

AddPac GSM VoIP Gateway AP-GS3300



VoiceFinder AP-GS3300 is a new cutting edge Digital or IP to Very High Density Multi-SIM GSM gateway supporting maximum 36 ports of GSM Voice interface. GSM and digital interface of AP-GS3300 provide an optimized call scenario when it interoperates with softswitch or conventional PBX. Compact cost effective design and system architecture of AP-GS3300 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. AddPac AP-GS3300 is a device that can support both GSM gateway service (Digital E1/T1->GSM or Internet->GSM) and VoIP gateway service (Digital E1/T1?Internet) simultaneously. It also supports SIP, H.323 Multiple VoIP signaling protocol, various voice codec support(G.711, G.726, G.729, G723.1), nine(9) module slots for GSM and digital VoIP module, one(1) module slot for CPU board, two(2) gigabit ethernet ports, 1-port RS232C console, and state-of-art technologies and services.AP-N1-GSM4S16 GSM VoIP

Module for AP-GS3300 is a new 4-Port GSM Module (16 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, Sixteen(16) SIM card slots per a GSM port, Hot-Swap Switch Button, One(1) GSM Antenna Interface. This GSM module has sixty-four(64) GSM SIM card slots internally. Also this module supports the Voice DSP for GSM call termination service based on VoIP Technology. 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

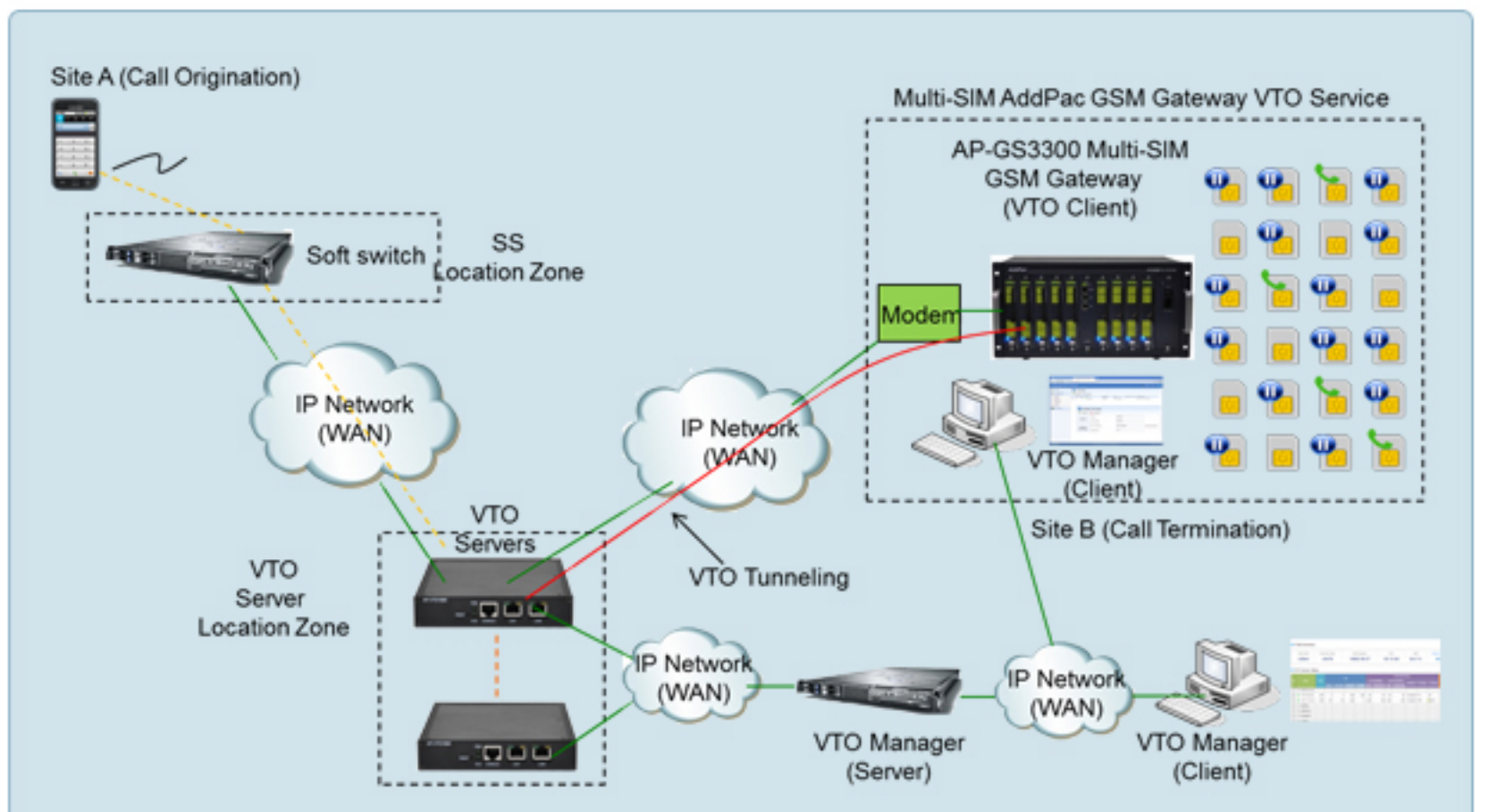
- > AddPac VTO(VoIP Traffic Optimzer) Service Client : VTO Plug & Play Service, Anti-VOIP, Bandwidth Compression, VoIP Traffic Monitoring (Packet loss, round-trip time), etc
- > Renew the Main Window
- > Mobile Statistics Display using Graph Chart (ASR, ACD, etc)
- > Several Enhancements for ASR Improvement (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 36- port GSM interface
- > Flash Memory: 512MB, DDR2 1Gbyte
- > GSM voice interface module : 4-Port GSM Module
- > Digital voice interface module: Digital E1/T1 Voice Module
- > Two(2) port 10/100/1000Mbps Gigabit Ethernet interface
- > SIP, H.323 concurrent dual 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



Multi-SIM GSM Module Comparison Table

Model	AP-N1-GSM2S4	AP-N1-GSM4S4	AP-N1-GSM4S8	AP-N1-GSM4S16	AP-N1-GSM8S4
Available Target	AP-GS3800 AP-GS3000 AP-GS2800 AP-GS2700 AP-GS2500 AP-GS916	AP-GS3800 AP-GS3000 AP-GS2800 AP-GS2700 AP-GS2500 AP-GS916	AP-GS3800 AP-GS3000 AP-GS2800 AP-GS2700 AP-GS2500 AP-GS916	AP-GS3300	AP-GS5000
GSM Channel	2-Port	4-Port	4-Port	4-Port	8-Port
SIM Card Slots	4 SIM Card Slots/a GSM Port Total = 4 x 2 = 8	4 SIM Card Slots/a GSM Port Total = 4 x 4 = 16	8 SIM Card Slots/a GSM Port Total = 8 x 4 = 32	16 SIM Card Slots/a GSM Port Total = 16 x 4 = 64	4SIM Card Slots/a GSM Port Total = 4 x 8 = 32
External Antenna	2	1 (Internal Antenna Combiner)	1 (Internal Antenna Combiner)	1 (Internal Antenna Combiner)	2 (Internal Antenna Combiner)
Hot-Swap	Support	Support	Support	Support	Support
VoIP Codec	G.711, G.729, G.723.1,etc	G.711, G.729, G.723.1,etc	G.711, G.729, G.723.1,etc	G.711, G.729, G.723.1,etc	G.711, G.729, G.723.1,etc

Network Diagram



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