

TAS6754-Q1 1L Modulation, 2MHz Digital Input 4-Channel Automotive Class-D Audio Amplifier with Current Sense and Real-time Load Diagnostics

1 Features

- AEC-Q100 qualified for Automotive applications – Temperature grade 1: -40° C to +125°C, T_A
- General operation
	- 4.5V to 19V supply voltage, 40V load dump
	- Low latency path reducing group delay by >70% at 48kHz
	- Support for 1.8V and 3.3V I/O's
	- I²C control with 8 address options
- 1L Modulation
	- Requires only one inductor per channel saving four inductors vs a traditional solution
	- Lower system component cost for inductors
- Smaller PCB footprint
- Audio Performance
	- THD+N 0.05% (4Ω, 1W, 1kHz)
	- $-$ Output noise: $35\mu V_{RMS}$ at 14.4V, A-weighting
	- Efficiency > 87% @ 4×25W, 4 Ω, 14.4V
	- Output current sensing by channel via $1²S$ or TDM
	- No external circuitry needed
- Real-time load diagnostics
	- Monitor output conditions while playing audio
	- Open Load , Shorted load, Short-to-power, and Short-to-ground detection
- DC and AC Standby load diagnostics
- Audio inputs
	- $-$ 2-4 channel via $1²S$ or 4-16 channel via TDM
	- Input sample rates: 44.1, 48, 96, 192 kHz
- Audio outputs
	- 4 channel bridge-tied load (BTL)
	- 2MHz output switching frequency
	- $-$ 4×30W (4 Ω, 14.4V, 1kHz 10% THD+N)
	- $-4 \times 50W$ (2 Ω, 14.4V, 1kHz 10% THD+N)
- Advanced spread-spectrum and selectable phase offset
- Protection and Monitoring
	- Cycle-by-cycle current limiting
	- Output short protection
	- Clip detection with configurable thresholds
	- Thermal foldback and PVDD foldback
	- $\,$ l²C temperature and supply voltage readout
	- Configurable overtemperature warning and individual channel shutdown
	- DC offset, undervoltage and overvoltage

2 Applications

- [Automotive head unit](https://www.ti.com/solution/automotive-head-unit)
- [Automotive external amplifier](http://www.ti.com/solution/automotive-external-amplifier)
- [Acoustic vehicle alerting system \(AVAS\)](https://www.ti.com/solution/acoustic-vehicle-alerting-system-avas)

3 Description

The TAS6754-Q1 is a four-channel digital-input Class-D audio amplifier that implements 1L modulation only requiring one inductor per BTL channel reducing system size and cost by removing four inductors compared to a traditional solution. Additionally, 1L modulation lowers switching losses compared to traditional Class-D modulation schemes.

The TAS6754-Q1 integrates DC and AC Load Diagnostics to determine the status of the connected loads. During audio playback this status can be monitored through output current sense which is available for each channel and reports the measurement to a host processor through TDM with minimal delay. The device monitors the output load condition while playing audio through real-time load diagnostics independent of the host and audio input. **Simple the control of the channel Schematic Channel Schematic through outier control of the connected through output current sense which the data be for each channel and reports the definite playing audio through real-tim**

The TAS6754-Q1 device features an additional low latency signal path for each channel, providing up to 70% faster signal processing at 48kHz which enables time-sensitive Active Noise Cancellation (ANC), Road Noise Cancellation (RNC) applications.

The device supports global temperature, channel temperature and PVDD values via I2C readout for easy system level thermal management.

The device is offered in a 56 pin HSSOP package with the exposed thermal pad up.

- (1) For all available packages, see the orderable addendum at the end of the data sheet.
- (2) The package size (length × width) is a nominal value and includes pins, where applicable.

An IMPORTANT NOTICE at the end of this data sheet addresses availability, warranty, changes, use in safety-critical applications, **AD** intellectual property matters and other important disclaimers. ADVANCE INFORMATION for preproduction products; subject to change without notice.

