



## Release Notes

### LifeSize Passport Series

#### Release v4.11.14

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Resolved Issues .....	1
Known Issues .....	1
Product Limitations .....	3
Interoperability .....	5
Interoperability Limitations .....	6
Documentation Errata .....	9
Contacting Technical Services .....	11

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For current product documentation, refer to [lifesize.com/support](http://lifesize.com/support). If you are using other LifeSize products with this release, read the latest release notes for those products for additional information.

## Resolved Issues

Following are the major resolved issues in this release. Numbers in parentheses are used for internal tracking.

- Stability and performance issues for interoperability with Skype calls have been addressed in this release. (END-21559)

## Known Issues

Following are known issues and their workarounds, if available. Numbers in parentheses are used for internal tracking.

### Video

- A Skype account with more than 100 contacts will show its presence status as offline to Skype clients and therefore, incoming Skype video calls to these Passport systems fail. (END-21559/21382)
- Calls are rejected even though auto answer is set. (END-19829)
- Audio and video corruption may occur in 5-way or greater calls with presentation, hosted by LifeSize Room 220. (END-21482)

### Audio

- G.722.1 is only supported for use with OCS and LifeSize Connections. (END-20495)

### Command Line Interface

- The `get audio codec` command erroneously lists `silk.24`, `silk.16`, `silk.12`, and `silk.8` in the list of supported audio codecs. These codecs are unsupported. (END-15164)
- System names that include special characters do not appear in full in the call history from the command line. (END-19068)

## Recording and Streaming

- Initiating a dial out recording from LifeSize Video Center to a LifeSize system with **Auto Record** enabled results in two recordings of the same content. (END-16743)
- When recording a multiway call (through LifeSize Video Center) from a system that is not the MCU, audio is not synchronized with video when the active speaker is at the near end. Audio is synchronized with video when the active speaker is at the far end. (END-16448)
- In a four-way call, the video recording icon might not appear for every participant. (END-19477)

## User Interface

- The five-minute setting for the in-call UI overlay fade out timer does not work. (END-17974)
- Call counter might not accurately represent remote calls with LifeSize Connections. (END-19146)
- Setting **Caller ID Timeout** to *Always On* does not maintain display of the ID. (END-19326)
- The presentation stream does not show the IP address of the presentation source. (END-19325)
- When H.323 tunneling is enabled, the gatekeeper username and password fields are not active.  
**Workaround:** Enter the gatekeeper username and password before enabling H.323 tunneling. (END-20873)

## Communications

- With STUN enabled and ICE disabled, systems detect and communicate using their public address (outside the router). Since most routers do not allow intra-LAN traffic through their public address, the media drops. (END-19656)
- UDP port 5070 is reserved for another use; therefore, you cannot use this port for SIP signaling. (END-20149)
- The SNMP management service on LifeSize systems may stop responding while communicating with SNMP management tools. If this problem occurs, disable the SNMP management service. You can also remove the LifeSize MIB from the SNMP management tool. It will not browse the LifeSize enterprise attributes causing the problem, which allows you to continue with generic monitoring of the LifeSize system. (END-15531)

## LifeSize Connections

- Applying a saved configuration in which LifeSize Connections was disabled resulted in Connections being set to enabled. **Workaround:** Manually reset Connections to disabled if this occurs. (END-18553)
- LifeSize Connections automatically becomes enabled when attempting to re-register with OCS. (END-18309)
- In a multiway call with LifeSize Connections participants, statistics are shown only for the first participant. (END-18368)
- LifeSize Connections participants cannot be entered in the Meetings directory. (END-18277)
- Passwords must be different from your Connections ID, or you can only log in using your email address. (CON-463)
- Multiple, subsequent changes to your password might result in lockout. (CON-467)
- Muted Connections participants become unmuted when a new participant joins. (CON-290)
- In a multiway call with presentation, when a second participant starts another presentation and a third participant changes the layout of the call, the presentation fails. (CON-622)
- If someone uses LifeSize Connections to participate in a conference call, the redial list is populated with the usernames of everyone who participated in the call. (END-18941)
- When a system is deactivated and reactivated again, it does not re-register to Connections automatically. You must manually re-register the device. (END-19224)

# Product Limitations

Following are known limitations with this software version. Numbers in parentheses are used for internal tracking.

