



## Wave App Connect Anywhere at Anytime

Wave is a mobile and desktop app that provides remote voice and video collaboration tools for businesses utilizing Grandstream's UCM6300 series IP PBXs. Ideal for remote workers, Wave offers an easy-to-use platform to remotely join, schedule and hold meetings, calls and conferences from anywhere. It also allows UCM6300 series users to directly call other extensions, landlines, and mobile numbers. Wave is free, available for Android and iOS devices as well as PC/Mac and can be quickly configured by scanning a barcode produced by the UCM6300 series. It pairs with Grandstream's UCM RemoteConnect cloud service, which ensures a fully secure connection between Wave and the UCM6300 series by providing automated NAT Traversal. Thanks to Wave, businesses can provide remote workers with a powerful mobile and desktop tool to meet and collaborate from anywhere, boosting productivity for dispersed organizations and remote workers.



Mobile app for Android & iOS devices; desktop for PC/Mac



Remotely join meetings, calls and conferences



Compatible with Wi-Fi and 2G/3G/4G/5G networks



Allows UCM6300 series users to call other extensions, landlines & mobile numbers



Schedule and start meetings from the app



Supports mobile devices' camera, built-in webcam & USB webcams



Supports video resolutions up to 1080p HD



Provides HD audio for crystal-clear meetings and calls



Supports use with Bluetooth, USB and other headsets



Download app & scan QR code from UCM6300 series to configure



Compatible with UCM RemoteConnect cloud service for secure remote connections

Protocols/Standards	SIP RFC3261, TCP/IP/UDP, RTP/RTCP, HTTP/HTTPS, DNS (A record, SRV, NAPTR), STUN/TURN/ICE, SIMPLE, LDAP, TLS, SRTP, IPv6
Network	Wi-Fi, 2G/3G/4G/5G
On-Screen Display	720P resolution or above
Camera	Supports mobile device's front and rear cameras
Bluetooth	Yes, with mobile device's Bluetooth support
Voice Codecs and Capabilities	Opus, G.711 A-law/U-law, G.722, G.726-32, G.729A/B, iLBC, GSM, FEC 2.0, NACK
DTMF	In-audio, RFC2833, SIP INFO
High Audio Quality	Full-duplex speaker, AEC, AGC, Noise Reduction, PLC, Adaptive JIB
Video Codecs and Capabilities	H.264, video resolution up to 1080P HD, on-screen-display, camera block, GS-FEC
Telephony Features	Hold, audio conference, video conference
UCM Integration	Feature code synchronization, call recording, transfer, conference room
Mobile Device Integration	Support background mode, proximity sensor for in-call touch screen and keys lock
	keys lock
Enterprise Features	
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QoS	Presence
QoS	Presence  Layer 3 (ToS, DiffServ, MPLS) QoS  English, Simplified Chinese
QoS Multi-language Security	Presence  Layer 3 (ToS, DiffServ, MPLS) QoS  English, Simplified Chinese  AES encrypted configuration file, SIP over TLS, SRTP (128-bit and 256-bit),

