

VoIP Gateway

WellGate 2504
4-FXS Gateway



- SIPv2 RFC 3261 Compliance
- G.168 Echo Cancellation
- Support Caller ID, FAX
- Local Dialing Plan & Digit Manipulation
- Flexible Routing Plan
- IPv6 / IPv4
- Auto Provision

WellGate 2540
4-FXO Gateway



- SIPv2 RFC 3261 Compliance
- G.168 Echo Cancellation
- Support Caller ID, FAX
- Support SIP Trunk
- Flexible Routing Plan
- Auto Release FXO port when line drop
- IPv6 / IPv4
- Auto Provision

WellGate 2522
2-FXO, 2-FXS Gateway



- SIPv2 RFC 3261 Compliance
- G.168 Echo Cancellation
- Support Caller ID, FAX
- Local Dialing Plan & Digit Manipulation
- Flexible Routing Plan
- PSTN Backup when Network Failure
- Migrate PABX to VoIP Call
- IPv6 / IPv4
- Auto Provision

WellGate 2608
8-port VoIP FXS Gateway



- SIPv2 RFC 3261 Compliance
- 1 WAN / 4 LAN
- G.168 Echo Cancellation
- Carrier-grade Features are Supported
- Interoperability Tested with Marketing Leading IP-PBX
- Support 8 ports FXS lines

WellGate 2680
8-port VoIP FXO Gateway



- SIPv2 RFC 3261 Compliance
- 1 WAN / 4 LAN
- G.168 Echo Cancellation
- Carrier-grade Features are Supported
- Interoperability Tested with Marketing Leading IP-PBX
- Auto Release FXO port when line drop

WellGate 2644
4-port FXO + 4-port FXS



- SIPv2 RFC 3261 Compliance
- 1 WAN / 4 LAN
- G.168 Echo Cancellation
- Carrier-grade Features are Supported
- Interoperability Tested with Marketing Leading IP-PBX
- Migrate digital PBX to VoIP Call
- 1 to 1 bypass directly when network or power is failure

WellGate 2644P
4-PSTN plus 4-FXS



- SIPv2 RFC 3261 Compliance
- G.168 Echo Cancellation
- Carrier-grade Features are Supported
- Interoperability Tested with Marketing Leading IP-PBX
- Migrate digital PBX to VoIP Call
- PSTN lines direct to FXS port
- PSTN lines restrict to access IP network
- 1 to 1 bypass directly when network or power is failure

WellGate 2424S
24-port High Density FXS Gateway



- SIPv2 RFC 3261 Compliance
- 1 WAN / 1 LAN
- G.168 Echo Cancellation
- Carrier-grade Features are Supported
- Interoperability Tested with Marketing Leading IP-PBX
- Support 24 ports FXS lines
- IPv6 / IPv4
- Auto Provision

ATA171 plus
1-line with 1WAN/1LAN NAT



- SIPv2 RFC 3261 Compliance
- Flexible Dial Plan
- STUN NAT Transversal
- G.168 Acoustic Echo Cancellation
- PPTP VPN
- Auto Provision
- Switch mode support wire line speed

ATA172 plus
2-line with 1WAN/1LAN NAT



- SIPv2 RFC 3261 Compliance
- Flexible Dial Plan
- STUN NAT Transversal
- G.168 Acoustic Echo Cancellation
- PPTP VPN
- Auto Provision
- Switch mode support wire line speed

ATA171P,171M
1-line with PSTN or FXO Adaptor



- SIPv2 RFC 3261 Compliance
- Flexible Dial Plan
- STUN NAT Transversal
- G.168 Acoustic Echo Cancellation
- PPTP VPN
- Auto Provision
- 1-FXS with 1-PSTN (ATA171P)
- 1-FXS with 1-FXO (ATA171M)



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IPv6+IPv4

DUAL STACK SIMULTANEOUSLY

ITSP Telephony Solution
IP-PBX
VoIP Gateway/ATA
Wi-Fi VoIP
IP Phone
TR-069 Provision