## Video

- Support for SIP dual video is subject to the following limitations:
  - Dual video is available in calls with LifeSize systems and Polycom SIP dual-video systems only.
  - In calls with bandwidth at 192 and 256 kb/s, the bandwidth for SIP dual video is 64 kb/s. In calls with bandwidth at 320 kb/s or greater, SIP dual video bandwidth is 128 kb/s. Adjusting bandwidth allocated to the presentation stream using the **Video Bandwidth Balance** preference has no effect.
  - SIP dual video is not supported with Cisco Unified Communications Manager. (END-10870)
- Some older laptops that use Intel or Nvidia graphics chipsets do not support the full set of resolutions that LifeSize supports. As a result, the resolution used on the LifeSize system is lower than that selected on the laptop. (END-15429)
- When all system inputs are connected to active devices, no video or corrupt video might appear in the video input selection box when you change video inputs. In a dual display configuration, the near end presentation video on the secondary display might be garbled. **Workaround:** Unplug one or more active devices. (END-15878, END-15877, END-15876)
- When you set **Administrator Preferences : Video : Video Preferences : Video Bandwidth Balance** to 10% / 90%, the presentation bandwidth is not actually 90% of the total, but is closer to 55%. (END-15884)

## Audio

- You might experience echoes if HDMI out is configured on both ends of a call. HDMI introduces a variable delay into the audio signal, resulting in acoustic echo. (END-14046)
- For audio only calls, dialing an IP address is successful even if voice dialing is set to *ISDN*. When voice calls are set to pulse, you can place an IP H.323 voice call. (END-9307)
- Adding an audio caller to a conference prior to the video calls becoming established might result in downgraded bandwidth. **Workaround:** Add additional audio calls after initial video participants have fully connected. (END-18869)

## Network/Communications

- When placing a call from a system behind a firewall (or without a static NAT configuration in the firewall) the call might complete and camera control from the system behind the firewall (the private system) to the system on the public internet (the public system) might work. However, FECC from the public system to the private system either does not work or works intermittently. LifeSize recommends deploying LifeSize Transit for this configuration. (END-12129)
- If your network does not support IPv6 auto configuration and you set the **IPv6** preference to *Enabled* and the **IPv6 Configuration** preference to *Auto* in **Administrator Preferences : Network : General**, upon reboot, the system fails to complete the initialization process. Restore the system configuration to its default values by pressing the reset button on the back of the codec or by turning off IPv6 in the user interface. Refer to the *LifeSize Video Communications Systems User and Administrator Guide* for more information about using the reset button. (END-13225)
- The LAN port might be unable to establish a link when you connect the LifeSize system to a switch and both devices are set to automatically negotiate the speed. If this event occurs, set the speed on the switch and the **Network Speed** preference on the LifeSize system to either 100 Mb/s (full duplex) or 10 Mb/s (full duplex). In rigid configurations, LifeSize recommends that you connect a switch to the network and then connect the device to this switch. (END-6539)
- OCS registration might fail from the web administration interface. **Workaround:** Register from the main screen interface of your video system.
- You must set the TLS port to a different value than the UDP and TCP signaling ports. Otherwise, you will be unable to place or receive SIP calls. (END-19088)

- Changing the UDP port range in **Administrator Preferences : Network : Reserved Ports** requires a system reboot to take effect. The system automatically reboots when the TCP port range is changed on this page, but not the UDP port range. **Workaround:** If you are changing only the UDP port range, reboot the system after making the change. (END-12524)
- Systems registered to OCS support only TCP transport for incoming calls. Incoming SIP calls from non-OCS clients fail. (END-20165)
- Because software upgrades require the system to communicate with the license server to perform a license check, DNS resolution must be enabled either through DHCP or by specifying **DNS Servers** in **Administrator Preferences : Network : General**. If you disable DHCP, set **DNS Servers** and specify the IP address, subnet mask, and gateway to facilitate software upgrades. (END-14192)

### LifeSize Connections

- A minimum of 30 UDP/TCP ports are required for the best performance in LifeSize Connections calls between the Client and a LifeSize device. (END-19286)
- Changing your Connections password ends the current session and requires you to log in again. (CON-305)
- For video calls, Connections attempts to connect before SIP PBX. For audio calls, SIP PBX attempts to connect before Connections. If the SIP PBX connects successfully, Connections is not tried and might appear to be unsuccessful if registered to a SIP server. (END-19495)
- If you register to the Connections server with greater than 1000 entries, the 1000 limit results in a subset that might differ on each system. (CON-680)

