# ılıılı cısco

# Cisco SPA232D Multi-Line DECT ATA

### Product Overview

The Cisco<sup>®</sup> SPA232D Multi-Line DECT ATA is a mobility enhanced, affordable, highly reliable voice gateway for connecting an analog phone or fax machine to a voice-over-IP (VoIP) service provider. It can also intelligently route calls to the public switched telephone network (PSTN).

The revolutionary advances in the Cisco SPA232D include an integrated DECT base station used exclusively with the Cisco SPA302D Multi-Line DECT Handset, providing a mobility solution for users who tend to frequently move about their location but must always be reachable around the premises. The SPA232D supports essential voice features such as caller ID, call transfer, call waiting, call forwarding, voicemail, and much more to provide a comprehensive, advanced, and highly innovative VoIP solution.

The SPA232D provides one RJ-11 FXS port to connect a standard analog phone or fax machine and one FXO port to connect to the PSTN. The SPA232D FXS and FXO lines can be independently configured through software by the service provider or the end user. Users can take full advantage of their broadband phone service by enabling intelligent "hop-on, hop-off" applications to route local calls from mobile phones and land lines over to their VoIP service provider, and conversely.

The Cisco SPA232D also includes two 100BASE-T RJ-45 Ethernet interfaces to connect to a home or business LAN, and an Ethernet port to connect to a broadband access device. It uses international standards for voice and data networking for reliable voice and fax operation, and it can be used in residential, small office or home office (SOHO), and business environments including full-featured hosted service providers using Broadsoft, MetaSwitch or open source IP-PBX solutions such as Asterisk.

#### Features and Benefits

The Cisco SPA232D Multi-Line DECT ATA (Figures 1 and 2) offers the following:

- High-quality feature-rich VoIP service through your broadband Internet connection
- One standard telephone port for use with analog phones or fax machines and one port for PSTN connectivity to route local calls from mobile phones and land lines over to VoIP service providers and conversely
- Cost-effective, multi line on-premises mobility with its integrated DECT base station designed exclusively for Cisco SPA302D Multi-Line DECT Handsets and capable of registering up to five handsets to a SPA232D ATA supporting four simultaneous active calls
- Highly configurable and secure remote provisioning capabilities to enable mass-scale service provider activation and deployment
- · Ideal solution for residential, SOHO, and business environments



Figure 2. Ports on Cisco SPA232D



Table 1 lists additional features and benefits of the Cisco SPA232D Multi-Line DECT ATA.

 Table 1.
 Features and Benefits of Cisco SPA232D Multi-Line DECT ATA

Feature	Benefit
Toll-quality voice and carrier-grade feature support	The SPA232D delivers clear, high-quality voice communication in diverse network conditions. Excellent voice quality in a demanding IP network is consistently achieved through our advanced implementation of standard voice-coding algorithms. The SPA232D is interoperable with common telephony equipment such as voicemail, fax, PBX, interactive-voice-response (IVR) systems, and many third-party call-control systems such as Broadsoft and Asterisk.
Large-scale deployment and management	The SPA232D offers all the important features and capabilities with which service providers can provide customized VoIP services to their subscribers. It can be remotely provisioned, and it is software-upgradable. A secure profile upload saves providers the time, expense, and hassle of managing and preconfiguring or reconfiguring subscriber equipment for deployment.
Outstanding security	The Cisco SPA232D supports highly secure, encryption-based methods for communication, provisioning, and servicing.
Comprehensive feature set	The standards-based Cisco SPA232D is compatible with essential Internet VoIP provider features such as caller ID, call waiting, call transfer, call forwarding, three-way conferencing, voicemail, and much more to provide a complete, affordable, and highly reliable VoIP solution.
Enhanced mobility	An integrated DECT base station with exclusive support for Cisco SPA302D Multi-Line DECT Handsets provides a simple and reliable mobility solution for users who tend to frequently move about their location but must be always reachable around the premises.
Easy installation and changes	The web-based configuration utility enables quick deployment and easy changes.
Peace of mind	Cisco solutions deliver the solid reliability you expect from Cisco. All solution components have been rigorously tested to help ensure easy setup, interoperability, and performance.

Figure 1. Cisco SPA232D Multi-Line DECT ATA

## **Product Specifications**

Table 2 gives the specifications of the Cisco SPA302D Multi-Line DECT Handsets.

#### **Table 2.**Product Specifications

\* Note: Many specifications are programmable within a defined range or list of options. Please refer to the Cisco SPA232D Administration Guide for details. The configuration profile is uploaded to the SPA232D at the time of provisioning.

