

UC2000-VF Multi-SIM GSM&3G<E VoIP Gateway

Overview

Dinstar UC2000-VF GSM/WCDMA/LTE VoIP Gateway is a multi-functional product used to effectively implement the smooth transition between PLMN (Mobile) and VoIP network. The UC2000-VF gateway is designed for comfort, ease-of-use to satisfied necessary of customers. It is not only inherits tradition of quality voice communication over IP network but reduce the human resource of VoIP network deployment.

UC2000-VF



It's a cost-effective gateway for call termination (VoIP to GSM) and origination (GSM to VoIP). It is fully compatible with leading Soft-switch and SIP server.

Key Features

- 16 Mobile Channels with 64 SIM Slots, Easy to Conduct SIM Rotation Locally
- Supports to work with different mobile networks:
- Compatible with LTE-FDD/LTE-TDD/WCDMA/GSM Networks
- Supports Make Call through VoLTE
- Auto Call/SMS Generation
- Auto Update SIM Number and Balance Check
- No Balance Alarm and Call Minutes Limitation



Voice Processing

Protocol: SIP v2.0, RFC3261, SIP TLS Codecs: G.711A/U law, G.723.1, G.729A/B Silence Suppression Comfort Noise Generator(CNG) Voice Activity Detection(VAD) Echo Cancellation: G.168 with up to 128ms Dynamic Jitter Buffer DTMF: RFC2833, SIP Info Call Progress Tone Generation Programmable Gain Control NAT: STUN, Dynamic NAT, Static NAT

Frequency Bands

Type A. FDD: B2/B4/B12 WCDMA: B2/B4/B5 Type E. FDD:B1/B3/B5/B7/B8/B20 TDD: B38/B40/B41 WCDMA: B1/B5/B8 GSM:900/1800MHz Type V. FDD:B4/B13 Type J. FDD:B1/B3/B8/B18/B19/B26 TDD:B41 WCDMA:B1/B6/B8/B19 Type AU. FDD:B1/B2/B3/B4/B5/B7/B8/B28 TDD:B40 WCDMA: B1/B2/B5/B8 GSM: 850/900/1800/1900 * UC2000-VF supports different frequency bands, please choose a dedicated type based on the region or operator.

Software Features

Calling Type:

Mobile to VoIP. VoIP to Mobile SIP Trunk and Trunk Group Caller/Called Number Manipulation Rules White/Black List **PSTN Hotline VoIP Hotline** Abnormal Call Monitor **Balance Check** Low Balance Alarm Auto Update SIM Number Random Call Interval Auto CLIP Signaling & RTP Encryption SMPP for SMS SIM Rotating by SIM Run Time

SIM Balance

Maintenance

About Us

Web/Telnet Configuration Configure Backup/Restore Firmware Upgrade by HTTP/TFTP Factory Reset CDR (10000 Lines CDRs Storage Locally) Syslog/Filelog Ping and Tracert Traffic statistics: TCP,UDP,RTP VoIP Call Statistics PSTN Call statistics: ASR, ACD, PDD Voice Diagnose with Loopback GSM Network Test **IVR** Customization Auto Provisioning SIP/RTP/PCM Capture

Mobile

USSD Encoding: Unicode/ASCII HTTP API for SMS Polarity Reversal PIN Modify CID/CLIR SMS Send and Receive SMS to Email Call Waiting/Call Back Call Forward **GSM Audio Coding:** HR, FR, EFR, AMR_FR, AMR_HR

Network

Static IP, DHCP Client, PPPoE

Network Protocols:

TCP/UDP,TFTP, FTP, RTP/ RTCP, SSH,ARP,RARP,ICMP Ping, NTP, HTTP, DNS Static/Dynamic ARP Request

Physical Interface

Mobile: 16 Channels

Antennas Connectors: SMA

Ethernet Interfaces:

2*10/100 Base-T, RJ45

LED Indicators:

Power, RUN, Signal Strength Indicators

Console:

1* RS232 115200bps

Environmental

Power Supply:100-240V,50-60Hz Power Consumption: 38w Temperature Operation: 0 °C ~ 45 °C Storage: -20 °C ~80 °C Operation Humidity: 10%-90% No Condensation Dimension(W/D/H): 440×270×44mm Weight: 4.7 kg Unit Weight: 3.8 kg

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Founded in 2011 in Shenzhen, DINSTAR is a leading global provider of IP Unified Communication products including VoIP Gateways, IP PBXs, IP Phones and SBCs, we have been delivering more agile, efficient and affordable communication solutions and unparalleled communication experiences to our customers with our reliable, innovative and future-proof products for years. Through our value-added distributors and resellers worldwide, now DINSTAR serves telecom operators, service providers, system integrators, enterprises, SMBs and OEM partners in over 100 countries.