

AXIS C1710 Network Display Speaker

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Solution overview

This manual describes how you make the device accessible to your audio system, and how to configure the device directly from its interface.

If you are using an audio or video management software, you can use that software for configuring the device. The following management software are available for controlling your audio system:

- **AXIS Audio Manager Edge** – Audio management software for small systems. Comes pre-installed on all audio devices with a firmware equal to or higher than 10.0.
 - *AXIS Audio Manager Edge user manual*
- **AXIS Audio Manager Pro** – Advanced audio management software for large systems.
 - *AXIS Audio Manager Pro user manual*
- **AXIS Camera Station Pro** – Advanced video management software for large systems.
 - *AXIS Camera Station Pro user manual*

For more information, see *Audio management software*.



To watch this video, go to the web version of this document.

An overview of how network audio works.

Installation

This video is an example of how to install AXIS C1710. For complete installation instructions as well as safety information, see the installation guide: axis.com/products/axis-c1710/support



To watch this video, go to the web version of this document.

Get started

⚠ WARNING

Flashing or flickering lights can trigger seizures in persons with photosensitive epilepsy.

Find the device on the network

To find Axis devices on the network and assign them IP addresses in Windows®, use AXIS IP Utility or AXIS Device Manager. Both applications are free and can be downloaded from axis.com/support.

For more information about how to find and assign IP addresses, go to *How to assign an IP address and access your device*.

Note

When the product is in factory defaulted state, its IP address will be shown on the display. Once you have opened the web interface and created a user, the display will show the clock and date instead.

Browser support

You can use the device with the following browsers:

	Chrome™	Firefox®	Edge™	Safari®
Windows®	recommended	✓	recommended	
macOS®	recommended	✓	recommended	✓*
Linux®	recommended	✓	recommended	
Other operating systems	✓	✓	✓	✓

Open the device's web interface

1. Open a browser and type the IP address or host name of the Axis device.
If you do not know the IP address, use AXIS IP Utility or AXIS Device Manager to find the device on the network.
2. Type the username and password. If you access the device for the first time, you must create an administrator account. See .

For descriptions of all the controls and options in the device's web interface, see .

Create an administrator account

The first time you log in to your device, you must create an administrator account.

1. Enter a username.
2. Enter a password. See .
3. Re-enter the password.
4. Accept the license agreement.
5. Click **Add account**.

Important

The device has no default account. If you lose the password for your administrator account, you must reset the device. See .

Secure passwords

Important

Use HTTPS (which is enabled by default) to set your password or other sensitive configurations over the network. HTTPS enables secure and encrypted network connections, thereby protecting sensitive data, such as passwords.

The device password is the primary protection for your data and services. Axis devices do not impose a password policy as they may be used in various types of installations.

To protect your data we strongly recommend that you:

- Use a password with at least 8 characters, preferably created by a password generator.
- Don't expose the password.
- Change the password at a recurring interval, at least once a year.

Make sure that no one has tampered with the device software

To make sure that the device has its original AXIS OS, or to take full control of the device after a security attack:

1. Reset to factory default settings. See .
After the reset, secure boot guarantees the state of the device.
2. Configure and install the device.

Configure your device

Calibrate and run a remote speaker test

You can run a speaker test to verify from a remote location that a speaker is working as intended. The speaker performs the test by playing a series of test tones that are registered by the built-in microphone. Every time you run the test, the registered values are compared with the values that were registered during the calibration.

Note

The test must be calibrated from its mounted position at the installation site. If the speaker is moved or if its local surroundings change, for instance if a wall is built or removed, the speaker should be re-calibrated.

During calibration, it is recommended that someone is physically present at the installation site to listen to the test tones and ensure that the test tones are not muffled or blocked by any unintended obstructions in the speaker's acoustic path. To avoid hearing damage, the installer should maintain a safe distance from the speaker during the test.

1. Go to the device interface > **Audio** > **Speaker test**.
2. To calibrate the audio device, click **Calibrate**.

Note

Once the Axis product is calibrated, the speaker test can be run at any time.

3. To run the speaker test, click **Run the test**.

Note

It is also possible to run the calibration by pressing the control button on the physical device. See to identify the control button.

Create a light profile

A profile is a collection of set configurations. You can have up to 30 profiles per light source with different priorities and patterns. You can only have one light source for a profile, so if you want profiles for both light sources, you need to create at least two profiles.

Create a new profile:

1. Go to **Light > Profiles** and click  **Create**.
2. Enter a **Name** and **Description**.
3. Select the **Light** settings that you want for your profile.
4. Set the light **Priority** and click **Save**.
5. To edit a profile, click  and select **Edit**.

Set up direct SIP (P2P)

Use peer-to-peer when the communication is between a few user agents within the same IP network and there is no need for extra features that a PBX-server could provide. To better understand how P2P works, see .

For more information about setting options, see .

1. Go to **System > SIP > SIP settings** and select **Enable SIP**.
2. To allow the device to receive incoming calls, select **Allow incoming calls**.
3. Under **Call handling**, set the timeout and duration for the call.
4. Under **Ports**, enter the port numbers.
 - **SIP port** – The network port used for SIP communication. The signaling traffic through this port is non-encrypted. The default port number is 5060. Enter a different port number if required.

- **TLS port** – The network port used for encrypted SIP communication. The signaling traffic through this port is encrypted with Transport Layer Security (TLS). The default port number is 5061. Enter a different port number if required.
 - **RTP start port** – Enter the port used for the first RTP media stream in a SIP call. The default start port for media transport is 4000. Some firewalls might block RTP traffic on certain port numbers. A port number must be between 1024 and 65535.
5. Under **NAT traversal**, select the protocols you want to enable for NAT traversal.

Note

Use NAT traversal when the device is connected to the network from behind a NAT router or a firewall. For more information see .

6. Under **Audio**, select at least one audio codec with the desired audio quality for SIP calls. Drag-and-drop to change the priority.
7. Under **Additional**, select additional options.
- **UDP-to-TCP switching** – Select to allow calls to switch transport protocols from UDP (User Datagram Protocol) to TCP (Transmission Control Protocol) temporarily. The reason for switching is to avoid fragmentation, and the switch can take place if a request is within 200 bytes of the maximum transmission unit (MTU) or larger than 1300 bytes.
 - **Allow via rewrite** – Select to send the local IP address instead of the router's public IP address.
 - **Allow contact rewrite** – Select to send the local IP address instead of the router's public IP address.
 - **Register with server every** – Set how often you want the device to register with the SIP server for the existing SIP accounts.
 - **DTMF payload type** – Changes the default payload type for DTMF.
8. Click **Save**.

Set up SIP through a server (PBX)

Use a PBX-server when user agents will communicate within and outside the IP network. Additional features could be added to the setup depending on the PBX-provider. To better understand how P2P works, see .

For more information about setting options, see .

1. Request the following information from your PBX provider:
 - User ID
 - Domain
 - Password
 - Authentication ID
 - Caller ID
 - Registrar
 - RTP start port
2. To add a new account, go to **System > SIP > SIP accounts** and click **+ Account**.
3. Enter the details you received from your PBX provider.
4. Select **Registered**.
5. Select a transport mode.
6. Click **Save**.
7. Set up the SIP settings the same way as for peer-to-peer. See for more information.

Set up rules for events

You can create rules to make your device perform actions when certain events occur. A rule consists of conditions and actions. The conditions can be used to trigger the actions. For example, the device can play an audio clip according to a schedule or when it receives a call, or send an email if the device changes IP address.

To learn more, check out our guide *Get started with rules for events*.

Send an email if a speaker test fails

In this example the audio device is configured to send an email to a defined recipient when a speaker test fails. The speaker test is configured to be performed 18:00 every day.

1. Set up a schedule for the speaker test:
 - 1.1. Go to the device interface > **System** > **Events** > **Schedules**.
 - 1.2. Create a schedule that starts at 18:00 and ends at 18:01 every day. Name it "Daily at 6pm".
2. Create an email recipient:
 - 2.1. Go to the device interface > **System** > **Events** > **Recipients**.
 - 2.2. Click **Add recipient**.
 - 2.3. Name the recipient "Speaker test recipients"
 - 2.4. Under **Type**, select **Email**.
 - 2.5. Under **Send email to**, enter the email addresses of the recipients. Use commas to separate multiple addresses.
 - 2.6. Enter the details for the email account of the sender.
 - 2.7. Click **Test** to send a test email.

Note

Some email providers have security filters that prevent users from receiving or viewing large attachments, from receiving scheduled emails and similar. Check the email provider's security policy to avoid delivery problems and locked email accounts.

- 2.8. Click **Save**.
3. Set up the automated speaker test:
 - 3.1. Go to the device interface > **System** > **Events** > **Rules**.
 - 3.2. Click **Add a rule**.
 - 3.3. Enter a name for the rule.
 - 3.4. Under **Condition**, select **Schedule** and select from the trigger list
 - 3.5. Under **Schedule**, select your schedule ("Daily at 6pm").
 - 3.6. Under **Action**, select **Run automatic speaker test**.
 - 3.7. Click **Save**.
4. Set up the condition for sending an email when the speaker test fails:
 - 4.1. Go to the device interface > **System** > **Events** > **Rules**.
 - 4.2. Click **Add a rule**.
 - 4.3. Enter a name for the rule.
 - 4.4. Under **Condition**, select **Speaker test result**.
 - 4.5. Under **Speaker test status**, select **Didn't pass the test**.
 - 4.6. Under **Action**, select **Send notification to email**.
 - 4.7. Under **Recipient**, select your recipient ("Speaker test recipients")
 - 4.8. Enter a subject and a message, and click **Save**.

Play audio when a camera detects motion

This example explains how to set up the audio device to play an audio clip when an Axis network camera detects motion.

Prerequisites

- The Axis audio device and Axis network camera are located on the same network.
 - The motion detection application is configured and running in the camera.
1. Prepare an audio clip link:
 - 1.1. Go to **Audio > Audio clips**.
 - 1.2. Click  > **Create link** for an audio clip.
 - 1.3. Set the volume and number of times to repeat the clip.
 - 1.4. Click the copy icon to copy the link.
 2. Create an action rule:
 - 2.1. Go to **System > Events > Recipients**.
 - 2.2. Click **+ Add recipient**.
 - 2.3. Type a name for the recipient, for example "Speaker".
 - 2.4. Select **HTTP** from the **Type** drop-down list.
 - 2.5. Paste the configured link from the audio device in the **URL** field.
 - 2.6. Enter the user name and password of the audio device.
 - 2.7. Click **Save**.
 - 2.8. Go to **Rules** and click **+ Add a rule**.
 - 2.9. Type a name for the action rule, for example "Play clip".
 - 2.10. From the **Condition** list, select a video motion detection alternative under **Applications**.

Note

If there are no options for video motion detection, then go to **Apps**, click **AXIS Video Motion Detection** and turn on motion detection.

- 2.11. From the **Action** list, select **Send notification through HTTP**.
- 2.12. Under **Recipient**, select your recipient.
- 2.13. Click **Save**.

Stop audio with DTMF

This example explains how to:

- Configure DTMF on a device.
 - Set up an event to stop the audio when a DTMF command is sent to the device.
1. Go to **System > SIP > SIP settings**.
 2. Make sure **Enable SIP** is turned on.
If you need to turn it on, remember to click **Save** afterwards.
 3. Go to **SIP accounts**.
 4. Next to the SIP account, click  > **Edit**.
 5. Under **DTMF**, click **+ DTMF sequence**.
 6. Under **Sequence**, enter "1".
 7. Under **Description**, enter "stop audio".

8. Click **Save**.
9. Go to **System > Events > Rules** and click **+ Add a rule**.
10. Under **Name**, enter "DTMF stop audio".
11. Under **Condition**, select **DTMF**.
12. Under **DTMF Event ID**, select **stop audio**.
13. Under **Action**, select **Stop playing audio clip**.
14. Click **Save**.

Set up audio for incoming SIP calls

You can set up a rule that plays an audio clip when you receive a SIP call.

You can also set up an additional rule that answers the SIP call automatically after the audio clip has ended. This can be useful in cases where an alarm operator wants to call the attention of someone near an audio device and establish a line of communication. This is done by making a SIP call to the audio device, which will play an audio clip to alert the persons near the audio device. When the audio clip has stopped playing, the SIP call is automatically answered by the audio device and communication between the alarm operator and the persons near the audio device can take place.

Enable SIP settings:

1. Go to the device interface of the speaker, by entering its IP address in a web browser.
2. Go to **System > SIP > SIP settings** and select **Enable SIP**.
3. To allow the device to receive incoming calls, select **Allow incoming calls**.
4. Click **Save**.
5. Go to **SIP accounts**.
6. Next to the SIP account, click  > **Edit**.
7. Uncheck **Answer automatically**.

