



## Optimized UC Features for SMBs

### UCM6200 series

Designed to provide a centralized solution for the communication needs of businesses, the UCM6200 series IP PBX appliance combines enterprise-grade voice, video, data, and mobility features in an easy-to-manage solution. This IP PBX series allows businesses to unify multiple communication technologies, such as voice, video calling, video conferencing, video surveillance, data tools, mobility options and facility access management onto one common network that can be managed and/or accessed remotely. The secure and reliable UCM6200 series delivers enterprise-grade features without any licensing fees, costs-per-feature or recurring fees.



**800 users**

Supports up to 800 users, 50 SIP trunk accounts, up to 100 concurrent calls

**zero CONFIG**

Zero configuration provisioning of Grandstream SIP endpoints



Strongest-possible security protection using SRTP, TLS and HTTPS encryption



**Gigabit**

Dual Gigabit network ports with integrated PoE+



Supports up to a 5-level IVR (Interactive Voice Response)



Built-in call recording server; recordings accessed via web user interface



Supports call queue for efficient call volume management



Built-in Call Detail Records (CDR) for tracking phone usage by line, date, etc.



Multi-language auto-attendant to efficiently handle incoming calls



Integrated LDAP and XML phonebooks, flexible dial plan



Supports any SIP video endpoint that using the H.264, H.263 and H.263+ codecs



Supports voicemail and fax forwarding to email

<b>Analog Telephone FXS Ports</b>	2 ports (both with lifeline capability in case of power outage)
<b>PSTN Line FXO Ports</b>	2 ports (UCM6202); 4 ports (UCM6204); 8 ports (UCM6208)
<b>Network Interfaces</b>	Dual Gigabit RJ45 ports with integrated PoE Plus (IEEE 802.3at-2009)
<b>NAT Router</b>	Yes (supports router mode and switch mode)
<b>Peripheral Ports</b>	USB, SD
<b>LED Indicators</b>	Power/Ready, Network, PSTN Line, USB, SD
<b>LCD Display</b>	128x32 graphic LCD with DOWN & OK button
<b>Reset Switch</b>	Yes
<b>Voice-over-Packet Capabilities</b>	LEC with NLP Packetized Voice Protocol Unit, 128ms-tail-length carrier grade Line Echo Cancellation, Dynamic Jitter Buffer, Modem detection & auto-switch to G.711
<b>Voice and Fax Codecs</b>	G.711 A-law/U-law, G.722, G.723.1 5.3K/6.3K, G.726, G.729A/B, iLBC, GSM, AAL2-G.726-32, ADPCM; T.38
<b>Video Codecs</b>	H.264, H.263, H263+
<b>QoS</b>	Layer 3 QoS, Layer 2 QoS
<b>DTMF Methods</b>	In Audio, RFC2833, and SIP INFO
<b>Provisioning Protocol &amp; Plug-and-Play</b>	TFTP/HTTP/HTTPS, auto-discovery & auto-provisioning of Grandstream IP endpoints via Zero-Config (DHCP Option 66 multicast SIP SUBSCRIBE mDNS), eventlist between local and remote trunk
<b>Network Protocols</b>	TCP/UDP/IP, RTP/RTCP, ICMP, ARP, DNS, DDNS, DHCP, NTP, TFTP, SSH, HTTP/HTTPS, PPPoE, SIP (RFC3261), STUN, SRTP, TLS, LDAP
<b>Disconnect Methods</b>	Call Progress Tone, Polarity Reversal, Hook Flash Timing, Loop Current Disconnect, Busy Tone
<b>Media Encryption</b>	SRTP, TLS, HTTPS, SSH
<b>Universal Power Supply</b>	Output: 12VDC, 1.5A; Input: 100 ~ 240VAC, 50 ~ 60Hz
<b>Dimensions</b>	UCM6202/6204: 226mm L x 155mm W x 34.5mm H UCM6208: 440mm L x 185mm W x 44mm H
<b>Weight</b>	UCM6202/6204: Unit weight 0.51kg, Package weight 0.94kg UCM6208: Unit weight 2.23kg, Package weight 3.09kg
<b>Environmental</b>	Operating: 32 ~ 104°F / 0 ~ 40°C, 10 ~ 90% (non-condensing); Storage: 14 ~ 140°F / -10 ~ 60°C
<b>Mounting</b>	Wall mount & Desktop (UCM6202/6204); Rack mount & Desktop (UCM6208)
<b>Multi-Language Support</b>	English/Simplified Chinese/Traditional Chinese/Spanish/French/Portuguese/German/Russian/Italian/Polish/Czech for Web UI; Customizable IVR/voice prompts for English, Chinese, British English, German, Spanish, Greek, French, Italian, Dutch, Polish, Portuguese, Russian, Swedish, Turkish, Hebrew, Arabic; Customizable language pack to support any other languages
<b>Caller ID</b>	Bellcore/Telcordia, ETSI-FSK, ETSI-DTMF, SIN 227 - BT
<b>Polarity Reversal/Wink</b>	Yes, with enable/disable option upon call establishment and termination
<b>Call Center</b>	Multiple configurable call queues, automatic call distribution (ACD) based on agent skills/availability/busy level, in-queue announcement
<b>Customizable Auto Attendant</b>	Up to 5 layers of IVR (Interactive Voice Response)
<b>Maximum Call Capacity</b>	-Registered SIP devices: supports up to 800 registered SIP devices/users (for UCM6208) and 500 registered SIP devices/users (for UCM6202/6204) -Concurrent SIP calls: Up to 30 (UCM6202), 45 (UCM6204), or 100 (UCM6208), or 66% performance if calls are SRTP encrypted
<b>Conference Bridges</b>	Up to 3 (UCM6202/6204) or 6 (UCM6208) password-protected conference bridges allowing up to 25 (UCM6202/6204) or 32 (UCM6208) simultaneous PSTN or IP participants
<b>Call Features</b>	Call park, call forward, call transfer, DND, ring/hunt group, paging/intercom etc.
<b>Compliance</b>	FCC: Part 15 (CFR 47) Class B, Part 68 CE: EN55022 Class B, EN55024, EN61000-3-2, EN61000-3-3, EN60950-1, TBR21, RoHS A-TICK: AS/NZS CISPR 22 Class B, AS/NZS CISPR 24, AS/NZS 60950, AS/ACIF S002 ITU-T K.21 (Basic Level); UL 60950 (power adapter)