Description	Specification
Voice gateway	Session Initiation Protocol (SIP) v2 (RFCs 3261, 3262, 3263, and 3264) SIP proxy redundancy: Dynamic through Domain Name System (DNS) SRV record Reregistration with primary SIP proxy server SIP support in Network Address Translation (NAT) networks (including Serial Tunneling [STUN]) Highly secure (encrypted) calling with Secure Real-Time Transport Protocol (SRTP) Codec name assignment G.722 G.711 (a-law and µ-law) G.726 (32 kbps) G.729 (b and ab) Dynamic payload Adjustable audio frames per packet Dual-tone multifrequency (DTMF): In-band and out-of-band (RFC 2833) (SIP INFO)
DECT	1910 to 1930 MHz (SPA232D-G1) 1880 to 1920 MHz (SPA232D-G7) Range: Indoor, 165 ft (50m); outdoor, 980 ft (300m) Five handset registrations on a SPA232D Four simultaneous active calls 11 individual SIP accounts (10 DECT and 1 FXS) Page/Registration button
DECT cordless handset* (SPA302D)	<ul> <li>* Please refer to the Cisco SPA302D Multi-Line DECT Handset data sheet for additional technical details.</li> <li>1.8-in. TFT (128 x 160 pixels), 65,000 colors, backlit with scratch-resistant lens</li> <li>Software upgradable over the air (SUOTA)</li> <li>White illuminated keypad backlight</li> <li>Administrative personal identification number (PIN) code support</li> <li>Dial keypad lock</li> <li>Speed dial: Eight programmable</li> <li>Private and shared phone books (50 records)</li> <li>Call history (50 records filtered by Outgoing, Incoming, and Missed)</li> <li>Visual Message-Waiting Indicator (VMWI)</li> <li>Five ringtones</li> <li>Call mute</li> <li>Call hold/resume</li> <li>New call (support for two active call segments)</li> <li>Redial</li> <li>Call park and unpark</li> <li>Intercom (handset-to-handset)</li> <li>Hearing Aid Compatibility (HAC)</li> <li>* Some features may require support by the call-control server.</li> </ul>

Description	Specification
Voice features	Quality of service (QoS) (Ethernet port upstream bandwidth control)
Voice realures	Independent configurable dial plans with interdigit timers and IP dialing (per line)
	Call progress tone generation
	Jitter buffer: Adaptive
	Frame-loss concealment
	Full-duplex audio
	Echo cancellation (G.165 and G.168)
	Voice activity detection (VAD)
	Silence suppression
	Comfort Noise Generation (CNG)
	Attenuation and gain adjustments
	Flash hook timer
	MWI tones
	VMWI through frequency shift keying (FSK)
	Polarity control
	Hook flash event signaling
	Caller ID generation (name and number): Bellcore, DTMF, and ETSI
	Music-on-hold (MOH) client
	Streaming audio server: Up to 10 sessions
	МОН
	Call waiting and call waiting caller ID
	Caller ID with name and number
	Caller ID blocking
	Selective and anonymous call rejection
	Call forwarding: no answer, busy, and all
	Do not disturb
	Call transfer, call return, and call back on busy
	Three-way conference calling with local mixing
	Per-call authentication and associated routing
	Call Blocking with Toll Restriction
	Distinctive ringing: Calling and called number
	Off-hook warning tone
	Advanced inbound and outbound call routing
	Hot line and warm line calling
	Long silence (configurable time setting) silence threshold
	Disconnect tone (for example, reorder tone)
	Configurable ring frequency
	Ring validation time setting
	Tip and ring voltage adjustment setting
	Ring indication delay setting
Fax capability	Fax tone detection pass-through
	Fax pass-through using G.711
	Real-time fax over IP through T.38 Fax Relay (T.38 support is dependent on fax machine and network and transport resilience.)
Enhanced actower	
Enhanced gateway authentication and routing	VoIP-to-PSTN (United States) service call origination and termination
features	PSTN-to-VoIP (United States) service call origination and termination
	Single- and two-stage dialing
	Forward calls to VoIP service: Selective, Authenticated, and All
	Forward calls to PSTN service: Selective, Authenticated, and All
	PSTN line sharing with multiple extensions
	Automatic PSTN fallback (loss of power or IP service to unit, with quiescence to normal operations)
	Advanced inbound and outbound call routing
	Independent configurable dial plans: Up to 8
	Force PSTN disconnection
	Sequential dialing support