Play audio when a SIP call is received:

1. Go to **Settings > System > Events > Rules** and add a rule.
2. Type a name for the rule.
3. In the list of conditions, select **State**.
4. In the list of states, select **Ringing**.
5. In the list of actions, select **Play audio clip**.
6. In the list of clips, select the audio clip you want to play.
7. Select how many times to repeat the audio clip. 0 means "play once".
8. Click **Save**.

Answer the SIP call automatically after the audio clip has ended:

1. Go to **Settings > System > Events > Rules** and add a rule.
2. Type a name for the rule.
3. In the list of conditions, select **Audio clip playing**.
4. Check **Use this condition as a trigger**.
5. Check **Invert this condition**.
6. Click **+ Add a condition** to add a second condition to the event.
7. In the list of conditions, select **State**.
8. In the list of states, select **Ringing**.
9. In the list of actions, select **Answer call**.

10. Click **Save**.

Show a message on the display

Show a text message on the display of an individual speaker:

1. Open the web interface by entering the IP address of the speaker in a web browser.
2. Go to **System > Events > Rules**.
3. Click **+ Add a rule**.
4. Name your rule.
5. Under **Condition**, select **Manual trigger**.
6. Under **Action**, select **Speaker display notification**.
7. Under **Message**, enter the text you want to show on the speaker.
8. Fill in the rest of the fields according to your needs.
9. Click **Save**.
10. Make sure the rule is turned on.
11. Go to **Manual triggers**.
12. Turn on **Trigger 1**.
The notification is now sent to the speaker.

Trigger the signaling LED with the PIR sensor

1. Open the web interface by entering the IP address of the speaker in a web browser.
2. Go to **System > Events > Rules**.
3. Click **+ Add a rule**.
4. Name your rule.
5. Under **Condition**, select **PIR sensor**.
6. Under **Action**, select **Light and siren > Run light profile while the rule is active**.
7. Under **Profile**, select one of the signaling LED profiles (see the tool tip when hovering an alternative).
To create a new profile, see .
8. Click **Save**.

The web interface

To reach the device's web interface, type the device's IP address in a web browser.

Note

Support for the features and settings described in this section varies between devices. This icon  indicates that the feature or setting is only available in some devices.



Show or hide the main menu.



Access the release notes.



Access the product help.



Change the language.



Set light theme or dark theme.



The user menu contains:

- Information about the user who is logged in.
-  **Change account** : Log out from the current account and log in to a new account.
-  **Log out** : Log out from the current account.



The context menu contains:

- **Analytics data**: Accept to share non-personal browser data.
- **Feedback**: Share any feedback to help us improve your user experience.
- **Legal**: View information about cookies and licenses.
- **About**: View device information, including AXIS OS version and serial number.

Status

Device info

Shows the device information, including AXIS OS version and serial number.

Upgrade AXIS OS: Upgrade the software on your device. Takes you to the Maintenance page where you can do the upgrade.

Time sync status

Shows NTP synchronization information, including if the device is in sync with an NTP server and the time remaining until the next sync.

NTP settings: View and update the NTP settings. Takes you to the **Time and location** page where you can change the NTP settings.

Security

Shows what kind of access to the device that is active, what encryption protocols are in use, and if unsigned apps are allowed. Recommendations to the settings are based on the AXIS OS Hardening Guide.

Hardening guide: Link to *AXIS OS Hardening guide* where you can learn more about cybersecurity on Axis devices and best practices.

Audio system info

This information is only shown for devices that belong to an AXIS Audio Manager Edge site.

AXIS Audio Manager Edge: Launch AXIS Audio Manager Edge.

Locate device

Shows the locate device information, including serial number and IP address.

Locate device: Plays a sound that helps you identify the speaker. For some products, the device will flash a LED.

Speaker test

Shows whether the speaker has been calibrated or not.

Speaker test:  : Calibrate the speaker. Takes you to the **Speaker test** page where you can do the calibration and run the speaker test.

Power status

Shows power status information, including current power, average power, and max power.

Power settings: View and update the power settings for the device. Takes you to the Power settings page where you can change the power settings.

Ongoing recordings

Shows ongoing recordings and their designated storage space.

Recordings: View ongoing and filtered recordings and their source. For more information, see



Shows the storage space where the recording is saved.

Connected clients

Shows the number of connections and connected clients.

View details: View and update the list of connected clients. The list shows IP address, protocol, port, state, and PID/process of each connection.

Display

Clock

Preview

The preview shows what the display will look like with the current settings.

Appearance

Use 24 hour clock: Turn on to show the clock with 24-hour format, e.g. 14:30. Turn off to show the clock with 12-hour format, e.g. 2:30 pm.

Show seconds: Turn on to show both hours, minutes and seconds. Turn off to show only hours and minutes.

Show date: Turn on to show the date and off to hide the date.

Font color: Set the color of the text that is shown on the display.

Background color: Set the color of the display background.

Availability

Off: Select to hide the clock. Messages triggered by rules will still appear on the display.

Always on: Select to keep the display on all the time.

Turn on display when presence is detected: Select to activate the display when presence is detected. If no presence is detected for the chosen standby time, the display will be turned off.

Turn on display according to schedule: Select to activate the display according to a schedule.

Invert schedule: Select to invert the schedule. For instance, if you schedule an event to play between 8.00 – 12.00 and invert the schedule, the event will play between 12.00 – 8.00 instead.

General

Brightness

Adaptive: Turn it on to use a range and turn it off to set a specific level.

- **Range:** The level will be set automatically by the speaker, based on ambient light. This usually saves energy.
- **Level:** Set a specific level.

Localization

Display language: Select a language for the date on the display.

Audio

AXIS Audio Manager Edge

AXIS Audio Manager Edge: Launch the application.

Audio site security

CA certificate: Select the certificate to use when you add devices to the audio site. You have to enable TLS authentication in AXIS Audio Manager Edge.

Save: Activate and save your selection.

Device settings

Input: Turn on or off audio input. Shows the type of input.

Gain: Use the slider to change the gain. Click the microphone icon to mute or unmute.

Input type  : Select the type of input.

Power type  : Select the type of power.

Output: Shows the type of output.

Gain: Use the slider to change the gain. Click the speaker icon to mute or unmute.

Stream

Encoding: Select the encoding to use for the input source streaming. You can only choose encoding if audio input is turned on. If audio input is turned off, click **Enable audio input** to turn it on.

Audio clips

 **Add clip:** Add a new audio clip. You can use .au, .mp3, .opus, .vorbis, .wav files.

 Play the audio clip.

 Stop playing the audio clip.

 The context menu contains:

- **Rename:** Change the name of the audio clip.
- **Create link:** Create a URL that, when used, plays the audio clip on the device. Specify the volume and number of times to play the clip.
- **Download:** Download the audio clip to your computer.
- **Delete:** Delete the audio clip from the device.

Listen and record

 Click to listen.

 Start a continuous recording of the live audio stream. Click again to stop the recording. If a recording is ongoing, it will resume automatically after a reboot.

Note

You can only listen and record if input is turned on for the device. Go to **Audio > Device settings** to make sure you turn on input.

 Shows the configured storage for the device. To configure the storage, you need to be logged in as an administrator.

Audio enhancement

Input

Ten Band Graphic Audio Equalizer: Turn on to adjust the level of different frequency bands within an audio signal. This feature is for advanced users with audio configuration experience.

Talkback range  : Choose the operational range to gather audio content. An increase to the operational range cause a reduction of simultaneous two-way communication capabilities.

Voice enhancement  : Turn on to enhance the voice content in relation to other sounds.

Speaker test

You can use the speaker test to verify remotely that the speaker works as intended.

Calibrate: You need to calibrate the speaker before its first test. During calibration, the speaker plays a series of test tones that are registered by the built-in microphone. When you calibrate the speaker, it must be installed in its final position. If you move the speaker later, or if its surroundings change, for example, if a wall is built or removed, you need to recalibrate the speaker.

Run the test: Play the same series of test tones that were played during calibration, and compare them with the calibration's registered values.

Peripherals

When you have connected devices, such as AXIS C8310 Volume Controller, then they will show up here.

Light

Overview

Light status

Shows the different light activities that run on the device. You can have up to 10 activities in the light status list at the same time. When two or more activities run at the same time, the activity with the highest priority shows the light status. That row will be highlighted in green in the status list.

Profiles

A profile is a collection of set configurations. You can have up to 30 profiles with different priorities and patterns. The profiles are listed to give an overview of name, priority and light settings.



Create: Click to create a profile.

Note

You can't have two profiles with the same name.

- **Name:** Enter a name of the profile.
- **Description:** Enter a description of the profile.
- **Light:** Select from the drop-down menu what kind of **LED** and **Pattern** you want.
- **Priority:** Set the priority of an activity to a number between 1 and 10. Activities with priority numbers higher than 10 can't be removed from the status list. There are three activities with higher priority than 10; **Maintenance** (11), **Identify** (12), and **Health check** (13).



Import: Add one or more profiles with predefined configuration.

- **Add**  : Add new profiles.
- **Delete and add**  : The old profiles are deleted, and you can upload new profiles.
- **Overwrite:** Updated profiles overwrite the existing profiles.

To copy a profile and save it to other devices, select one or more profiles and click **Export**. A .json file is exported.



Start a profile. The profile and its activities appear in the status list.



Choose to **Edit**, **Copy**, **Export**, or **Delete** the profile.

Recordings



Click to filter the recordings.

From: Show recordings done after a certain point in time.

To: Show recordings up until a certain point in time.

Source  : Show recordings based on source. The source refers to the sensor.

Event: Show recordings based on events.

Storage: Show recordings based on storage type.

Ongoing recordings: Show all ongoing recordings on the device.

- Start a recording on the device.



Choose which storage device to save to.

- Stop a recording on the device.

Triggered recordings will end when manually stopped or when the device is shut down.

Continuous recordings will continue until manually stopped. Even if the device is shut down, the recording will continue when the device starts up again.



Play the recording.



Stop playing the recording.



Show or hide information and options about the recording.

Set export range: If you only want to export part of the recording, enter a time span. Note that if you work in a different time zone than the location of the device, the time span is based on the device's time zone.

Encrypt: Select to set a password for exported recordings. It will not be possible to open the exported file without the password.



Click to delete a recording.

Export: Export the whole or a part of the recording.

Apps



Add app: Install a new app.

Find more apps: Find more apps to install. You will be taken to an overview page of Axis apps.



Allow unsigned apps : Turn on to allow installation of unsigned apps.



View the security updates in AXIS OS and ACAP apps.

Note

The device's performance might be affected if you run several apps at the same time.

Use the switch next to the app name to start or stop the app.

Open: Access the app's settings. The available settings depend on the application. Some applications don't have any settings.



The context menu can contain one or more of the following options:

- **Open-source license:** View information about open-source licenses used in the app.
- **App log:** View a log of the app events. The log is helpful when you contact support.
- **Activate license with a key:** If the app requires a license, you need to activate it. Use this option if your device doesn't have internet access.
If you don't have a license key, go to axis.com/products/analytics. You need a license code and the Axis product serial number to generate a license key.
- **Activate license automatically:** If the app requires a license, you need to activate it. Use this option if your device has internet access. You need a license code to activate the license.
- **Deactivate the license:** Deactivate the license to replace it with another license, for example, when you change from a trial license to a full license. If you deactivate the license, you also remove it from the device.
- **Settings:** Configure the parameters.
- **Delete:** Delete the app permanently from the device. If you don't deactivate the license first, it remains active.

System

Time and location

Date and time

The time format depends on the web browser's language settings.

Note

We recommend you synchronize the device's date and time with an NTP server.

Synchronization: Select an option for the device's date and time synchronization.

- **Automatic date and time (manual NTS KE servers):** Synchronize with the secure NTP key establishment servers connected to the DHCP server.
 - **Manual NTS KE servers:** Enter the IP address of one or two NTP servers. When you use two NTP servers, the device synchronizes and adapts its time based on input from both.
 - **Max NTP poll time:** Select the maximum amount of time the device should wait before it polls the NTP server to get an updated time.
 - **Min NTP poll time:** Select the minimum amount of time the device should wait before it polls the NTP server to get an updated time.
- **Automatic date and time (NTP servers using DHCP):** Synchronize with the NTP servers connected to the DHCP server.
 - **Fallback NTP servers:** Enter the IP address of one or two fallback servers.
 - **Max NTP poll time:** Select the maximum amount of time the device should wait before it polls the NTP server to get an updated time.
 - **Min NTP poll time:** Select the minimum amount of time the device should wait before it polls the NTP server to get an updated time.
- **Automatic date and time (manual NTP servers):** Synchronize with NTP servers of your choice.
 - **Manual NTP servers:** Enter the IP address of one or two NTP servers. When you use two NTP servers, the device synchronizes and adapts its time based on input from both.
 - **Max NTP poll time:** Select the maximum amount of time the device should wait before it polls the NTP server to get an updated time.
 - **Min NTP poll time:** Select the minimum amount of time the device should wait before it polls the NTP server to get an updated time.
- **Custom date and time:** Manually set the date and time. Click **Get from system** to fetch the date and time settings once from your computer or mobile device.

