



ANANT UCS is an enterprise-grade high capacity, high durability, high scalability and high efficacy software-based communication solution. ANANT UCS offers a range of enterprise grade calling features that improves employees' productivity and speeds up the decision-making process. With an in-call redundant system, ANANT UCS boosts of providing the best customer engagement. The hardware-independent platform provides customers with a choice to select a reliable server as per his requirement.

| TECHNICAL SPECIFICATIONS | |
|---|---|
| Туре | IP at core |
| Signaling | SIP |
| SIP Extensions | 5000 |
| SIP Trunks | 99 |
| Max. DRTP/RTP Relay Audio Calls | 1024 |
| Max. Simultaneous Transcoding Audio Calls | 1024 |
| Max. DRTP/RTP Relay Video Calls | 102 |
| Max. Three-party Conference | 512 |
| Max. Conference Participants (System Wide) | 1536 |
| Max. Participants in a Single Conference | 64 |
| Max. Voice Mail Channels | 64 |
| Max. Call Recording/Tapping | 64 |
| Redundancy | 1+1 Active-Stand-by Redundancy |
| Supported IETF RFCs | 3261, 3262, 3389, 3550, 3551, 3311, 3265, 4733, 2915, 2916, 2327, 2833, 2806, 768, 793, 1034 |
| Configuration | Web-based Intuitive GUI |
| Security | MD5 Algorithm for SIP TLS and SRTP over SIP Password Encrypted with AES128 Algorithm in Configuration Files |
| Web Browser compatibility | ChromeFirefoxInternet Explorer |

| | VOIP |
|---------------------|--|
| Туре | Built-in Software |
| Network Protocol | IPv6, IPv4, TCP, UDP, SNTP, STUN, ARP, ICMP, PPP, DNS, SMTP |
| SIP | Maximum 99 SIP Accounts per System, Out Bound Proxy Support, Display Name, User Name, Password, URL, Proxy URL, Register Interval |
| Voice | Dynamic Jitter Buffer (Adaptive), Comfort Noise Generation and Voice Activity Detection |
| NAT | STUN and NAT Keep Alive |
| Voice CODECs | G.729, G.711 u-law, G.711 A-law, iLBC – 30 ms, iLBC – 20 ms, G.722 |
| Call Progress Tones | Dial tone, Ring Back Tone, Busy Tone, Error Tone, Call Waiting Tone, Confirmation Tone, Feature Tone, Programming Tone, |
| Quality of Service | Support of Layer 2/3 CoS/QoS for SIP Support of Layer 3 QoS for RTP |

| GENERIC SERVER SPECIFICATIONS | |
|-------------------------------|-------------------------------------|
| Server Manufacturer | Any |
| Processor | Intel Xeon E-2134 or above |
| CPU Speed | 3.5 GHz or above |
| CPU Core/Thread | 4/8 |
| RAM | 8 GB DDR4 ECC |
| Hard Disk | 1 TB (1 No.) - Without RAID Support |
| | 1 TB (2 Nos.) - With RAID1 Support |
| | 1 TB (3 Nos.) - With RAID5 Support |
| NIC | 1 Gbps (2 Nos.) |

| VOICEMAIL | |
|-----------|-------------------|
| | Built-in Software |

Туре

| VMS Channels | 64 |
|-----------------|---|
| Voicemail Box | Dedicated Mail Box for each Extension |
| Voice Recording | Recording up to 37,316 hours with 1 TB Hard |
| | Drive |

ANANT UCS PLATFORM - Features:

- Abbreviated Dialing (Global and Personal)
- Access Codes
- Account Codes
- Alarms
- Alternate Number Dialing
- Apple Push Notification Service Support
- Auto Call Back
- Auto Configuration for SIP Phones and UC Clients
- Auto Redial
- Auto-attendant
- Automatic Number Translation
- Authority Codes
- Background Music (DKP and SLT)
- Backup SMDR
- Backup System Configuration
- Backup System Software
- Barge-in
- Busy Lamp Field for Trunks
- Call Back on Trunk Ports
- Call Budget on Extensions
- Call Budget on Trunks
- Call Chaining
- Call Cost Calculation and Display
- Call Duration Control and Display
- Call Forward
- Call Park
- Call Pick Up
- Call Progress Tones
- Call Taping
- Call Transfer
- Call Toggle
- Calling Line Identification and Presentation (CLIP)

SUPPORTED MATRIX IP PHONES AND SOFT CLIENTS

SPARSH VP310E

- Calling Line Identity Restriction (CLIR)
- Cancel All Station Features
- Class of Service (COS)
- CLI based Routing
- Closed User Group
- Conference Multiple Participants
- Conference Dial-in
- Conflict Dialing
- Continued Dialing
- Conversation Recording
- COSEC Integration
- Customer Name
- Date and Time Format
- Daylight Saving Time (DST)
- Day Night Mode
- Department Call

- Dial By Name
- Dialed Number Directory
- Dial Plan for SIP Extension
- Digest Authentication (on SIP)
- Direct Dialing-in (DDI)
- Direct Inward System Access (DISA)
- Direct Station Selection Console (DSS Console)
- Distinctive Rings
- Do-Not-Disturb (DND)
- DSS Call Pick-up
- Dynamic Lock
- Email Notification (VMS)
- Emergency Calls Detection and Reporting
- Emergency Conference
- Emergency Number Dialing
- Emergency Calls (911) Reporting to PSAP
- Extended IP Phone/VARTA UC Client
- Firebase Cloud Messaging (FCM) Support
- Flexible Numbers (Up to 6 Digits)
- Floor Service