Description	Specification
	VoIP to PSTN:
	VolP-to-PSTN gateway enable and disable
	<ul> <li>VolP caller authentication method (None, PIN, and HTTP digest)</li> </ul>
	VolP PIN maximum retry setting
	One-stage dialing enable and disable
	VolP caller ID pattern matching
	VolP access-allowed caller list (no further authentication)
	VolP caller PIN and associated dial plan
	PSTN to VolP:
	PSTN to VoIP.     PSTN-to-VoIP gateway enable and disable
	VoIP caller authentication method (None, PIN, and HTTP digest)
	Ring through to FXS enable and disable
	Ring-through tone: Configurable
	Caller ID (Bellcore Type 1) for VoIP service access
	Caller ID enable and disable
	PIN maximum retry settings
	Access-allowed caller list (no further authentication)
	Caller PIN and associated dial plan
	Least-cost routing (through outbound VoIP: Line1 dial plan
FXO control settings	VoIP answer-delay timer
	PSTN answer-delay timer
	VoIP PIN digit timeout timer
	PSTN PIN digit timeout timer
	PSTN-to-VoIP call maximum-duration timer
	<ul> <li>VolP-to PSTN call maximum-duration timer</li> </ul>
	PSTN ring-through delay timer
	PSTN dialing delay timer
	VolP DLG (Dialog) refresh-interval timer
	PSTN ring timeout timer
	PSTN Disconnection Detection
	<ul> <li>CPC (Calling Party Control) (removal of tip and ring voltage momentarily)</li> </ul>
	Polarity reversal
	Long silence (configurable time setting)
	<ul> <li>Disconnect tone (for example, reorder tone)</li> </ul>
	Silence threshold
	International Control
	FXO port impedance: Settings by country
	Ring frequency: Configurable
	<ul> <li>SPA-to-PSTN and PSTN-to-SPA gain settings</li> </ul>
	Ring frequency: Maximum setting     Ding unlideting time setting
	Ring validation time setting     Tip and ring values a division and setting
	Tip and ring voltage adjustment setting
	Ring indication delay setting
	Ring timeout setting
	Ring threshold
	Line-in-use voltage setting
Security	Password-protected system reset to factory default
	Password-protected administrator and user access authority
	Provisioning, configuration, and authentication
	Secure HTTP (HTTPS) with factory-installed client certificate
	HTTP digest: Encrypted authentication with (MD5; RFC 1321)
	Up to 256-bit Advanced Encryption Standard (AES) encryption
	SIP Transport Layer Security (TLS)
	Reset button bypass (provisionable)
	21 (f)

Description	Specification
Data networking	MAC address (IEEE 802.3)
Data networking	IPv4 (RFC 791)
	Address Resolution Protocol (ARP)
	DNS-A record (RFC 1706) and SRV record (RFC 2782)
	Dynamic Host Configuration Protocol (DHCP) server and client (RFC 2131)
	DHCP client reservation
	DHCP Option 159 and Option 160
	Point-to-Point Protocol over Ethernet (PPoE) client (RFC 2516)
	Internet Control Message Protocol (ICMP; RFC 792)
	TCP (RFC 793)
	User Datagram Protocol (UDP; RFC 768)
	Real-Time Transport Protocol (RTP; RFCs 1889 and 1890)
	Real-Time Control Protocol (RTCP; RFC 1889)
	Differentiated Services (DiffServ) (RFC 2475) and type of service (ToS; RFCs 791 and1349)
	VLAN tagging (IEEE 802.1p)
	Simple Network Time Protocol (SNTP) (RFC 2030)
	Upload data rate limiting: Static and automatic
	QoS: Voice packet prioritization over other packet types
	MAC address cloning
	Port forwarding
	SIP channels support for both UDP and TCP transport
	VPN pass-through with IP Security encapsulating security payload (IPsec ESP), Point-to-Point Tunneling
	Protocol (PPTP), and Layer 2 Tunneling Protocol (L2TP)
Provisioning, administration, and	Web browser administration and configuration through integral web server
maintenance	Telephone keypad configuration with IVR prompts
	Automated provisioning and upgrade through HTTPS, HTTP, and Trivial File Transfer Protocol (TFTP)
	TR-069
	Asynchronous notification of upgrade availability using NOTIFY
	Nonintrusive, in-service upgrades
	Report generation and event logging
	Statistics in BYE message
	Syslog and debug server records: Per-line configurable web browser
	Configuration management: Backup and restore
	Support for Bonjour
	Support for Link Layer Discovery Protocol (LLDP) and Cisco Discovery Protocol (CDP)
Physical interfaces	One WAN 100BASE-T RJ-45 Ethernet port (IEEE 802.3)
	One LAN 100BASE-T RJ-45 Ethernet port (IEEE 802.3)
	One RJ-11 FXS phone port for analog circuit telephone device (tip and ring)
	One RJ-11 FXO phone port for PSTN or PBX connection
	Reset button
Subscriber-line interface circuit (SLIC)	Ring voltage: 40–90 Vpk configurable
	Ring frequency: 16–50 Hz
	Ring waveform: Trapezoidal or sinusoidal
	Maximum ringer load: Five ringer equivalence numbers (RENs)
	On-hook and off-hook characteristics:
	On-hook voltage (tip/ring): -46 ~ -56V
	Off-hook current: 25 mA
	Terminating impedance: Multiple
	150 nF complex impedance
	Frequency response: 300–3400 Hz
	Return loss (600 ohm, 300–3400 Hz) >20 dB
	Idle channel noise: <10 dB (typical)
	Longitudinal balance: 60 dB (typical)
Regulatory compliance	FCC (Part 15 Class B), CE, ICES-003, A-Tick certification, Restriction of Hazardous Substances (RoHS), and UL

