

FAG3108

FIBERME FAG3108 Analog Gateway converts analog PSTN messages into a format suitable for transmission over standard IP networks. The device comes with 8 FXS ports, Ideally suited for small and medium businesses, Call centers, and office businesses. FAG3108 supports the standard SIP protocol and it's compatible with leading SIP- based IP telephony systems.

- · Cost effective gateway with 8 FXS ports
- Fax over IP (T.38 and Pass-Through)
- Support IPv4 and IPv6
- TR069 and SNMP

- Multiple codecs: G.711A/U,G.723.1,G.729A/B,AMR, etc.
- Fully compatible with leading IMS/NGN, SIP based IP telephony system

Fit to Diverse Applications

Used to connect IP telephones to a legacy PBX, integrate network-hosted applications with the PBX, extend the PBX to branch offices, and integrate various voice and call processing capabilities in an enterprise LAN or WAN environment.

Flexible Call Routing

Calls from the switched network to a VoIP destination on the IP network or conversely, and support the following call routing options: TDM to IP or IP to TDM, with IP load balancing, IP fault tolerance

Controlled Migration to IP

Compatible with general FXS lines, and a variety of popular PBX manufacturers and protects investment in legacy telecommunications equipment.

User-Friendly Management

Supports configuration via serial, Telnet, and a web browser including context-sensitive help.

Easy to install, configure, debug, and maintain.

Enhanced Voice/Fax Processing

Supports a variety of compression algorithms, including G.711 A-law and µ-law, G.729A/B; Transcode fax from T.30 fax protocol (supporting V.17) to T.38 for transmission over a packet network.

Web Server Interface & Hot Swap

Each gateway unit is delivered with a web server interface, allowing configuration and software upgrades via a web browser; Allows gateway units to be added or removed without affecting other gateway.



Capacity	8 FXS Ports
Ethernet Interfaces	2x LAN Ports, RJ45
Console Port	1x RS232, 115200bps
Voice and Fax	 G.711A/U law, G.723.1, G.729 A/B, AMR,iLBC Silence Suppression Comfort Noise Generation(CNG) Voice Activity Detection(VAD) Echo Cancellation(G.168), with up to 128ms Adaptive (Dynamic) Jitter Buffer Hook Flash Programmable Gain Control T.38/Pass-through High-speed fax up to 14.4kbps Modem/POS DTMF mode: Signal/RFC2833/INBAND VLAN 802.1P/802.1Q (data/voice/management VLANs) Layer3 QoS and DiffServ
FXS Connector	 RJ11 Dial Mode: DTMF and Pulse Caller ID: DTMF/FSK CLI Presentation Max Cable Length: 5 km Reversed Polarity Programmable Call Progress Tone
Software Features	 Hunting Group Web ACL PPPoE/IPv4/IPv6 Digitmap Routing Rules based Prefixes Caller/Called Number Manipulation
Maintenance	 SNMP v1/v2/v3 TR069 Auto Provisioning Web/Telnet Configuration Backup/Restore Firmware Upgrade via Web CDR Syslog(Emerg,alert, critical,error warning,notice,info, debug) Ping/Tracert Test Network Capture NTP/Daylight Saving Time
VoIP Protocols	 SIP v2.0 (UDP/TCP),RFC3261 SDP,RTP(RFC2833), RFC3262, 3263,3264,3265,3515,2976,3311 SIP TLS/SRTP RTP/RTCP, RFC2198, 1889 RFC4028 Session Timer DNS SRV/ A Query SIP Trunk Early Media/Early Answer NAT:STUN, Static/Dynamic NAT
Power Supply	12V – 1.0A
Storage Temperature	20 °C ~85 °C
Humidity	10%-90% Non-Condensing
FC CE Rolls	FIBERME Communications LLC 1749 Old Meadow Rd., Suite 401 McLean, VA 22102, USA

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