

## AddPac Multi-SIM GSM Gateway AP-GS3800



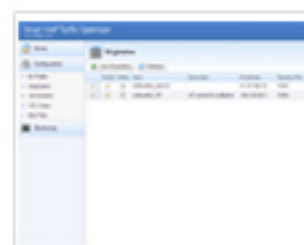
VoiceFinder AP-GS3800 Multi-SIM GSM gateway is a cutting edge IP to GSM gateway supporting maximum 44 ports of GSM Voice interface. Compact cost effective design and system architecture of AP-GS3800 provides customer satisfaction in high quality, performance and system reliance. This product uses the state-of-art technology voice compressed algorithm and unique QoS algorithm of AddPac to maintain the maximum voice quality under fast internet line and slow internet line as well. AP-GS3800 Multi-SIM GSM Gateway supports various Multi-SIM GSM VoIP Modules like as AP-N1-GSM4S8, AP-N1-GSM4S4, etc. AP-N1-GSM4S8 GSM VoIP Module is a new 4-Port GSM Module (8 SIM card slots per GSM Port) for GSM Call Termination Service. This GSM VoIP Module supports Quad-Band (850/900/1800/1900MHz) GSM frequency, Eight(8) SIM card slots per GSM port, Hot-Swap Switch Button, One(1) GSM Antenna Interface. This GSM module has thirty-two(32) GSM SIM card slots internally. Also this module supports the Voice DSP for GSM call termination service based on VoIP Technology. AP-N1-GSM4S4 GSM VoIP Module is a new 4-Port GSM Module

(4 SIM card slots per a GSM Port) for GSM Call Termination Service. This GSM VoIP Module supports Quad-Band (850/900/1800/1900MHz) GSM frequency, Four(4) SIM card slots per a GSM port, Hot-Swap Switch Button, one(1) GSM Antenna Interface. This GSM module has sixteen(16 = 4x4) GSM SIM card slots internally. Also this module supports the Voice DSP for GSM call termination service based on VoIP Technology. AddPac's various GSM, VoIP Gateway series and multimedia network devices have been fully recognized in terms of its performances and stability throughout the world. AP-GS3800, with our accumulated experiences and know-how in the enterprise and service provider markets, will provide full satisfactions for customers who ask for a next-generation GSM based call termination device. AddPac VTO service is a high performance VoIP Traffic Optimization service which can reduce bandwidth usage in VOIP call termination. This means that call termination service provider can deliver improved VoIP calling service with much lower VoIP traffic cost. Also, AddPac VTO service supports the VoIP anti-blocking service features and can help to provide mobile VoIP termination service in locations where VoIP service is completely blocked. In addition to basic VoIP Traffic Optimization service features, AddPac VTO service supports real-time VoIP traffic monitoring service such as VoIP packet loss rate, up/down link status, etc in between VTO server and VTO client site each. This can help to provide real-time VTO technical support service AddPac VTO service is very easy to use because AddPac GSM VoIP gateway provides the VTO client service features internally besides GSM VoIP gateway service features. This means there is no need an external Linux server for VTO service and complex installation procedure. User just does new firmware upgrade for AddPac GSM gateway VTO client package. GSM VoIP gateway service concept supporting VTO client service is a new architecture and design concept. This concept reduces the hop count and enhances the VoIP QoS like as jitter, delay. As a result, this architecture increases the ASR, ACD that are most important factors in call termination service area.

## Major Feature

### Overall Features

- AddPac VTO(VoIP Traffic Optimizer) Service Client : VTO Plug & Play Service, Anti-VOIP, Bandwidth Compression, VoIP Traffic Monitoring (Packet loss, round-trip time), etc
- Renew the Main Window
- Mobile Statistics Display using Graph Chart (ASR, ACD, etc)
- Several Enhancements for ASR Improvement (zombie call, etc)
- Human Behavior Call Pattern Modeling Algorithm for Anti-SIM Block
- IMEI change, BTS Selection, etc
- High-performance VoIP Modular Architecture
- Supports maximum 44- port GSM interface
- Flash Memory: 512MB, DDR2 128MB
- GSM voice interface module : 4-Port GSM Module
- Two(2) port 10/100Mbps fast Ethernet interface
- SIP, H.323 dual VoIP signaling stack
- Support Voice Processing Features
- VAD, DTMF,CNG,G.168, and T.38 G3 Fax Relay
- G.723.1,G.729A,G.711 Voice Compressions
- High-performance IP-Routing Capability with Reliability
- Static and IEEE 802.1Q VLAN Routing Protocols including VRRP
- Traffic Queuing, and SNMP MIB v2 for Network Management Features
- Standard & Extended Access List for Security Functions
- Essential Scalability Features such as DHCP Server & Relay, NAT/PAT, IEEE Transparent Bridging, IP Accounting, and Debugging/Diagnostics, etc.
- DNSProxy Support
- UPnP Support
- MAC Address Filter Service



## GSM SIM Server Service Diagram

Model	AP-GS916	AP-GS3000	AP-GS3800	AP-GS3300	AP-GS5000
Available Modules	AP-N1-GSM4 AP-N1-FXS8 AP-N1-FXO8 AP-N1-FXS4O4	AP-N1-GSM4 AP-N1-FXS8 AP-N1-FXO8 AP-N1-FXS4O4 AP-N1-E1	AP-N1-GSM4 AP-N1-FXS8 AP-N1-FXO8 AP-N1-FXS4O4	AP-N1-E1	AP-N1-GSM8 AP-N1-FXS8 AP-N1-FXO8 AP-N1-E1 AP-N1-2E1
Multi-SIM GSM Modules	AP-N1-GSM4S4 AP-N1-GSM4S8	AP-N1-GSM4S4 AP-N1-GSM4S8	AP-N1-GSM4S4 AP-N1-GSM4S8	AP-N1-GSM4S16	AP-N1-GSM8S4
GSM Channel	Up to 16 Ch.	Up to 36 Ch.	Up to 44 Ch.	Up to 36 Ch.	Up to 80 Channel
GSM Antenna	One(1) / 4 Channel GSM Module (AP-N1-GSM4)	One(1) / 4 Channel GSM Module (AP-N1-GSM4)	One(1) / 4 Channel GSM Module (AP-N1-GSM4)	One(1) / 4 Channel GSM Module (AP-N1-GSM4S16)	Two(2) / 8 Channel GSM Module(AP-N1-GSM8)
Module Slot	Four(4) Module Slots for GSM	Nine(9) Module Slots for GSM	Eleven(11) Module Slots for GSM	Nine(9) Module Slots for GSM	Ten(10) Module Slots for GSM, E1/T1 Module Slot
LAN Port	2	2	2	2	2
Console	1	1	1	1	1
Power	Single PSU	Single PSU	Single PSU	Single PSU(Module)	Dual PSU (module)

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