# Vega 3050G Analog Gateway



Take advantage of the newest Voice over Internet (VoIP) technology and achieve wide-ranging benefits, such as reduced telephone charges and transition costs, by seamlessly integrating your existing equipment with a Vega 3050G High Density Analog Gateway from Sangoma.

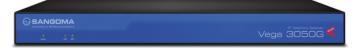
# The High Density 50 Port Vega 3050G Analog Gateway

The Vega 3050G is designed to simplify the integration of legacy phone systems to a VoIP network. The Vega 3050G connects a large quantity of analog handsets and devices to an IP network, typically either an Internet Telephony Service Provider (ITSP) or a corporate network.

"Perfect solution for multi-dwelling units, hotels, and hospitals to seamlessly integrate a large density of analog handsets to an IP core, eliminating the cost of re-wiring infrastructure to migrate to IP phones."

# Quick Facts

- 50 FXS Ports
- » Voice, FAX and Modem Support
- EXCLUSIVE Local Survivability
- Support for 2 FXO Ports using Expansion Module for Emergency PSTN Backup
- >> Flexible Call Routing for Fallback & Least Cost Routing
- Interoperability with a Wide Range of Legacy and IP Equipment
- Optional Annual Support & Software Maintenance Plans



**Front** 



Back

# **ADVANCED CAPABILITIES**

# **Local Survivability**

This option enables continuity of service during WAN/SIP outages and may be configured to operate in a number of ways including:

- Standalone proxy
- » IP device survivability
- » IP device call routing
- » SIP to SIP call routing
- » Emergency call routing

# **Open, Non-Proprietary Interfaces**

#### The Vega 3050G Supports SIP and T.38 Fax

This gateway can be configured for different country requirements, such as tones and line impedance.

All Vega gateways have proven interoperability with a wide range of existing telecommunications and VoIP equipment.

# **Lifeline PSTN Backup**

When fitted with the 2 FXO port expansion module, under power failure conditions they provide a hard-wired bypass to the FXS ports allowing PSTN calls.

### **USE CASES**

# **Service Provider Applications**

- Connect analog phones to hosted telephony platforms
- » Low-density PSTN gateway
- » Survivability for IP phones in case of wide area network (WAN) disruptions

# **Enterprise Applications**

# **Enterprise VolP Networking**

- » Enterprise IP telephony gateway
- Compatibility of legacy phones with IP-PBX and UC (Unified Communications) platforms

# **Density**

With up to 50 FXS ports, and 2 FXO ports (with optional expansion board) in a chassis that is only 1U-high and half a rack deep, these gateways provide exceptional density.

#### **TECHNICAL SPECIFICATIONS**

#### **Interfaces**

# **VoIP Interface:**

- » SIP
- Fax support up to G3 FAX, using T.38
- Modem support up to V.90, using G.711
- » Up to 50 VoIP channels
- » Audio Codecs:
  - > G.711 (a-law/µ-law)
  - > G.729a (8kbps)
  - > G.723.1 (5.3/6.4 kbps)
  - > G.726
  - > T.38

# **Telephony Interface:**

- » 2x 25 FXS ports on an RJ-21 connector
- » 2x RJ-11 FXO ports on optional FXO expansion board
- » 600R, 900R or CTR-21 line impedance

#### **LAN Interface:**

- » 1x RJ-45 Gigabit Ethernet
- » 1000BaseT / 100BaseTx / 10BaseT, full / half duplex

#### **USB Interface:**

» 1x USB 2.0 for extra storage

#### **Features**

#### **Telephony Features:**

- » Call waiting
- » Call forward unconditional, busy, no-answer
- » Call transfer blind, consultative
- >> 3-way conference
- » Do not disturb
- » Message waiting indicator audible, visual
- » Music on hold
- » Executive barge
- » Caller ID presentation UK, DTMF, Bellcore GR30, ETSI
- » Caller ID screening
- » SIP registration & digest authentication

## **Operations, Maintenance & Billing:**

- » HTTP(S) web server
- » RADIUS accounting & login
- » Remote firmware upgrade
- » Auto configuration upgrade
- » SNMP V1, V2 & V3
- » TR-069
- » Syslog
- >> TFTP/FTP support
- » VT100 RS232/Telnet/SSH
- » Voice readback of IP parameters

# **Routing & Numbering:**

- » Direct Dialing In (DDI)
- » SIP registration to multiple proxies
- » Dial planner sophisticated call routing capabilities, standalone or gatekeeper/proxy integration

#### **Call Quality:**

- » Adaptive jitter removal
- » Comfort noise generation
- » Silence suppression
- » 802.1p/Q VLAN tagging
- » Differentiated Services (DiffServ)
- > Type of Service (ToS)
- » QoS statistics reporting
- » Echo cancellation (G.168 up to 128ms tail)

# Security & Encryption: \*Optional

- \*Media SRTP/ \*SIP TLS
- » Management HTTPS, SSH Telnet
- » Configurable user login passwords
- » Local Survivability

# **Hardware**

#### **Certification:**

- >> EMC (CLASS B)
  - > EN55022
  - > EN55024
  - > FCC Part 15
  - > AS/NZS3548
- » Safety
  - > EN60950
  - > IEC60950
  - > UL60950
  - > AS/NZS60950

#### **Environmental:**

- » EMC (CLASS B)
  - > 0° .. 40°C
  - > 0% .. 90% humidity (non-condensing)

#### **FXS Line Length:**

» 8 km at 1 REN, depending on environment

# **LED Indicators:**

- » System: Power/System Ready/Activity
- » LAN: Speed/Activity

#### **Dimensions:**

- > 1U: 440mm (W) x 250mm (D) x 45mm (H)
- » Weight: 2.7kgs (6lbs)
- » Rackmount brackets supplied: 483mm

# **Power Supply:**

» Internal PSU 100..240 VAC, 47..63 Hz, 1..0.5 A

