

AP1100FN

4-Port FXS 4-Port FXO VoIP Gateway

High Performance VoIP Gateway Solution









AddPac Technology

Sales and Marketing

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Product Overview

- H.323/SIP Dual Concurrent Stack Embedded
- High Performance RISC & Programmable DSP Architecture
- One(1) 10/100Mbps Fast Ethernet, One(1) 10Mbps Ethernet
- High Performance LAN-to-LAN Routing Capability
- G.711/G.726/G.723/G.729, T.38 Fax, VAD, etc
- Powerful Network Protocols (PPPoE, DHCP, Static Routing, etc)
- Firmware Upgradeable Architecture
- Light SIP Proxy Support (Mini Call Manager Service)
- SNMS (VoIP Plug&Play Management System) for Large Scale Deployment
- Advanced Voice QoS Mechanism
- Small, Light and Compact Design
- Power Switch for Stability and Status LEDs
- 4-Port FXS 4-Port FXO Interface



Product Highlights

AP1100FN 4-Port FXS 4-Port FXO VoIP Gateway

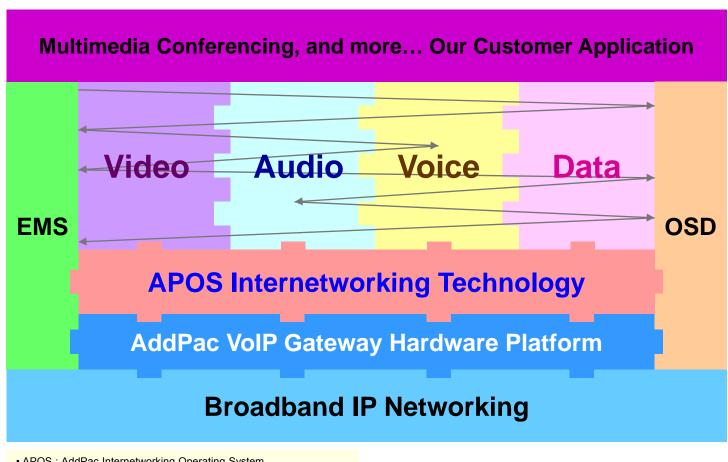
State-of-art Signaling H.323, SIP **Concurrent Dual Stack SNMS Large Scale VolP** Small, Light and **Gateway Management Support Compact Design High Performance RISC CPU Excellent Voice Quality** + Programmable DSP G.729/G.723/G.726/G.711 Voicefinder **Broadband IP Networking** Web based With 10/100Mbps **Fast Ethernet** Smart Easy Setup™ **APOS**TM**Technology High Performance Firmware Upgradeable Architecture LAN-to-LAN Packet Routing Processing Powerful Network Protocol Support** 4-Port FXS, 4-Port FXO



Advanced Voice
Traffic QoS Mechanism
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Analog Interface

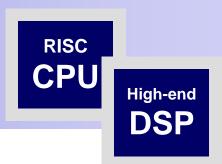
APOS Technology



- APOS : AddPac Internetworking Operating System
- · OSD : On- Screen Display
- · EMS: Element Management System



Hardware Specification



- RISC Microprocessor Computing Power
 - + High End Programmable DSP
- VoIP Analog Interface
 - Quad(4) FXS Port
 - Quad(4) FXO Port
- Network Interface
 - One(1) 10/100Mbps Fast Ethernet (RJ45)
 - One(1) 10Mbps Ethernet (RJ45)
- Power Supply
 - External Power Adaptor (5V, 5A)
- Power On/Off Switch
- Power LED, LAN LED, FXS Port LED



