

AXIS C1410 Network Mini Speaker

Speaker with wide-area coverage for voice messages

AXIS C1410 Network Mini Speaker is a discrete and affordable speaker that fits into smaller spaces and provides wide-area audio coverage. It allows users to remotely warn off intruders before they commit a crime, to deliver instructions during an emergency or to make general announcements. Built-in memory supports pre-recorded messages, or personnel can respond to notifications with live speak. Digital signal processing (DSP) ensures clear sound. Open standards support easy integration with network video, access control, analytics, and VoIP. AXIS C1410 is a standalone unit that can be placed almost anywhere, which supports a flexible, scalable and cost-effective approach to system design.

- > [All-in-one speaker system](#)
- > [Connects to standard network](#)
- > [Simple installation with PoE](#)
- > [Remote health testing](#)
- > [PIR sensor for motion detection](#)



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Audio hardware		Call: DTMF, State changes
Enclosure	One way enclosure with 2.5 inches broadband dynamic cone speaker	Event actions
Max sound pressure level	90 dB	Play audio clip, send SNMP trap, send Auto Speaker Test, status LED, File upload via HTTP, network share and email, Notification via email, HTTP, HTTPS and TCP
Frequency response	100 Hz - 20 kHz	Built-in installation aids
Coverage pattern	145° coaxial	Test tone verification and identification
Audio input/output	Built-in microphone (can be disabled)	Cybersecurity
Built-in microphone specification	50 Hz - 12 kHz	Edge security
Amplifier description	Built-in 7 W Class D amplifier	Software: Signed firmware, brute force delay protection, digest authentication and OAuth 2.0 RFC6749 OpenID Authorization Code Flow for centralized ADFS account management, password protection
Digital signal processing	Built-in and pre-configured	Network security
Audio management		IEEE 802.1X (EAP-TLS, PEAP-MSCHAPv2), HTTPS/HSTS, TLS v1.2/v1.3, Network Time Security (NTS), X.509 Certificate PKI, host-based firewall
AXIS Audio Manager Edge	Built in: – Content management for music and live/pre-recorded announcements. – Scheduling to decide when and where to play specific content. – Content prioritization to ensure urgent messages interrupt the schedule. – Zone management allowing you to divide up to 200 speakers into 20 zones. – Health monitoring for remote discovery of system errors. – User management to control who has access to what features. See separate datasheet for more details.	Documentation
AXIS Audio Manager Pro	For larger and more advanced systems. Sold separately. See separate datasheet for specifications.	<i>AXIS OS Hardening Guide</i> <i>Axis Vulnerability Management Policy</i> <i>Axis Security Development Model</i> AXIS OS Software Bill of Material (SBOM) To download documents, go to axis.com/support/cybersecurity/resources To read more about Axis cybersecurity support, go to axis.com/cybersecurity
Audio software		General
Audio streaming	One-way/two-way with optional half-duplex echo cancellation. Mono.	Casing
Audio encoding	AAC LC 8/16/32/48 kHz, G.711 PCM 8 kHz, G.726 ADPCM 8 kHz, Axis μ -law 16 kHz, WAV, MP3 in mono/stereo from 64 kbps to 320 kbps. Constant and variable bit rate. Sampling rate from 8 kHz up to 48 kHz.	Plastic casing, IP40 rated Colors: White Repaintable. For repainting instructions and impact on warranty, contact your Axis partner.
System on chip (SoC)		Power
Model	i.MX GULL	Power over Ethernet (PoE) IEEE 802.3af/802.3at Type 1 Class 3 (max 12.95 W)
Memory	256 MB RAM, 512 MB Flash	Connectors
Network		RJ45 10BASE-T/100BASE-TX PoE
Network protocols	IPv4/v6 ^a , HTTP, HTTPS, SSL/TLS, QoS Layer 3 DiffServ, FTP, SFTP, CIFS/SMB, SMTP, Bonjour, UPnP™, SNMP v1/v2c/v3 (MIB-II), DNS, DynDNS, NTP, RTSP, RTP, TCP, UDP, IGMPv1/v2/v3, RTCP, ICMP, DHCP, ARP, SOCKS, SSH, NTCIP, SIP (Cisco, Avaya, Asterisk)	Reliability
System integration		Designed for 24/7 operation.
Application Programming Interface	Open API for software integration, including VAPIX®, One-click cloud connection, AXIS Camera Application Platform (ACAP).	LED indicators
VoIP	Support for Session Initiation Protocol (SIP) for integration with Voice over IP (VoIP) systems. Peer to peer or integrated with SIP/PBX. Tested with: SIP client such as Cisco, Bria and Grandstream and PBX suppliers such as Cisco and Asterisk. Supported SIP features: secondary SIP server, IPv6, SRTP, SIPS, SIP TLS, DTMF (RFC2976 and RFC2833), NAT (ICE, STUN, TURN) Supported codecs: PCMU, PCMA, opus, L16/16000, L16/8000, speex/8000, speex/16000, G.726-32	PIR sensor
Intelligent audio	Auto Speaker Test (verification via built-in microphone)	Passive infrared (PIR) motion sensor. Max range: 6 m (19.7 ft)
Event triggers	Virtual inputs, AXIS Camera Application Platform (ACAP)	Operating conditions
		-20 °C to 50 °C (4 °F to 122 °F) Humidity 10–85% RH (condensing)
		Storage conditions
		-40 °C to 65 °C (-40 °F to 149 °F) Humidity 5–95% RH (non-condensing)
		Approvals
		EMC EN 55032 Class B, EN 55035, EN 61000-3-2, EN 61000-3-3, EN 61000-6-1, EN 61000-6-2, FCC Part 15 Subpart B Class B, ICES-3(B)/NMB-3(B), RCM AS/NZS CISPR 32 Class B, VCCI Class B Safety IEC/EN/UL 62368-1
		Dimensions
		Ø182 X 52 mm (7.2 x 2.1 in)
		Weight
		600 g (1.3 lb)
		Included accessories
		Installation Guide, AVHS authentication key
		Optional accessories
		AXIS T91A33 Lighting Track Mount, AXIS T91A23 Tile Grid Ceiling Mount, AXIS TC1601 Universal Mount For more accessories, see axis.com
		Mounting option
		When you mount the device on a soft tile ceiling, use a plate behind the tile for reinforcement.
		Video management software
		AXIS Companion, AXIS Camera Station, video management software from Axis' Application Development Partners available on axis.com/techsup/software
		Languages
		English, German, French, Spanish, Italian
		Warranty
		5-year warranty, see axis.com/warranty

a. Audio synchronization with IPv4 only.