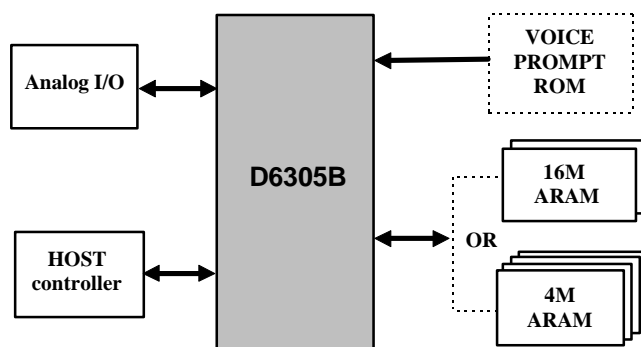


## EASYTAD™ Chip for an All- Digital Telephone Answering Machine



**Figure 1. D6305B Block Diagram**

The D6305B chip is a digital speech/signal processing subsystem that implements all of the functions of speech compression, telephone line signal processing, memory management, and TRUESPEECH® natural sounding voice prompting for an all digital answering machine. The D6305B is fully controlled by the system host through a simple interface protocol. The host controller provides activation and control of all system functions such as speech recording and playback, DTMF and tone generation, DTMF and call progress tone detection, and voice prompting.

The fully digital solution has no moving parts, yielding high reliability. The chip's programmability enables a full range of answering device features for the end user. The D6305B is a single-chip, digital signal processor that includes all of the necessary circuitry to interface with an external HOST, memory, and an analog front end.

### FEATURES

- High-quality, variable low rate TRUESPEECH® digital speech compression allowing 14-15 minutes of recording time per each 4 MB ARAM (Audio grade DRAM)
- Flexible storage of incoming messages (ICM) and outgoing messages (OGM). Supports multiple OGMs and multiple mailboxes
- TRUESPEECH® natural-sound voice prompting, for Day/Time stamp and voice instructions
- DTMF generation and detection with near-end echo cancellation for superior performance
- Fully backwards compatible with D6305A. Host software compatible, pin-to-pin replaceable
- Uses more relaxed ARAM specifications (standard ARAM)
- Will work with 1M x 4 (4MB) or 4M x 4 (16 MB) ARAMs
- Supports "offset playback" for jumps within a message
- Programmable Tone Generation and Tone Generation from tables
- 8-bit parallel HOST interface
- Storage of 32 telephone numbers for speed dialing
- Low power consumption
- Special commands for production line testing

# EASYTAD - D6305B Product Overview

## CHIP CONFIGURATION

- D6305B-11 Digital Speech Processor (80-pin PQFP)
  - 1 each

## ADDITIONAL SYSTEM COMPONENTS

(supplied by the customer according to DSP Group's specifications)

- D0000-29 Analog I/O Interface (16-pin DIP)
  - 1 each
- D0000-35B 4-MB ARAM Message Memory (SOJ) or D0000-36B 16-MB ARAM
  - Up to four devices per chipset - Up to two devices per chipset

## NOTES

For Voice Prompt storage an external EPROM/ROM (access time of 400ns or less) is required. Each 16K x 8 block will support 19.5 seconds of voice prompt, up to 64K x 8 (for up to 78 seconds of voice prompt).

TECHNICAL SPECIFICATIONS			
PARAMETERS	MIN/TYP/MAX	DATA	COMMENTS
Voice Bandwidth		300-3,400 Hz	
Voice Recording Time	typical	15 - 17 minutes	4 MB of ARAM storage
DTMF Detection			
Signal Level	minimum	-35 dB *	
Accept Duration	minimum	40 ms	
Reject Duration	maximum	23 ms	
VOX Detection Level	minimum	-37 dB *	
VCC Supply		5V±10%	
ICC Supply Current, Operating Mode	typ	45 mA	
Power Down Mode	typ	3.5 mA	
Operating Temperature Range		0° C to 70° C	
Crystal	typical	29.4912 MHz	
Package		80-pin PQFP	

\* 0.707 V<sub>RMS</sub> is defined as the 0 dB0



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