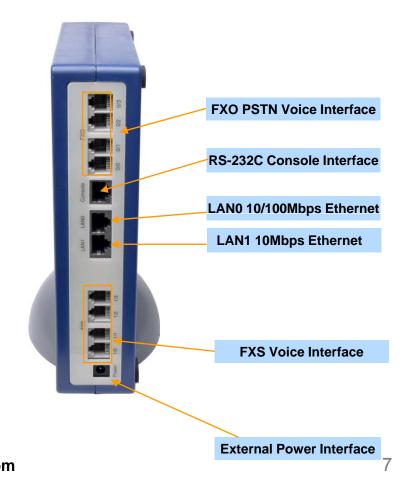
Hardware Specification

AP1100FN 4-Port FXS 4-Port FXO VoIP Gateway

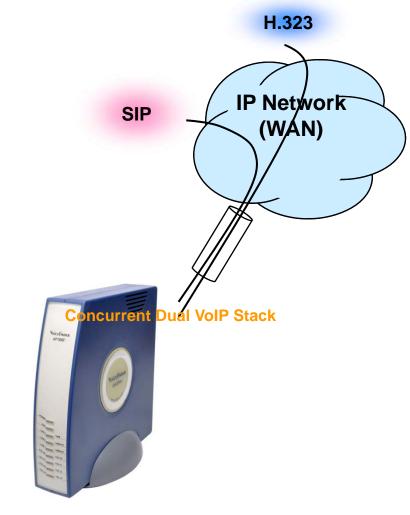
Hardware Specifications

AP1100F VoIP Series Model A Type	Basic Specifications
CPU	32Bit RISC Microprocessor
Analog Voice Interface	4-Port FXS Voice Interface (4xRJ11) + 4-Port FXO Voice Interface (4xRJ11)
Ethernet Interface	1-Port 10/100Mbps Fast Ethernet Interface(1xRJ45) 1-Port 10Mbps Ethernet Interface(1xRJ45)
Flash Memory	8Mbyte High-speed Flash Memory
Base Memory	64Mbyte High-speed SDRAM
Power Requirement	External Power Supply Adaptor / VAC 110~220V, 50/60Hz, 25Watt
Operating Temperature	0°C ~ 45°C (32 °F ~ 122°F)
Storage Temperature	-40°C ~ 85°C (-40°C ~ 185°F)
Relative Humidity	5% ~ 95% (Non-condensing)
Dimensions	50mm x 210mm x 210mm (H x W x D)

Network interface Configurations



- H.323, SIP Dual Stack
- H.323
 - ITU-T Standard H.323 v3 Support
 - Support H.245 Tunneling
 - Including H.235 Security Features
- SIP
 - IETF RFC3261 or RFC2543 SIP Standard









AP1100FN 4-Port FXS 4-Port FXO VoIP Gateway

H.323

- Fast connect, normal connect support
- H.245 tunneling support
- Q.931 response message setting for inbound VoIP calls
- H.245 logical channel open timing selection function
- Start H.245 procedure support
- DTMF / Hook flash relay with H.245 alphanumeric / signal
- Secondary gatekeeper support
- Gatekeeper assignment according to the domain name
- Gatekeeper discovery with multicast
- Lightweight RRQ support
- Signaling TCP port assignment
- Resource threshold setting with RAI
- H.235 clear-token, crypto-token support
- canMapAlias support
- Technical prefix (supported prefix) support
- Public IP assignment in NAT environment

• SIP

- Gateway-based / Endpoint-based registration support
- Secondary proxy-server assignment function
- SIP signaling port change function
- SIP proxy server assignment according to the domain name
- T.38 real-time fax relay support
- DTMF relay support with RFC2833 / OPTION message
- Re-INVITE support





AP1100FN 4-Port FXS 4-Port FXO VoIP Gateway

Voice Codec

- G.711 A-Law, G.711 U-Law
- G.726 r16, G.726 r32
- G.729A
- G.723.1 r63, G.723.1 r53
- VAD (Voice Activity Detection) function support
- DTMF relay support (H.323, SIP, MGCP common) based on RFC2833

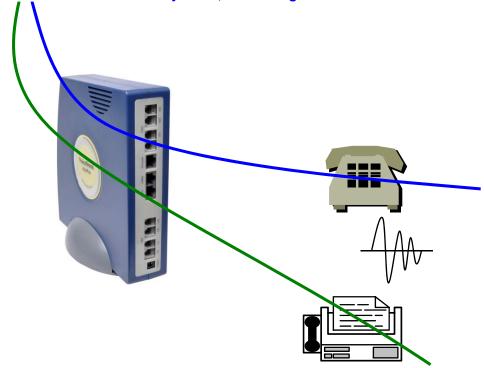
RTP

- Redundant RTP packet transmission in case of severe packet loss
- Dynamic jitter buffer management and RPT packet jitter and loss compensation with heuristic & DSP error concealment
- Static jitter buffer setting support
- Voice frame per RTP packet number control for each codec
- In-band ring-back tone support
- Virtual ring-back tone support
- Tone parameter change support

FAX

VolP

- Fax relay mode supporting T.38, inband-T.38, bypass mode
- Lost packet compensation with redundant setting in case of T.38 fax relay
- Fax relay mode, rate setting for remote end





