♦ 30 SIP Extensions

♦ 15 Concurrent Calls

♦ Fax to Email

♦ Voice Mail to Email

♦ Paging and Intercom

♦ HD Voice Codec G.722

Supported ♦ Web Call via

Google Chrome

UT-VPX-20 IP Phone system

The UT-VPX-20 IP Phone System is the second generation of Mini IP Phone System launched by Uniteli, meanwhile it's the first Mobile IP Phone System with the ability to use the GSM module to connect to the mobile network for data and voice transfer.

UT-VPX-20 is a next-generation smart IP PBX System designed and optimized for today's business telecommunication needs. UT-VPX-20 is an affordable, scalable, interoperable, secure and reliable solution which provides not only traditional PBX features such as Automated Attendant and Voicemail, but also many advanced features including Remote Extensions, Remote Office Connection, Conference Bridge, Call Recording, Call Detail Records (CDR), Automatic Call Distribution (ACD),Unified Messaging (Voicemail to Email), and many more features.

Technical Data

Highlights & Features

- 30 IP Phone Registers/Extensions & up to 15 Concurrent Calls for cost-effective communication
- Fax to Email/Email to Fax
- Voicemail to Email
- Paging and Intercom function strengthens the working efficiency
- Monitor and Whisper
- HD Voice Codec G.722 for perfect voice quality
- Strong security features protect your system from hacking

Model	FXO	FXS
VPX-20-2	2	0
VPX-20-1S/1	1	1

GSM is optional for configuration

System Specifications

Call Features
- Call Back/Call Forward/Call Hold/Call Paging and Intercom/Call Park/Call Pickup/Call Queue/Call Recording Call Routing/Blind Transfer/Attended Transfer/Call Waiting/Caller ID/Dial by Name/Music On Hold/Transfer 3-Way Conference/Video Call

PBX Features
- Black List/BLF (Busy Lamp Field)/CDR (Call Detailed Record)/Conference Room (3 Rooms)
Call Monitoring
DID (Direct Inward Dialing Number)
DISA (Direct Inward System Access)
Distinctive Ringtone/DND (Do Not Disturb)
DNIS (Dialed Number Identification Service)
Feature Codes
FOP (Flash Operation Panel) Status Monitoring
Follow Me/IVR (Interactive Voice Responses)
Mobility Extension/Multi-Language Prompts
Multi-Language GUI/One Touch Recording
Phone Book
Phone Provisioning/Pin Set/Record File Download
Ringgroup/SIP Register with UDP/TCP/TLS/SIP Trunk
Skype for SIP/Smart DID/Speed Dial/Spy
SRTP (Secure Realtime Transport Protocol)/T.38 Fax (Pass-through)
Time Based Rule/Virtual Fax
Voicemail & Voicemail to Email

System Capacities
15 Concurrent Calls
Up to 30 IP Phone Registers/Extensions
Recording: 21,000 mins (.gsm); 4,000 mins (.wav)
Voicemail: 21,000 mins (.gsm); 4,000 mins (.wav)
Codecs & Protocols
Audio Codecs: G.722/G.711-Ulaw/G.711-Alaw/ G.726/G.729/GSM/SPEEX
Video Codecs: H.261/H.263/H.263+/H.264
Protocols: SIP (RFC3261)/IAX2
DTMF: RFC2833/SIP INFO/In-Band
Network Features
DDNS Client/DHCP Server/IPv4/IPv6/IEEE802.1Q
IP Assignment (PPPoE/DHCP/Static)
SNMP v1/v2
TR069/Static Route Table
Trouble Shooting (Ping/Traceroute)
PoE (Power over Ethernet)
VPN Client (N2N/L2TP/PPTP/OpenVPN/IP Sec)
VPN Server (PPTP/L2TP/OpenVPN/IP Sec, Up to 10 Users)
Security Features
Refuse SIP Register DoS
Refuse Abort Invite DoS
Refuse SSH Login DoS
Firewall
Logs
PBX Log/Web Access LogPBX Debug Log
Hardware Interfaces
1 Reset Button
1 Power Interface
2 Ethernet Interfaces
2 Analog Ports (FXO/FXS)
1 Interface for Optional GSM Module
1 USB interface
Hardware Specifications
Processor: DualCore A7 1GMhz DSP
SDRAM: DDR3 512MB
Storage: 8GB SD Card
Power Supply: Input AC 100~240V, 50/60Hz;
Output DC 12V/1A
Environment
Working Temperature: 0 ~ 40°C
Storage Temperature: -20 ~ 55°C
Humidity: 5 ~ 95% Non-condensing
Packings
Dimension: 295 × 155 × 65mm
Weight/Unit: 0.8kg
Unit /Carton: 10 Units
Weight/Carton: 9kg