IP-PBX



SIPPBX 6200S/GS/N
Enterprise IP-PBX

- SIPv2 RFC 3261 Compliance
- Up-to 1,000 Registration Users
- 1+1 High Availability Redundant
- Auto Attendant / Voice Mail / Conference Bridge
- Support Up-to 12 Branch Offices
- Support Operator Console & Softphone
- Compatible With SIP Door Phone LP803



SIPPBX 6200A
Enterprise IP-PBX

- SIPv2 RFC 3261 Compliance
- Up-to 200 Registration, 50 Concurrent Calls
- PABX Telephony Service
- Auto Attendant / Voice Mail / Conference Bridge
- Support Asterisk SIP Stack
- IPv6 / IPv4



ePBX100A-128
Small IP-PBX

- SIPv2 RFC 3261 Compliance
- Up-to 30 Registration Users, 15 Concurrent Calls
- Build-in 1GB Flash for voice mail
- PABX Telephony Service
- Auto Attendant / Voice Mail / Conference Bridge
- Embedded Asterisk
- Support BLF (Busy Lamp Field)

Enhance Service



WellRec 5600
Lawful Interception (LI) / VoIP Recorder

- SIPv2 RFC 3261 Compliance
- Up-to 120 Channels, 30,000 Hours Storage
- Subscriber/Target Based Voice Recording
- Support Lawful Interception/Call Logging
- Support IETF H12/H13
- Easy Management & Call Retrieve
- Built-in SBC Inside



IPCentx 6850GS
SIP based IP Centrex

- Based on SIIIVR 6800GS
- Up-to 120 Channels, 10,000 Subscribers
- Auto Attendant, Voice Mail, Coloring Ring Back Tone and Announcement Service
- Supplementary ISR Call Flow: Callback, Calling Card, Balance Query, Welcome Call
- Up-to 9999 Companies per Chassis



SIIIVR 6800GS
SIP Based IVR Server

- SIPv2 RFC 3261 Compliance
- Up-to 120 Universal VOIP IVR Channels
- Drag and Drop Call Flow Editor
- Real Time Status/Variable Debugger
- Rich-set of Predefined Components
- Support G.711, G.729A
- Support External API Interface

SIP Proxy Server



WellSIP 6500 V3.0
Telephony SIP Proxy Server

- SIPv2 RFC 3261 Compliance
- Up-to 20K Users
- 1+1 High Availability Redundant
- NAT On-Demanded Traversal
- Rich Telephony Service
- Flexible Call Routing
- Support SIP/RTP Encryption
- Support ENUM
- Support BLF (RFC 4235)



WellSIP 6550 V3.0
High Capacity Telephony SIP Proxy Server

- SIPv2 RFC 3261 Compliance
- Up-to 100K Users
- 1+1 High Availability Redundant
- NAT On-Demanded Traversal
- Rich Telephony Service
- Flexible Call Routing
- Support SIP/RTP Encryption
- Support ENUM
- Support BLF (RFC 4235)



WellSIP 6500B
Telephony SIP Server with Billing

- SIPv2 RFC 3261 and RADIUS AAA
- Up-to 1000 Users
- NAT On-Demanded Traversal
- Rich Telephony Service
- Flexible Call Routing
- Support Prepaid/Postpaid
- Flexible Rate Plan Support

IP Phone



IP Phone LP600N
SIP IP Phone

- Power Over Ethernet (Optional)
- SIPv2 RFC 3261 Compliance
- Built-in Headset Jack and Provide Optional Headset
- Full-Duplex Hand-Free Speaker AEC
- 2 Multi-lines Registration
- 3 Programmable Keys with BLF
- 3-way Conference
- For IP-PBX Application with auto provision



IP Phone LP801
SIP IP Feature Phone

- SIPv2 RFC 3261 Compliance
- 128X64 LCD Display
- Chinese and English Display
- Power Over Ethernet (Optional)
- Built-in Headset Jack and Provide Optional Headset
- Full-Duplex Hand-Free Speaker AEC
- 2 Multi-lines Registration
- 3-way Conference
- Support Auto Provision