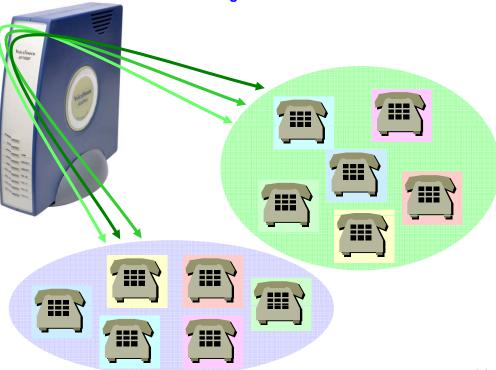
AP1100FN 4-Port FXS 4-Port FXO VoIP Gateway

VolP Call Controls

- Hot line connection function with PLAR (Private Line Auto Ring Down)
- Leased line emulation function
- Connection monitoring function
- Fault tolerant with Redundancy and Call Distribution among Gateways for load balancing
- Call attempt with IP address
- H.323, SIP, MGCP inbound call connection for each voice port
- Multiple E.164 setting for one voice port
- One E.164 or digit pattern can be assigned to more than one voice port
- Hunting with Longest match/ priority/ sequence/ random
- One stage call setup by Digit forwarding
- Call barring with specific digit patterns
- Calling and called number conversion for PSTN outbound calls
- PSTN rerouting in case of VoIP call attempt failure

VolP Call Controls (cont.)

- Call transfer for internal calls
- Call pickup for internal calls
- Calling and called number conversion for VoIP outbound calls
- Calling and called number conversion for VoIP inbound calls
- Fax broadcasting call control





Advanced QoS Features

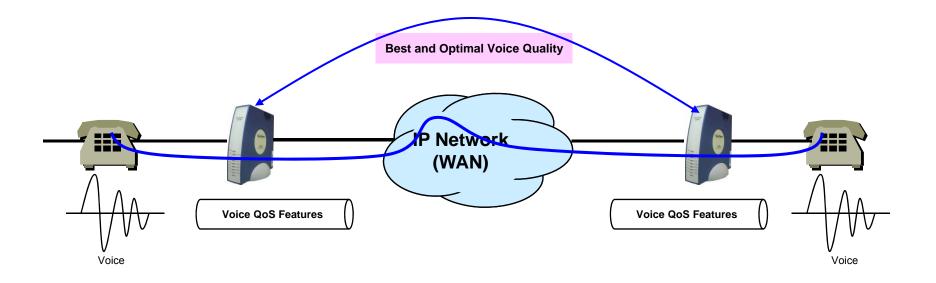
AP1100FN 4-Port FXS 4-Port FXO VoIP Gateway

Enhances Transmit Voice QoS Features

- Voice Traffic Priority Queuing
- QoS Service Profiling
- Providing Virtual Network Transmit Algorithm
- Real-time Voice Traffic QoS Support
- RTP Packet Transmit Interval Control
- Supporting RTP Packet Redundancy Scheme
- IP Header Control such as ToS, Diffserv

Enhances Receive Voice QoS Features

- Dynamic Jitter Buffer Management
- Error Concealment
- Support T.38 FAX Data Error Recovery Scheme





Network Protocols

AP1100FN 4-Port FXS 4-Port FXO VoIP Gateway

Basic Network Protocols

- ARP, IPv4, TCP, UDP, ICMP, SCTP, IGMP, MLD

Routing Protocol

- IPv4 : Static

Service Protocol

- FTP, Telnet, TFTP, DHCP Server/Relay, SNMP Server
- CDP (Cisco Discovery Protocol)
- DNS Resolver, DDNS(nsupdate)
- Bridge
- Syslog

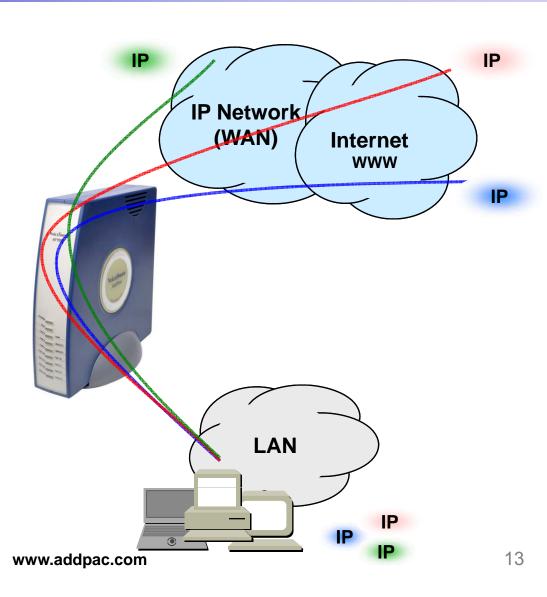
IPv4 Address Configuration

- Fixed (Static)
- DHCP
- PPPoE

Miscellaneous

- -Cisco Style CLI
- Standard & Extended IPv4 Access List
- Multi-level User Account Management
- IP accounting
- STUN Client



