HD81802

HITACHI/ (MCU/MPU)

ADPCM Speech Encoding/Decoding LSI

The HD81802 is a digital recording and playing LSI. It compresses and codes PCM (Pulse Code Modulation) speech data into ADPCM (Adaptive Differential Pulse Code Modulation) speech data in digital recording, and decodes ADPCM data into PCM data in digital playing. The ADPCM speech data is compressed into less than half of the PCM speech data without being affected in the speech quality, which saves the speech memory capacity and the cost of communication channel.

Features

- Both high speech quality and high data compression rate realized by ADPCM coding/decoding.
- Data compression rates: 32 kbits/s for higher speech quality
- Quick recovery from data transmission error
- · Power output of recording speech
- Speech power detecting function provided to record the necessary speech utterance only.
- Easy construction of a system including A/D, D/A and filter by connecting a CODEC for telephone directly.
- Direct data output to telephone lines using A-law/μ-law PCM CODEC.
- · Four types of CODEC available
- · System bus compatible with 8/16 bit microprocessor

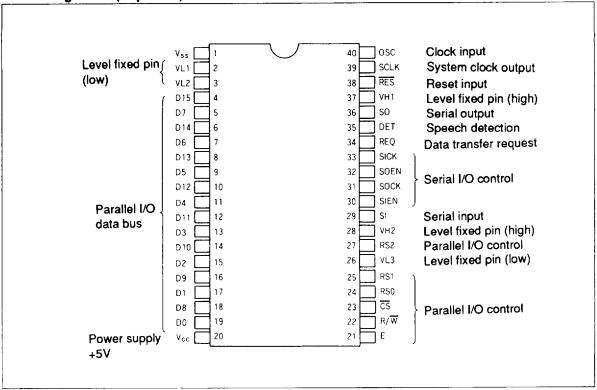
Specifications

Items	Contents
Coding process	ADPCM (Adaptive Differential Pulse Code Modulation)
Sampling frequency	8-10 kHz
Coding bit count	4 bits
Data rate	32 kbits/s
Serial interface	• μ-law PCM CODEC (ex. HD44238C/HD44278I
(CODEC interface)	• A-law PCM CODEC (ex. HD44237C/HD44277F
	 Linear A/D, D/A converter (two's complement)
	 Linear A/D, D/A converter (offset binary)
Parallel interface	8-bit microprocessor
(microcomputer interface)	16-bit microprocessor
Supplementary functions	Speech part detection
	Speech power output
	Recovery from transmission error
Process	CMOS
Power dissipation	150 mW (Typ)
Power supply voltage	A single power supply of +5 V
Package	DILP-40

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Block Diagram

