







ANANT

Server-based Unified Communication Solution



Why Server-based PBX?

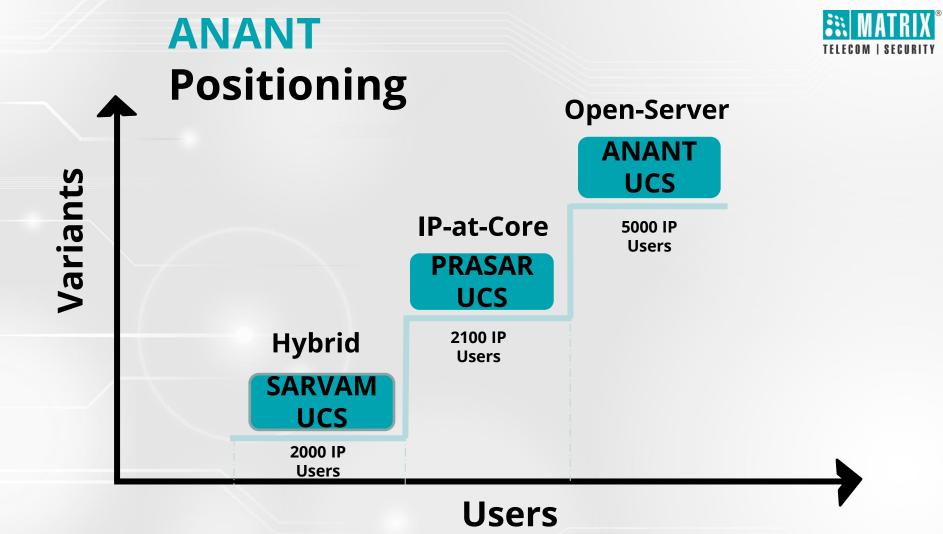
- Future-proof Technology
- Hardware Independent
- Scalable
- Reduced IT Infrastructure
- Optimum Resource Utilization



What is **ANANT UCS**?

A Unified Communication Solution that is

- Enterprise-grade
- Hardware Independent
- Deployable on any Reliable COTS Server
- Highly Scalable





Product Highlights

5000 IP Users

1024

512

Concurrent Calls Three Party Conference 64

Simultaneous Call Recording





Hardware Independence

- Hardware of Choice
- Industry Standard Server
- Maximum Resource Utilization

Key Features



Active-standby Redundancy

- Uninterrupted
 Communication
- Increased Productivity



Mobility

- Anywhere, Anytime Connectivity
- Faster Decision Making

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RAID1 and RAID5 Support

- Fault Tolerance
- Easy Data Recovery





Auto Sign-in for Softphone

- One Touch Configuration
- Reduced Installation Time

Key Features



LDAP Client Support

- Centralized Directory
- Easy Management



Built-in Auto-attendant and Voicemail

- No Missed Messages
- Customer Retention



Auto Sync Configuration

- Quick Configuration
- Smoother Operations





Integration with VoIP Gateways

- Integrated Solution
- Utilize Legacy Infrastructure

Key Features



Scheduled Backup of Voicemessages

- Secured Messages
- Hassle-free Back-up



Security with ANANT UCS



- SIP Security with TLS and SRTP
- Encrypted Communication
- Blacklisting Unauthorized Access
- Allow SIP Traffic only from a trusted source
- Password Ageing
- Role Based System Access and Management