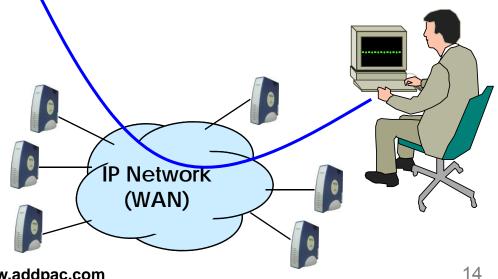


Network Management

- SNMP
 - Standard Simple Network Management Protocol(SNMP) Agent support
 - MIB v1 and v2 Support
- Web-based Management
 - Smart Easy Setup
 - **Standard Voice Interface**
 - Standard PSTN Back-up Interface
- Watch-dog Function
 - Hardware, Software watch-dog services
- Remote Management
 - Telnet
 - Rlogin
- Auto Upgrade Service
 - HTTP server based APOS image and configuration file auto-upgrade support
- Batch Job Function
 - Text based script downloading

- Interoperable with AP-VPMS Service
 - AddPac VoIP Plug & Play Management System (AP-VPMS)

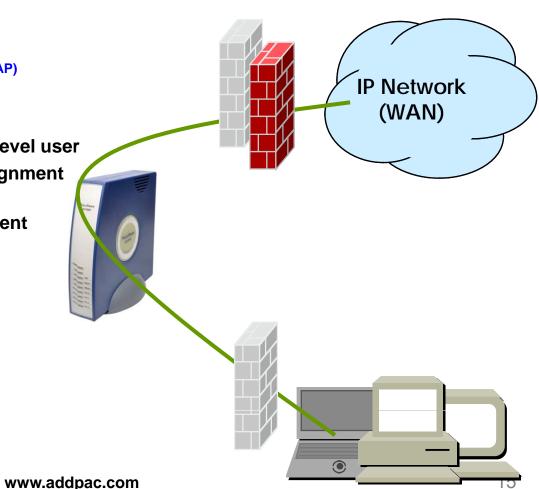






Security Management

- IP packet filtering
- IP access list
- User authentication function
 - Password Authentication Protocol (PAP)
 - Challenge Handshake Authentication Protocol (CHAP)
- Enable/Disable specific protocols
- Auto-square connect of Telnet session
- Account Management function for multi-level user
- SNMP/TELNET/FTP/HTTP/TFTP port assignment function
- SNMP/TELNET/FTP access list management
- Boot mode security checking function





SNMS (Smart Network Management System)

AP1100FN 4-Port FXS 4-Port FXO VoIP Gateway

AddPac AP-SNMS Support - AddPac Smart Network Management System - Best EMS Solutions for AddPac VoIP G/W

- Requirements Hardware Platform
 - Over 1GHz Intel Pentium 4 Machine

 - Microsoft Windows Advanced Server or Windows 2000 Server or Windows XP Professional





FXO Service Feature List

FXO Service Features	Polarity inverse detection function
	Caller-ID detection function
	Clear down tone registration and detect function
	Hook flash timing setting function
	Ring detect timeout setting function
	Ring number setting function
	Voice-confirmed connect function