### Recording and Streaming

- Although LifeSize Video Center can generate 10-digit recording keys, LifeSize video communications systems cannot accept them and, instead, produce an error message. **Workaround:** Limit your recording keys to nine digits. (END-15471)
- When the MCU is recording a multiway call, the recording might produce blank video for a presentation started on one non-MCU system before an initial presentation from another non-MCU system has ended. **Workaround:** When sharing multiple presentations from different non-MCU systems while recording a multiway call, ensure that one presentation ends before starting the next one. (END-15604)
- The default recording layout on LifeSize Video Center determines whether video from the near end, far end, or both are recorded. This setting defaults to **Use the video system setting**, located at **Administrator Preferences : Video : Record and Stream : Default Recording**. The setting in the recording key overrides the setting on the LifeSize video system initiating the recording. Selecting **Far video only** in a multiway call records only the first far end caller. (END-16964)
- Recordings display the red recording indicator during playback in multiway calls. **Workaround:** Initiate the recording from a participant system instead of the host. (END-18526)

### User Interface

- The call statistics in the user interface of a LifeSize system that is participating in a multiway call hosted by another LifeSize system do not match the call statistics that appear in the web administration interface. The call statistics that appear in the web administration interface show the received resolution and bandwidth for the connection between the participant and the MCU. The statistics that appear in the user interface show resolutions for each participant in the call and bandwidths that are the result of dividing the received bandwidth by the number of participants in the call. (END-13946)
- LifeSize v3.5.x with Flash Player v10 fails during upgrades. If you are using Flash Player v10 with LifeSize v3.5.x, downgrade to Flash Player v9 before upgrading to LifeSize v4.x. (END-9548)
- Caller ID of PSTN calls during call waiting is not supported. (END-1201)
- Calls placed from the **Call Manager** in the web administration interface always appear on the **Redial** list with *Auto* as the bandwidth and protocol, regardless of the actual bandwidth and protocol specified when the call was first placed. (END-6497)

- In calls with systems that use IPv6 addresses, call statistics incorrectly show zero as the value of the packet loss for transmitted video. (END-6127)
- When recording a multiway SIP call, virtual multiway is not available after a presentation ends.  
**Workaround:** On the systems on which virtual multiway is unavailable, press the mute button on the remote control twice (mute and unmute). Virtual multiway layout data reappears.
- The corporate directory may not load in the web administration interface if it contains greater than the 1000 limit of entries. **Workaround:** View the directory from the main screen. (END-19761, END-19584)

## Command Line Interface

- Error 02, file error is returned in the automation command line interface if you use `set camera position -P` to a preset that has not been set. The proper error code is 0d, No data available. (END-16273)
- Redial list does not appear for specific bandwidth bitrates when dialed from the command line. **Workaround:** Dial the call from the main screen. (END-18794)
- If a call placed from the meetings directory using the command line interface is not answered within three seconds by the first participant dialed, all other participants in the meetings entry are then dialed but become unavailable. If the first participant subsequently answers, the call becomes a two-way call.  
**Workaround:** Add the remaining participants to the call from the video system's user interface. (END-13657)
- You are unable to dismiss a rejected or invalid call from the command line, even though entry 0 is present.  
**Workaround:** Dismiss the call from the user interface. (END-12246)

## Interoperability

LifeSize video communications systems with this software release are supported with the following devices.

Avaya	SIP Enablement Services: 5.1.x Communication Manager: 6.1.x 1-X Communicator: 6.1.x
Asterisk	Asterisk: 1.4.22.1
Browser support	Microsoft Internet Explorer v7, v8 Apple Safari for Mac v4.0.4 Adobe Flash Player v9, v10
Cisco	IOS GK: v12.4(17a) ASA 5510 Firewall: v8.0(4) UCM: v7.13.10000-11e
Codian	MCU 4220: 4.3(2.18) MCU 4505: 4.3(2.18)
Microsoft	Office Communications Server 2007: 3.5.6907.0 Office Communicator 2007: 3.5.6907.206 Lync 2010 v4.0.7577.0 (OS-Windows Vista, Windows 7, Lync 2010 for Mac) Mac 14.0.1 (111018) (Mac OS 10.7.2)
Polycom	HDX Series: 3.0.4 RMX: 7.6.0 VSX: 9.0.6.2 BFCP for HDX: 3.0.2 <i>220s only:</i> VVX: 3.3.1 IP 4000: 3.1.6 IP 7000: 3.3.1
sipX	sipXecs: 4.2.1