Mobility with **ANANT UCS**



Voicemail to Email



Instant Messaging





Screen Sharing



Outlook Integration







Analog Phone

Flexible Phone Options





Android/iOS Softphone Clients

PC Soft Phones





Integration with **VoIP Gateways**

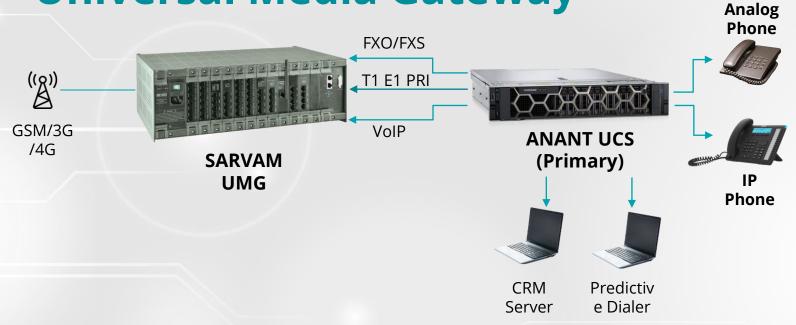


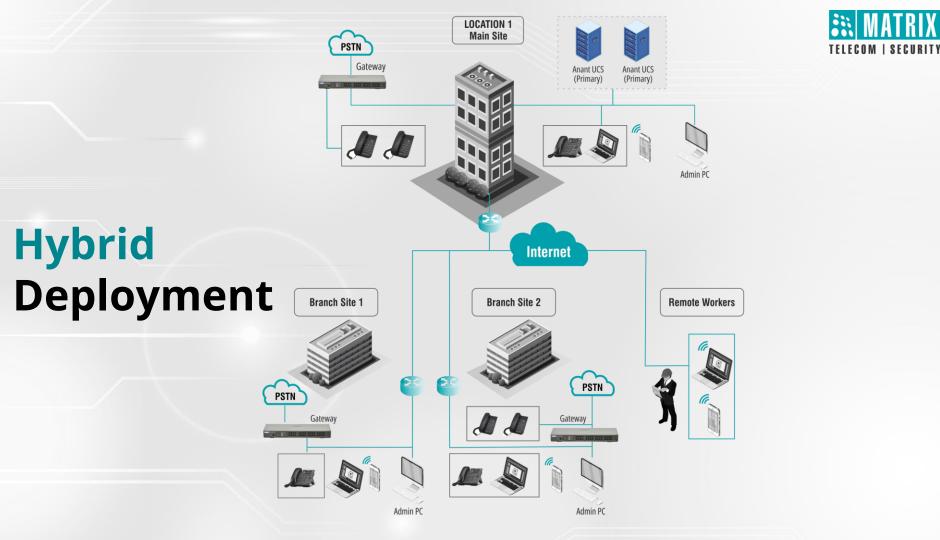
VoIP - PRI Gateway

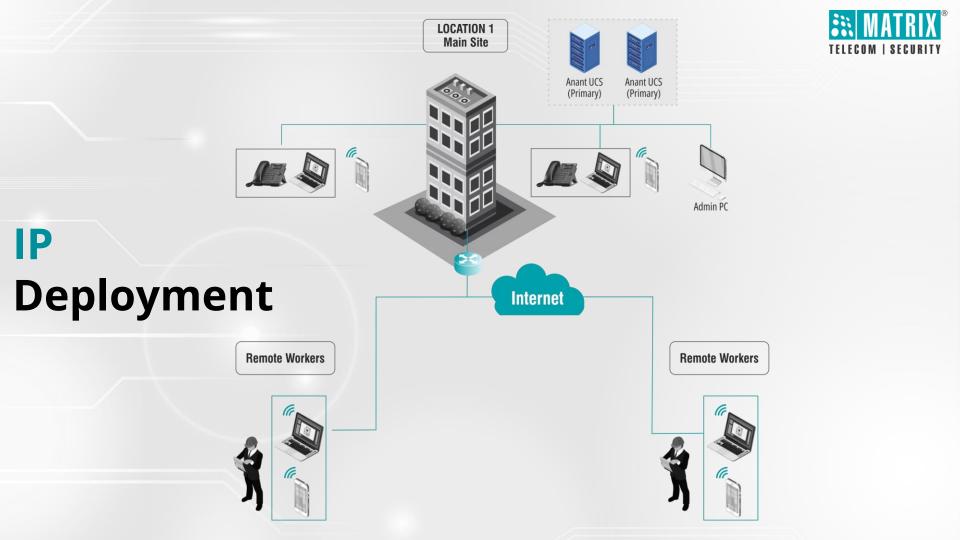
GSM - FCT Gateway



ANANT with Universal Media Gateway









Technical Specifications

Parameters	Details
SIP Extensions	5000
SIP Trunks	99
Maximum Transcoding Audio Calls	Based on Resources (Max. 1024)
Maximum DRTP/RTP Relay Audio Calls	1024
Maximum DRTP/RTP Relay Video Calls	102
Maximum Voicemail Channels	64
Maximum Conference Participants (System wide)	1536



Technical Specifications

Parameters	Details
Maximum 3-Party Conference	512
Maximum Participants in a Single Multi- Party or Dial-In Conference	64
Maximum Call Tapping or Recordings	64



Server Specifications

Parameters	Details
Server Model	Any
Processor Model	Intel Xeon E-2134 or above
CPU Speed	3.5 GHz or above
CPU Core/Thread	4/8
Max Turbo Frequency	4.5 GHz or above
Cache	8 MB Smart Cache
RAM	8 GB DDR4 ECC



Server Specifications

Parameters	Details
Hard Disk	1 TB (1 No.) - Without RAID Support 1 TB (2 Nos.) - With RAID1 Support 1 TB (3 Nos.) - With RAID5 Support
Form Factor	1U (RACK)
Raid Controller (Optional)	H330 Raid Controller*



Industrial Applications

Manufacturing

COSEC Door Integration
PA System Integration
Multi-location Connectivity

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TELECOM I SECURITY

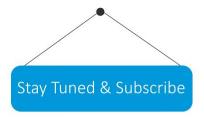
Large Enterprise

- 1024 Concurrent Calls
- Conference
- Central Call Processing and Management
- Hot Redundancy

TELECOM I SECURITY

Hospitality

- Intercom
- Auto-attendant
- Voicemail
- Integration with renowned PMS Software







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THANK YOU!



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