

Mobile Dialer (Softphone)

SIP/VoIP softphone has these features:

- Superior quality two-way calls to fixed and mobile telephone subscribers and PC users
- Calling to PSTN or CS, SIP buddy or dial an IP address
- Free voice and video IP calls
- Conferencing of multiple users
- Create, share and maintain a phone book
- Access to a computer-based voicemail box via Web browser
- Seamless roaming between Wi-Fi, GPRS (UTMS), GSM and 3G
- Send DTMF or the so-called touch tones (full support, incl. SIP INFO DTMF and RFC 2833)
- Instant/automated redialing
- Access to the account balance information

Softphone features:

- Volume of sound shown as an animated bar
- Volume control (the one that you hear)
- Voice auto-gain control
- Microphone volume and muting controls (the way you are heard by your partner)
- Ring-tone volume and muting
- Voicemail captures incoming messages if the user is offline
- Voicemail interface with onscreen buttons: Play, Delete
- Voicemail integration
- Do-not-disturb (DND)
- Last-number redial
- Auto answer
- Dialing via the computer keyboard or screen dial pad (GUI),
- Transfer of incoming calls when the user is offline, including a "new message received" indicator
- Automated initiation of outgoing telephone calls via "autodial" and the user's database of contacts
- Call forwarding: route your incoming calls to a mobile/cell phone, landline or IP-based destination
- Works on all PCs with a soundcard and headset/speakers/microphone
- Conference calling can involve up to 20 callers in one session



Possible modifications:

- Rebranded Look
- Own Logo
- Personalized Icon
- Hardcoded SIP server
- and many other functions!



Supported platforms (OS):

- Windows 2000/XP/Vista/7
- Android
- Symbian
- Windows Mobile

Supported protocols:

- SIP proxy authorization support and HTTP digest authentication (RFC 2617)
- Intelligent handling of NAT and firewall setups, which works reliably over SIP and RTP ports
- Detecting NAT using STUN

Connectivity: any broadband IP, wired or wireless (DSL, Wi-Fi, 3G, WiMAX)

Codecs: GSM (06.10), G.729, G.711

Communication protocols and standards:

- SIP (RFC 3261)
- http auth (RFC 2617)
- SDP (RFC2327)
- RTP/RTCP (RFC 3550)
- SIP INFO (RFC 2976)
- STUN (RFC 3489)
- Presence (RFC 3856)





Price:

• 1500 EUR per OS



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