Skype	Passport Only: Windows 7 with Skype version 6.7.60.102 Mac OS 10.7.2 with Skype version 6.0.59.126 Mac Air 10.6.8 with Skype version 6.5.0.158  Samsung Galaxy S3 with Skype version 3.2.0.6673 Apple iPad 3 with Skype version 4.8.324
Sony	Sony PCS XG-80 2.34
Tandberg	Edge 95 MXP: F9.0.2 C Series: TC4.2.3 Tandberg VCS X5.1.1
USB devices	Sewell: AP1102 StarTech: ICUSB232

## Interoperability Limitations

Following are the known limitations with third party products. Numbers in parentheses are for internal tracking.

### General

- Enabling static NAT on a LifeSize system and then placing a call through a router with an application-level gateway or protocol fixup that modifies call control traffic might result in no video and/or audio at either the near end or far end of the call. Depending on the router, disabling static NAT on the LifeSize system might resolve this issue. LifeSize recommends disabling fixup on the router. (END-6920)
- The mute button on a third party microphone connected to the microphone input on a LifeSize system might not function properly. For best results, use a LifeSize MicPod when connecting a microphone to the microphone input. (END-8860)

### Avaya

- LifeSize video systems registered to Avaya PBX using G722.1c and SIP may experience audio degradation. (END-21308)

### Cisco

- LifeSize does not support the Cisco proprietary SCCP protocol that is required to use call forwarding or voicemail with the Cisco IP Phone. (END-3320)
- SIP dual video is not available in SIP calls between LifeSize video communications systems connected through Cisco Unified Communications Manager. (END-10870)
- H.239 might not work through your Cisco PIX or ASA (Adaptive Security Appliance) firewall/ASA device. The Cisco fixup protocol did not recognize H.239 and terminated a call if it attempted to open an H.239 stream.  
**Workaround:** Upgrade to ASA v8.2.1 or later. (END-1611)

### Codian

- In a multiway call hosted by a Codian 4220, the MCU first uses the H.263+ protocol and then switches to H.264. The frame rate remains at 15 f/s for the duration of the call. **Workaround:** Disable H.263 and H.263+ on the Codian 4220 MCU. (END-17361)
- In a multiway call with a Codian MCU, video and text that appear in the display might appear cropped on the bottom or sides of the image. **Workaround:** Add the LifeSize system to the directory on the Codian MCU and adjust the border size to 2 or 3, depending on your display. You can adjust the border size from the LifeSize system during a call by using far end camera control. With the far end camera of the Codian MCU selected, press the zoom out key on the remote control, ensure that **Border width** is selected, and then press the right arrow key to change the border width. (END-9248)

- You might experience poor quality audio in calls with a Codian MCU 42XX that has system software earlier than v2.1. To resolve this issue, upgrade the Codian MCU to v2.1 or later. (END-5858)
- When creating a dial-out conference on the Codian MCU, the first two systems connect without issues, but any participant after that is reduced to 256k. (END-12277)
- The Codian 4505 MCU does not support 1080p decode. It can support 1080p encode only if the peer device supports it. LifeSize systems can receive 1080p30 video from the Codian MCU only if it is in 2x2 layout. If video is set to full screen, it displays 1280x720p30 receive and transmit. (END-12220)
- A known issue with the Codian MCU results in distorted video on a LifeSize system in a 40-device conference to the Codian 4520. (END-10794)

