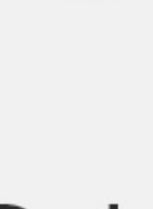




KNTD-1200



UC FEATURES FOR UP TO 1200 USERS

The KNTD-1200 IP PBX appliance is designed to bring leading edge voice, video, data, and ability features to enterprises, small and medium businesses, retail and residential environments in an easy-to-manage fashion. This enterprise-grade on premise IP PBX supports E1, T1 and J1 networks and offers scalability by supporting up to 1200 users. This secure and reliable IP PBX delivers unified communication features at an unprecedented price point without any licensing fees, costs-per-feature, or recurring fees.



MIX 1200 users
Supports up to 1200 users, 120 SIP trunk accounts, up to 600 concurrent calls



Zero configuration
provisioning of KNTech SIP End points



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



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



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

Model	KNTD-1200 IPPBX Enterprise - class unified communication system	
HARDWARE INTERFACE		
FXS port	2 x RJ11 (support power escape) Optional to add external configuration	
FXO port	2 x RJ11 (support power escape) Optional to add external configuration	
Network Interface	Using dual Gigabit interface (10/100 / 1000M Ethernet adaptive) and integrated POE (IEEE 802.3at-2009) The third Gigabit port for the hot backup cluster	
NAT router	Support (user configuration)	
Reset switch	Support, Web configuration	
AUDIO / VIDEO CAPABILITY		
Speech algorithm	Support fax tone detection and automatic switching G.711, 7.29, 7.23, 7.26 encoding	
Voice	G.711 A-law / U-law, G.722, G.723.1, G.726, G.729A / B, AAL2-G.726-32	
Video coding	H.264, H.263, H263 +	
QoS	Layer 3 QoS, Layer 2 QoS	
SIGNALING PROTOCOL		
DTMF detection method		Support three, In Audio, RFC2833, SIP INFO
Digital signal	PRI, SS7, MFC / R2	
Terminal management and automatic deployment capabilities	Support for SIP terminal plug-and-play auto-deployment (including DHCP Option 66 / multicast SIP SUBSCRIBE / MDNS Automatic detection and configuration), support TFTP / HTTP / HTTPS upgrade, local upgrade	
Network protocol	TCP / UDP / IP, RTP / RTCP, ICMP, ARP, DNS, DDNS, DHCP, NTP, TFTP, SSH, HTTP / HTTPS, SIP (RFC3261), STUN, SRTP, TLS / SIP, LDAP	
PSTN disconnection method	Support the PSTN line intelligent detection and automatic matching function, support five kinds of signaling detection methods (1, Busy Tone detection, 2, CPT detection, 3, polarity reversal detection, 4, Hook Flash Timing detection, 5, Loop Current Disconnect detection)	
SAFETY		
Media encryption	SRTP, TLS, HTTPS, SSH	
Advanced defense	Fail2ban, Alert events, whitelists, blacklists, encrypted access controls	
PHYSICAL CHARACTERISTICS		
Global universal power supply	Input: 100 ~ ~ 240VAC, 50 / 60Hz;	
weight	Equipment weight: 2.165 Kg; Package weight: 3.012 Kg	
Use the environment	Operating environment: 32 - 113°F / 0 ~ 45°C, humidity 10 - 90% (No condensation) Storage environment: 14 - 140°F / - 10 ~ ~ 60°C, humidity 10 - 90% (No condensation)	
size	440mm (L) x 185mm (W) x 44mm (H) IU standard	
installation	Desktop or cabinet installation	
ADDITIONAL FEATURES		
Multi-language support	Web user interface supports English / Simplified Chinese, customizable IVR / voice prompts in English, Chinese, english English	
Caller ID detection	ETSI-FSK, ETSI-DTMF	
Polarity reversal/ Wink	Support, option on / off control call setup and termination	
Call Center	Supports multiple call queues and queue queuing sounds	
Customize Auto Attendant	Supports Layer 3 IVR (Interactive Voice Response)	
Concurrent call capability	Up to 5,000 SIP terminals can be registered, up to 500 concurrent calls	
Meeting room capacity	Supports up to 10 voice conference rooms, up to 100 voice conference members	
Call function	Call Park, Call Forward, Call Forward, Do Not Disturb, DISA, ring Group, Pickup Group, blacklist, call / intercom, etc.	

KNTECH IP PABX MODEL NUMBER

SIMULTANEOUS CALLS

REGISTER USER	SIMULTANEOUS CALLS	MODEL NUMBER
50 line	15 line	KNTD-50
100 line	30 line	KNTD-100
300 line	90 line	KNTD-300
500 line	150 line	KNTD-500
800 line	400 line	KNTD-800
1000 line	300 line	KNTD-01K
3000 line	900 line	KNTD-03K
5000 line	1500 line	KNTD-05K