GSM Gateway ETS-8x8G Data Sheet



1.1 Overview

GSM VoIP gateway ETS8x8G is newly designed IP to GSM gateway supporting maximum 8 ports 64SIMs of GSM Voice interface, it can effectively realize the smooth transition between PLMN(GSM) and VoIP network. Compact cost effective design and system architecture of GSM Gateway ETS-8x8G provides customer satisfaction in high quality , performance and system reliance.

Mostly important, GSM Gateway ETS-8x8 features with new functions such as Multi-SIM rotation, Human behavior, BTS rotation, proxy server encryption for anti IP block, ETS bandwidth optimization (SBO), Auto IMEI change and generation, auto activation SIM card etc.

This product uses the state-of-art technology voice compression and Smart QoS of ETS to maintain the maximum voice quality under fast internet line and slow internet line as well, thus It is an ideal gateway for heavy duty VoIP call termination (VoIP to GSM) and Origination (GSM to VoIP), it is fully compatible with leading soft switch and SIP server.

1.2 Main features

Support 8 GSM ports, up to 8 concurrent calls (1 Ports 8 SIM card, total 64 SIM card)

Support GSM: Quad-band 850/900/1800/1900Mhz

CDMA: 450/ 800/ 1900Mhz optional 3G/UMTS: 850/900/2100Mhz optional

Support Multi SIM card rotation to avoid SIM block

Support Human behavior function

Support SBO (Switch Bandwidth Optimization)

Support BTS rotation and lock

Support encryption for VOS for anti-block of IP port

Support SMS Sending / batch SMS Sending / receiving

Support USSD balance inquiry

Auto activation SIM card and recharge

Support IMEI change, auto IMEI change and generation

Automatically lock/open SIM card/ port according to its balance or alarm

Support Codec: G.711a/u law, G.723.1, G726,G.729AB

Support bandwidth optimization (optional)

Built-in WiFi router, can directly connect to Modem

Built-in 4G module (optional), can use through 4G network if 4G module inserted

If with 4G module and mobile battery /power supply, then it can do call termination with mobility

Sys log output by USB interface for tracking records

User friendly web management interface

HTTP Web support for configuration and upgrade

SIM swapping

HTTP Web support ASR, ACD, PDD, SIM balance inquiry

Convert the number as preset rules /Number translation

Call routing / digit map

CDR management

IVR customized

BCCH management

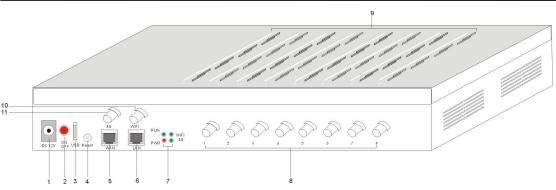
SIM card rotation conditions:

- 1) According to accumulated call duration check (talk time)
- 2) According to accumulated idle&talking check(use time)
- 3) According to accumulated calls check (call counter)
- 4) According to consecutive call failure check(call failure)
- 5) According to consecutive low-duration calls check (Low duration)

Human behavior conditions:

- 1) According to accumulated call duration check (talk time)
- 2) According to accumulated idle&talking check(use time)
- 3) According to accumulated calls check (call counter)

1.3 Interface instruction



S/N	
1	Power port DC 12V
2	Switch ON/OFF
3	USB PORT
4	Reset key
5	WAN Port
6	LAN port
7	LED Indicator for POWER, RUN,WiFi,4G
8	Antenna port (SMA) (8 pieces
9	SIM card slots (every port 8 ,total 64)
10	WiFi Antenna port (SMA)
11	4G Antenna port (SMA) optional

1.4 Specifications

Interfaces

Mobile Ports: 8 ports GSM/CDMA channels

Two(2) 10/100Mbps Ethernet Interface (2xRJ45)

USB: 1 port

SIM Card interface: 64 pieces + 4G SIM card 1 piece (Optional)

4G interface: 1 channel

Wifi port: 1Voice Processing

• Voice Codec: G.711a/u law, G.723.1, G726,G.729AB

QoS: Diff Serve, T oS, VAD, PLC, CNG

Call Features

Calling Type: Terminate/Originate calls

● IVR Voice Prompt: Two stage dialing, Customized IVR

Call handling: One stage dialing, Configurable dial plan, digit map

● DTMF: RFC2833, SIP

Mobile Features

• General Feature: SMS, USSD, IMEI/PIN modify, Call minutes restriction, Carrier select, BCCH, Reversal Polarity Network

Network Mode: NAT router or switch mode

• Network Protocols: IP,TCP, UDP,TFTP, FTP, RTP, RTCP, ARP, RARP,ICMP Ping, NTP, SNTP, HTTP, DNS, PPPoE, DHCP

NAT traversal: Static NAT, STUN

Protocol

Protocol: SIP V2.0 RFC3261

SIP Characteristics: By port/device registration, Two183 mode

Configuration Management

Management: TFTP, HTTP, Sys log, CDR

• Web GUI: Configuration, firmware upgrade, call status, CDR, Configuration backup/restore

Hardware Specifications

Power Supply:AC100~240V 50/60HZ DC12V/2A External

• Humidity: $5\% \sim 90\%$ RH, Power Consumption: 24W

Product Appearance: steel cabinet

Product Dimensions: 37(W) x 17.2(D) x 6.5 (H) cm

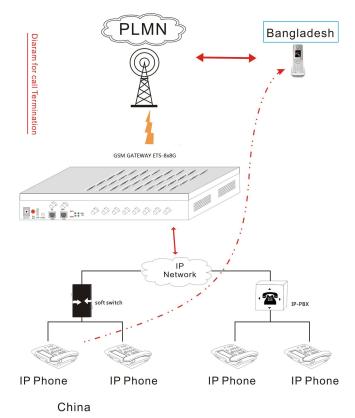
Product Net weight: 2.65kg

Carton box Dimensions: 41(W) x 25.5(D) x 15(H) mm

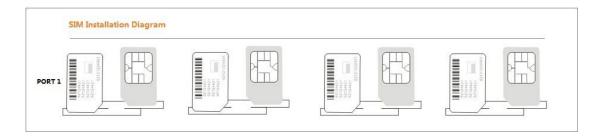
Gross weight: 3.5kg

1.5 Application

International VoIP call termination



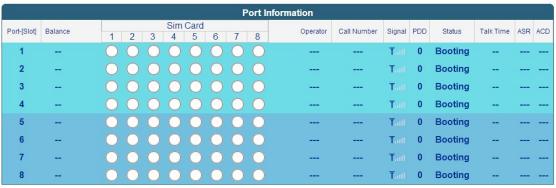
1.6 SIM Card Installation Diagram



1.7 Call traffic examples









1.8 Photos