## Polycom

- Audio only calls fail in calls hosted by Polycom RMX. (END-17382)
- Distorted video appears on LifeSize systems in a 384 kb/s call hosted by Polycom RMX. (END-17162, END-17163)
- Distorted video appears on a LifeSize system in an encrypted 1024 kb/s call hosted by Polycom RMX. (END-17165)
- A participant on a LifeSize system joining a call in progress that is hosted by Polycom RMX is unable to view an ongoing presentation. **Workaround:** Add all call participants before starting the presentation. (END-17243)
- Audio is not synchronized with video in 1080p30 calls with Polycom HDX 8000 and Polycom 8006 systems. (END-12251, END-17318)
- During a two-way call between LifeSize and Polycom HDX 4000 systems, audio is not synchronized with video on the Polycom HDX 4000 system. After approximately six minutes into the call, audio and video are synchronized. (END-17173)
- Audio is not synchronized with video in calls hosted by Polycom RMX when the LifeSize systems are connected at different speeds. Latency increases as the duration of the call increases. (END-7012)
- Video goes blank on Polycom HDX 9002 in a multiway call when video is renegotiated from H263+ to H264. (END-15426)
- LifeSize video communications systems might not have far end camera control on Polycom VSX 8000 in two-way or multiway calls. (END-15854)
- When a LifeSize system dials the E.164 address for a Polycom system through a gatekeeper, the audio might be distorted because of a byte swap issue on G722.1C codecs. **Workaround:** Contact Technical Services to override the byte swap. With this fix enabled, you might experience distorted audio on previously functioning G722.1C codecs. (END-13752)
- A LifeSize system in a multiway call with Polycom VSX 8000 or VSX 7000 as the MCU cannot send a presentation from a device connected to the SD input due to limitations in negotiating a compatible resolution for the video. The same issue occurs if the presentation device is connected to the VGA input on the LifeSize system. **Workaround:** If the VGA input is used, change the resolution on the VGA input device to 1024x768 or greater. (END-7611) (END-9357)
- LifeSize systems do not receive a presentation from Polycom systems when Polycom RMX is the MCU due to features sent from the MCU that are not supported on LifeSize systems. (END-10310)
- Distorted video appears on a LifeSize system when calling a Polycom device (for encoded resolutions that do not match the source aspect ratio). (END-12002)
- SIP calls from Polycom HDX to LifeSize systems fail through the sipXecs registrar. (END-18828)
- Far end camera control is not supported between LifeSize systems and Polycom HDX4000. (END-18532)
- Polycom HDX 9002 disconnects when VSX 7000 joins an AES call with LifeSize systems and presentation. **Workaround:** Add participants (HDX and VSX) first and then start presentation.
- In a call between a Polycom VSX 8000 and LifeSize Room 220 hosted by a Polycom RMX, the presentation video on the LifeSize Room 220 displays colored video artifacts. (END-19634)

## ShoreTel

- When a call is placed between two LifeSize systems that are using the ShoreTel PBX, either the call does not connect or the presentation fails. Also, a SIP call to LifeSize Multipoint 230 using the ShoreTel PBX disconnects after the first ring. **Workaround:** Disable presentations on the LifeSize system. (END-17079, END-12263, END-18893)
- Call transfer is unsuccessful using the ShoreTel PBX. (END-15969)

## SipX

- A LifeSize system registers successfully with SipXecs PBX despite being unauthorized. (END-11883)

## Sony

- Audio is not synchronized with video from a Sony XG80 system in a three-way call with two LifeSize systems. (END-17420)
- A LifeSize system in a two-way call with Sony PCS-G70 (v2.63) can start and stop only one presentation during the call. Attempting to start a subsequent presentation fails. The same issue occurs if the presentation is started and restarted on the Sony PCS-G70. **Workaround:** Hang up the call, place the call again, and start the presentation. (END-10874, END-15411, END-15332)
- Frozen video and packet loss might occur in H.323 calls with Sony XG-80. (END-18506)
- You must disable network presentations to successfully place calls to Sony XG-80. (END-18212)

## Tandberg

- Registration with a Tandberg VCS gatekeeper is initially shown as successful on a LifeSize system after removing the gatekeeper authentication credentials on the LifeSize system. (END-17266)
- H.460 enabled LifeSize systems registered to Tandberg VCS Expressway have their presentations blocked. **Workaround:** Enable H.460.19 demultiplexing mode in Tandberg VCS Expressway.
  1. Navigate to **VCS Configuration : Expressway : Locally registered devices**.
  2. Set H.460.19 demultiplexing mode to *On*. (END-14559)
- Audio is not synchronized with video in call to Tandberg Edge 95 MXP. (END-14795)
- A SIP call placed from a LifeSize system configured to use UDP/TCP signaling for SIP calls to a Tandberg MXP device using TLS and security set to *Auto* fails. **Workaround:** Place the call from the Tandberg device or disable the auto feature on the Tandberg device. (END-10462)
- When a two-way ISDN call is dialed from a LifeSize system to a Tandberg 6000 MXP, the message "No incoming video" flashes on the Tandberg side just after call setup. After a couple of seconds this message is cleared and video appears. (END-9724)
- Tandberg Edge 95 systems receive a maximum resolution of 720x400 in calls with LifeSize systems. (END-12440, END-15849)

# Documentation Errata

## Registering LifeSize Systems with LifeSize Connections

You must add a LifeSize Passport Connect system to the Connections registry before it can use LifeSize Connections to place or receive a call. Use the Connections Administrator Console to register systems individually or as a group, depending on the level of system information to which you have access.

