

AXIS Audio Manager Pro

What's new

AXIS Audio Manager Pro

For a full description of what is contained in each release, see *AXIS Audio Manager Pro release notes*

AXIS Audio Manager Pro

AXIS Audio Manager Pro 4.6.2

AXIS Audio Manager Pro 4.6.2

September, 2024

- Support for AXIS OS 12.0. This release is ready for the breaking changes introduced in AXIS OS 12.0, particularly the removal of root privileges and new signing requirements for ACAPs. This version requires devices to run AXIS OS version 11.11 or later (or AXIS OS version 10.12 for older devices with no 11-track available).
- Added echo cancellation setting to device source and talkback configuration. Enable on supported devices to prevent feedback and echo during two-way communication.

AXIS Audio Manager Pro

AXIS Audio Manager Pro 4.6

AXIS Audio Manager Pro 4.6

May, 2024

- New version of the operational API v1.1 that enables you to programmatically interact with an existing audio site. See API specification for details. News include possibility to page into the system via SIP, RTP and HTTP, volume controllers and device health information.

See *AXIS Audio Manager Pro API*

- Support of AXIS devices:
 - *AXIS C1310-E Mk II Network Horn Speaker*
 - *AXIS C1410 Mk II Network Mini Speaker*
 - *AXIS XC1311 Explosion-Protected Network Horn Speaker*
- Six new languages: Czech, Swedish, Finnish, Turkish, Thai, Vietnamese

AXIS Audio Manager Pro

AXIS Audio Manager Pro 4.5

AXIS Audio Manager Pro 4.5

February, 2024

- Adaptations to the AXIS ACAP root user changes communicated at axis.com/developer-community/news/axis-os-root-acap-signing.
- Software Bill of Material (SBOM) document published with this and future releases.
- Switch to Pipewire (from ALSA) for low level audio routing and processing.
- SIP improvements
 - Max call length option added. SIP connections longer than this value will be automatically stopped. The settings can be used to automatically resolve the problem if someone forgets to hang up a SIP call.


× Edit SIP server 

SETTINGS

CLIENTS

Authorization

Require authorization

 Make sure that you entered authentication ID and password

Registration

Registration - min (seconds)

30



Registration - max (seconds)

180

Call handling

End calls after (seconds)

3600



Infinite call duration

SAVE

CANCEL

✕ Edit SIP server

SETTINGS CLIENTS TRUNKS

Trunks are used for connecting remote PBX servers to the SIP server

NAME	STATUS	REMARKS
MyTrunk	Remote server unreachable	