Time zone: Select which time zone to use. Time will automatically adjust to daylight saving time and standard time.

- **DHCP:** Adopts the time zone of the DHCP server. The device must be connected to a DHCP server before you can select this option.
- **Manual:** Select a time zone from the drop-down list.

Note

The system uses the date and time settings in all recordings, logs, and system settings.

Device location

Enter where the device is located. Your video management system can use this information to place the device on a map.

- **Format:** Select the format to use when you enter your device's latitude and longitude.
- **Latitude:** Positive values are north of the equator.
- **Longitude:** Positive values are east of the prime meridian.
- **Heading:** Enter the compass direction that the device is facing. 0 is due north.
- **Label:** Enter a descriptive name for your device.
- **Save:** Click to save your device location.

Network

IPv4

Assign IPv4 automatically: Select to let the network router assign an IP address to the device automatically. We recommend automatic IP (DHCP) for most networks.

IP address: Enter a unique IP address for the device. Static IP addresses can be assigned at random within isolated networks, provided that each address is unique. To avoid conflicts, we recommend you contact your network administrator before you assign a static IP address.

Subnet mask: Enter the subnet mask to define what addresses are inside the local area network. Any address outside the local area network goes through the router.

Router: Enter the IP address of the default router (gateway) used to connect devices that are attached to different networks and network segments.

Fallback to static IP address if DHCP isn't available: Select if you want to add a static IP address to use as fallback if DHCP is unavailable and can't assign an IP address automatically.

Note

If DHCP isn't available and the device uses a static address fallback, the static address is configured with a limited scope.

IPv6

Assign IPv6 automatically: Select to turn on IPv6 and to let the network router assign an IP address to the device automatically.

Hostname

Assign hostname automatically: Select to let the network router assign a hostname to the device automatically.

Hostname: Enter the hostname manually to use as an alternative way of accessing the device. The server report and system log use the hostname. Allowed characters are A–Z, a–z, 0–9 and –.

Enable dynamic DNS updates: Allow your device to automatically update its domain name server records whenever its IP address changes.

Register DNS name: Enter a unique domain name that points to your device's IP address. Allowed characters are A–Z, a–z, 0–9 and –.

TTL: Time to Live (TTL) sets how long a DNS record stays valid before it needs to be updated.

DNS servers

Assign DNS automatically: Select to let the DHCP server assign search domains and DNS server addresses to the device automatically. We recommend automatic DNS (DHCP) for most networks.

Search domains: When you use a hostname that is not fully qualified, click **Add search domain** and enter a domain in which to search for the hostname the device uses.

DNS servers: Click **Add DNS server** and enter the IP address of the DNS server. This provides the translation of hostnames to IP addresses on your network.

HTTP and HTTPS

HTTPS is a protocol that provides encryption for page requests from users and for the pages returned by the web server. The encrypted exchange of information is governed by the use of an HTTPS certificate, which guarantees the authenticity of the server.

To use HTTPS on the device, you must install an HTTPS certificate. Go to **System > Security** to create and install certificates.

Allow access through: Select if a user is allowed to connect to the device through the HTTP, HTTPS, or both HTTP and HTTPS protocols.

Note

If you view encrypted web pages through HTTPS, you might experience a drop in performance, especially when you request a page for the first time.

HTTP port: Enter the HTTP port to use. The device allows port 80 or any port in the range 1024–65535. If you are logged in as an administrator, you can also enter any port in the range 1–1023. If you use a port in this range, you get a warning.

HTTPS port: Enter the HTTPS port to use. The device allows port 443 or any port in the range 1024–65535. If you are logged in as an administrator, you can also enter any port in the range 1–1023. If you use a port in this range, you get a warning.

Certificate: Select a certificate to enable HTTPS for the device.

Network discovery protocols

Bonjour®: Turn on to allow automatic discovery on the network.

Bonjour name: Enter a friendly name to be visible on the network. The default name is the device name and MAC address.

UPnP®: Turn on to allow automatic discovery on the network.

UPnP name: Enter a friendly name to be visible on the network. The default name is the device name and MAC address.

WS-Discovery: Turn on to allow automatic discovery on the network.

LLDP and CDP: Turn on to allow automatic discovery on the network. Turning LLDP and CDP off can impact the PoE power negotiation. To resolve any issues with the PoE power negotiation, configure the PoE switch for hardware PoE power negotiation only.

Global proxies

Http proxy: Specify a global proxy host or IP address according to the allowed format.

Https proxy: Specify a global proxy host or IP address according to the allowed format.

Allowed formats for http and https proxies:

- `http(s)://host:port`
- `http(s)://user@host:port`
- `http(s)://user:pass@host:port`

Note

Restart the device to apply the global proxy settings.

No proxy: Use **No proxy** to bypass global proxies. Enter one of the options in the list, or enter several separated by a comma:

- Leave empty
- Specify an IP address
- Specify an IP address in CIDR format
- Specify a domain name, for example: `www.<domain name>.com`
- Specify all subdomains in a specific domain, for example `.<domain name>.com`

One-click cloud connection

One-click cloud connection (O3C) together with an O3C service provides easy and secure internet access to live and recorded video from any location. For more information, see axis.com/end-to-end-solutions/hosted-services.

Allow O3C:

- **One-click:** This is the default option. To connect to O3C, press the control button on the device. Depending on the device model, either press and release or press and hold, until the status LED flashes. Register the device with the O3C service within 24 hours to enable **Always** and stay connected. If you don't register, the device will disconnect from O3C.
- **Always:** The device continuously attempts to connect to an O3C service over the internet. Once you register the device, it stays connected. Use this option if the control button is out of reach.
- **No:** Disconnects the O3C service.

Proxy settings: If needed, enter the proxy settings to connect to the proxy server.

Host: Enter the proxy server's address.

Port: Enter the port number used for access.

Login and Password: If needed, enter username and password for the proxy server.

Authentication method:

- **Basic:** This method is the most compatible authentication scheme for HTTP. It's less secure than the **Digest** method because it sends the username and password unencrypted to the server.
- **Digest:** This method is more secure because it always transfers the password encrypted across the network.
- **Auto:** This option lets the device select the authentication method depending on the supported methods. It prioritizes the **Digest** method over the **Basic** method.

Owner authentication key (OAK): Click **Get key** to fetch the owner authentication key. This is only possible if the device is connected to the internet without a firewall or proxy.

SNMP

The Simple Network Management Protocol (SNMP) allows remote management of network devices.

SNMP: Select the version of SNMP to use.

- **v1 and v2c:**
 - **Read community:** Enter the community name that has read-only access to all supported SNMP objects. The default value is **public**.
 - **Write community:** Enter the community name that has read or write access to all supported SNMP objects (except read-only objects). The default value is **write**.
 - **Activate traps:** Turn on to activate trap reporting. The device uses traps to send messages for important events or status changes to a management system. In the web interface, you can set up traps for SNMP v1 and v2c. Traps are automatically turned off if you change to SNMP v3 or turn off SNMP. If you use SNMP v3, you can set up traps through the SNMP v3 management application.
 - **Trap address:** Enter the IP address or host name of the management server.
 - **Trap community:** Enter the community to use when the device sends a trap message to the management system.
 - **Traps:**
 - **Cold start:** Sends a trap message when the device starts.
 - **Link up:** Sends a trap message when a link changes from down to up.
 - **Link down:** Sends a trap message when a link changes from up to down.
 - **Authentication failed:** Sends a trap message when an authentication attempt fails.

Note

All Axis Video MIB traps are enabled when you turn on SNMP v1 and v2c traps. For more information, see *AXIS OS Portal > SNMP*.

- **v3:** SNMP v3 is a more secure version, which provides encryption and secure passwords. To use SNMP v3, we recommend you to activate HTTPS, as the password is then sent through HTTPS. This also prevents unauthorized parties' access to unencrypted SNMP v1 and v2c traps. If you use SNMP v3, you can set up traps through the SNMP v3 management application.
 - **Password for the account "initial":** Enter the SNMP password for the account named "initial". Although the password can be sent without activating HTTPS, we don't recommend it. The SNMP v3 password can only be set once, and preferably only when HTTPS is enabled. Once the password is set, the password field is no longer displayed. To set the password again, you must reset the device to factory default settings.

Security

Certificates

Certificates are used to authenticate devices on a network. The device supports two types of certificates:

- **Client/server certificates**
A client/server certificate validates the device's identity, and can be self-signed or issued by a Certificate Authority (CA). A self-signed certificate offers limited protection and can be used before a CA-issued certificate has been obtained.
- **CA certificates**
You can use a CA certificate to authenticate a peer certificate, for example to validate the identity of an authentication server when the device connects to a network protected by IEEE 802.1X. The device has several pre-installed CA certificates.

These formats are supported:

- Certificate formats: .PEM, .CER, and .PFX
- Private key formats: PKCS#1 and PKCS#12

Important

If you reset the device to factory default, all certificates are deleted. Any pre-installed CA certificates are reinstalled.



Add certificate : Click to add a certificate. A step-by-step guide opens up.

- **More**  : Show more fields to fill in or select.
- **Secure keystore**: Select to use **Trusted Execution Environment (SoC TEE)** or **Secure element** to securely store the private key. For more information on which secure keystore to select, go to [help.axis.com/axis-os#cryptographic-support](https://axis.com/axis-os#cryptographic-support).
- **Key type**: Select the default or a different encryption algorithm from the drop-down list to protect the certificate.



The context menu contains:

- **Certificate information**: View an installed certificate's properties.
- **Delete certificate**: Delete the certificate.
- **Create certificate signing request**: Create a certificate signing request to send to a registration authority to apply for a digital identity certificate.

Secure keystore  :

- **Trusted Execution Environment (SoC TEE)**: Select to use SoC TEE for secure keystore.
- **Secure element (CC EAL6+, FIPS 140-3 Level 3)**: Select to use secure element for secure keystore.

Network access control and encryption

IEEE 802.1x

IEEE 802.1x is an IEEE standard for port-based network admission control providing secure authentication of wired and wireless network devices. IEEE 802.1x is based on EAP (Extensible Authentication Protocol).

To access a network protected by IEEE 802.1x, network devices must authenticate themselves. The authentication is performed by an authentication server, typically a RADIUS server (for example, FreeRADIUS and Microsoft Internet Authentication Server).

IEEE 802.1AE MACsec

IEEE 802.1AE MACsec is an IEEE standard for media access control (MAC) security that defines connectionless data confidentiality and integrity for media access independent protocols.

Certificates

When configured without a CA certificate, server certificate validation is disabled and the device tries to authenticate itself regardless of what network it is connected to.

When using a certificate, in Axis' implementation, the device and the authentication server authenticate themselves with digital certificates using EAP-TLS (Extensible Authentication Protocol - Transport Layer Security).

To allow the device to access a network protected through certificates, you must install a signed client certificate on the device.

Authentication method: Select an EAP type used for authentication.

Client certificate: Select a client certificate to use IEEE 802.1x. The authentication server uses the certificate to validate the client's identity.

CA certificates: Select CA certificates to validate the authentication server's identity. When no certificate is selected, the device tries to authenticate itself regardless of what network it is connected to.

EAP identity: Enter the user identity associated with the client certificate.

EAPOL version: Select the EAPOL version that is used in the network switch.

Use IEEE 802.1x: Select to use the IEEE 802.1x protocol.

These settings are only available if you use **IEEE 802.1x PEAP-MSCHAPv2** as the authentication method:

- **Password:** Enter the password for your user identity.
- **Peap version:** Select the Peap version that is used in the network switch.
- **Label:** Select 1 to use client EAP encryption; select 2 to use client PEAP encryption. Select the Label that the network switch uses when using Peap version 1.

These settings are only available if you use **IEEE 802.1ae MACsec (Static CAK/Pre-Shared Key)** as the authentication method:

- **Key agreement connectivity association key name:** Enter the connectivity association name (CKN). It must be 2 to 64 (divisible by 2) hexadecimal characters. The CKN must be manually configured in the connectivity association and must match on both ends of the link to initially enable MACsec.
- **Key agreement connectivity association key:** Enter the connectivity association key (CAK). It should be either 32 or 64 hexadecimal characters long. The CAK must be manually configured in the connectivity association and must match on both ends of the link to initially enable MACsec.