Polarity Inverse Detection Function

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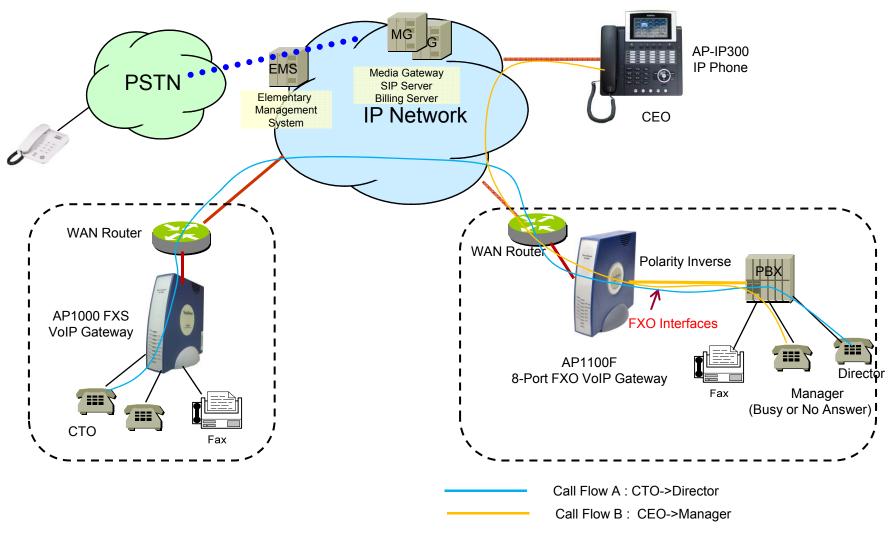
- Polarity inverse detection function
 - The FXO port detects the polarity inverse signal coming from Legacy PBX
 - When there is an incoming VoIP call via the FXO port to Legacy PBX, the gateway sends call connect message to Softswitch after detecting the polarity inverse signal on the FXO port.
 - Using Polarity Inverse Signal, a accurate billing service is available.

When polarity inverse function is enabled

- In case of A flow, Billing is start when the director hooks off.
- In case of B flow, Billing is not start because manager port is busy or no answer



Polarity Inverse Detection Function



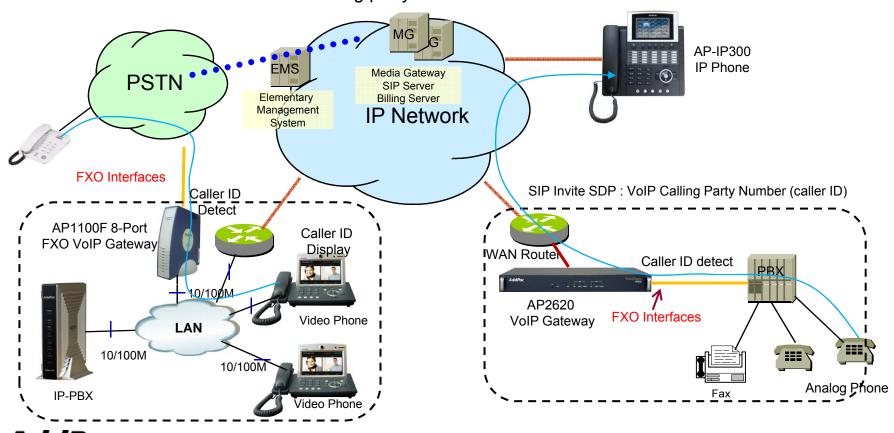


Caller ID Detect Function

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Caller-ID detection function

- The FXO port is connected to PSTN or PBX, and is able to detect Caller-ID.
- When a VoIP call is originated from the FXO port, the FXO port detects the caller-ID and uses the number as the VoIP calling party number.



