

Product Brief

DA7210 Ultra low power stereo audio codec with True-GND headphone driver

General Description

The DA7210 is a high definition audio codec with integrated true-ground capless headphone driver suitable for a variety of low power, digital portable audio products.

Featuring a high efficiency headphone amplifier and minimum supply voltage of 1.8V to simplify interfacing to digital processors, the ultra-low 2.5mW power consumption extends music playback time for battery operated equipment.

The integrated PLL uses a FRACT-N PLL architecture that supports a large range of input and output frequencies. This is in addition to standard mobile phone/USB system clock frequencies - enabling audio data synchronization when no master clock is readily available.

Eight analogue input pins allow multiple audio sources to be internally mixed, eliminating the need for external switches. Both single-ended and fully-differential line and microphone inputs are supported with built-in variable gain amplifiers to optimize dynamic range prior to digitisation. This allows a diverse variety of analogue audio sources such as baseband voice, mobile TV, WiFi and FM radio to be managed. Input and output mixers with stereo-to-mono conversion also support mono configurations such as headset/baseband line outputs. Beside the fully differential mono RX channel, DA7210 provides two volume controlled differential/single-ended stereo line out drivers and ground centered stereo amplifiers to directly drive standard 3-wire 16ohm headphones. For example the dc-coupled, dedicated pop-free drivers may be connected to stereo headphones, stereo speaker and mono line out, all simultaneously and without external switches.

All filtering and sidetone functions are performed digitally including 5-band EQ and a digital input AGC with programmable attack and decay parameters. A configurable signal processing engine allows various audio enhancements and effects i.e. acoustic filtering, wind noise suppression and 3D sound

The multi-slot I2S/PCM interface supports all common sample rates between 8-96kHz in master or slave modes.

Features

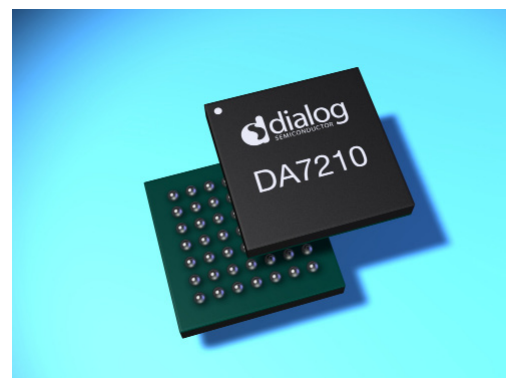
- High performance audio codec with integrated PLL
- True-GND capless Class G 40mW headphone driver with integrated charge pump
- Flexible clocking capability to minimise master clock circuit board routing
- Fully differential mono voice channel
- Pop & Click suppression circuitry
- Multi mode audio routers and mixers & volume control

Functions

- Stereo Playback/Record: 2.5mW & 3.5mW @ 1.8V
- Audio performance@ 2.5V
- - DAC: 100dB/-85db THD
- - ADC: 96db SNR/-89dB THD
- Sample rates up to 96kHz supported via multi-slot I2S/PCM interface
- Stereo fully-differential microphone amplifiers with 5uV input noise and bias
- DSP 5 Band EQ, Input ALC, programmable noise and acoustic enhancement filters

Target Applications

- Smart phones
- Personal Media Players
- Multimedia Handsets
- Personal Navigation Devices



Available in 49 bump WLCSP package