Prevent brute-force attacks

Blocking: Turn on to block brute-force attacks. A brute-force attack uses trial-and-error to guess login info or encryption keys.

Blocking period: Enter the number of seconds to block a brute-force attack.

Blocking conditions: Enter the number of authentication failures allowed per second before the block starts. You can set the number of failures allowed both on page level and device level.

Firewall

Firewall: Turn on to activate the firewall.

Default Policy: Select how you want the firewall to handle connection requests not covered by rules.

- **ACCEPT:** Allows all connections to the device. This option is set by default.
- **DROP:** Blocks all connections to the device.

To make exceptions to the default policy, you can create rules that allows or blocks connections to the device from specific addresses, protocols, and ports.

+ New rule: Click to create a rule.

Rule type:

- **FILTER:** Select to either allow or block connections from devices that match the criteria defined in the rule.
 - **Policy:** Select **Accept** or **Drop** for the firewall rule.
 - **IP range:** Select to specify a range of addresses to allow or block. Use IPv4/IPv6 in **Start** and **End**.
 - **IP address:** Enter an address that you want to allow or block. Use IPv4/IPv6 or CIDR format.
 - **Protocol:** Select a network protocol (TCP, UDP, or Both) to allow or block. If you select a protocol, you must also specify a port.
 - **MAC:** Enter the MAC address of a device that you want to allow or block.
 - **Port range:** Select to specify the range of ports to allow or block. Add them in **Start** and **End**.
 - **Port:** Enter a port number that you want to allow or block. Port numbers must be between 1 and 65535.
 - **Traffic type:** Select a traffic type that you want to allow or block.
 - **UNICAST:** Traffic from a single sender to a single recipient.
 - **BROADCAST:** Traffic from a single sender to all devices on the network.
 - **MULTICAST:** Traffic from one or more senders to one or more recipient.
- **LIMIT:** Select to accept connections from devices that match the criteria defined in the rule but apply limits to reduce excessive traffic.
 - **IP range:** Select to specify a range of addresses to allow or block. Use IPv4/IPv6 in **Start** and **End**.
 - **IP address:** Enter an address that you want to allow or block. Use IPv4/IPv6 or CIDR format.
 - **Protocol:** Select a network protocol (TCP, UDP, or Both) to allow or block. If you select a protocol, you must also specify a port.
 - **MAC:** Enter the MAC address of a device that you want to allow or block.
 - **Port range:** Select to specify the range of ports to allow or block. Add them in **Start** and **End**.
 - **Port:** Enter a port number that you want to allow or block. Port numbers must be between 1 and 65535.
 - **Unit:** Select the type of connections to allow or block.
 - **Period:** Select the time period related to **Amount**.
 - **Amount:** Set the maximum number of times a device is allowed to connect within the set **Period**. The maximum amount is 65535.
 - **Burst:** Enter the number of connections allowed to exceed the set **Amount** once during the set **Period**. Once the number has been reached, only the set amount during the set period is allowed.
 - **Traffic type:** Select a traffic type that you want to allow or block.
 - **UNICAST:** Traffic from a single sender to a single recipient.
 - **BROADCAST:** Traffic from a single sender to all devices on the network.

- **MULTICAST:** Traffic from one or more senders to one or more recipient.

Test rules: Click to test the rules that you have defined.

- **Test time in seconds:** Set a time limit for testing the rules.
- **Roll back:** Click to roll back the firewall to its previous state, before you have tested the rules.
- **Apply rules:** Click to activate the rules without testing. We don't recommend that you do this.

Custom signed AXIS OS certificate

To install test software or other custom software from Axis on the device, you need a custom signed AXIS OS certificate. The certificate verifies that the software is approved by both the device owner and Axis. The software can only run on a specific device which is identified by its unique serial number and chip ID. Only Axis can create custom signed AXIS OS certificates, since Axis holds the key to sign them.

Install: Click to install the certificate. You need to install the certificate before you install the software.



The context menu contains:

- **Delete certificate:** Delete the certificate.

Accounts

Accounts



Add account: Click to add a new account. You can add up to 100 accounts.

Account: Enter a unique account name.

New password: Enter a password for the account. Passwords must be 1 to 64 characters long. Only ASCII printable characters (code 32 to 126) are allowed in the password, for example, letters, numbers, punctuation, and some symbols.

Repeat password: Enter the same password again.

Privileges:

- **Administrator:** Has full access to all settings. Administrators can also add, update, and remove other accounts.
- **Operator:** Has access to all settings except:
 - All **System** settings.
- **Viewer:** Doesn't have access to change any settings.



The context menu contains:

Update account: Edit the account properties.

Delete account: Delete the account. You can't delete the root account.

Anonymous access

Allow anonymous viewing: Turn on to allow anyone access the device as a viewer without logging in with an account.

Allow anonymous PTZ operating  : Turn on to allow anonymous users to pan, tilt, and zoom the image.

SSH accounts



Add SSH account: Click to add a new SSH account.

- **Enable SSH:** Turn on to use SSH service.

Account: Enter a unique account name.

New password: Enter a password for the account. Passwords must be 1 to 64 characters long. Only ASCII printable characters (code 32 to 126) are allowed in the password, for example, letters, numbers, punctuation, and some symbols.

Repeat password: Enter the same password again.

Comment: Enter a comment (optional).



The context menu contains:

Update SSH account: Edit the account properties.

Delete SSH account: Delete the account. You can't delete the root account.

Virtual host



Add virtual host: Click to add a new virtual host.

Enabled: Select to use this virtual host.

Server name: Enter the name of the server. Only use numbers 0-9, letters A-Z, and hyphen (-).

Port: Enter the port the server is connected to.

Type: Select the type of authentication to use. Select between **Basic**, **Digest**, and **Open ID**.



The context menu contains:

- **Update:** Update the virtual host.
- **Delete:** Delete the virtual host.

Disabled: The server is disabled.

Client Credentials Grant Configuration

Admin claim: Enter a value for the admin role.

Verification URI: Enter the web link for the API endpoint authentication.

Operator claim: Enter a value for the operator role.

Require claim: Enter the data that should be in the token.

Viewer claim: Enter the value for the viewer role.

Save: Click to save the values.

OpenID Configuration

Important

If you can't use OpenID to sign in, use the Digest or Basic credentials you used when you configured OpenID to sign in.

Client ID: Enter the OpenID username.

Outgoing Proxy: Enter the proxy address for the OpenID connection to use a proxy server.

Admin claim: Enter a value for the admin role.

Provider URL: Enter the web link for the API endpoint authentication. Format should be https://[insert URL]/well-known/openid-configuration

Operator claim: Enter a value for the operator role.

Require claim: Enter the data that should be in the token.

Viewer claim: Enter the value for the viewer role.

Remote user: Enter a value to identify remote users. This assists to display the current user in the device's web interface.

Scopes: Optional scopes that could be part of the token.

Client secret: Enter the OpenID password

Save: Click to save the OpenID values.

Enable OpenID: Turn on to close current connection and allow device authentication from the provider URL.

Events

Rules

A rule defines the conditions that triggers the product to perform an action. The list shows all the currently configured rules in the product.

Note

You can create up to 256 action rules.



Add a rule: Create a rule.

Name: Enter a name for the rule.

Wait between actions: Enter the minimum time (hh:mm:ss) that must pass between rule activations. It is useful if the rule is activated by, for example, day-night mode conditions, to avoid that small light changes during sunrise and sunset activate the rule repeatedly.

Condition: Select a condition from the list. A condition must be met for the device to perform an action. If multiple conditions are defined, all of them must be met to trigger the action. For information about specific conditions, see *Get started with rules for events*.

Use this condition as a trigger: Select to make this first condition function only as a starting trigger. It means that once the rule is activated, it remains active for as long as all the other conditions are met, no matter the state of the first condition. If you don't select this option, the rule will simply be active whenever all the conditions are met.

Invert this condition: Select if you want the condition to be the opposite of your selection.



Add a condition: Click to add an additional condition.

Action: Select an action from the list and enter its required information. For information about specific actions, see *Get started with rules for events*.

Your product may have some of the following pre-configured rules:

Front-facing LED Activation: LiveStream: When the microphone is turned on and a live stream is received, then the front-facing LED on the audio device will turn green.

Front-facing LED Activation: Recording : When the microphone is turned on and a recording is ongoing, then the front-facing LED on the audio device will turn green.

Front-facing LED Activation: SIP : When the microphone is turned on and a SIP call is active, then the front-facing LED on the audio device will turn green. You must enable SIP on the audio device before it can trigger this event.

Pre-announcement tone: Play tone on incoming call: When a SIP call is made to the audio device, then the device plays a pre-defined audio clip. You must enable SIP for the audio device. For the SIP caller to hear a ring tone while the audio device plays the audio clip, you must configure the SIP account for the device to not answer the call automatically.

Pre-announcement tone: Answer call after incoming call-tone: When the audio clip has ended, the incoming SIP-call is answered. You must enable SIP for the audio device.

Loud ringer : When a SIP call is made to the audio device, a pre-defined audio clip is played as long as the rule is active. You must enable SIP for the audio device.

Recipients

You can set up your device to notify recipients about events or send files.

Note

If you set up your device to use FTP or SFTP, don't change or remove the unique sequence number that's added to the file names. If you do that, only one image per event can be sent.

The list shows all the recipients currently configured in the product, along with information about their configuration.

Note

You can create up to 20 recipients.



Add a recipient: Click to add a recipient.

Name: Enter a name for the recipient.

Type: Select from the list:

- **FTP** 
 - **Host:** Enter the server's IP address or hostname. If you enter a hostname, make sure that a DNS server is specified under **System > Network > IPv4 and IPv6**.
 - **Port:** Enter the port number used by the FTP server. The default is 21.
 - **Folder:** Enter the path to the directory where you want to store files. If this directory doesn't already exist on the FTP server, you will get an error message when uploading files.
 - **Username:** Enter the username for the login.
 - **Password:** Enter the password for the login.
 - **Use temporary file name:** Select to upload files with temporary, automatically generated filenames. The files get renamed to the desired names when the upload completes. If the upload is aborted/interrupted, you don't get any corrupt files. However, you probably still get the temporary files. This way you know that all files that have the desired name are correct.
 - **Use passive FTP:** Under normal circumstances, the product simply requests the target FTP server to open the data connection. The device actively initiates both the FTP control and data connections to the target server. This is normally needed if there is a firewall between the device and the target FTP server.
- **HTTP**
 - **URL:** Enter the network address to the HTTP server and the script that will handle the request. For example, `http://192.168.254.10/cgi-bin/notify.cgi`.
 - **Username:** Enter the username for the login.
 - **Password:** Enter the password for the login.
 - **Proxy:** Turn on and enter the required information if a proxy server must be passed to connect to the HTTP server.
- **HTTPS**
 - **URL:** Enter the network address to the HTTPS server and the script that will handle the request. For example, `https://192.168.254.10/cgi-bin/notify.cgi`.
 - **Validate server certificate:** Select to validate the certificate that was created by HTTPS server.
 - **Username:** Enter the username for the login.
 - **Password:** Enter the password for the login.
 - **Proxy:** Turn on and enter the required information if a proxy server must be passed to connect to the HTTPS server.
- **Network storage** 

You can add network storage such as NAS (network-attached storage) and use it as a recipient to store files. The files are stored in the Matroska (MKV) file format.

 - **Host:** Enter the IP address or hostname for the network storage.
 - **Share:** Enter the name of the share on the host.
 - **Folder:** Enter the path to the directory where you want to store files.
 - **Username:** Enter the username for the login.
 - **Password:** Enter the password for the login.

