

# V9 Featurelist

## PBX System Features

*For all innovaphone products based on V7, V8 and V9*

- Access to exchange via group number
- Access to separate exchange via individual number
- ACD groups
- Active Directory Replication
- Administration via uniform interface
- Alarm and error events
- Alternate route selection
- Announcement before pickup (\*)
- Announcements (\*)
- Authorisation groups (class of service)
- Automatic break through for call protection for preferred numbers
- Automatic call distribution (ACD)
- Automatic night switching with week clock
- Automatic route selection (ARS)
- Bootcode with minifirmware (not supported with IP3000)
- Branches (virtual central PBX)
- Bunched circuit
- Call back
- Call back also if internal call
- Call detail recording (CDR)
- Calling line identification restriction (CLIR)
- Calling party identification (CLIP)
- Calling to E.164
- Calling via ENUM (H.323 and SIP)
- Centralized backup / firmware / config upload

- Circle call with circle call query (#)
- Clock synchronization via IP
- Config templates for PBX User
- Configurable tones
- Configure / change / delete call diversion via web
- Conversation return to operator
- Cross connection via IP
- Cross connection via ISDN
- Different call (external, internal)
- Direct dial (with and without number reassessment)
- Door bell interface with door opener function (#)
- DTMF detection disabling
- DTMF-post dial/direct dial
- Emigration of endpoint
- Enhanced error reporting (SIP, QOS parameter, router error reporting)
- Enhanced H.323 features according to H.450.x
- Extended DTMF Object (feature set via DTMF dial)
- External communication
- Firmware upload in DRAM (not supported with IP3000)
- Floating license model
- Flexible dial plan
- General night switching
- Group call (all, circular)
- Hide Objects from LDAP
- Home office integration
- HTTP / HTTPS allowed stations
- Incoming direct call to subscriber or group
- Interactive call acceptance (\*)
- Interface for external announcement construction (#)
- Internal communication

- Internal music on hold
- LDAP replication
- Least cost routing (LCR)
- Log forwarding
- Message center (H.450.7 or DTMF)
- Message Waiting Indication (MWI)
- Messaging (SIP and H.323)
- Mixed dial method (overlap and en-block dialing)
- MS-chap V1/V2
- Multi conferencing (IP800, IP6000, IP0010, IP1060, IP3010, IP6010)
- Multi protocol (H.323, SIP)
- Multilevel admin access (detailed multilevel user rights for the PBX configuration)
- Multiple system admin (multiple admin and viewer rights possible for the whole system)
- Multistage login (multi-user)
- Music / announcement in standby from analogue resource (#)
- Music / announcement in standby from HTTP resource (\*)
- Network wide features (Master/slave with virtual PBX)
- Only calling to registered subscribers (passive control)
- Overflow of subscribers and cross connection
- Overlapped sending
- PBX directory search (LDAP search via DTMF)
- Phonebook either from internal or from external source (LDAP)
- QSIG
- QSIG inter-working
- QSIG trunking (point to point and point to multipoint)
- Redundancy of the ISDN-lines
- Redundancy of the PBX
- Remote monitoring and remote diagnostics
- Remote software upgrade
- RTP Proxy

- SDP Transit (for video and other codecs)
- Short dial (individual, group, system wide)
- Silent intrusion
- SIP/H.323 Medial Relay (RTP stream routing via gateway)
- SIP trunking
- SIP via TCP
- SIPS (SIP-Secure) – SIP via TCP/TLS (not supported with IP3000)
- SNMP Traps (alarm events)
- SOAP-API
- SRTP (Voice encryption) for H.323 and SIP (not supported with IP3000)
- Subscriber authorisation
- Standard protocols to the endpoint
- Subscriber groups with progressive call (parallel, circular, sequential)
- Switch between day/night with code digit from subscribers
- System time/date via timeserver (LAN or web)
- TAPI / Multi line TAPI (via several PBX's / locations)
- Telephone number assignment
- Time based subscriber authorisation
- Twin Phones
- US standard (T1 and CAS)
- VoIP-CAPI
- Waiting queue – call busy operators / CFU disable operators
- 802.1x

