



# DP750

## Long-range DECT VoIP Base Station

The DP750 is a powerful DECT VoIP base station that pairs with up to 5 of Grandstream's DP720 DECT handsets to offer mobility to business and residential users. It supports a range of 300 meters outdoors and 50 meters indoors to give users the freedom to move around their work or home space, delivering efficient flexibility. This DECT VoIP base station supports up to 10 SIP accounts and 4 concurrent calls while also offering 3-way voice conferencing, full HD audio and integrated PoE. A shared SIP account on all handsets will add seamless unified features that gives users the ability to answer all calls regardless of location in real-time. The DP750 supports a variety of auto-provisioning methods and TLS/SRTP/HTTPS encryption security. When paired with Grandstream's DP720, the DP750 offers a powerful DECT VoIP base station that allows any business or residential user to create a cordless VoIP solution.



Up to 10 SIP accounts per system; up to 10 lines per handset



TLS and SRTP security encryption technology to protect calls and accounts



3-way audio conferencing for easy conference calls



DP750 is fully compliant with the SIP/DECT standard



Automated provisioning options include TR-069 and XML config files



Use with Grandstream's UCM series of IP PBXs for Zero Config provisioning



Supports advanced telephony features, including call transfer, call forward, call-waiting, do not disturb, message waiting indication, multi-language prompts, flexible dial plan and more



Built-in PoE to power the device and give it a network connection

<b>Air Interface</b>	<p><b>Telephony standards:</b> DECT</p> <p><b>Frequency bands:</b> 1880 – 1900 MHz (Europe), 1920 – 1930 MHz (US), 1910 – 1920 MHz (Brazil), 1786 – 1792 MHz (Korea), 1893 – 1906 MHz (Japan), 1880 – 1895 MHz (Taiwan)</p> <p><b>Number of Channels:</b> 10 (Europe), 5 (US, Brazil or Japan), 3 (Korea), 8 (Taiwan)</p> <p><b>Range:</b> up to 300 meters outdoors and 50 meters indoors</p>
<b>Peripherals</b>	<p>5 LED indicators: Power, Network, Register, Call, DECT</p> <p>Reset button, Pairing/Paging button</p> <p>One 10/100 Mbps auto-sensing Ethernet port with integrated PoE</p>
<b>Protocol/Standards</b>	SIP RFC3261, TCP/IP/UDP, RTP/RTCP, HTTP/HTTPS, ARP/RARP, ICMP, DNS (A record, SRV, NAPTR), DHCP, PPPoE, SSH, TFTP, NTP, STUN, SIMPLE, LLDP-MED, LDAP, TR-069, 802.1x, TLS, SRTP, IPv6 (Pending)
<b>Voice Codecs</b>	G.711μ/a-law, G.723.1, G.729A/B, G.726-32, iLBC, G.722, OPUS, G.722.2/AMR-WB (special order), in-band and out-of-band DTMF (in audio, RFC2833, SIP INFO), VAD, CNG, PLC, AJB
<b>Telephony Features</b>	Hold, transfer, forward, 3-way conference, downloadable phonebook (XML, LDAP, up to 3000 entries), call waiting, call log (up to 300 records), auto answer, flexible dial plan, music on hold, server redundancy and fail-over
<b>Sample Applications</b>	Weather, Currency (Pending)
<b>QoS</b>	Layer 2 QoS (802.1Q, 802.1P) and Layer 3 QoS (ToS, DiffServ, MPLS)
<b>Security</b>	User and administrator level access control, MD5 and MD5-sess based authentication, 256-bit AES encrypted configuration file, TLS, SRTP, HTTPS, 802.1x media access control, DECT authentication & encryption
<b>Multi-language</b>	Chinese Simple, Chinese Tradition, Czech, Danish, Dutch, English, Estonian, Finnish, French, German, Hebrew, Hungarian, Japanese, Korean, Norwegian, Portuguese, Romanian, Spanish, Swedish, Turkish
<b>Upgrade/ Provisioning</b>	Firmware upgrade via TFTP/HTTP/HTTPS, mass provisioning using TR-069 or AES encrypted XML configuration file
<b>Multiple SIP Accounts</b>	<p>Up to ten (10) distinct SIP accounts per system</p> <p><b>Each handset</b> may map to any SIP account(s)</p> <p><b>Each SIP account</b> may map to any handset(s)</p>
<b>Ring Group</b>	<p>Flexible options when multiple handsets share the same SIP account</p> <p><b>Circular Mode:</b> all phones ring sequentially from the phone next to the one that answered last</p> <p><b>Linear Mode:</b> all phones ring sequentially in the predesignated order</p> <p><b>Parallel Mode:</b> all phones ring concurrently and after one phone answers, the remaining available phones can make new calls</p> <p><b>Shared Mode:</b> all phones ring concurrently and always share the same line similar to analog phones</p>
<b>Power &amp; Green Energy Efficiency</b>	Universal Power Supply Input AC 100-240V 50/60Hz; Output 5VDC 1A; Micro-USB connection; PoE: IEEE802.3af Class 1, 0.44W–3.84W
<b>Package Content</b>	Base Unit, Universal Power Supply; Ethernet cable; Quick Start Guide, GPL statement
<b>Dimensions (H x W x D)</b>	28.5 x 130 x 90 mm
<b>Weight</b>	Base unit: 143g; Universal Power Supply: 50g; Package: 360g
<b>Temperature and Humidity</b>	<p>Operation: -10 to 50°C (14 to 122°F); Storage: -20° to 60°C (-4 to 140°F)</p> <p>Humidity: 10% to 90% non-condensing</p>
<b>Compliance</b>	<p>FCC: Part 15D, Part 15B</p> <p>CE: EN60950, EN301489-1-6, EN301406</p> <p>RCM: AS/NZS60950</p> <p>ANATEL: #2288-16-9452</p>