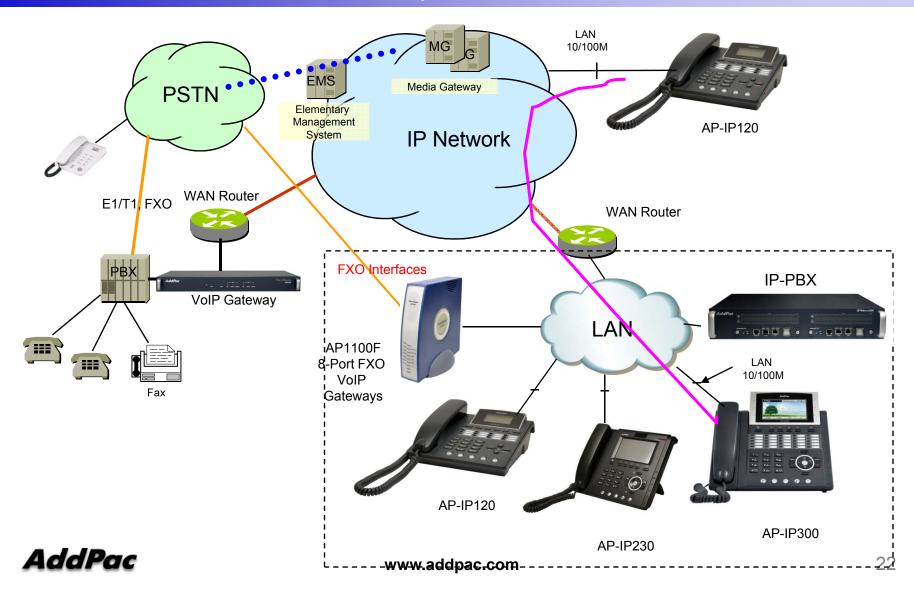
FXO Service Description

Features	Description
Voice-confirmed connect function	When FXO port is connected to PBX extension and the subscriber does take the call, connect message is not sent to sender side and billing is not included.
Ring number setting function	Use this command to set the maximum number of rings to be detected before answering a call over an FXO voice port. In that case, the FXO interface would answer if the equipment online did not answer the incoming call in the configured number of rings.
Clear down tone registration and detect function	Clear-down-tone detects call termination of FXO port connected to and generated from PSTN or PBX. The value of clear-down-tone (busy tone, fast busy tone) is different for each PSTN and PBX. So use voice class clear-down-tone for registration process in global configuration mode.
Hook flash timing setting function	Different from call-transfer, you need to press hook-flash button twice for conference call. Basically, it takes 500 ms (0.5 sec) to recognize hook-flash button from the AddPac gateway. If you think 500ms (0.5 sec) is too short, you can change hook-flash detect timeout value when hook-flash duration time of PBX is more than 500ms.



Network Diagram for FXO Call

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Light Call Manager Service

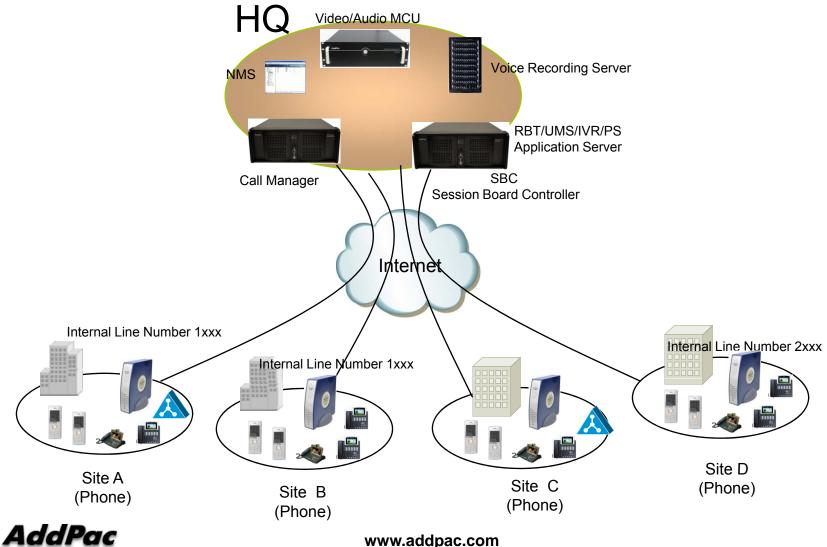


Light Call Manager Service Features

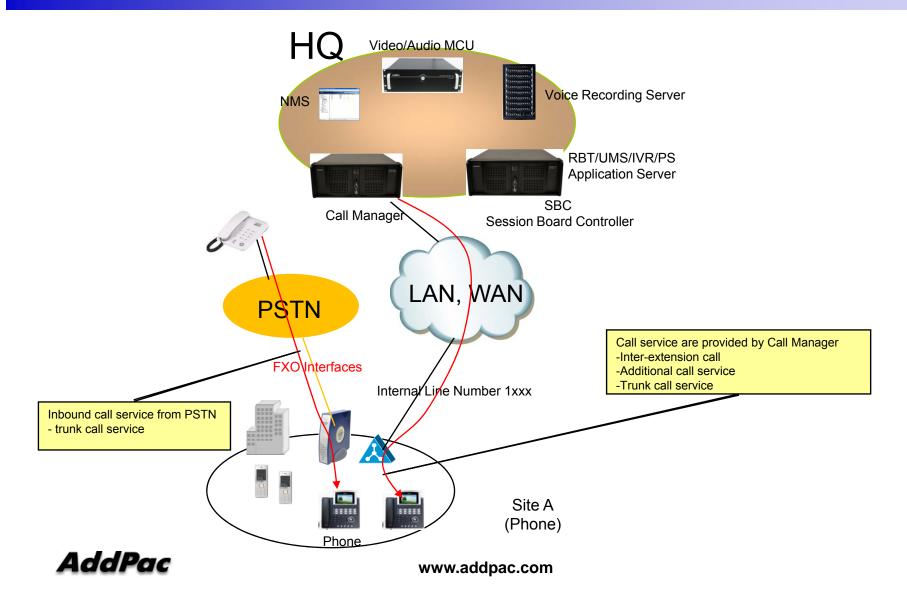
- Support basic SIP registration service
 - Support basic SIP register service (no authentication).
- Support basic call service
 - Support inter-extension call service
 - NOT support addition call service hold resume pickup, etc.
- Support PSTN call service
 - Support Inbound and outbound call service
- Can be used as a backup call manager
 - If a main VoIP call manager is not available, extensions are registered to light call manager(AP1100FN).