Table of Contents

Figure 4-1. DKQ Package, 56-Pin HSSOP with exposed Thermal Pad Up, Top View

Table 4-1. Pin Functions

Copyright © 2024 Texas Instruments Incorporated *[Submit Document Feedback](https://www.ti.com/feedbackform/techdocfeedback?litnum=SLOSE88&partnum=TAS6754-Q1)* 3

Table 4-1. Pin Functions (continued)

(1) DI = digital input, DO = digital output, DI/O = digital input/output, GND = ground, NC = no connect, NO = negative output, PO = positive output, PWR = power

5 Specifications

5.1 Absolute Maximum Ratings

over operating free-air temperature range (unless otherwise noted) (1)

(1) Operation outside the Absolute Maximum Ratings may cause permanent device damage. Absolute Maximum Ratings do not imply functional operation of the device at these or any other conditions beyond those listed under Recommended Operating Conditions. If used outside the Recommended Operating Conditions but within the Absolute Maximum Ratings, the device may not be fully functional, and this may affect device reliability, functionality, performance, and shorten the device lifetime.

5.2 Recommended Operating Conditions

5.3 ESD Ratings

(1) AEC Q100-002 indicates that HBM stressing shall be in accordance with the ANSI/ESDA/JEDEC JS-001 specification.

5.4 Thermal Information

(1) For more information about traditional and new thermalmetrics, see the [Semiconductor and IC Package Thermal Metrics](https://www.ti.com/lit/pdf/SPRA953) application report.

(2) JEDEC Standard, 4-Layer PCB.

5.5 Electrical Characteristics

Test conditions (unless otherwise noted): T $_{\rm C}$ = 25°C, PVDD = 14.4V, VBAT = 14.4V, DVDD = 1.8V, R $_{\rm L}$ = 4 Ω , $P_{\sf out}$ = 1W/ch, $f_{\sf out}$ = 1kHz, F_{sw} = 2.048MHz, AES17 Filter, reconstruction filter inductor used: 3.3µH-VCMT053T-3R3MN5 and 1μ F, default I²C settings + start-up script, see application diagram

Copyright © 2024 Texas Instruments Incorporated *[Submit Document Feedback](https://www.ti.com/feedbackform/techdocfeedback?litnum=SLOSE88&partnum=TAS6754-Q1)* 7

Test conditions (unless otherwise noted): T $_{\rm C}$ = 25°C, PVDD = 14.4V, VBAT = 14.4V, DVDD = 1.8V, R $_{\rm L}$ = 4 Ω , $P_{\sf out}$ = 1W/ch, $f_{\sf out}$ = 1kHz, F_{sw} = 2.048MHz, AES17 Filter, reconstruction filter inductor used: 3.3µH-VCMT053T-3R3MN5 and 1μ F, default I²C settings + start-up script, see application diagram

5.6 Typical Characteristics

Test conditions (unless otherwise noted): TC = 25°C, PVDD = 14.4V, VBAT = 14.4V, DVDD = 1.8V, RL = 4Ω, Pout = 1W/ch, ƒout = 1kHz, Fsw = 2.048MHz, AES17 Filter, reconstruction filter as described in Parameter Measurement Information, default I2C settings + start-up script, see application diagram

5.6 Typical Characteristics (continued)

Test conditions (unless otherwise noted): TC = 25°C, PVDD = 14.4V, VBAT = 14.4V, DVDD = 1.8V, RL = 4Ω, Pout = 1W/ch, ƒout = 1kHz, Fsw = 2.048MHz, AES17 Filter, reconstruction filter as described in Parameter Measurement Information, default I2C settings + start-up script, see application diagram

5.6 Typical Characteristics (continued)

Test conditions (unless otherwise noted): TC = 25°C, PVDD = 14.4V, VBAT = 14.4V, DVDD = 1.8V, RL = 4Ω, Pout = 1W/ch, ƒout = 1kHz, Fsw = 2.048MHz, AES17 Filter, reconstruction filter as described in Parameter Measurement Information, default I2C settings + start-up script, see application diagram

6 Parameter Measurement Information

For measurements, the LC reconstruction filter Cyntec VCMT053T-3R3MN5 3.3 µH inductor + 1 µF capacitor is used.