## New in V8

- Announcing the queue position in a waiting queue
- Call completion on busy/free to ISDN
- Customized Visual Appearance (individually designable user interface)
- Dual Forking (OCS Integration)
- Dynamic PBX (more PBXs on one Device ) PBX hosting / PBX virtualization

- Kerberos based authentication and Cross Realm authentication
- Mobility (GSM integration) - Call-waiting - hold - transfer - park - group login/logout - mobility on/off - setting presence status
- Off load Dialtone (external gateway)
- Off load Media Relay (external gateway)
- Off load MOH (external gateway)
- Presence
- Recording at the ISDN and analogue interfaces
- Roaming Phone Profiles
- RSTP (Rapid Spanning Tree)
- Send Message to group
- SIP error detection (alarm/event updated)
- SIP Federation
- SIP video multisite support

## New in V9

- Executive / secretary enhancements
  - Secretary can initiate calls with CLIP from the executive
  - Executive can see if the secretary is joined or unjoined in the primary secretary group
  - Executive sees if a call diversion is set at the secretary
  - Breakthrough disabled call waiting at secretary
- innovaphone Reporting
  - Based on internal or external Linux Application Platform
  - Based on PBX CDR's
  - Predefined filter
  - Reports (export as PDF, XML) from filter, combined filter, date and time
- UC client (myPBX) web-based and Windows launcher

Languages supported: Danish, English, Estonian, Finnish, French, Italian, Latvian, Dutch, Norwegian, Polish, Portuguese, Spanish, Czech – additional languages on demand.

- CTI (initiate call, accept call, terminate call, hold call, retrieve call, transfer call, park call, 3PTY conference)
- Manage own diversions
- Call list
- Presence (manage own presence state)
- Buddylist (including partners presence and call status)
- Chat (multiparty chat)
- Audioconference (via innovaphone audio conference Interface)
- Videoconference (via integration of Webex or Microsoft OLM)
- Application sharing (via integration of Webex or Microsoft OIM)
- Emergency call priority
- Enhanced Boolean (call forward, filter and diversion filter) based also on date
- Import/Export PBX user database (CSV/XML)
- Individual MOH (context sensitive based on object, node or PBX)
- IP filters for registrations with or without password
- IPv6 support (IPv6 routing, 6to4 routing and HTTP access, telnet)
- LDAP via TLS
- Maximum call duration
- MTLS
- NTP redundancy
- Partnerkeys via separate PBX's
- Wideband audio

## Features IP Phones – IP110 / IP200 / IP230 / IP240 / IP240-1000 / IP241 / IP222 / IP232

- Adjustable loudness
- Adjustable ringing tones also possible, import RTTTL and midi files
- Alphanumeric phonebook
- Announcement via loudspeaker
- Automatic callback on busy and free
- Automatic redialing
- Automatic redialing to busy bunched circuit

- Blind transfer
- Call diversion to internal phone number
- Call hold / retrieve call diversion to external phone number
- Call intrusion
- Call list (outgoing, incoming, combined, not reached, not answered)
- Call park and enquiry from other subscribers
- Call prevention
- Call transfer / pickup (selective and from a group)
- Call waiting (acoustical and optical signaling of waiting calls)
- Caller identification (name/number)
- Changeover of the assigned partners
- Conference (3 way)
- Configuration of the voice recording via HTTP
- Connect to busy and free subscribers without notification
- Counter and list of missed calls
- Date and time display (automatic summer/winter time adjustment)
- Delete call back
- Delete variable call diversion
- Dialing an IP address (with and without RAS)
- Dialing with code key from an unauthorized phone
- Direct dialing (immediate, internal/external)
- Display duration of call
- Display calling number
- Emigration of subscriber/endpoint
- Fixed call diversion (unconditional, no answer (time adjustable) on busy)
- Forced announcement (priority call)
- Function keys – multitude possible – exact list can be found in the technical specification of each IP phone
- Hands free
- Headset
- Individual configuration