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- Follow Me
- Forced Answer
- Forced Call Disconnection
- Help Desk
- Handover and Handoff
- Hot Desking
- Hotline (Immediate and With Delay)
- Holiday Table
- IM (Chat)
- Installation Wizard
- Incoming CLI Modification
- Intercom
- Internal Call Restriction
- Interrupt Request
- IPv6 Ready
- Last Caller Recall
- Last Number Redial
- Least Cost Routing (Carrier Pre-Selection)
- License Management
- Lightweight Directory Access Protocol (LDAP)
- Live Call Supervision
- Logical Partitioning
- Meet Me Paging
- Message Wait Indication
- Mobility Extension
- Multi-stage Dialing
- Music-On-Hold
- Mute

Number Lists

Paging

Privacy

Quick Dial

Reminder

Raid

Redundancy

Real Time Clock

Room Monitor

Routing Group

Self Ring Test

SMTP Settings

Station Groups

System Security

Time Zone Display

Trunk Landing Group

User Absent/Present

Time Tables

Toll Control

User Password

Virtual Extensions

Web based Programming

Video Calling

Voicemail

SPARSH VP510E

Response Mapping

Selective Port Access

Shared Call Appearance

SIP and RTP QoS (VOIP)

SIP over TCP and TLSS

Static Routing Table

Station Message Detail Record

System Activity Log and Display

System Engineer (SE) Mode

System Fault Log and Display

System Administrator (SA) Mode

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NAT and STUN (VoIP)

Peer-to-Peer Calling

Preset Call Forward

Priority (Intercom and Trunk)

Return Call to Original Caller (RCOC)

Simple Network Management Protocol (SNMP)

SMDR Posting (Call Accounting System Interface)

Walk-in Class of Service (Single/Multiple calls)

SPARSH VP710

Simple Network Time Protocol (SNTP)

Presence Sharing

PIN Dialing

One Touch TransferOnline SMDR



SPARSH VP110

| | ANANT UCS PLATFORM |
|----------------------|--|
| ANANT UCS PLATFORM | License for Unified Communication Server for Open Server Platform |
| ANANT VMS CHNL4 | License for Voicemail Channels for ANANT UCS to support 4/16/32 simultaneous voice mail sessions |
| ANANT VMS CHNL16 | |
| ANANT VMS CHNL32 | |
| ANANT IPSUB10 | License of IP Subscribers for ANANT UCS to create 5/10/50/100/500 VoIP subscribers |
| ANANT IPSUB50 | |
| ANANT IPSUB100 | |
| ANANT IPSUB500 | |
| ANANT VARTA USER5E | License for 5/10/50/100 VARTA Soft Clients with Essential features. ANANT UCS needs this license to register Android/iOS/Windows Desktop |
| ANANT VARTA USER10E | UCS Clients. This license is not required for Hardware IP-Phones. |
| ANANT VARTA USER50E | |
| ANANT VARTA USER100E | |
| ANANT VARTA USER5P | License for 5/10/50/100 VARTA Soft Clients with Professional features. ANANT UCS needs this license to register Android/iOS/Windows |
| ANANT VARTA USER10P | Desktop UCS Clients. This license is not required for Hardware IP-Phones. |
| ANANT VARTA USER50P | |
| ANANT VARTA USER100P | |

SPARSH VP330E

| License for 5/10/50/100 VARTA Soft Clients with Premium Collaboration features. ANANT UCS needs this license to register |
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| Android/iOS/Windows Desktop UCS Clients. This license is not required for Hardware IP-Phones. |
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| License for 8/16/32 Conference Channels for larger conferences. ANANT UCS needs this license for more number of participants in a single |
| conference. |
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| License to enable Hospitality functions suite for 10/50/100/500 users in ANANT UCS to support hospitality functions and features used in |
| Hotel-Motel. |
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| License to enable Property Management System interface for ANANT UCS to integrate with third party Property Management System (PMS) |
| used in Hotel-Motel. |
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| License to enable Direct Dialing of Emergency Number 911 without any prefix for 10/50/100/500 users in ANANT UCS. Applicable only for US |
| market. |
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| |
| License to enable Redundancy functionality for 10/50/100/500 users in ANANT UCS. Additional (redundant) server hardware is required for |
| Redundancy function to work. No license is required to be activated on the redundant hardware for Redundancy functionality. |
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| License for Annual Upgradation Package (AUP) for 10/50/100/500 Users valid for 12 months. This license is for software support. All system |
| will come with built in 1-year software support license. During support period, all the releases/ upgrades/ patches release for the product |
| shall be available to the customer free of cost. |
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Note: Dell Server can be ordered at Matrix. Refer pricelist for specific product requirement.

ABOUT MATRIX

Established in 1991, Matrix is a leader in security and telecom solutions for modern businesses and enterprises. Matrix, an innovative, technology driven and customer focused organization, is committed to keep pace with the revolutions in security and telecom industries. With more than 40% of its human resources dedicated to the development of new products, Matrix has launched cutting-edge products such as Unified Communications, IP-PBX, Universal Gateways, Convergence, VOIP Gateways, GSM Gateways, IP Video Surveillance, Access Control and Time-Attendance. These solutions are feature-rich, reliable and conform to the international standards. Having global footprints in Europe, North America, South America, Africa and Asia through an extensive network of more than 1000 system integrators, Matrix ensures that the products serve the needs of its customers faster and longer. Matrix has gained trust and admiration of customers representing the entire spectrum of industries. Matrix has won many national and international awards for its innovative products.



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