When enabling Real-Time Load Diagnostics with the integrated pilot tone, an analog balanced input filter must be used to avoid misleading measurement results. An elliptic high pass filter as provided by the APx500 series with a cutoff frequency of 20 Hz and a low-pass filter such as AES17 (20 kHz) are recommended. If the test equipment does not support this filter type, it is recommended to turn off the Real-Time Load Diagnostics for accurate performance measurements.

7 Detailed Description

7.1 Overview

The TAS6754-Q1 is a four-channel digital input Class-D audio amplifier, specifically tailored for use in the automotive industry. The device is designed for vehicle battery operation up to 19 V with up to 40 V load dump support. The ultra-efficient Class-D technology with advanced 1L modulation allows for lower switching losses, and a smaller PCB area. The device realizes a high-fidelity audio sound system design with smaller size, lower weight and advanced functionality.

The core design blocks are:

- Serial audio port
- PLL and Clock management
- Dual Core Audio DSP subsystem
- Pulse width modulator (PWM) with output stage feedback
- Gate drive
- Power FETs
- Current Sense
- Diagnostics including Real-Time Load Diagnostics
- **Protection**
- Power supply
- I²C serial communication bus
- **Sensing**

7.2 Functional Block Diagram

7.3 Feature Description

7.3.1 Power Supply

The device has three power supply inputs, DVDD, VBAT and PVDD, which are described as follows:

- **DVDD** This is a 1.8V or 3.3 V supply that is connected at the DVDD pin and provides power to the digital circuitry.
- **VBAT** This pin is a higher voltage supply that can be connected to the vehicle battery or the regulated voltage rail in a boosted system within the recommended limits. For best performance, this rail should be 10 V or higher. See the *[Recommended Operating Conditions](#page-5-0)* table for the maximum supply voltage. This supply rail is used for higher voltage analog circuits but not the output FETs.
- **PVDD** This pin is a high current supply that can either be connected to the vehicle battery or to another voltage rail in a boosted system. The PVDD pin supplies the power to the output FETs and can be within the *[Recommended Operating Conditions](#page-5-0)*, even if that is below the VBAT supply, to allow for dynamic voltage systems.

Several on-chip regulators are included generating the voltages necessary for the internal circuitry. The external pins are provided only for bypass capacitors to filter the supply and should not be used to power other circuits.

The device can withstand fortuitous open ground and power conditions within the absolute maximum ratings for the device. Fortuitous open ground usually occurs when a speaker wire is shorted to ground, allowing for a second ground path through the body diode in the output FETs.

7.3.1.1 Power-Supply Sequence

In a typical system, the VBAT and PVDD supplies are both connected to the vehicle battery and power up at the same time.

7.3.1.1.1 Power-Up Sequence

At power-up, the PD pin is recommended to be kept low until all three power supply rails (VBAT, PVDD, DVDD) are within the *[Recommended Operating Conditions](#page-5-0)*.

When all power rails are applied and ready, releasing the \overline{PD} pin will power up the internal digital circuitry. After releasing the PD pin a minimum 4 ms wait time is recommended before releasing the STBY pin. Releasing the STBY pin powers up the internal analog circuitry. If this sequence is not possible and PD as well as STBY have to be released at the same time, the following state transition might take up to 6 ms in which the device powers up the analog circuitry and ensures a proper internal boot procedure. **Example the colleges necessary for the internal circuity. The external product Folder Internal Control of the internal circuity. The external constrained process when a speaker wire is shorted to ground, allowing for a g**

Any time a power fault happens after the analog circuitry is powered up (VBAT ready and STBY released), TAS6754-Q1 will leave PLAY or other states and go back to the [Auto Recovery \(AUTOREC\) State,](#page-35-0) until the power fault disappears.

Applying DVDD before VBAT will leads to a reported "VBAT Undervoltage Fault" which needs to be cleared

Figure 7-2. TAS6754-Q1 Power-Up Sequence

7.3.1.1.2 Power-Down Sequence

To power-down the device, first set the \overline{STBY} pin or PD pin low for at least 10 ms before removing PVDD, VBAT or DVDD. After 10 ms, the power supplies can be removed. Removing PVDD and VBAT first is recommended before removing the DVDD supply.

