

FAG3108

**FXS Gateway** 



# 8 FXS Gateway



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







