

PANASONIC KX-UT SERIES SIP TERMINALS FEATURE SHEET

The Panasonic KX-UT range of SIP telephony terminals enhance personal communications through excellent HD quality audio on every phone, combined with easy access to powerful supportive features and applications. The range - from standard phones, office key-sets, executive terminals and touch-screen Smart Desk application phones - addresses all requirements. Panasonic's reputation for design, quality, reliability and care for the environment, ensures an exceptional user experience wherever the terminals are deployed - as part of a "cloud based" service or with an IP PBX - in a business environment or in the home.





KEY FEATURES

High Definition "HD" Audio

The KX-UT series of SIP terminals offers 'best in class' audio quality, meaning fewer repeated conversations and misheard calls. Offering Wideband High Definition Audio as standard across the range, the KX-UT series offer G.722, G.711 and G.729a Codecs. Coupled with Enhanced Echo Cancellation and an Expanded Acoustic Chamber, the KX-UT series of SIP terminals offers a superior audio experience to users over handset, speakerphone and optional headsets. The entire range features wideband compliant, hearing aid compatible handsets, and built-in, high quality speaker and microphone.

LCD Displays

Large, clear LCD displays with intuitive User Interface offer fast access to phonebooks and features.

Electronic Hook Switch (EHS)

A built in Electronic Hook Switch (Plantronics compliant) port allows the KX-UT133 and KX-UT136 SIP Terminals to have access to the Plantronics range of wireless headsets. This offers a range of portability and comfort as frequent users are able to move around freely, without being tied down by handsets.

ECO Friendly

Low power consumption, combined with an advanced ECO standby mode means lower energy costs. Consumption can be less than 1 watt in ECO mode! (KX-UT 1xx series).

Plug and Play Configuration

Extensive provisioning options, including automatic configuration via a provisioning server, means a smaller administration overhead, saving time and money. The range is certified compatible with Digium Asterisk and Broadsoft Broadworks, ensuring excellent compatibility with leading soft switch suppliers.









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	KX-UT113	KX-UT123	KX-UT133	KX-UT136	KX-UT248*	KX-UT670*
DISPLAY						
LCD Display	Monochrome Graphical	Monochrome Graphical	Monochrome Graphical	Monochrome Graphical	Greyscale Graphical	Colour 262k
LCD Size		242 x 55 pixels – 3 lines		242 x 109 pix. – 6 lines	4.4 inch	7 inch touch No
LCD Contrast		6 levels				
LCD Backlight HD Video	None	None On/Auto/Off No No				
	No 720p					
INSTALL OPTIONS	N	2		Voo 7 m	onitiona	
Wall mount	No Yes - 2 positions KX-A432 (optional) KX-A433 (optional)					KX-A434 (optional)
Power adaptor						• KX-A422
NETWORKING						
Ethernet Ports	1 - 10/100	- 10/100 2 - 10/100 2 - 10				
Power over Ethernet (PoE)		Yes				
Bluetooth	No Yes					Optional
AUDIO FEATURES						
Handset, Speaker, Headset Volume	8 levels (includes echo cancellation and distortion prevention)					15 levels
Ringtones	32					
Ringer Volume	6 levels + Off 7 levels + Off 7					
Headset Port Electronic Hook	2.5 mm 3.5 mm Plantronics compatible					
Switch Control Port						
Audio Codec	G.711, G.722, G.726, G.729a G.711, G.722, G.729a					
KEYS						
Soft Keys	Yes - 4					Yes - screen
Programmable Keys 0-9, *, # keys	024Yes			24	24 (3 pages of 8 keys)	24 (4 pages of 6 keys) Yes - screen keys
Navigator & Cancel Key	Yes					On screen + menu keys
SOFTWARE FEATURES						
Phone Book (Entries)	100 – each with 5 numbers 500 - each with 5 numbers					Depends on memory *2
Call Log Entries	None 30 incoming calls + 30 outgoing calls					100 in + 100 out
Conferencing	3 parties (within terminal – multi-party dependent on server)					
XML	XML application support *3					
Music on Hold	Supplied by Host Service (PBX / SIP Server)					
IP FEATURES						
SIP Accounts	2			4		6
SIP Compatibility	RFC 3261 Standard SIP Server, Asterisk, Broadsoft, Panasonic SIP PBX					
IP Version	Pv4					
DHCP Client DNS	Yes Yes					
HTTP	Yes					
HTTPS	Yes					
SNTP Client	Yes					
VLAN (802.1q)	Yes					
QoS (DiffServ)	Yes					
802.1x	No Yes					
CONFIGURATION						
Plug & Play Configuration Manual Configuration	Server based configuration, TR-069, Panasonic Redirect Server Internal web Configurator, Local (LCD based) network configuration					
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*Preliminary Specifications *2 up to 500 connected to NS1000 *3 Beginning of 2012

