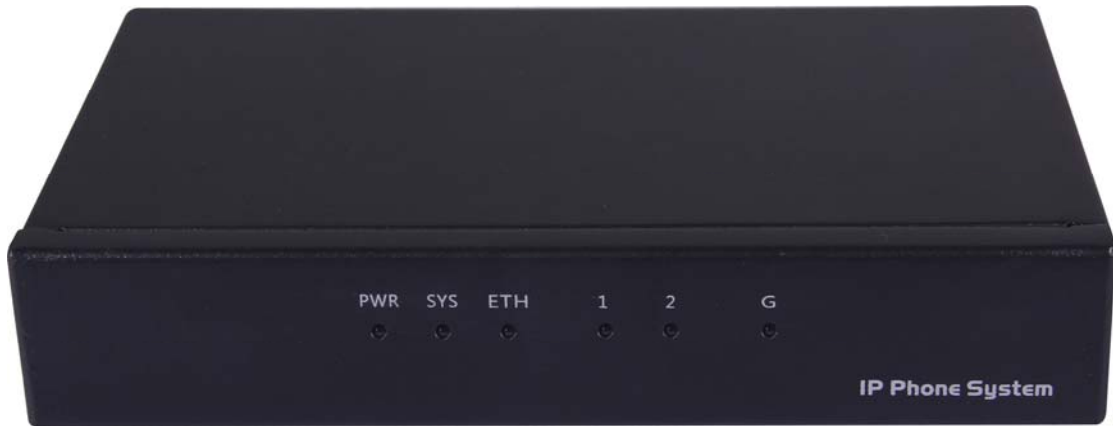


# Small & Medium Office Business

## IP PBX

### UTT-200 series





**Model**

Model	FXO	FXS
UTT-200-A202	2	0
UTT-200-A211	1	1

GSM or UMTS (3G) is optional for configuration.

**Applicable modules**



**FXO-200**



**FXS-200**



**FXOS**



**1GSM Module**



**1UM (3G)**

### Quick Details

- Place of Origin: Guangdong, China (Mainland)
- Brand Name: Ultrative or OEM
- Model Number: UTT-200 series
- Type: VoIP PBX
- 3-Way Conference: supports
- Video Call: supports
- BLF (Busy Lamp Field): supports
- Mobility Extension: supports
- Multi-Language Prompts/GUI: supports
- SIP Trunk: supports
- Concurrent Calls: 10
- IP Phone registers/extensions: 30
- Audio Codecs: G.722, G.711, G.726, G.729, GSM, SPEEX
- Video Codecs: H.261, H.263, H.263+, H.264

### Specifications

30 IP Phone Registers/Extensions & up to 10 Concurrent Calls  
Fax to Email/Email to Fax, Voicemail to Email  
Paging and Intercom

### Product Description

UTT-200 series IP Phone system is the second generation of Mini IP Phone System, meanwhile it's the first Mobile IP Phone system with the ability to use the GSM or UMTS (3G) module to connect to the mobile network for data transport.

UTT-200 series is a next-generation smart IP PBX system designed and optimized for today's business telecommunication needs.

UTT-200 series is an affordable, scalable, interoperable, secure and reliable solution which not only provides traditional PBX functions 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 (voice mail to email), and many more features

### Features

<b>Call Features</b> Call Back	<b>System Capacities</b> 10(Max) Concurrent Calls
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<p>Call Forward</p> <p>Call Hold</p> <p>Call Paging and Intercom</p> <p>Call Park</p> <p>Call Pickup</p> <p>Call Queue</p> <p>Call Recording</p> <p>Call Routing</p> <p>Blind Transfer</p> <p>Attended Transfer</p> <p>Call Waiting</p> <p>Caller ID</p> <p>Dial by Name</p> <p>Music On Hold/Transfer</p> <p>3-Way Conference</p> <p>Video Call</p> <p><b>PBX Features</b></p> <p>Black List</p> <p>BLF (Busy Lamp Field)</p> <p>CDR (Call Detailed Record)</p> <p>Conference Room (3 Rooms)</p> <p>Call Monitoring</p> <p>DID (Direct Inward Dialing Number)</p> <p>DISA (Direct Inward System Access)</p> <p>Distinctive Ringtone</p> <p>DND (Do Not Disturb)</p> <p>DNIS (Dialed Number Identification Service)</p> <p>Feature Codes</p> <p>FOP (Flash Operation Panel) Status Monitoring</p> <p>Follow Me</p> <p>IVR (Interactive Voice Responses)</p> <p>Mobility Extension</p> <p>Multi-Language Prompts</p> <p>Multi-Language GUI</p> <p>One Touch Recording</p> <p>Phone Book</p> <p>Phone Provisioning</p> <p>Pin Set</p> <p>Record File Download</p> <p>Ringgroup</p> <p>SIP Register with UDP/TCP/TLS</p> <p>SIP Trunk</p> <p>Skype for SIP</p>	<p>Up to 30 IP Phone Registers/Extensions</p> <p>Recording: 21,000 mins (.gsm); 3,000 mins (.wav)</p> <p>Voicemail: 21,000 mins (.gsm); 3,000 mins (.wav)</p> <p><b>Codecs &amp; Protocols</b></p> <p>Audio Codecs: G.722 /G.711-Ulaw /G.711-Alaw /G.726 /G.729 /GSM /SPEEX</p> <p>Video Codecs: H.261 /H.263 /H.263+ /H.264</p> <p>Protocols: SIP (RFC3261) /IAX2</p> <p>DTMF: RFC2833/SIP INFO/In-Band</p> <p><b>Network Features</b></p> <p>DDNS Client</p> <p>DHCP Server</p> <p>IPv4/IPv6/IEEE802.1Q</p> <p>IP Assignment (PPPoE/DHCP/Static)</p> <p>SNMP v1/v2</p> <p>TR069</p> <p>Static Route Table</p> <p>Trouble Shooting (Ping/Traceroute)</p> <p>PoE (Power over Ethernet)</p> <p>VPN Client (N2N/L2TP/PPTP/OpenVPN)</p> <p>VPN Server (PPTP/L2TP/OpenVPN, Up to 10 Users)</p> <p><b>Security Features</b></p> <p>Refuse SIP Register DoS</p> <p>Refuse Abort Invite DoS</p> <p>Refuse SSH Login DoS</p> <p>Firewall</p> <p><b>Logs</b></p> <p>PBX Log</p> <p>Web Access Log</p> <p>PBX Debug Log</p>
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Smart DID	
Speed Dial	
Spy	
S RTP (Secure Realtime Transport Protocol)	
T.38 Fax (Pass-through)	
Time Based Rule	
Virtual Fax	
Voicemail & Voicemail to Email	

## Specifications

### Hardware Interfaces

- 1 Reset Button
- 1 Power Interface
- 1 Ethernet Interface
- 2 Analog Ports (FXO/FXS)
- 1 GSM/UMTS Interface (Onboard)

### Hardware Specifications

Processor: DualCore 500Mhz DSP

SDRAM: DDR2 128MB

Storage: 4GB 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

\*PSTN is not perfectly supported temporarily in UK

### Packing

Inner Box Dimension	320 mm × 170 mm × 80 mm
G.W./Unit	0.8 kg
Carton Dimension	340 mm × 330 mm × 330 mm
Units/Carton	10 Units
G.W./Carton	10 kg