


Specifications

HX4 VoIP IAD Series

Item	Features
Port configurations	HX411: 1 FXS and 1 FXO ports, 2 Ethernet ports
	HX422: 2 FXS and 2 FXO ports, 2 Ethernet ports
	HX402: 2 FXS port, 2 Ethernet ports
Case	
	Desktop and wall mountable
	Size: 150 x 109 x 30mm
	Weight: 300g
	LED: power, Ethernet, FXS, FXO, device status
Connector type	Ethernet: RJ45
	FXS: RJ11
	FXO: RJ11
Hardware	ATMEL AT91SAM9G20B with 400MHz
	32MB SDRAM
	4MB Flash memory
Powe supply	AC/DC Adaptor with input: 100-240V AC and output: 9V/0.66A DC
	Power consumption: 6W
Operating system	Linux kernel 2.6.27

Item	Features
SIP protocol RFC3261 RFC3262 RFC3264 RFC3311 RFC3515 RFC3581 RFC 3966 RFC4028	SIP registration (per trunk, per gateway)
	Registration expire setting
	SIP trunk
	Backup SIP proxy (up to 10 proxies)
	Peer-to-peer communication
	SIP-to-SIP relay
	Hook flash relay (INFO)
FXS	Polarity reversal generation
	Caller ID generation (Bellcore and ETSI FSK, DTMF, before/after ring)
	Ring cadence setting
	Ring frequency setting
	Volume control
	Hook flash timing setting
	Message waiting indicator (FSK, polarity inverse)
FXO	Hot line
	Relay to FXS extension
	Redirect to SIP server
	Interactive Voice Response (IVR) and second-stage dialing
	Outbound calls from either FXS or service network
	Battery reversal detection
	Caller ID detection (Bellcore and ETSI FSK, DTMF, before/after ring)
	Busy tone detection
	DTMF out-pulsing timing setting
	Volume control
	Ring parameter setting
	Busyout when FXS is not available

Item	Features
Codec/FAX/RTP	G.711ALaw, G.711Ulaw, G.729 (Note1) and G.729B (Note1)
	T.30 fax transparent, T.38 fax relay (Note2)
	Echo cancellation
	Dynamic jitter buffer management
	Static jitter buffer
	DTMF relay (RFC2833, SIP/INFO, inband)
	Adjustable packetization period, 10/20/30/40/50/60ms
	Call progress tones compliant with multi-nation standards
Voice QoS	IEEE 802.1p tag
	DiffServ code point (TOS) bits
Call control	Blind transfer
	Explicit transfer
	Call forward on busy
	Call forward on no answer
	Call forward variable
	Call waiting
	Caller ID
	Caller ID blocking
	Caller ID on call waiting
	Distinctive ring
	Do not disturb
	Music on hold
	Color ring back tone
	Call progress tone (configurable)
	Release control (called party control, calling party control)
	Built-in 3-way calling

Item	Features
	Speed dialing
	Calling and called number based routing
	Call number modification (add, delete, and replace)
	Hunt group (sequential and circular selection)
	Ring group
	Digit map
	PSTN failover (upon IP network break or failure to reach SIP proxy, or power break)
Networking	DHCP client
	DNS/DDNS client
	PPPoE client
	NAT traversal (STUN)
Security	IP filtering list (IP table)
	SIP/RTP/Telnet/HTTP/TFTP port screening
	Web-utility access privilege (admin and user)
System management	TR069-based management (TR069, TR104, and TR106) (Note1)
	Standard SNMP agent, MIB v2 (Note1)
	Web-utility for configuration, data import/export, and firmware upgrade
	Auto-provisioning for configuration and firmware upgrade
	Log management (8 levels)
	Syslog
	Debugging and call trace
	TCP dump
	System status monitoring and statistics (TR069, SNMP, Web)

Note1: The feature will be released in Q2 of 2011.

Note2: The feature will be released in Q3 of 2011.