IP Phone LP803
SIP IP Phone

- Support up to 3 Register Accounts at three SIP Servers
- Portable IP Phone for Office, Home, Traveler and Door phone
- Excellent Loud Speaker Phone
- Auto Provision feature
- Color : Black and White
- Support PoE (LP803A)

Session Boarder Controller



WellIBG 5800/5800S
SIP-aware Session Border Controller

- SIPv2 RFC 3261 Compliance
- Support Multiple SIP Service Port
- Support Audio and Video NAT Traversal
- N+1 Redundant
- Support Voice Recording
- Local Surviving Client Side SBC
- Softswitch Network Hiding



WellIRTP 5100
External RTP Resource Server

- Support RTP/RTCP/T.38
- Up-to 384 Full Duplex RTP Proxy
- Support Voice/Video RTP
- Support WellSIP 6500 and WellIBG 5800 Series



WellBilling 6600 V2.0
RADIUS Prepaid/Postpaid Billing Server

- RADIUS AAA Support
- Up-to 500K Subscribers Support
- 1+1 High Availability Redundant
- Support Prepaid/Postpaid
- Flexible Rate Plan Support
- Support Least Cost Routing
- Support QoS Route
- Provide Operation Reports

Billing System



IP Phone LP399
SIP IP Feature phone

- SIPv2 RFC 3261 Compliance
- Full-Duplex Hand-Free Speaker AEC
- 3 Multi-lines Registration
- STUN NAT Transversal
- Build-in 3-way Conference Function
- PPTP VPN
- Auto provision



IP Phone LP389
IPv6 and IPv4 SIP IP Phone

- IPV6 and IPv4 Dual Stack simultaneously
- Support up-to 3 Register Accounts at three SIP Servers
- Rich Telephony Features
- Support Receive Instant Message (IM)
- Intelligent Phone Book Name Dialing
- Dual Flash Memory image backup
- Auto Provision feature
- Color : Blue, Black and Grey
- Support PoE (LP389A)

Trunking Gateway



WellGate 5260
T1/E1 Universal VOIP Gateway

- Support SIP RFC 3261 and ITU-T H.323 V5 Simultaneously
- Support G.711, GSM, G.729A, G.723.1
- Up to 4 Programmable E1/T1 Trunks
- PSTN Signaling: ISDN/PRI, CAS, MFC R2, QSIG
- Support up-to 16 Multiple SIP Proxy Servers
- Support SIP Proxy, Gatekeeper and P2P Calls Simultaneously
- Flexible Call Flow Editor



WellGate 5290
A Universal VOIP/SS7 Gateway

- Support SIP RFC 3261 and ITU-T H.323 V5 Simultaneously
- Support G.711, GSM, G.729A, G.723.1
- Support 4*Voice E1s
- PSTN Signaling: ISDN/PRI, CAS, MFC R2, QSIG, SS7
- Support SS7 only or SS7 with Voice E1
- Support up-to 16 Multiple SIP Proxy Servers
- Support SIP Proxy, Gatekeeper and P2P Calls Simultaneously
- Flexible Call Flow Editor



WellISC 5900
Universal SIP/H.323 Signal Converter

- SIP RFC 3261 and H.323 V5
- H.323 to H.323 Calls
- SIP to H.323 Calls
- SIP to SIP Calls
- H.323 to SIP Calls
- Support H.323 and SIP Video
- Support Route and un-Route RTP Mode
- Support DTMF Relay Conversion

Signal Converter

WiFi Phone



WP589
WiFi Phone Standby time 160 hrs

- Comply with IEEE 802.11 b/g WiFi client mode
- Support up to 4 Register SIP Servers
- Comply with SIP RFC3261 and any SIP Server, IPPBX
- Color LCD display with Long Durable talk time
- Roaming among WiFi AP
- Bilingual Display : English and Chinese

Wireless VoIP Gateway



WellGate 3512
Wi-Fi VoIP Router + Triple play

- SIPv2 RFC 3261 Compliance
- Build-in Wi-Fi AP, Wi-Fi Client, Router, and Voip Gateway
- SOHO Router (1 Wan / 4 Lan) With Voice Priority
- Support WDS and Universal Repeater
- Triple play service
- Security Protect by Firewall and Denial of Service.