7.3.1.2 Device Initialization and Power-On-Reset (POR)

The device initializes when either the system first powers up, the PD pin is pulled high, or when the DVDD voltage falls below the POR threshold and then comes back to normal condition.

During device initialization all ${}^{12}C$ registers are set to default values.

The I²C device address is determined from the I2C_ADDR pin. See I²[C Address Selection](#page-37-0) for details.

7.3.2 Serial Audio Port

The Serial Audio Interface can receive data in left-justified, I²S or DSP mode formats. In addition, time-division multiplexing (TDM) can be implemented to enable multichannel operation with support up to TDM16.

The pins SDIN1 and SDOUT1 are available for the data transfer, while any of the GPIO pins can be assigned to become SDIN2 and SDOUT2 if required. Refer to [GPIO Pins](#page-30-0) for more details.

ĒΥΔS

7.3.2.1 Left-Justified Timing

Left-Justified timing uses the FSYNC pin to define when the data is being transmitted for the left channel or the right channel. The MSB of the left channel is valid on the rising edge of the serial clock (SCLK) following the rising edge of the audio frame clock (FSYNC). Similarly, the MSB of the right channel is valid on the rising edge of SCLK clock following the falling edge of FSYNC. A channel offset can be configured and is identical across all channels.

7.3.2.2 I ²S Mode

²S mode uses the FSYNC pin to define when the data is being transmitted for the left channel and when the data is being transmitted for the right channel. In ^{12}S mode, the MSB of the left channel is valid on the second rising edge of the serial clock (SCLK) after the falling edge of the audio frame clock (FSYNC). Similarly the MSB of the right channel is valid on the second rising edge of SCLK after the rising edge of FSYNC. A channel offset can be configured and is identical for across channels.

Figure 7-6. Timing Diagram for I2S Mode with Offset = 1

 $Offset1 = 1$ $Offset1 = 1$ $Offset1 = 1$

Channel 2

 \blacktriangleright

 \blacktriangleright

N-1 | N-2 | | | | 3 | 2 | 1 | 0

Channel 4

Channel 1

Channel 3

SDIN2 $\overline{\left\| N-1 \right\| N-2}$ $\left\| N-2 \right\| \left\| 3 \right\| 2 \left\| 1 \right\| 0$

Channel 1

 $\mathbf{||}$

Channel 3

N-1 N-2

 $\overline{ }$

d

7.3.2.3 DSP Mode

DSP mode uses the FSYNC pin to define the start of the audio data, but not to differentiate between channels. The rising edge of the audio frame clock (FSYNC) starts the data transfer with the left channel data first and is immediately followed by the right channel data. Each data bit is valid on the rising edge of the serial clock (SCLK). A 10-bit channel offset can be configured and is identical across all channels.

7.3.2.4 TDM Mode

TDM mode supports 4, 8, or 16 channels of audio data through SDIN. The data can be received through a single pin or split across SDIN1 and SDIN2. The data format follows the [DSP Mode.](#page-18-0)

Figure 7-11. Timing Diagram for TDM mode with Audio Channels Offset1, Low Latency Channels Offset2

7.3.2.5 SDOUT - Data Output

TAS6754-Q1 can transmit selected data in either I2S mode or TDM mode. The audio input serial clock (SCLK) and audio frame clock (FSYNC) is reused, and the outgoing data has the same sampling frequency and maximum audio frame size as the audio input signal.

7.3.2.6 Device Clocking

The TAS6754-Q1 has a flexible clocking system. Internally, the device requires several additional clocks, mostly at related clock rates to function correctly. All of these clocks can be derived from the Serial Audio Interface.

Figure 7-12 shows the basic data flow and clock distribution.

Figure 7-12. Audio Flow with Respective Clocks

The Serial Audio Interface typically has 3 connection pins which are listed as follows:

- SCLK (Audio Serial Clock)
- FSYNC (Frame Sync in TDM or Left/Right in 1^2S)
- SDIN (Input Data). In TDM mode, a single SDIN is used while I2S requires two SDIN pins to provide audio data for 4 channels
- Optional: SDOUT for outgoing data transmission. Up to two SDOUT pins can be configured

The device has an internal PLL which uses SCLK as reference clock and creates the higher rate clocks required by the DSP and the DAC clock.