To access the Administrator Console, open a web browser and enter the URL that was provided with your authentication credentials when you purchased LifeSize Connections. Enter the administrator username and password and click **Login**.

### Registering Systems Individually

If you know the serial numbers of your LifeSize Passport Connect systems, use them to add your systems to the Connections registry. If you do not know the serial numbers of your systems, create Connections IDs and obtain registration keys for each of them.

1. In the Administrator Console, click the **Rooms** tab.

2. Click **Add Rooms Individually**.

If you do not know the serial number of your system, proceed to step 3.

If you know the serial number of your system, proceed to step 4.

3. Click **Provision Manually**.

The **Registration Key** text box populates with a numeric value. Use this key, along with the Connections ID that you specify in the next step, to connect to LifeSize Connections.

4. Specify the following information:

- A name that describes the location or function of the system.
- The serial number of the system, if applicable.
- A unique Connections ID that is associated with the system.

A valid Connections ID is between 4 and 32 alphanumeric characters in length and begins with a lowercase letter that is followed by any combination of the following characters:

- lowercase letters (a-z)
- numbers (0-9)
- periods (.)
- underscores (\_)

Dashes can be used after the fourth character in a Connections ID.

5. Click **Add**.

The registered system appears in the **Room List** and can connect to LifeSize Connections. Repeat the previous four steps as necessary to register your remaining systems.

## Connecting a LifeSize System to LifeSize Connections

Connect to LifeSize Connections from your LifeSize system by setting **Administrator Preferences : Communications : LifeSize Connections: Connections** to *Enabled* and clicking **Connect**. Do not enter any additional information at this point. If the serial number of your LifeSize system is included in the Connections registry, the LifeSize Connections service recognizes the number and establishes a connection to your system. If the Connections registry does not include the serial number, specify a valid Connections ID and registration key and click **Connect**. If an attempt to connect to LifeSize Connections is unsuccessful, a message identifies the condition that prevented the connection.

A Connections icon that appears next to an entry in your contacts, the directory, or the redial list indicates that the system is logged in to LifeSize Connections. The color of the icon indicates the status of the system's connection to LifeSize Connections, as follows:

- Green – Available
- Orange – In a call
- Red – Do not disturb
- Silver – Inactive
- Gray – Offline
- Gray crosshatch – Away
- Black – No presence-related information is available

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To access the Administrator Console, open a web browser and enter the URL that was provided with your authentication credentials when you purchased LifeSize Connections. Enter the administrator username and password and click **Login**.

The following table lists the messages that might appear if you are unable to connect to LifeSize Connections.

Message	Description	Recommendation
Authentication Unnecessary	You do not need a Connections ID or registration key to access LifeSize Connections.	Enable Connections and click <b>Connect</b> .
Conferencing Unavailable	Conferencing is temporarily unavailable.	Try again later.
Conflicting Call	You might be trying to place a Connections call while a call that was not placed through Connections is in progress. Conversely, you might be trying to place a call through a means other than Connections while a Connections call is in progress.	Disconnect the responsible call and try again.
Invalid Connections ID	The Connections ID is invalid.	Contact your administrator.
Invalid Registration Key	The registration key is invalid.	Contact your administrator.
Server Unavailable	LifeSize Connections is temporarily unavailable.	Try again later.
Server Unreachable	LifeSize Connections is temporarily unavailable.	Check your network connection and try again later.

SIP Disabled	SIP is disabled.	Enable SIP and try again.
Subscription Expired	You do not have permission to access LifeSize Connections.	Contact your administrator.
Transit Conflict	The system is connected to LifeSize Transit.	Disable Transit and try again.
Unregistered System	The system is not registered with LifeSize Connections.	Contact your administrator.

## Contacting Technical Services

LifeSize Communications welcomes comments about our products and services. Send feedback about this or any LifeSize product to [feedback@lifesize.com](mailto:feedback@lifesize.com). Refer to [lifesize.com/support](http://lifesize.com/support) for additional ways to contact LifeSize Technical Services.