

Cisco ATA 192 Multiplatform Analog Telephone Adapter

The Cisco® ATA 192 Multiplatform Analog Telephone Adapter is a 2-port handset-to-Ethernet adapter that brings traditional analog devices into the IP world.



Product overview

The Cisco ATA 192 Multiplatform Analog Telephone Adapter turns traditional telephone, fax, and overhead paging communications devices into IP devices for greater cost-effectiveness. Customers can take advantage of IP telephony applications by connecting their analog devices to Cisco analog telephone adapters.

The ATA 192 is the preferred solution to address the needs of customers who connect to enterprise networks, small offices, or unified communications

as a service from the cloud. It has two standard FXS ports, which can be configured independently as two Session Initiation Protocol (SIP) registrations. It also has two 100BASE-T ports with an integrated high-performance router to extend local network connectivity. With the ATA 192, customers can protect and extend their existing investment in analog systems, as well as smooth their migration to pure voice over IP in a more affordable and reliable way.

Platform support information

The Cisco ATA 192 Multiplatform Analog Telephone Adapter is designed to work with Cisco Webex Calling, BroadCloud and BroadWorks, and other third-party call control system Metaswitch, Asterisk.

Licensing information

No license is required to connect the Cisco ATA 192 Multiplatform Analog Telephone Adapter to third-party call control systems.

Features and benefits

Feature	Benefit
Voice quality	Offers clear, natural-sounding voice quality via advanced preprocessing, high-performance echo cancellation, voice activity detection, and comfort noise generation
Cloud provisioning	Enables zero-touch provisioning via TR-069 and XML configuration files
Security	Provides a complete security solution for both media and signaling
Problem reporting (PRT)	Improves serviceability with a dedicated PRT button for problem reporting and log collection
IPv6	Enables IPv6 dual stack to help with migration to IPv6

Product specifications

Feature	Specifications
Physical dimensions (H×W×D)	3.9 x 3.9 x 1.1 in. (100x100x28 mm)
Weight	4.7 oz (132.1 g)
Hardware	Interface: Two RJ-11 FXS ports, two 10/100 Mbps RJ-45 Ethernet ports Button: Reset / Problem Reporting (PRT) LED indicators: Power, Network, Phone 1, Phone 2, PRT Wall mountable
Subscriber Line Interface Circuit (SLIC)	Ring voltage: 40 to 90 Vpk configurable Ring frequency accuracy: 1% Ring waveform: Trapezoidal or sinusoidal Maximum ringer load: 3 Ringer Equivalence Numbers (RENs) On-hook voltage (tip and ring): -46 to -56V Off-hook current: 25mA +/- 10% Terminating impedance: 600 ohm resistive, 900 ohm resistive, or 220 ohm + 820 ohm 120 nF complex impedance Frequency response: 300 to 3400 Hz Return loss (600 ohm, 300 to 3400 Hz): up to 26 dB Idle channel noise: <-65 dBm Op Longitudinal balance: 58 dB (typical) Voice quality Mean Opinion Score (MOS): >4.0 Voice quality jitter: <150 ms

Feature	Specifications
Networking	MAC address IPv4 only IPv6 only IPv4/IPv6 dual stack Session Initiation Protocol (SIP) Transmission Control Protocol (TCP) User Datagram Protocol (UDP) Real Time Protocol (RTP) Real Time Control Protocol (RTCP) HTTP HTTPS Trivial File Transfer Protocol (TFTP) Address Resolution Protocol (ARP) DNS A/AAAA and SRV records Dynamic Host Configuration Protocol (DHCP) client Internet Control Message Protocol (ICMP) Simple Network Time Protocol (SNTP) Cisco Discovery Protocol Link Layer Discovery Protocol (LLDP) Point-to-Point Protocol over Ethernet (PPPoE)
Routing	Routing and bridging Static and dynamic address assignment Network Address Translation (NAT) DHCP client reservation MAC address cloning Port forwarding DMZ mode VPN pass-through: IP Security (IPsec) Encapsulating Security Payload (ESP), Point-to-Point Tunneling Protocol (PPTP), Layer 2 Tunneling Protocol (L2TP)
Quality of Service (QoS)	IEEE 802.1p/Q (QoS and VLAN tagging) Differentiated Services (DiffServ), Type of Service (ToS)
Telephony	Anonymous call and call blocking Call forwarding: No answer, busy, and all Call hold and resume Caller ID blocking Caller ID generation (name and number): Bellcore, BT, and European Telecommunications Standards Institute (ETSI) Caller ID with name and number Call pickup and group pickup Call transfer, call return, and call back on busy

