



KX-UT Series **SIP Telephones**

- HD Voice and PoE Enabled
- Two Data Ports* for Simplified Wiring and Connecting a Second Device
- Certified for Use with BroadSoft® BroadWorks® and Digium® Asterisk®
- Environmentally Friendly—Power Consumption 1W in ECO Mode
- Two-year Limited Warranty





*Select models.



The Panasonic SIP Phone UT Series

The new UT Series combines state-of-the-art telephony architecture with business-friendly features. Choose from entry-level models with large, clear LCDs and conferencing capability to executive desk terminals with 6-line displays and wireless headset compatibility. Designed to yield **low total cost of ownership**, UT Series phones are PoE enabled and include extra data ports so there's less wiring expense. An energy-saving Eco Mode feature lets you reduce power consumption while a powerful two-year limited warranty protects your investment well into the life of the phone.

KX-UT Series Specifications

	 KX-UT136-B	 KX-UT133-B	 KX-UT123-B	 KX-UT113-B
LCD Display	Monochrome Graphical			
LCD Size	242 x 109 pix. – 6 lines	242 x 55 pixels – 3 lines		
LCD Contrast	6 levels			
LCD Backlight	On/Auto/Off			None
INSTALL OPTIONS				
Desk Mount Tilt	Yes - 2 Positions		No	
Wall Mount	KX-A433-B (optional)		KX-A432-B (optional)	
Power Adaptor	KX-A239 (optional)			
NETWORKING				
Ethernet Ports	2 – 10/100			1 – 10/100
Power Over Ethernet (PoE)	Yes (Class 2)			
AUDIO FEATURES				
Audio Codec	G.711, G.722, G.726, G.729a			
Handset, Speaker, Headset Volume	8 levels (includes echo cancellation and distortion prevention) 15 levels			
Ringtones	27			
Ringer Volume	6 levels + Off			
Headset Port	2.5 mm			
Electronic Hook Switch Control Port	3.5mm Plantronics compatible		None	
Hearing Aid	Compatible with Hearing Aid TIA-8013			
KEYS				
Keys (total*)	52		28	
FF Keys	24		-	
Navigator & Cancel Key	Yes			
SOFTWARE FEATURES				
Phone Book Entries	500 - each with 5 numbers			100 - each with 5 numbers
Call Log Entries	30 incoming calls + 30 outgoing calls			
Conferencing	3 parties (within terminal – multi-party dependent on server)			
IP FEATURES				
SIP Accounts	4		2	
SIP Compatibility	RFC 3261 Standard SIP Server, Asterisk®, BroadSoft®			
IP Version	IPv4			
DHCP Client	Yes			
DNS	Yes			
HTTP/HTTPS	Yes			
SNTP Client	Yes			
VLAN (802.1q)	Yes			
QoS (DiffServ)	Yes			
CONFIGURATION				
Plug & Play Configuration	Server based configuration, TR-069, BroadSoft® Device Management Server			
Manual Configuration	Internal web Configurator, Local (LCD based) network configuration			

*52 keys include 24 keys, 12 dial keys, 4 soft keys and 12 feature keys. Feature keys include 2 navigation keys (direction and center key).