- **SFTP** 
 - **Host:** Enter the server's IP address or hostname. If you enter a hostname, make sure that a DNS server is specified under **System > Network > IPv4 and IPv6**.
 - **Port:** Enter the port number used by the SFTP server. The default is 22.
 - **Folder:** Enter the path to the directory where you want to store files. If this directory doesn't already exist on the SFTP server, you will get an error message when uploading files.
 - **Username:** Enter the username for the login.
 - **Password:** Enter the password for the login.
 - **SSH host public key type (MD5):** Enter the fingerprint of the remote host's public key (a 32-digit hexadecimal string). The SFTP client supports SFTP servers using SSH-2 with RSA, DSA, ECDSA, and ED25519 host key types. RSA is the preferred method during negotiation, followed by ECDSA, ED25519, and DSA. Make sure to enter the right MD5 host key that is used by your SFTP server. While the Axis device supports both MD5 and SHA-256 hash keys, we recommend using SHA-256 due to stronger security over MD5. For more information on how to configure an SFTP server with an Axis device, go to the *AXIS OS Portal*.
 - **SSH host public key type (SHA256):** Enter the fingerprint of the remote host's public key (a 43-digit Base64 encoded string). The SFTP client supports SFTP servers using SSH-2 with RSA, DSA, ECDSA, and ED25519 host key types. RSA is the preferred method during negotiation, followed by ECDSA, ED25519, and DSA. Make sure to enter the right MD5 host key that is used by your SFTP server. While the Axis device supports both MD5 and SHA-256 hash keys, we recommend using SHA-256 due to stronger security over MD5. For more information on how to configure an SFTP server with an Axis device, go to the *AXIS OS Portal*.
 - **Use temporary file name:** Select to upload files with temporary, automatically generated filenames. The files get renamed to the desired names when the upload completes. If the upload is aborted or interrupted, you don't get any corrupt files. However, you probably still get the temporary files. This way, you know that all files that have the desired name are correct.
- **SIP or VMS**  :
 - SIP:** Select to make a SIP call.
 - VMS:** Select to make a VMS call.
 - **From SIP account:** Select from the list.
 - **To SIP address:** Enter the SIP address.
 - **Test:** Click to test that your call settings works.
- **Email**
 - **Send email to:** Enter the email address to send emails to. To enter multiple addresses, use commas to separate them.
 - **Send email from:** Enter the email address of the sending server.
 - **Username:** Enter the username for the mail server. Leave this field empty if the mail server does not require authentication.
 - **Password:** Enter the password for the mail server. Leave this field empty if the mail server does not require authentication.
 - **Email server (SMTP):** Enter the name of the SMTP server, for example, smtp.gmail.com, smtp.mail.yahoo.com.
 - **Port:** Enter the port number for the SMTP server, using values in the range 0-65535. The default value is 587.
 - **Encryption:** To use encryption, select either SSL or TLS.
 - **Validate server certificate:** If you use encryption, select to validate the identity of the device. The certificate can be self-signed or issued by a Certificate Authority (CA).

- **POP authentication:** Turn on to enter the name of the POP server, for example, pop.gmail.com.

Note

Some email providers have security filters that prevent users from receiving or viewing large amount of attachments, from receiving scheduled emails and similar. Check the email provider's security policy to avoid your email account being locked or missing out on your expected emails.

- **TCP**
 - **Host:** Enter the server's IP address or hostname. If you enter a hostname, make sure that a DNS server is specified under **System > Network > IPv4 and IPv6**.
 - **Port:** Enter the port number used to access the server.

Test: Click to test the setup.



The context menu contains:

View recipient: Click to view all the recipient details.

Copy recipient: Click to copy a recipient. When you copy, you can make changes to the new recipient.

Delete recipient: Click to delete the recipient permanently.

Schedules

Schedules and pulses can be used as conditions in rules. The list shows all the schedules and pulses currently configured in the product, along with information about their configuration.



Add schedule: Click to create a schedule or pulse.

Manual triggers

You can use the manual trigger to manually trigger a rule. The manual trigger can, for example, be used to validate actions during product installation and configuration.

MQTT

MQTT (Message Queuing Telemetry Transport) is a standard messaging protocol for the Internet of Things (IoT). It was designed for simplified IoT integration and is used in a wide variety of industries to connect remote devices with a small code footprint and minimal network bandwidth. The MQTT client in Axis device software can simplify integration of data and events produced in the device to systems which are not video management software (VMS).

Set up the device as an MQTT client. MQTT communication is based on two entities, the clients and the broker. The clients can send and receive messages. The broker is responsible for routing messages between clients.

You can learn more about MQTT in *AXIS OS Knowledge base*.

ALPN

ALPN is a TLS/SSL extension that allows for the selection of an application protocol during the handshake phase of the connection between the client and server. This is used to enable MQTT traffic over the same port that is used for other protocols, such as HTTP. In some cases, there might not be a dedicated port open for MQTT communication. A solution in such cases is to use ALPN to negotiate the use of MQTT as the application protocol on a standard port, allowed by the firewalls.

MQTT client

Connect: Turn on or off the MQTT client.

Status: Shows the current status of the MQTT client.

Broker

Host: Enter the hostname or IP address of the MQTT server.

Protocol: Select which protocol to use.

Port: Enter the port number.

- 1883 is the default value for **MQTT over TCP**
- 8883 is the default value for **MQTT over SSL**
- 80 is the default value for **MQTT over WebSocket**
- 443 is the default value for **MQTT over WebSocket Secure**

ALPN protocol: Enter the ALPN protocol name provided by your MQTT broker provider. This is only applicable with MQTT over SSL and MQTT over WebSocket Secure.

Username: Enter the username that the client will use to access the server.

Password: Enter a password for the username.

Client ID: Enter a client ID. The client identifier is sent to the server when the client connects to it.

Clean session: Controls the behavior at connection and disconnection time. When selected, the state information is discarded at connect and disconnect.

HTTP proxy: A URL with a maximum length of 255 bytes. You can leave the field empty if you don't want to use an HTTP proxy.

HTTPS proxy: A URL with a maximum length of 255 bytes. You can leave the field empty if you don't want to use an HTTPS proxy.

Keep alive interval: Enables the client to detect when the server is no longer available without having to wait for the long TCP/IP timeout.

Timeout: The time interval in seconds to allow a connect to complete. Default value: 60

Device topic prefix: Used in the default values for the topic in the connect message and LWT message on the MQTT client tab, and in the publication conditions on the **MQTT publication** tab.

Reconnect automatically: Specifies whether the client should reconnect automatically after a disconnect.

Connect message

Specifies if a message should be sent out when a connection is established.

Send message: Turn on to send messages.

Use default: Turn off to enter your own default message.

Topic: Enter the topic for the default message.

Payload: Enter the content for the default message.

Retain: Select to keep the state of client on this Topic

QoS: Change the QoS layer for the packet flow.

Last Will and Testament message

The Last Will Testament (LWT) lets a client provide a testament along with its credentials when connecting to the broker. If the client disconnects ungracefully at some point later (maybe because his power source died), it

can let the broker deliver a message to other clients. This LWT message has the same form as an ordinary message and gets routed via the same mechanics.

Send message: Turn on to send messages.

Use default: Turn off to enter your own default message.

Topic: Enter the topic for the default message.

Payload: Enter the content for the default message.

Retain: Select to keep the state of client on this Topic

QoS: Change the QoS layer for the packet flow.

MQTT publication

Use default topic prefix: Select to use the default topic prefix, that is defined in the device topic prefix in the MQTT client tab.

Include topic name: Select to include the topic that describes the condition in the MQTT topic.

Include topic namespaces: Select to include ONVIF topic namespaces in the MQTT topic.

Include serial number: Select to include the device's serial number in the MQTT payload.



Add condition: Click to add a condition.

Retain: Defines which MQTT messages are sent as retained.

- **None:** Send all messages as non-retained.
- **Property:** Send only stateful messages as retained.
- **All:** Send both stateful and stateless messages as retained.

QoS: Select the desired level for the MQTT publication.

MQTT subscriptions



Add subscription: Click to add a new MQTT subscription.

Subscription filter: Enter the MQTT topic that you want to subscribe to.

Use device topic prefix: Add the subscription filter as prefix to the MQTT topic.

Subscription type:

- **Stateless:** Select to convert MQTT messages into a stateless message.
- **Stateful:** Select to convert MQTT messages into a condition. The payload is used as the state.

QoS: Select the desired level for the MQTT subscription.

SIP

Settings

Session Initiation Protocol (SIP) is used for interactive communication sessions between users. The sessions can include audio and video.

SIP setup assistant: Click to set up and configure SIP step by step.

Enable SIP: Check this option to make it possible to initiate and receive SIP calls.

Allow incoming calls: Check this option to allow incoming calls from other SIP devices.

Call handling

- **Calling timeout:** Set the maximum duration of an attempted call if no one answers.
- **Incoming call duration:** Set the maximum time an incoming call can last (max 10 min).
- **End calls after:** Set the maximum time that a call can last (max 60 minutes). Select **Infinite call duration** if you don't want to limit the length of a call.

Ports

A port number must be between 1024 and 65535.

- **SIP port:** The network port used for SIP communication. The signaling traffic through this port is non-encrypted. The default port number is 5060. Enter a different port number if required.
- **TLS port:** The network port used for encrypted SIP communication. The signaling traffic through this port is encrypted with Transport Layer Security (TLS). The default port number is 5061. Enter a different port number if required.
- **RTP start port:** The network port used for the first RTP media stream in a SIP call. The default start port number is 4000. Some firewalls block RTP traffic on certain port numbers.

NAT traversal

Use NAT (Network Address Translation) traversal when the device is located on an private network (LAN) and you want to make it available from outside of that network.

Note

For NAT traversal to work, the router must support it. The router must also support UPnP®.

Each NAT traversal protocol can be used separately or in different combinations depending on the network environment.

- **ICE:** The ICE (Interactive Connectivity Establishment) protocol increases the chances of finding the most efficient path to successful communication between peer devices. If you also enable STUN and TURN, you improve the ICE protocol's chances.
- **STUN:** STUN (Session Traversal Utilities for NAT) is a client-server network protocol that lets the device determine if it is located behind a NAT or firewall, and if so obtain the mapped public IP address and port number allocated for connections to remote hosts. Enter the STUN server address, for example, an IP address.
- **TURN:** TURN (Traversal Using Relays around NAT) is a protocol that lets a device behind a NAT router or firewall receive incoming data from other hosts over TCP or UDP. Enter the TURN server address and the login information.

Audio

- **Audio codec priority:** Select at least one audio codec with the desired audio quality for SIP calls. Drag-and-drop to change the priority.

Note

The selected codecs must match the call recipient codec, since the recipient codec is decisive when a call is made.

- **Audio direction:** Select allowed audio directions.

Additional

- **UDP-to-TCP switching:** Select to allow calls to switch transport protocols from UDP (User Datagram Protocol) to TCP (Transmission Control Protocol) temporarily. The reason for switching is to avoid fragmentation, and the switch can take place if a request is within 200 bytes of the maximum transmission unit (MTU) or larger than 1300 bytes.
- **Allow via rewrite:** Select to send the local IP address instead of the router's public IP address.
- **Allow contact rewrite:** Select to send the local IP address instead of the router's public IP address.

- **Register with server every:** Set how often you want the device to register with the SIP server for the existing SIP accounts.
- **DTMF payload type:** Changes the default payload type for DTMF.
- **Max retransmissions:** Set the maximum number of times the device tries to connect to the SIP server before it stops trying.
- **Seconds until failback:** Set the number of seconds until the device tries to reconnect to the primary SIP server after it has failed over to a secondary SIP server.

Accounts

All current SIP accounts are listed under **SIP accounts**. For registered accounts, the colored circle lets you know the status.

- The account is successfully registered with the SIP server.
- There is a problem with the account. Possible reasons can be authorization failure, that the account credentials are wrong, or that the SIP server can't find the account.

The **peer to peer (default)** account is an automatically created account. You can delete it if you create at least one other account and set that account as default. The default account is always used when a VAPIX® Application Programming Interface (API) call is made without specifying which SIP account to call from.



Add account: Click to create a new SIP account.

- **Active:** Select to be able to use the account.
- **Make default:** Select to make this the default account. There must be a default account, and there can only be one default account.
- **Answer automatically:** Select to automatically answer an incoming call.
- **Prioritize IPv6 over IPv4**  : Select to prioritize IPv6 addresses over IPv4 addresses. This is useful when you connect to peer-to-peer accounts or domain names that resolve in both IPv4 and IPv6 addresses. You can only prioritize IPv6 for domain names that are mapped to IPv6 addresses.
- **Name:** Enter a descriptive name. This can, for example, be a first and last name, a role, or a location. The name is not unique.
- **User ID:** Enter the unique extension or phone number assigned to the device.
- **Peer-to-peer:** Use for direct calls to another SIP device on the local network.
- **Registered:** Use for calls to SIP devices outside the local network, through a SIP server.
- **Domain:** If available, enter the public domain name. It will be shown as part of the SIP address when calling other accounts.
- **Password:** Enter the password associated with the SIP account for authenticating against the SIP server.
- **Authentication ID:** Enter the authentication ID used for authenticating against the SIP server. If it is the same as the user ID, you don't need to enter the authentication ID.
- **Caller ID:** The name which is presented to the recipient of calls from the device.
- **Registrar:** Enter the IP address for the registrar.
- **Transport mode:** Select the SIP transport mode for the account: UDP, TCP, or TLS.
- **TLS version (only with transport mode TLS):** Select the version of TLS to use. Versions **v1.2** and **v1.3** are the most secure. **Automatic** selects the most secure version that the system can handle.
- **Media encryption (only with transport mode TLS):** Select the type of encryption for media (audio and video) in SIP calls.
- **Certificate (only with transport mode TLS):** Select a certificate.
- **Verify server certificate (only with transport mode TLS):** Check to verify the server certificate.
- **Secondary SIP server:** Turn on if you want the device to try to register on a secondary SIP server if registration on the primary SIP server fails.
- **SIP secure:** Select to use Secure Session Initiation Protocol (SIPS). SIPS uses the TLS transport mode to encrypt traffic.
- **Proxies**
 -  **Proxy:** Click to add a proxy.
 - **Prioritize:** If you have added two or more proxies, click to prioritize them.