Feature	Specifications
	<ul style="list-style-type: none"> Call waiting Configurable ring frequency Configurable tones and cadences Disconnect tone Distinctive ringing: Calling and called number Do not disturb Forced Authorization Code (FAC)/Client Matter Code (CMC) Flash hook timer Hook flash event signaling Hotline and warm line calling Message Waiting Indicator (MWI) tones Music on hold Off-hook warning tone Polarity control Redial Selective and anonymous call rejection SIP redundancy Speed dial Streaming audio server: Up to 4 sessions Three-way conference calling with local mixing Tip and ring voltage adjustment setting Visual Messaging Waiting Indicator (VMWI) using frequency shift keying (FSK) Network Address Translation (NAT) Session Traversal Utilities for NAT (STUN)
Audio	<ul style="list-style-type: none"> Codec: G.711 a-law, G.711 u-law, G.729a, G.729ab, G.726 Codec name assignment Full-duplex audio Echo cancellation Voice activity detection Silence suppression Configurable silence threshold Comfort noise generation Adaptive jitter buffer Frame loss concealment Adjustable audio frames per packet Call progress tone generation Impedance and gain adjustment Dynamic audio payload
Fax	<ul style="list-style-type: none"> Real-time fax over IP via T.38 fax relay (Group 3) Fax pass-through via G.711 (Group 3) Fax tone detection and pass-through Automatic negotiation on transmission rate

Feature**Specifications**

Provisioning and management

Cloud provisioning (remote configuration)
Web-based administration
Interactive Voice Response (IVR)
Automated provisioning and upgrading via HTTP, HTTPS,
and TFTP TR-069
SSH access
Simple Network Management Protocol (SNMPv3)
Report generation and event logging
Dedicated PRT button
Support for RTP statistics
Syslog (multilevel granularity)
Ping and trace route diagnostics
Configuration management: Backup and restore
Dual image

Security

Password-protected system reset to factory default
Password-protected administrator and user access authority
Provisioning, configuration, and authentication
HTTPS with factory-installed client certificate
Advanced Encryption Standard (AES) encryption
SIP over Transport Layer Security (TLS1.1 and TLS1.2)
Secure (encrypted) calling using Secure RTP (sRTP)
Encrypted configuration files
Image authentication
Secure boot
Secure Shell (SSH)

Power

DC input voltage: 5V DC at 2.4A maximum
Power consumption: 5W
Switching type (100-240V) automatic
Power adapter: 100-240V and 50-60 Hz (26-34 VA) AC input,
with 1.8m cord

Reliability

Mean Time Between Failures (MTBF): 300,000 hours
Operating temperature: 32° to 104°F (0° to 40°C)
Nonoperating temperature: 14° to 140°F (-10° to 60°C)
Humidity: Operating 10% to 90%, noncondensing / nonoperating 10%
to 95%, noncondensing

Compliance (regulatory)

CE Markings per directives 2014/30/EU and 2014/35/EU

Feature**Specifications**

Compliance (safety)

UL 60950 Second Edition
CAN/CSA-C22.2 No. 60950 Second Edition
IEC 60950-1:2005 (Second Edition) + A1:2009 + A2:2013
and/or AS/NZS 60950.1:2015

Compliance (safety)

AS/NZS CISPR 32: 2015 Class B
CISPR 32: 2015 Class B
EN 55032: 2015 Class B
EN 61000-3-2: 2014 Class A
EN 61000-3-3: 2013
EN 55024: 2010+A1: 2015
EN 61000-4-2: 2009
EN 61000-4-3: 2006+A1:2008+A2:2010
EN 61000-4-4: 2012
EN 61000-4-5: 2014
EN 61000-4-6: 2014+AC: 2015
EN 61000-4-8: 2010
EN 61000-4-11: 2004
FCC Part 15, Subpart B
ANSI C63.4-2014
ICES-003 Issue 6: 2016
ANSI C63.4-2014
VCCI-TECHNICAL REQUIREMENTS (VCCI-CISPR 32: 2016)/
CISPR 32: 2015 class B

Ordering information

Part number	Product description
ATA192-3PW-K9	2-port analog telephone adapter with router for multiplatform
ATA191-PWR	Spare power adapter for ATA 191 and ATA 192

Warranty information

The Cisco ATA 192 Multiplatform Analog Telephone Adapter is covered by a Cisco 1-year limited hardware warranty.

Learn more

For additional details on the Cisco ATA 192 Multiplatform Analog Telephone Adapter, go to <https://www.cisco.com/c/en/us/products/unified-communications/ata-190-series-analog-telephone-adapters/index>

For more information, please contact a sales representative. Visit [ringcentral.com](https://www.ringcentral.com) or call 855-774-2510.

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