international line from North America).
match=2;pre="9";	For the second pattern, prefix 9 to access an outside line.
match=3;pre="91";	For the third pattern, prefix 9 and 1 to access an outside line and enter the long-distance code.
match=4;prestrip=1;pre="9011"	For the fourth pattern, remove the + and prefix 9011 to access an outside line and enter the international code.

C Location of Files

C:\Documents and Settings\<user name>\Local Settings\Application Data\CounterPath Corporation\X-Lite

C:\Users\<user name>\AppData\Local\CounterPath Corporation\X-Lite

D Glossary

AEC	Acoustic echo cancellation. Processing of the audio or video signal to reduce the echo effect that can arise with a speakerphone or that can arise if the sound from the speakerphone or headphone leaks into the microphone.
AGC	Automatic gain control. Processing of the audio or video signal to adjust the microphone volume level so that the other party does not hear the distortion that might be caused by too high a microphone input or too low volume (due to too low input level).
AVI	Audio Video Interleave. A multimedia container format. AVI files contain both audio and video data in a standard container that allows simultaneous playback.
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 softphone address or 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.
HID	Human interface device. In X-Lite, if the headset is HID-compliant, the user can configure the buttons on the device to invoke functions on X-Lite such as answering an incoming call.
IM	Instant Messaging. A technology that lets users send text message and files for near instantaneous delivery and display on each others' computers.
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.
Presence	An instant messaging feature that allows users to share information about their availability, mood, location and so on.
Proxy	See SIP account.
PSTN	Public Switch Telephone Network. The traditional land-line phone network.

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RFC	Request for Comment. A document that describes an aspect of an internet technology. An RFC may be a proposed, draft or full internet standard.
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.
SIMPLE protocol	Session Initiation Protocol for Instant Messaging and Presence Leveraging Extensions. The instant messaging (IM) protocol followed by X-Lite. It encapsulate the rules for exchanging instant messages.
SIP	Session Initiation Protocol. The signaling protocol followed by X-Lite for handling phone calls.
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.
softphone address	The address used to connect to a SIP endpoint. In other words, the "phone number" used in a VoIP phone call. For example, sip:joseph@domainA.com.
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.
ТСР	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.
URI	Uniform Resource Identifier. A name or address that identifies a location on the world wide web. A softphone address is a type of URI.
URL	Uniform Resource Locator. A URI that both identifies a name or address and indicates how to locate it.
USB device	Universal Serial Bus device. A device that follows a specific communications standard. A headset may be a "USB type" of headset.
VAD	Voice Activity Detection. A technology that detects if audio is a human voice or background noise. X-Lite includes a feature (Preserve bandwidth on the Network panel of the Preferences window) that controls whether audio is transmitted when VAD determines that no one is actually speaking.
VoIP	Voice over Internet Protocol. A variation of IP used for sending voice data over the internet, in other words, used for making phone calls over the internet.
WAV	Or WAVE. A file format standard for storing audio on PCs.