- **Server address:** Enter the IP address of the SIP proxy server.
- **Username:** If required, enter the username for the SIP proxy server.
- **Password:** If required, enter the password for the SIP proxy server.
- **Video** ⓘ
 - **View area:** Select the view area to use for video calls. If you select none, the native view is used.
 - **Resolution:** Select the resolution to use for video calls. The resolution affects the required bandwidth.
 - **Frame rate:** Select the number of frames per second for video calls. The frame rate affects the required bandwidth.
 - **H.264 profile:** Select the profile to use for video calls.

DTMF

 **Add sequence:** Click to create a new dual-tone multifrequency (DTMF) sequence. To create a rule that is activated by touch-tone, go to **Events > Rules**.

Sequence: Enter the characters to activate the rule. Allowed characters: 0–9, A–D, #, and *.

Description: Enter a description of the action to be triggered by the sequence.

Accounts: Select the accounts that will use the DTMF sequence. If you choose **peer-to-peer**, all peer-to-peer accounts will share the same DTMF sequence.

Protocols

Select the protocols to use for each account. All peer-to-peer accounts share the same protocol settings.

Use RTP (RFC2833): Turn on to allow dual-tone multifrequency (DTMF) signaling, other tone signals and telephony events in RTP packets.

Use SIP INFO (RFC2976): Turn to include the INFO method to the SIP protocol. The INFO method adds optional application layer information, generally related to the session.

Test call

SIP account: Select which account to make the test call from.

SIP address: Enter a SIP address and click  to make a test call and verify that the account works.

Access list

Use access list: Turn on to restrict who can make calls to the device.

Policy:

- **Allow:** Select to allow incoming calls only from the sources in the access list.
- **Block:** Select to block incoming calls from the sources in the access list.

 **Add source:** Click to create a new entry in the access list.

SIP source: Type the caller ID or SIP server address of the source.

Multicast controller

User multicast controller: Turn on to activate multicast controller.

Audio codec: Select an audio codec.



Source: Add a new multicast controller source.

- **Label:** Enter the name of a label that is not already used by a source.
- **Source:** Enter a source.
- **Port:** Enter a port.
- **Priority:** Select a priority.
- **Profile:** Select a profile.
- **SRTP key:** Enter an SRTP key.



The context menu contains:

Edit: Edit the multicast controller source.

Delete: Delete the multicast controller source.

Storage

Network storage

Ignore: Turn on to ignore network storage.

Add network storage: Click to add a network share where you can save recordings.

- **Address:** Enter the IP address or host name of the host server, typically a NAS (network-attached storage). We recommend you to configure the host to use a fixed IP address (not DHCP since a dynamic IP address can change) or that you use DNS. Windows SMB/CIFS names are not supported.
- **Network share:** Enter the name of the shared location on the host server. Several Axis devices can use the same network share since each device gets its own folder.
- **User:** If the server requires a login, enter the username. To log in to a specific domain server, type DOMAIN\username.
- **Password:** If the server requires a login, enter the password.
- **SMB version:** Select the SMB storage protocol version to connect to the NAS. If you select **Auto**, the device tries to negotiate one of the secure versions SMB: 3.02, 3.0, or 2.1. Select 1.0 or 2.0 to connect to older NAS that don't support higher versions. You can read more about SMB support in Axis devices [here](#).
- **Add share without testing:** Select to add the network share even if an error is discovered during the connection test. The error can be, for example, that you didn't enter a password even though the server requires one.

Remove network storage: Click to unmount, unbind, and remove the connection to the network share. This removes all settings for the network share.

Unbind: Click to unbind and disconnect the network share.

Bind: Click to bind and connect the network share.

Unmount: Click to unmount the network share.

Mount: Click to mount the network share.

Write protect: Turn on to stop writing to the network share and protect recordings from being removed. You can't format a write-protected network share.

Retention time: Select how long to keep recordings, to limit the amount of old recordings, or to comply with regulations regarding data storage. If the network storage becomes full, old recordings are removed before the selected time period passes.

Tools

- **Test connection:** Test the connection to the network share.
- **Format:** Format the network share, for example, when you need to quickly erase all data. CIFS is the available file system option.

Use tool: Click to activate the selected tool.

Onboard storage

Important

Risk of data loss and corrupted recordings. Do not remove the SD card while the device is running. Unmount the SD card before you remove it.

Unmount: Click to safely remove the SD card.

Write protect: Turn on to stop writing to the SD card and protect recordings from being removed. You can't format a write-protected SD card.

Autoformat: Turn on to automatically format a newly inserted SD card. It formats the file system into ext4.

Ignore: Turn on to stop storing recordings on the SD card. When you ignore the SD card, the device no longer recognizes that the card exists. The setting is only available to administrators.

Retention time: Select how long to keep recordings to limit the amount of old recordings or comply with data storage regulations. When the SD card is full, it deletes old recordings before their retention time has passed.

Tools

- **Check:** Check for errors on the SD card.
- **Repair:** Repair errors in the file system.
- **Format:** Format the SD card to change the file system and erase all data. You can only format the SD card to the ext4 file system. You need a third-party ext4 driver or application to access the file system from Windows®.
- **Encrypt:** Use this tool to format the SD card and enable encryption. This erases all data stored on the SD card. Any new data you store on the SD card will be encrypted.
- **Decrypt:** Use this tool to format the SD card without encryption. This erases all data stored on the SD card. Any new data you store on the SD card will not be encrypted.
- **Change password:** Change the password required to encrypt the SD card.

Use tool: Click to activate the selected tool.

Wear trigger: Set a value for the SD card wear level at which you want to trigger an action. The wear level ranges from 0–200%. A new SD card that has never been used has a wear level of 0%. A wear level of 100% indicates that the SD card is close to its expected lifetime. When the wear-level reaches 200%, there is a high risk of the SD card malfunctioning. We recommend setting the wear trigger between 80–90%. This gives you time to download any recordings as well as replace the SD card in time before it potentially wears out. The wear trigger allows you to set up an event and get a notification when the wear level reaches your set value.

ONVIF

ONVIF accounts

ONVIF (Open Network Video Interface Forum) is a global interface standard that makes it easier for end-users, integrators, consultants, and manufacturers to take advantage of the possibilities offered by network video technology. ONVIF enables interoperability between different vendor products, increased flexibility, reduced cost and future-proof systems.

When you create an ONVIF account, you automatically enable ONVIF communication. Use the account name and password for all ONVIF communication with the device. For more information see the Axis Developer Community at axis.com.



Add accounts: Click to add a new ONVIF account.

Account: Enter a unique account name.

New password: Enter a password for the account. Passwords must be 1 to 64 characters long. Only ASCII printable characters (code 32 to 126) are allowed in the password, for example, letters, numbers, punctuation, and some symbols.

Repeat password: Enter the same password again.

Role:

- **Administrator:** Has full access to all settings. Administrators can also add, update, and remove other accounts.
- **Operator:** Has access to all settings except:
 - All **System** settings.
 - Adding apps.
- **Media account:** Allows access to the video stream only.



The context menu contains:

Update account: Edit the account properties.

Delete account: Delete the account. You can't delete the root account.

ONVIF media profiles

An ONVIF media profile consists of a set of configurations that you can use to change media stream settings. You can create new profiles with your own set of configurations or use preconfigured profiles for a quick setup.



Add media profile: Click to add a new ONVIF media profile.

Profile name: Add a name for the media profile.



Audio source : Select the audio input source for your configuration.

- **Select configuration:** Select a user-defined configuration from the list and adjust the audio settings. The configurations in the drop-down list correspond to the device's audio inputs. If the device has one audio input, it's user0. If the device has several audio inputs, there will be additional users in the list.



Audio encoder : Select the audio encoding format for your configuration.

- **Select configuration:** Select a user-defined configuration from the list and adjust the audio encoding settings. The configurations in the drop-down list act as identifiers/names of the audio encoder configuration.



Audio decoder : Select the audio decoding format for your configuration.

- **Select configuration:** Select a user-defined configuration from the list and adjust the settings. The configurations in the drop-down list act as identifiers/names of the configuration.



Audio output : Select the audio output format for your configuration.

- **Select configuration:** Select a user-defined configuration from the list and adjust the settings. The configurations in the drop-down list act as identifiers/names of the configuration.

Metadata: Select the metadata to include in your configuration.

- **Select configuration:** Select a user-defined configuration from the list and adjust the metadata settings. The configurations in the drop-down list act as identifiers/names of the metadata configuration.

Create: Click to save your settings and create the profile.

Cancel: Click to cancel the configuration and clear all settings.

profile_x: Click on the profile name to open and edit the preconfigured profile.

Detectors

Audio detection

These settings are available for each audio input.

Sound level: Adjust the sound level to a value from 0–100, where 0 is the most sensitive and 100 the least sensitive. Use the activity indicator as a guide when you set the sound level. When you create events, you can use the sound level as a condition. You can choose to trigger an action if the sound level rises above, falls below or passes the set value.

Power settings

Power status

Shows power status information. Information varies depending on the product.

Power meter

Energy usage

Shows the current power usage, average power usage, maximum power usage, and power consumption over time.



The context menu contains:

- **Export:** Click to export the chart data.

Accessories

I/O ports

Use digital input to connect external devices that can toggle between an open and closed circuit, for example, PIR sensors, door or window contacts, and glass break detectors.

Use digital output to connect external devices such as relays and LEDs. You can activate connected devices through the VAPIX® Application Programming Interface or the web interface.

Port

Name: Edit the text to rename the port.

Direction:  indicates that the port is an input port.  indicates that it's an output port. If the port is configurable, you can click the icons to change between input and output.

Normal state: Click  for open circuit, and  for closed circuit.

Current state: Shows the current state of the port. The input or output is activated when the current state is different from the normal state. An input on the device has an open circuit when it's disconnected or when there is a voltage above 1 VDC.

Note

During restart, the output circuit is open. When the restart is complete, the circuit goes back to the normal position. If you change any settings on this page, the output circuits go back to their normal positions regardless of any active triggers.

Supervised  : Turn on to make it possible to detect and trigger actions if someone tampers with the connection to digital I/O devices. In addition to detecting if an input is open or closed, you can also detect if someone has tampered with it (that is, cut or shorted). To supervise the connection requires additional hardware (end-of-line resistors) in the external I/O loop.

Logs

Reports and logs

Reports

- **View the device server report:** View information about the product status in a pop-up window. The Access Log is automatically included in the Server Report.
- **Download the device server report:** It creates a .zip file that contains a complete server report text file in UTF-8 format, as well as a snapshot of the current live view image. Always include the server report .zip file when you contact support.
- **Download the crash report:** Download an archive with detailed information about the server's status. The crash report contains information that is in the server report as well as detailed debug information. This report might contain sensitive information such as network traces. It can take several minutes to generate the report.

Logs

- **View the system log:** Click to show information about system events such as device startup, warnings, and critical messages.
- **View the access log:** Click to show all failed attempts to access the device, for example, when a wrong login password is used.

Remote system log

Syslog is a standard for message logging. It allows separation of the software that generates messages, the system that stores them, and the software that reports and analyzes them. Each message is labeled with a facility code, which indicates the software type generating the message, and assigned a severity level.



Server: Click to add a new server.

Host: Enter the hostname or IP address of the server.

Format: Select which syslog message format to use.

- Axis
- RFC 3164
- RFC 5424

Protocol: Select the protocol to use:

- UDP (Default port is 514)
- TCP (Default port is 601)
- TLS (Default port is 6514)

Port: Edit the port number to use a different port.

Severity: Select which messages to send when triggered.

Type: Select the type of logs you want to send.

Test server setup: Send a test message to all servers before you save the settings.

CA certificate set: See the current settings or add a certificate.

Plain config

Plain config is for advanced users with experience of Axis device configuration. Most parameters can be set and edited from this page.

Maintenance

Maintenance

Restart: Restart the device. This does not affect any of the current settings. Running applications restart automatically.

