

### KNTD-50M

### UC features for up to 50 users

The KNTD-50M 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 SIP Trunk, RJ45 networks and offers scalability by supporting up to 50 users. This hication features at an unprecedented rise secure and reliable IP PBX delivers unified communi point it hout any licensing fees, costs-per-feature, or recurring fees. Host Standby N+1 redundancy System design.

KNTD-50M server is a computer device that centrally processes and manages information, with multiple functions such as storing, processing, sharing, and protecting data.

The KNTD-50M server can be equipped with three major systems: telephone, broadcasting, and access control, and has efficient and stable performance to meet your various needs.



#### 50 users Supports up to 50 users, 10 SIP trunkacCounts, up to 50concurrent calls.



Supports IVR (Interactive VoiceResponse)



Three-In-One Optional, Telephone system, Broadcasting system, Access control system.



Built-in call recordingserver recordingsaccessed via webuser interface



Strongest-possible security protectionusing HTTP



Supports call queue for efficient call volume management





### **Technical Parameter**

Hardware interface Host Standby N+1 redundancy System design  Network Interface RJ45 Gigabit interface (10/100 / 1000M Ethernet adaptive)  Reset switch Support, Web configuration  Network Interface RJ45 Gigabit interface (10/100 / 1000M Ethernet adaptive)  Audio / video capability  Speech algorithm G.711, 7.29, 7.23,7.26 encoding  Voice G.711 A-law/ U-law, G.7231, G.726, G.729A  Video coding H.264.  Signaling protocol  DTMF detection method Support RFC2833, SIP INFO			
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Signaling protocol			
DTMF detection method Support RFC2833, SIP INFO			
Digital signal SIP2.0			
Terminal management and automatic deployment capabilities  Support for SIP terminal plug -and-play auto-deployment (including DHCP / multicast SIP , support Remote upgrade,local upgrade)			
Network protocol TCP   UDP   IP, RTP DHCP,SSH,HTTP,SIP			
Safety			
Advanced defense Alert events, whitelists, blacklists			
Physical characteristics			
Global universal power supply Input: 100~ ~ 240VAC, 50 / 60Hz;			
weight Equipment weight: 3.5 Kg; Package weight: 4,4Kg			
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)			
Installation Desktop or cabinet installation			

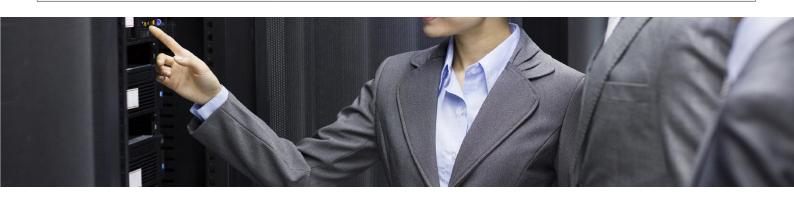


### Additional features

Multi language support	Web user interface supports English / Simplified Chinese, customizable IVR / voice prompts in English, Chinese
Caller ID	ETSI-DTMF
Call Center	Supports call queues.
Customize Auto	Supports  VR (Interactive Voice Response)
Concurrent call	Up to 50 SIP terminals can be registered, up to 10 concurrent calls
Meeting room	Supports up to 10 voice conference rooms, up to 100 voice conference members
members	Calls waiting, Call transfer, Ringing group, Subgroup, meeting, broadcast, blacklist, call / intercom, etc

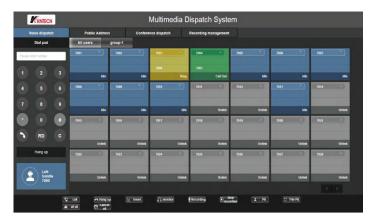
### KNTECHIP PABX MODEL NUMBER

REGISTER USER	SIMULTANEOUS CALLS	MODEL NUMBER
50 line	10 line	KNTD-50M
100 line	20 line	KNTD-100
300 line	60 line	KNTD-300
500 line	100 line	KNTD-500
800 line	160 line	KNTD-800
1000 line	200 line	KNTD-01K
3000 line	600 line	KNTD-03K
5000 line	1000 line	KNTD-05K



### Voice dispatch-Telephone call

Blue color means on line, green color means ringing, yellow color means busy.





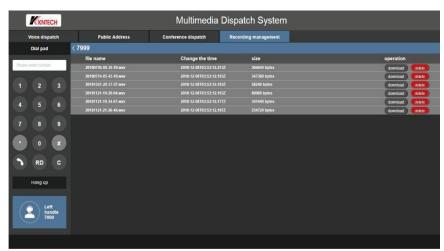
### Conference dispatch-Meeting room



### **Public Address**



### **Configuration Manual**



### Emergency telephone system