Network Diagram

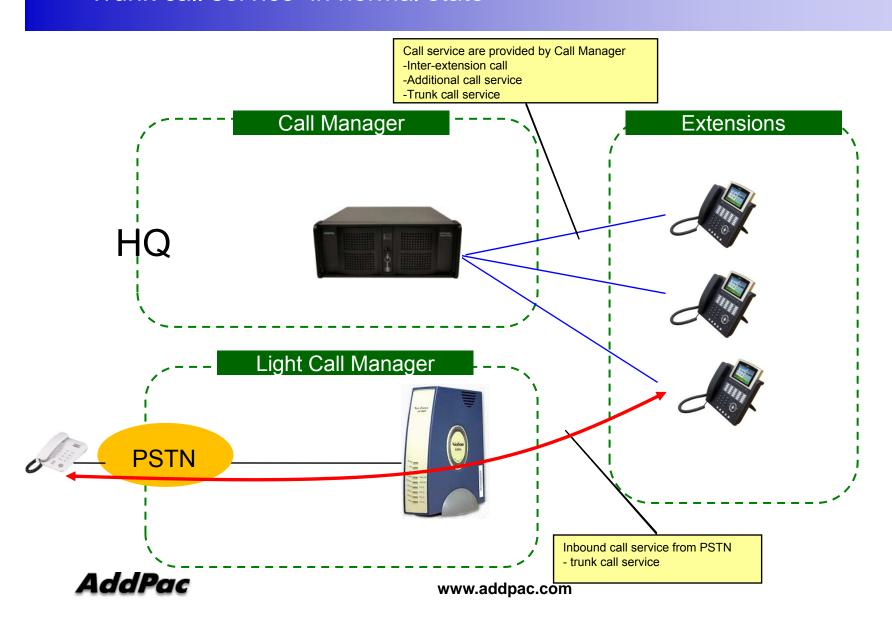


Network Diagram (PSTN Inbound)