Restore: Return most settings to the factory default values. Afterwards you must reconfigure the device and apps, reinstall any apps that didn't come preinstalled, and recreate any events and presets.

Important

The only settings saved after restore are:

- Boot protocol (DHCP or static)
- Static IP address
- Default router
- Subnet mask
- 802.1X settings
- O3C settings
- DNS server IP address

Factory default: Return all settings to the factory default values. Afterwards you must reset the IP address to make the device accessible.

Note

All Axis device software is digitally signed to ensure that you only install verified software on your device. This further increases the overall minimum cybersecurity level of Axis devices. For more information, see the white paper "Axis Edge Vault" at axis.com.

AXIS OS upgrade: Upgrade to a new AXIS OS version. New releases can contain improved functionality, bug fixes, and completely new features. We recommend you to always use the latest AXIS OS release. To download the latest release, go to axis.com/support.

When you upgrade, you can choose between three options:

- **Standard upgrade:** Upgrade to the new AXIS OS version.
- **Factory default:** Upgrade and return all settings to the factory default values. When you choose this option, you can't revert to the previous AXIS OS version after the upgrade.
- **Automatic rollback:** Upgrade and confirm the upgrade within the set time. If you don't confirm, the device reverts to the previous AXIS OS version.

AXIS OS rollback: Revert to the previously installed AXIS OS version.

Troubleshoot

Reset PTR  : Reset PTR if for some reason the **Pan**, **Tilt**, or **Roll** settings aren't working as expected. The PTR motors are always calibrated in a new camera. But calibration can be lost, for example, if the camera loses power or if the motors are moved by hand. When you reset PTR, the camera is re-calibrated and returns to its factory default position.

Calibration  : Click **Calibrate** to recalibrate the pan, tilt, and roll motors to their default positions.

Ping: To check if the device can reach a specific address, enter the hostname or IP address of the host you want to ping and click **Start**.

Port check: To verify connectivity from the device to a specific IP address and TCP/UDP port, enter the hostname or IP address and port number you want to check and click **Start**.

Network trace

Important

A network trace file might contain sensitive information such as certificates or passwords.

A network trace file can help you troubleshoot problems by recording activity on the network.

Trace time: Select the duration of the trace in seconds or minutes and click **Download**.

Learn more

Voice over IP (VoIP)

Voice over IP (VoIP) is a group of technologies that enables voice communication and multimedia sessions over IP networks, such as the internet. In traditional phone calls, analog signals are sent through circuit transmissions over the Public Switched Telephone Network (PSTN). In a VoIP call, analog signals are turned into digital signals to make it possible to send them in data packets across local IP networks or the internet.

In the Axis product, VoIP is enabled through the Session Initiation Protocol (SIP) and Dual-Tone Multi-Frequency (DTMF) signaling.

Session Initiation Protocol (SIP)

The Session Initiation Protocol (SIP) is used to set up, maintain and terminate VoIP calls. You can make calls between two or more parties, called SIP user agents. To make a SIP call you can use, for example, SIP phones, softphones or SIP-enabled Axis devices.

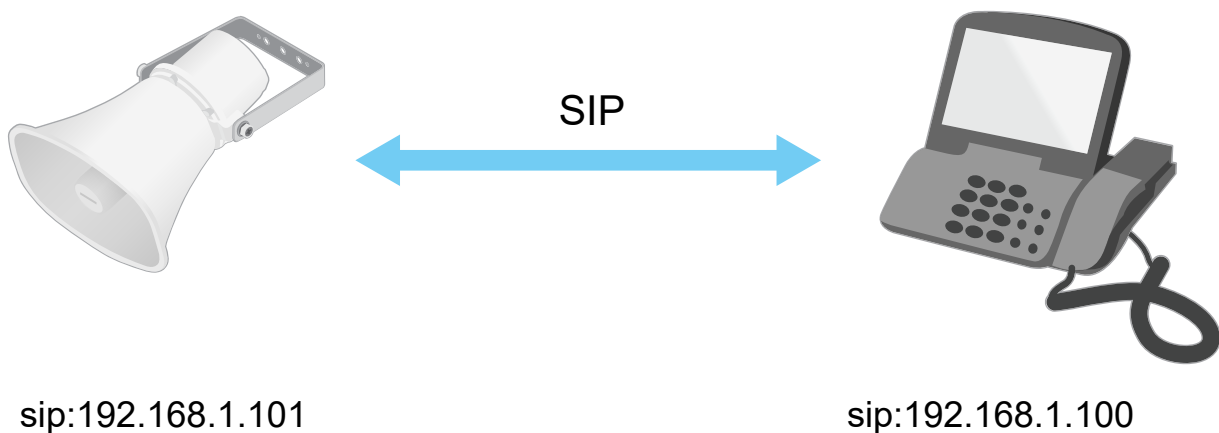
The actual audio or video is exchanged between the SIP user agents with a transport protocol, for example RTP (Real-Time Transport Protocol).

You can make calls on local networks using a peer-to-peer setup, or across networks using a PBX.

Peer-to-peer SIP (P2PSIP)

The most basic type of SIP communication takes place directly between two or more SIP user agents. This is called peer-to-peer SIP (P2PSIP). If it takes place on a local network, all that's needed are the SIP addresses of the user agents. A typical SIP address in this case would be `sip:<local-ip>`.

Example:



You can set up a SIP-enabled phone to call an audio device on the same network using a peer-to-peer SIP setup.

Private Branch Exchange (PBX)

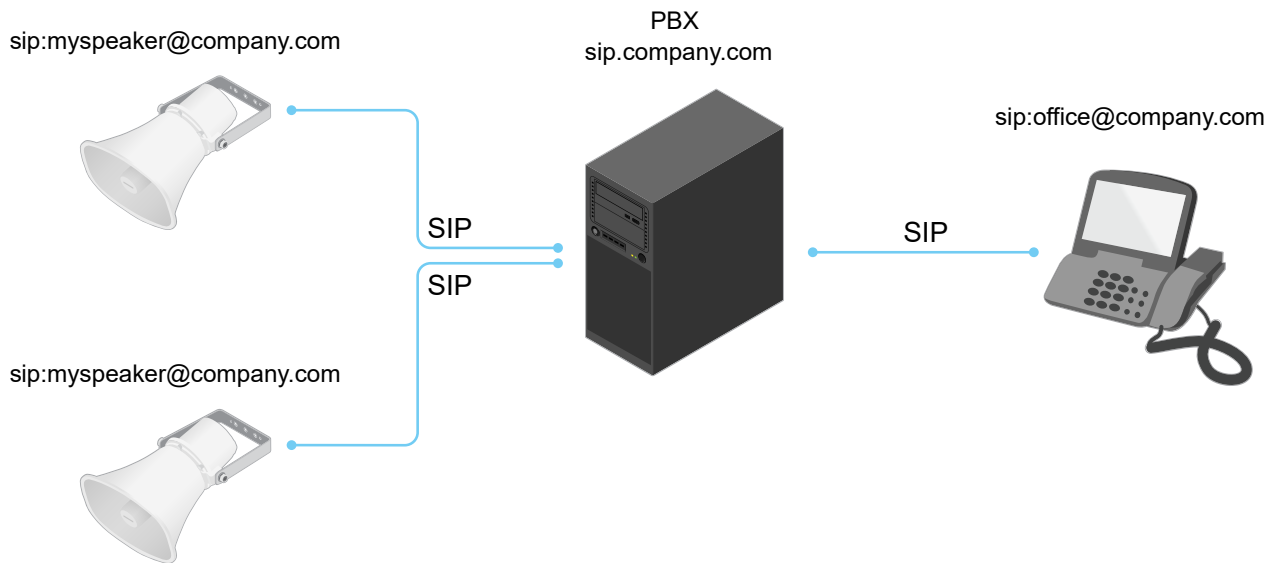
When you make SIP calls outside your local IP network, a Private Branch Exchange (PBX) can act as a central hub. The main component of a PBX is a SIP server, which is also referred to as a SIP proxy or a registrar. A PBX works like a traditional switchboard, showing the client's current status and allowing for example call transfers, voicemail, and redirections.

The PBX SIP server can be set up as a local entity or offsite. It can be hosted on an intranet or by a third party provider. When you make SIP calls between networks, calls are routed through a set of PBXs, that query the location of the SIP address to be reached.

Each SIP user agent registers with the PBX, and can then reach the others by dialing the correct extension. A typical SIP address in this case would be `sip:<user>@<domain>` or `sip:<user>@<registrar-ip>`.

The SIP address is independent of its IP address and the PBX makes the device accessible as long as it is registered to the PBX.

Example:



NAT traversal

Use NAT (Network Address Translation) traversal when the Axis device is located on an private network (LAN) and you want to access it from outside of that network.

Note

The router must support NAT traversal and UPnP®.

Each NAT traversal protocol can be used separately or in different combinations depending on the network environment.

- **ICE** The ICE (Interactive Connectivity Establishment) protocol increases the chances of finding the most efficient path to successful communication between peer devices. If you also enable STUN and TURN, you improve the ICE protocol's chances.
- **STUN** - STUN (Session Traversal Utilities for NAT) is a client-server network protocol that lets the Axis device determine if it is located behind a NAT or firewall, and if so obtain the mapped public IP address and port number allocated for connections to remote hosts. Enter the STUN server address, for example, an IP address.
- **TURN** - TURN (Traversal Using Relays around NAT) is a protocol that lets a device behind a NAT router or firewall receive incoming data from other hosts over TCP or UDP. Enter TURN server address and the login information.

Applications

With applications, you can get more out of your Axis device. AXIS Camera Application Platform (ACAP) is an open platform that makes it possible for third parties to develop analytics and other applications for Axis devices. Applications can be preinstalled on the device, available for download for free, or for a license fee.

To find the user manuals for Axis applications, go to help.axis.com.

Cybersecurity

For product-specific information about cybersecurity, see the product's datasheet at axis.com.

For in-depth information about cybersecurity in AXIS OS, read the *AXIS OS Hardening guide*.

Axis security notification service

Axis provides a notification service with information about vulnerability and other security related matters for Axis devices. To receive notifications, you can subscribe at [axis.com/security-notification-service](https://www.axis.com/security-notification-service).

Vulnerability management

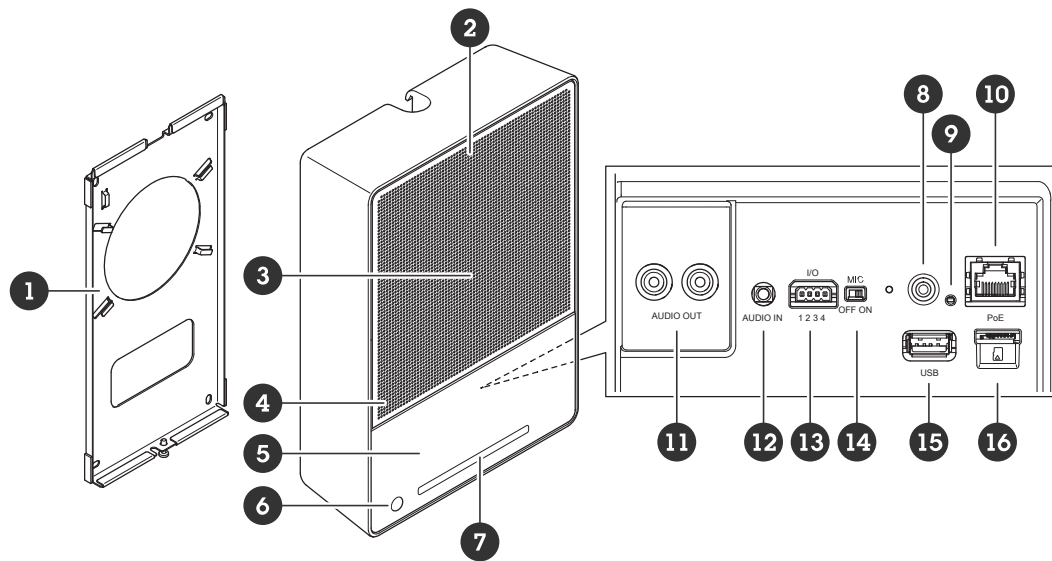
To minimize customers' risk of exposure, Axis, as a **Common Vulnerability and Exposures (CVE) numbering authority (CNA)**, follows industry standards to manage and respond to discovered vulnerabilities in our devices, software, and services. For more information about Axis vulnerability management policy, how to report vulnerabilities, already disclosed vulnerabilities, and corresponding security advisories, see [axis.com/vulnerability-management](https://www.axis.com/vulnerability-management).