Description	Specification	
Power supply	DC input voltage: 5V DC at 2.0A maximum Power consumption: 5W Switching type (100–240V) automatic Power adapter: 100–240V, 50–60 Hz (26–34 VA) AC input, 6 ft.(1.8m) cord	
Indicator lights and LED	Line, phone, Internet, and power	
Documentation	Quick Start Guide Administration Guide: Available online Provisioning Guide: Available online	
Environmental	Environmental	
Dimensions (H x W x D)	3.98 x 3.98 x 1.10 in. (101 x 101 x 28 mm)	
Unit weight	5.40 oz. (153 g)	
Operating temperature	32 to 113°F (0 to 45°C)	
Storage temperature	-77 to 158°F (-25 to 70°C)	
Operating humidity	10 to 90% noncondensing	
Storage humidity	10 to 90% noncondensing	
Package contents       Cisco SPA232D Multi-Line DECT ATA         5V/2A power adapter       5V/2A power adapter         6-ft (1.83m) Ethernet cable       RJ-11 telephone cable         Quick Start Guide       CD with documentation including license and warranty		

#### Warranty Information

Cisco SPA232D is covered by a Cisco standard 1-year limited hardware warranty with return to factory replacement and 90-day limited software warranty. To download software updates, please visit: <a href="http://www.cisco.com/go/smallbiz">www.cisco.com/go/smallbiz</a>.

#### Ordering Information

#### **Table 3.**Ordering Information

Product Name	Part Number
Cisco SPA232D Multi-line DECT ATA	SPA232D-G1 (NA)
	SPA232D-G7 (EU, APAC, ANZ)

#### **Cisco Services**

The Cisco Small Business Support Service provides "peace of mind" coverage at an affordable price and helps you get the most value from your Cisco Small Business solution. This device-level, subscription-based service includes software upgrades and updates, extended access to the Cisco Small Business Support Center, and next-business-day hardware replacement as necessary. It also provides community-based support to enable small businesses to share knowledge and collaborate using online forums and wikis to help boost business efficiency, identify and reduce risks, and serve customers better.

#### For More Information

For more information about Cisco Small Business solutions, visit <u>www.cisco.com/go/smallbusiness</u>. For more information about the Cisco SPA232D and other voice gateways and analog telephone adapters (ATAs), visit <u>www.cisco.com/go/gateways</u> or contact your local Cisco account representative.



Americas Headquarters Cisco Systems, Inc. San Jose, CA Asia Pacific Headquarters Cisco Systems (USA) Pte. Ltd. Singapore Europe Headquarters Cisco Systems International BV Amsterdam, The Netherlands

Cisco has more than 200 offices worldwide. Addresses, phone numbers, and fax numbers are listed on the Cisco Website at www.cisco.com/go/offices.

Cisco and the Cisco logo are trademarks or registered trademarks of Cisco and/or its affiliates in the U.S. and other countries. To view a list of Cisco trademarks, go to this URL: www.cisco.com/go/trademarks. Third party trademarks mentioned are the property of their respective owners. The use of the word partner does not imply a partnership relationship between Cisco and any other company. (1110R)

Printed in USA