




We change the shape of the world

| Featurelist NovaTec S3 / Cisco VG-2BRI |  |
|---|---|
| Width x Depth x Height | Desktop or wall-mounted chassis: 211 x 155 x 40 mm |
| Approximate weight | 1 up to 2 kg |
| Input power supply | 90 to 264 V~ |
| PRI connection | No |
| Max. number of BRI (S/T) interfaces, non-blocking | 2 |
| Max. analogue subscriber line (optional) according to ITU-T G.712 and Q.552 | 2 |
| Max. number of U_{P0} interfaces (optional) | 2 |
| Number of media converter for VoIP | 4 |
| Transparent ISDN over IP | Yes |
| SIP (Session Initiation Protocol) <ul style="list-style-type: none"> • RFC 3261 (UDP, TCP-IP support) • Digest Authentication (RFC2617) • Reliability of provisional responses (RFC3262) • Offer/answer (RFC3264) • STUN client and server (NAT traversal RFC3489) • Firewall NAT routing (fixed tables) • In-Band/Out-of-Band DTMF (RFC2833) • Session timer (RFC4028) • SIP Extensions for Caller Identity and Privacy • SIP media inactivity Timer • ITU-T T.38 fax • E164 (RFC3824) • DNS name resolution (RFC 1034, 1035) • ENUM (RFC 3761) • HTTP(S) Server for diagnostics, statistics and trace SIP Call Features <ul style="list-style-type: none"> • Hold • Call Transfer • Music on Hold • ISDN Calling Name Security <ul style="list-style-type: none"> • sRTP for speech and data (RFC3711) • sRTP fallback to RTP and RTP to sRTP (configurable) • X505v3 certificates can be loaded, created (RFC 5280) | Yes |
| SIP trunking with Cisco Communications Manager from release 7.1.2 on as Cisco end device and with Cisco ISR for data and language applications. | Yes |
| Budget assignment per user with user blockage when budget is spent | Yes |
| E-Mail message incl. CLIP information in case of no answer on the mobile, fixed-line or IP phone (direct response or callback possible) | Yes |
| Offer/answer model with Session Description Protocol (SDP) | Yes |
| Proxy usage (outgoing call) | Yes |
| Remote configuration via IP & ISDN | Yes |
| Hold and off hold & Call transfer | Yes |
| Forwarding: forward on busy, unconditional forward and forward on no answer | Yes |
| Parallel and serial forking | Yes |



| Featurelist | S3 Versions / VG-2BRI-R |
|---|-------------------------|
| Re-INVITE: codec change, hold, off hold, session timer | Yes |
| Supported header field "Supported" | Yes |
| Number Mapping | Yes |
| Common DDI for mobile, fixed-line and IP phone | Yes |
| ISDN cause codes in call detail records | Yes |
| SIP cause codes in call detail records | Yes |
| T.38-Fax: Fax over IP according to ITU-T recommendation T.38 and Annex D | Yes |
| Codec and Speech Compression <ul style="list-style-type: none"> • G.711 incl. Annex I (BFI) and Annex II (VAD/CNG) • G.726 incl. VAD/CNG, BFI Error Concealment and Payload Support RTP according to "RFC 3551" • G.728, 16 kbit/s • G.729 A/B, 8 kbit/s • Fax Relay, T.38 Support V.21, V.27ter, V.29 and V.17 • 30 ms Voice Packet Size (all Codecs, upstream) • Adaptive / Fix Jitter Buffer max. 200 msec • Jitter Buffer Inband Modem Support • RTP/sRTP protocol support according to RFC3550 and RFC3711 Payload Byte Counter (H248.1 Annex E) • X-CCD & Clear Mode for data transmission. • Silence Compression • Comfort Noise Generation | Yes |
| Analogue Signalling <ul style="list-style-type: none"> • The Near Line Echo Canceller (16 msec) is compatible with the available ITU-T G.165 and G.168 standards. • Caller ID Sender (CIDS), V.23 and Bel202 • Caller ID Receiver (CIDR), V.23 and Bel202 • DTMF/AT Generator • DTMF Receiver (DTMFR) according to ITU-T Q.23. • Universal Tone Generator (UTG) • Universal Tone Detector (UTD) according to ITU-T V.8 • Text Phone V.18 A Detector • Call Progress Tone Detector (CPTD) • Answering Tone Detector (ATD) • Digital Identification Signal (DIS) V.21 Detector • DTMF Event Support according to RFC2833 | Yes |
| VLAN (Virtual LAN) according to 802.1Q | Yes |
| MD5 authentication | Yes |
| NAT mapping tables | Yes |
| STUN Server / Client | Yes |
| HTTP diagnostics interface | Yes |
| User accounting | Yes |
| Remote maintenance & diagnostics | Yes |
| Local time from ISDN | Yes |
| NTP: UTC Server access for automatic time setting | Yes |
| Summer / winter time | Yes |
| Static IP / Dynamic IP | Yes |
| Basic & supplementary call set up | Yes |
| Session timer as described in the IETF Internet Draft <i>ietf sip session timer-04.txt</i> | Yes |



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| Featurelist | NovaTec / TransNova® SIP Gateway S3 |
|--|--|
| Public and private IP address support for use with NAT firewalls | Yes |
| Configurable SIP and RTP/RTCP ports | Yes |
| Encryption (SRTP) AES 128 Key (activation through NovaTec necessary) | Yes |
| Encryption (TLS): (activation through NovaTec necessary) <ul style="list-style-type: none">• TLS1.0 / SSL3.0 according to RFC2246• RC2, RC4 DES and AES according to RFC3268• X509v3 certificates according to RFC3647• RSA key exchange for asymmetric crypto system | Yes |
| SNMP <ul style="list-style-type: none">• Presently SNMP-V1 and SNMP-V2C without traps is supported.• SNMP Mapper additionally supports SNMP-V3 and V3-Traps. | Yes |
| Ethernet 802.3 (10 BASE-T & 100 BASE-TX) | Yes |
| Ethernet half duplex | Yes |
| Ethernet full duplex | Yes |
| Synchronization with RMCS | Only 1F8000-R |
| RMCS server functionality | Only 1F8000-R |