

## GIPS VoiceEngine™ IP Phone

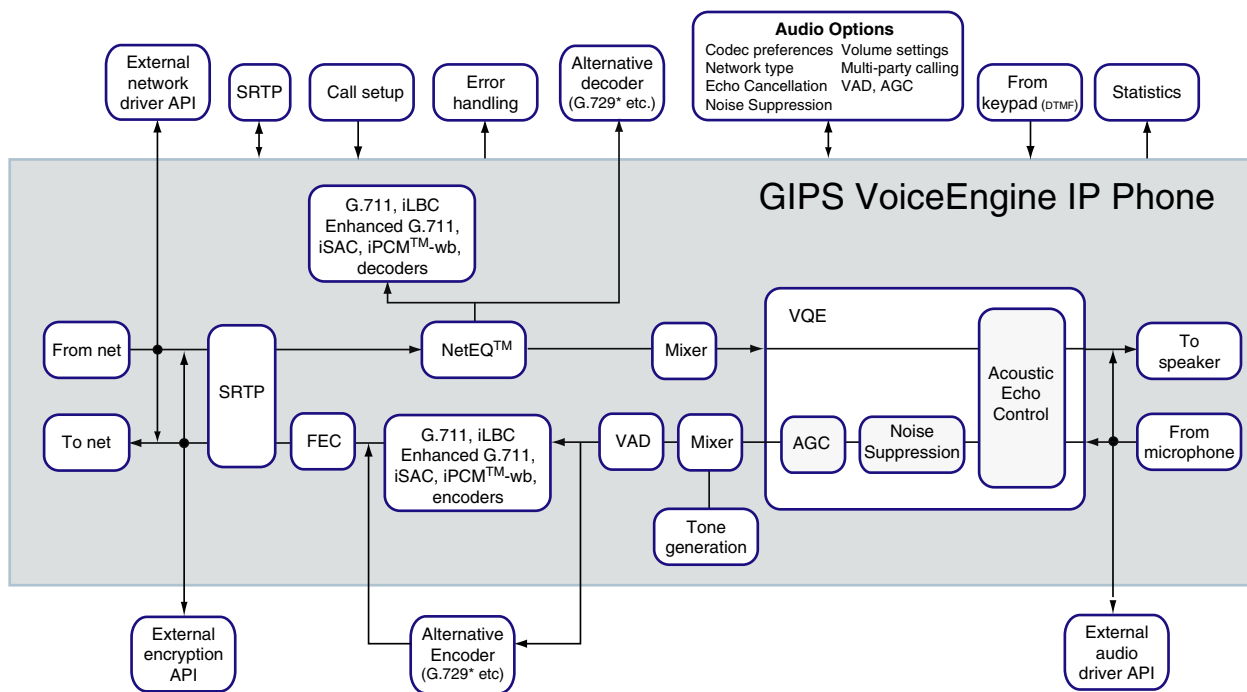
### Consistent and unmatched voice quality across platforms

GIPS VoiceEngine IP Phone is an extension to the world-renowned GIPS VoiceEngine product family. Building on the success of GIPS VoiceEngine, Global IP Solutions has created an embedded solution that meets the highest requirements of IP phone manufacturers. This member of the VoiceEngine family is specifically designed to deliver the highest quality voice processing and ease of integration in CPU based IP phones.

The comprehensive GIPS VoiceEngine Embedded package provides:

- Accelerated time-to-market and reduced development costs
- Simplified integration of sound-processing software with high-level API
- Field-proven, high quality solution
- Sophisticated speech enhancements including acoustic echo cancellation
- Flexible and rich feature set

GIPS VoiceEngine IP Phone provides manufacturers with a solution that solves all of the complex voice processing functions in one integral piece of software that guarantees the best voice quality with the lowest possible delay. GIPS VoiceEngine IP Phone takes full advantage of Global IP Solutions' patented technology and expertise to bring to market industry-leading VoIP enabled devices.





GLOBAL IP SOLUTIONS

## GIPS VoiceEngine™ IP Phone

Feature	VoiceEngine IP Phone <sup>1</sup>
Same high level API on all platforms	✓
Low latency and high packet loss robustness with GIPS NetEQ and GIPS codecs	✓
GIPS narrowband and wideband codecs	✓
Standard narrowband and wideband codecs	Optional
Forward Error Correction (FEC), RFC 2198	✓
Re-samplings filters	✓
VAD/DTX/CNG	✓
Send DTMF	✓
Speech level indicator	✓
Microphone and speaker volume controls	✓
Compatible with any call setup protocol (e.g. SIP, H.323)	✓
RTP Handling or support for customer specific media transport protocol	✓
RTCP and statistics	✓
SRTP	✓
Support for addition of external encryption technology	✓
Voice Quality Enhancement	
- Acoustic echo suppression	✓
- Acoustic echo cancellation	✓
- Noise Suppression	✓
- Automatic gain control	✓
Audio mixing/Multi-party calling (between 3 and 5 participants)	✓
Multiple channel support (call waiting etc.)	✓
Video synchronization	✓
Support for multiple hardware <sup>2</sup> and OS platforms <sup>3</sup>	✓*

1 = Options may be limited by OS or hardware configuration

2 = Currently available for ARM, MIPS, XScale and ColdFire CPUs

3 = Currently available for Linux, eCOS and VxWorks operating systems

\* As GIPS is constantly innovating, please contact a sales representative for the most recent list of supported platforms