- Individual phonebook
- Intercom with automatic call pick up
- Interlinked and limitless multiple conference
- Manager/assistant feature
- Message waiting indication (MWI)
- Messaging
- Multi language
- Multi-protocol and multi-registration (SIP and H.323 concurrent)
- Multiple registrations with selective de-registration
- Mute (switch off the microphone)
- Name dialing
- Open listening
- Over plugging (connect call to a busy subscriber)
- Phone lock
- Preferred subscribers
- Project code
- Redialing of the last 100 dialed numbers
- Silent Intrusion
- SIP via TCP
- SIPS (Secure SIP via TCP/TLS)
- SRTP (SIP and H.323)
- Status indication of the subscriber (free/busy/call)
- Subscriber with several lines
- Subscriber with several numbers
- Transfer with consultation
- Variable call diversion
- 802.1x



## New in V9

- Colour display (IP241 / IP222 / IP232)
- Individual background picture (IP241 / IP222 / IP232)
- Individual audible signal for announcement calls / multicast announcements and messaging
- Wideband audio (IP222 / IP232)
- 802.3az (IP222 / IP232)

## Features: 3<sup>rd</sup> party Phones connected to analogue (POTS) port, DECT phones IP52 / IP54 / IP55 / IP61 / IP63 / IP65

- Call completion on busy
- Call completion on free
- Call diversion to internal/external number
- Call hold/retrieve
- Call park and enquiry from other subscribers
- Call protection
- Call transfer/pickup (selective and from a group)
- Call waiting (acoustic and optical signal of the waiting call)
- Clip to an analogue phone
- Confirmation tone (positive and negative)
- Connect to a free and busy subscriber without notification
- Delete variable call diversion
- Dial with code key from an unauthorized phone
- Direct dial immediate (internal, external)
- Do not disturb (internal and external calls)
- Fixed call diversion; unconditional, no answer (time adjustable), on busy
- Lock phone
- Message Waiting Indication (MWI) on DECT Phones
- Preferred subscriber
- Project code
- SIP via TCP

- Special tone when active feature
- Transfer with consultation

## Features innovaphone Voice mail (\*)

*Languages supported: German / English / Italian / French / Dutch / Spanish / Norwegian / Danish / Polish / Swedish Czech – additional languages on demand*

*Voicemail runs on Compact Flash (CF slot in IP800, IP305, IP302, IP2000, IP6000, IP1060, IP3010, IP6010) or on external Web Server*

- Email notification (with or without attachment)
- Listen to /delete / store voice mails by means of DTMF input
- MWI (according to H.450.7)
- Personal greeting message
- Voice recording

## News in V9

- Voicemail support SRTP

## Features Operator (V8) software based

*Localized versions: German / English / Italian / Dutch – additional languages on demand*

- Busylampfield (divided into groups)
- Call journal
- Call transfer (with/without consultation (blind))
- Drag&Drop or with keypad use
- Integrated help
- Monitor blind transferred calls
- Monitor of waiting queue
- Name display for incoming calls in LDAP directory
- Park/unpark
- Send messages to PBX users
- Searching within a LDAP directory
- Searching for PBX users in all PBX's

- Set/change call forward for all PBX users
- Set/change presence status for all PBX users
- Show busy state
- Show call forward
- Show current call
- Show presence status
- State-of-the-art user interface
- Support of Master/Slave scenarios

## Features Interactive Voice Response (IVR) (\*)

- Announcements depending on the caller (e.g. automatic or manual language selection)
- Call handling according to lines and/or calling number (CLI)
- Listen to the menu also from internal subscribers
- Multi level menu
- Overflow from the operator to an automatic operator
- Voice menu at extension (if no answer, busy, unconditional)
- Voice menu with information and automatic dialing
- Voice menu with predefined destinations
- Voice menu with manual dialing of the destination (DTMF)

## New in V8

- Announcing the queue position in a waiting queue

Note:

- (\*) External or internal (CF-card) web server required
- (#) innovaphone IP21 required