Secure operation of Axis devices

Axis devices with factory default settings are pre-configured with secure default protection mechanisms. We recommend using more security configuration when installing the device. To learn more about Axis' approach to cybersecurity, including best practices, resources, and guidelines for securing your devices, go to <https://www.axis.com/about-axis/cybersecurity>.

Specifications

Product overview



- 1 Back plate
- 2 Audio LED
- 3 Speaker
- 4 Microphone
- 5 Display
- 6 PIR sensor
- 7 Signaling LED
- 8 Control button
- 9 Status LED
- 10 Network connector
- 11 Audio out (RCA connector)
- 12 Audio connector
- 13 I/O connector
- 14 Microphone disable switch
- 15 USB connector (disabled, reserved for future use)
- 16 SD card slot

LED indicators

Status LED	Indication
Unlit	Unlit for normal operation.
Green	Steady for 10 seconds for normal operation after startup completed.
Amber	Steady during startup. Flashes during device software upgrade or reset to factory default.
Amber/Red	Flashes if network connection is unavailable or lost.
Red	Flashes slowly if upgrade failed.
Red/Green	Flashes fast when Locate device is selected.

SD card slot

NOTICE

- Risk of damage to SD card. Don't use sharp tools, metal objects, or excessive force when inserting or removing the SD card. Use your fingers to insert and remove the card.
- Risk of data loss and corrupted recordings. Unmount the SD card from the device's web interface before removing it. Don't remove the SD card while the product is running.

For SD card recommendations, see axis.com.



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Buttons

Control button

The control button is used for:

- Calibrating the speaker test. Press and release the control button and a test tone is played.
- Resetting the product to factory default settings. See .

Microphone disable switch

For location of the microphone disable switch, see .

The microphone disable switch is used to mechanically turn the microphone **ON** or **OFF**. The factory default setting for this switch is **ON**.

Connectors

Network connector

RJ45 Ethernet connector with Power over Ethernet (PoE).

NOTICE

The product shall be connected using a shielded network cable (STP). All cables connecting the product to the network shall be intended for their specific use. Make sure that the network devices are installed in accordance with the manufacturer's instructions. For information about regulatory requirements, see the Installation Guide at www.axis.com.

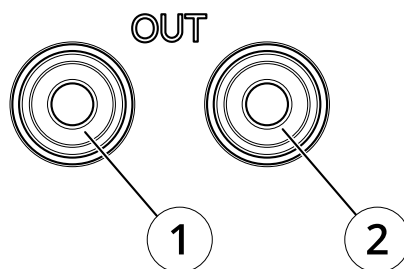
Audio connector



Audio input

1 Tip	2 Ring	3 Sleeve
Unbalanced microphone (with or without electret power) or line	Electret power if selected	Ground
Balanced microphone (with or without phantom power) or line, "hot" signal	Balanced microphone (with or without phantom power) or line, "cold" signal	Ground
Digital signal	Ring power if selected	Ground
Stereo unbalanced microphone (with or without electret power) or line, "left"	Stereo unbalanced microphone (with or without electret power) or line, "right"	Ground

Audio output (RCA connector)



	1 White connector	2 Red connector
Audio output	Audio out (left)	Audio out (right)

I/O connector

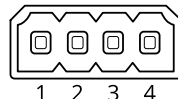
Use the I/O connector with external devices in combination with, for example, motion detection, event triggering, and alarm notifications. In addition to the 0 VDC reference point and power (12 V DC output), the I/O connector provides the interface to:

Digital input – For connecting devices that can toggle between an open and closed circuit, for example PIR sensors, door/window contacts, and glass break detectors.

Supervised input – Enables possibility to detect tampering on a digital input.

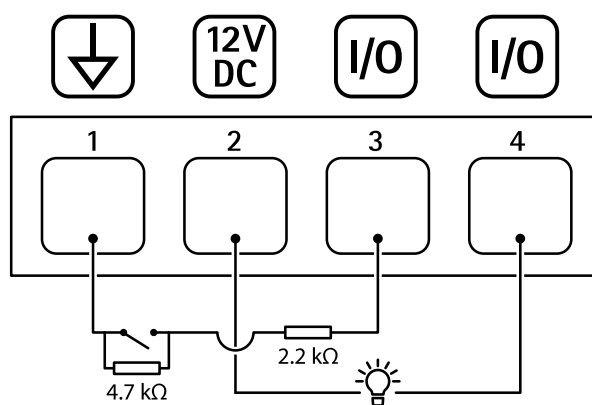
Digital output – For connecting external devices such as relays and LEDs. Connected devices can be activated by the VAPIX® Application Programming Interface, through an event or from the device's web interface.

4-pin terminal block



Function	Pin	Notes	Specifications
DC ground	1		0 VDC
DC output	2	 Can be used to power auxiliary equipment. Note: This pin can only be used as power out.	12 VDC Max load = 50 mA
Configurable (Input or Output)	3–4	Digital input or Supervised input – Connect to pin 1 to activate, or leave floating (unconnected) to deactivate. To use supervised input, install end-of-line resistors. See connection diagram for information about how to connect the resistors.	0 to max 30 VDC
		Digital output – Internally connected to pin 1 (DC ground) when active, and floating (unconnected) when inactive. If used with an inductive load, e.g., a relay, connect a diode in parallel with the load, to protect against voltage transients.	0 to max 30 VDC, open drain, 100 mA

Example:



- 1 DC ground
- 2 DC output 12 V, max 50 mA
- 3 I/O configured as supervised input
- 4 I/O configured as output

Troubleshooting

Reset to factory default settings

Important

Reset to factory default should be used with caution. A reset to factory default resets all settings, including the IP address, to the factory default values.

To reset the product to the factory default settings:

1. Disconnect power from the product.
2. Press and hold the control button while reconnecting power. See .
3. Keep the control button pressed for 10 seconds until the status LED indicator turns amber for the second time.
4. Release the control button. The process is complete when the status LED indicator turns green. If no DHCP server is available on the network, the device IP address will default to one of the following:
 - Devices with AXIS OS 12.0 and later: Obtained from the link-local address subnet (169.254.0.0/16)
 - Devices with AXIS OS 11.11 and earlier: 192.168.0.90/24
5. Use the installation and management software tools, assign an IP address, set the password, and access the product.

You can also reset parameters to factory default through the device's web interface. Go to **Maintenance > Factory default** and click **Default**.

AXIS OS options

Axis offers device software management according to either the active track or the long-term support (LTS) tracks. Being on the active track means continuously getting access to all the latest product features, while the LTS tracks provide a fixed platform with periodic releases focused mainly on bug fixes and security updates.

Using AXIS OS from the active track is recommended if you want to access the newest features, or if you use Axis end-to-end system offerings. The LTS tracks are recommended if you use third-party integrations, which are not continuously validated against the latest active track. With LTS, the products can maintain cybersecurity without introducing any significant functional changes or affecting any existing integrations. For more detailed information about Axis device software strategy, go to axis.com/support/device-software.

Check the current AXIS OS version

AXIS OS determines the functionality of our devices. When you troubleshoot a problem, we recommend that you to start by checking the current AXIS OS version. The latest version might contain a correction that fixes your particular problem.

To check the current AXIS OS version:

1. Go to the device's web interface > **Status**.
2. Under **Device info**, see the AXIS OS version.

Upgrade AXIS OS

Important

- Preconfigured and customized settings are saved when you upgrade the device software (provided that the features are available in the new AXIS OS) although this is not guaranteed by Axis Communications AB.
- Make sure the device remains connected to the power source throughout the upgrade process.

Note

When you upgrade the device with the latest AXIS OS version in the active track, the product receives the latest functionality available. Always read the upgrade instructions and release notes available with each new release before you upgrade. To find the latest AXIS OS version and the release notes, go to axis.com/support/device-software.

1. Download the AXIS OS file to your computer, available free of charge at axis.com/support/device-software.
2. Log in to the device as an administrator.
3. Go to **Maintenance > AXIS OS upgrade** and click **Upgrade**.

When the upgrade has finished, the product restarts automatically.

Technical issues, clues, and solutions

If you can't find what you're looking for here, try the troubleshooting section at axis.com/support.

Problems upgrading AXIS OS

AXIS OS upgrade failure	If the upgrade fails, the device reloads the previous version. The most common reason is that the wrong AXIS OS file has been uploaded. Check that the name of the AXIS OS file corresponds to your device and try again.
Problems after AXIS OS upgrade	If you experience problems after the upgrade, roll back to the previously installed version from the Maintenance page.

Problems setting the IP address

The device is located on a different subnet	If the IP address intended for the device and the IP address of the computer used to access the device are located on different subnets, you cannot set the IP address. Contact your network administrator to obtain an IP address.
The IP address is being used by another device	Disconnect the Axis device from the network. Run the ping command (in a Command/DOS window, type <code>ping</code> and the IP address of the device): • If you receive: <code>Reply from <IP address>: bytes=32; time=10...</code> this means that the IP address may already be in use by another device on the network. Obtain a new IP address from the network administrator and reinstall the device. • If you receive: <code>Request timed out</code>, this means that the IP address is available for use with the Axis device. Check all cabling and reinstall the device.
Possible IP address conflict with another device on the same subnet	The static IP address in the Axis device is used before the DHCP server sets a dynamic address. This means that if the same default static IP address is also used by another device, there may be problems accessing the device.

The device can't be accessed from a browser

Can't log in	When HTTPS is enabled, ensure that the correct protocol (HTTP or HTTPS) is used when attempting to log in. You may need to manually type <code>http</code> or <code>https</code> in the browser's address field. If the password for the root account is lost, the device must be reset to the factory default settings. See .
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The IP address has been changed by DHCP	IP addresses obtained from a DHCP server are dynamic and may change. If the IP address has been changed, use AXIS IP Utility or AXIS Device Manager to locate the device on the network. Identify the device using its model or serial number, or by the DNS name (if the name has been configured). If required, a static IP address can be assigned manually. For instructions, go to axis.com/support.
Certificate error when using IEEE 802.1X	For authentication to work properly, the date and time settings in the Axis device must be synchronized with an NTP server. Go to System > Date and time .

The device is accessible locally but not externally

To access the device externally, we recommend you to use one of the following applications for Windows®:

- AXIS Camera Station Edge: free of charge, ideal for small systems with basic surveillance needs.
- AXIS Camera Station 5: 30-day trial version free of charge, ideal for small to mid-size systems.
- AXIS Camera Station Pro: 90-day trial version free of charge, ideal for small to mid-size systems.

For instructions and download, go to axis.com/vms.

Problems with sound files

Can't upload media clip	The following audio clip formats are supported: • au file format, encoded in μ-law and sampled with 8 or 16 kHz. • wav file format, encoded in PCM audio. It supports encoding as 8 or 16-bit mono or stereo and sample rate of 8 to 48 kHz. • mp3 file format, in mono or stereo with bitrate of 64 kbps to 320 kbps and sample rate of 8 to 48 kHz.
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Media clips are played with different volumes

A sound file is recorded with a certain gain. If your audio clips have been created with different gains, they will be played with a different loudness. Make sure that you use clips that have the same gain.

Can't connect over port 8883 with MQTT over SSL

The firewall blocks traffic using port 8883 as it's deemed insecure.	In some cases the server/broker might not provide a specific port for MQTT communication. It may still be possible to use MQTT over a port normally used for HTTP/HTTPS traffic. • If the server/broker supports WebSocket/WebSocket Secure (WS/WSS), typically on port 443, use this protocol instead. Check with the server/broker provider to see if WS/WSS is supported and which port and basepath to use. • If the server/broker supports ALPN, the use of MQTT can be negotiated over an open port, such as 443. Check with your server/broker provider to see if ALPN is supported and which ALPN protocol and port to use.
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Problems with the sound

The device is not as loud as expected	Check that the device is closed correctly and that there are no obstructions in the horn or on the speaker element.
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Problems with the light

The device is not as bright as expected	Check that a PoE class 6 power supply is used. Check the device's ambient temperature. If the device is installed in a high temperature environment, the lights will dim automatically.
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Performance considerations

When setting up your system, it is important to consider how various settings and situations affect the amount of needed bandwidth (the bitrate) required.

The following factors are the most important to consider:

- Heavy network utilization due to poor infrastructure affects the bandwidth.
- Running multiple AXIS Camera Application Platform (ACAP) applications simultaneously may affect the general performance.

Power considerations

Capabilities can be limited depending on what PoE class you use.

PoE class 6	No limitations.
PoE class 4	Limited sound pressure level, display brightness and signaling LED brightness.
PoE class 3	You can configure a product to prepare it for installation, but the product must have PoE class 4 or 6 to be used in the final installation.

Contact support

If you need more help, go to axis.com/support.

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