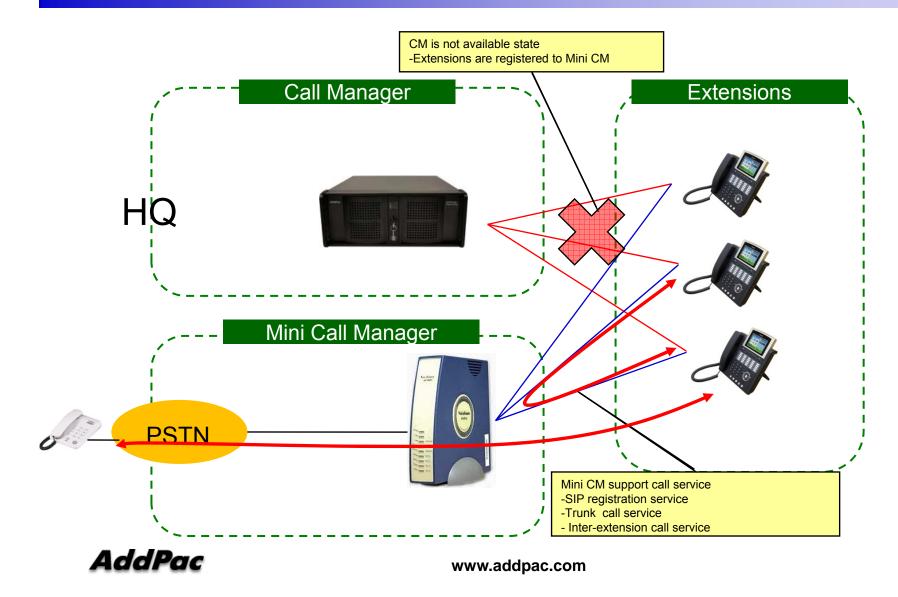
Call Service Diagram

- Trunk call service in normal state



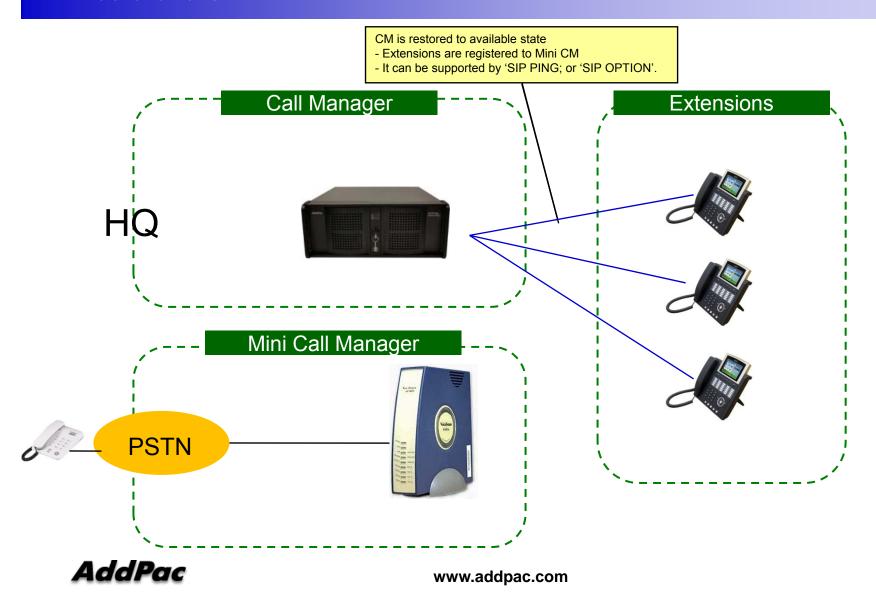
Call Service Diagram

- backup call service in abnormal state



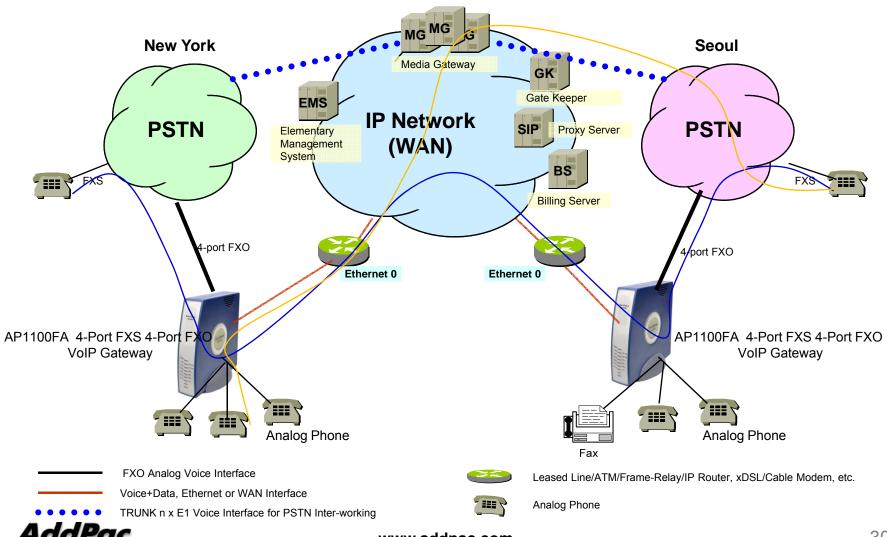
Call Service Diagram

- restore to CM



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Standard Application



Ordering Information

AP1100FN 8-Port FXO VoIP Gateway Hardware

- AP1100FN VoIP Gateway Main Body (Model C Type)
- RISC Microprocessor with High-end Programmable DSP Architecture
- 1-Port 10/100Mbps Fast Ethernet(RJ45)
- 1-Port 10Mbps Ethernet(RJ45)
- 8-Port FXO (RJ11)
- Including Network Cable & Ext. Power Supply, etc.
- Built-in APOS Internetworking Software for AP1100FN
- Including 1 Year Hardware Warranty
- Product Documents
 - Install and Operation Guide (PDF)
- Pricing
 - AddPac Technology Regional Sales Manager
 - Authorized Sales and Marketing Representatives
 - Please Contact www.addpac.com



VoIP Gateway Series

Thank you!

AddPac Technology Co., Ltd. Sales and Marketing

Phone +82.2.568.3848 (KOREA) FAX +82.2.568.3847 (KOREA) E-mail sales@addpac.com