The TAS6754-Q1 has an audio sampling rate detection circuit that automatically senses the sampling frequency. Common audio sampling frequencies of 44.1kHz – 48 kHz, 88.2 kHz – 96 kHz and 192 kHz are supported. The sampling frequency detector sets the clock for DAC and DSP automatically.

7.3.2.6.1 Clock Rates

The serial audio interface port is a 3-wire serial port with the signals SCLK, FSYNC and SDIN_1 as well as an optional SDIN 2 in I²S Mode.

SCLK is the serial audio bit clock used to clock the serial data present on SDIN x into the serial shift register of the audio interface. Serial data is clocked into the TAS6754-Q1 device with SCLK.

The FSYNC pin is the serial audio left/right word clock or frame sync when the device is operated in TDM Mode.

SDIN 1 is the TDM data input. In ^{12}S mode, SDIN 1 is the data input for channels 1 and 2 and a GPIO pin needs to be configured as SDIN_2 to receive the data input for channels 3 and 4.

7.3.2.6.2 Clock Halt Auto-recovery

Certain host processors halt the audio clock when no audio is playing. When the clock is halted, the device puts all channels into the Hi-Z state and issues a latched error report in . The transition to Hi-Z occurs gracefully by holding the last received sample from the audio interface and ramping down the volume. This behavior can be

changed . The latched error report clears once read. After audio clock recovery, the device automatically returns to the previous state.

7.3.2.6.3 Sample Rate on the Fly Change

TAS6754-Q1 supports an on-the-fly change of the FSYNC rate. When changing FSYNC, for example from 48 kHz to 96 kHz, the host processor needs to put the FSYNC/SCLK to halt state for at least 30ms before changing to the new sample rate. During this halt state a clock error is reported. See [Clock Halt Auto-recovery](#page-20-0) section for further details.

7.3.2.7 Clock Error Handling

After [Power-On-Reset \(POR\)](#page-35-0) the device assumes that a clock error exists, but does not assert the clock error flag until the clock error detection result is valid.

If any input clock changes are detected, the auto detect system immediately requests the device to mute, while it continues to monitor and identify a new stable condition.

7.3.3 Digital Audio Processing

TAS6754-Q1 offers advanced digital audio processing capabilities including:

- High-Pass Filter / DC Blocking
- Digital Volume Control
- PVDD Foldback / AGL
- Thermal Foldback
- Gain Compensation Biquads
- Real-Time Load Diagnostics
- Clip Detect
- Low Latency Path

The availability of specific features is dependent on the selected sampling frequency. Higher sampling frequencies reduce the available processing time of the integrated DSP and limit the amount of functions that can operate in parallel. It is essential at all sampling frequencies that the total processing need of the enabled functions does not exceed the available amount of processing time.

7.3.3.1 PVDD Foldback

PVDD Foldback applies a smooth compression to the audio signal to maintain a consistent dynamic range while the supply voltage (PVDD) varies. This feature helps to prevent unexpected output clipping and distortion in systems where the audio signal would exceed the supply headroom and can also be described as Automatic Gain Limiter (AGL).

Figure 7-13. PVDD Foldback Example

7.3.3.2 High-Pass Filter

To protect speakers connected to the TAS6754-Q1, a DC blocking high pass filter is build into the audio processing path.

7.3.3.3 Analog Gain

TAS6754-Q1 allows the user to set the analog gain for each channel pair and in increments of 0.5 dB.

A gain setting of 0dB corresponds with a peak output voltage of 28V/FS at full scale digital input. It is recommended to select the lowest possible gain for the expected PVDD operation to optimize output noise and dynamic range performance.

The Analog Gain setting should only be changed while the affected channel is in [DEEP SLEEP](#page-34-0) or [SLEEP](#page-34-0) state.

As the device enters [PLAY State](#page-34-0) it gradually ramps the analog gain to the desired value in steps of 0.5 dB.

7.3.3.4 Digital Volume Control

The output channels have a digital-volume control with a range from -103 dB to 0 dB with 0.5 dB steps.

7.3.3.4.1 Auto Mute

When detecting a consecutive stream of zero samples at the audio input, the device can automatically set channels into mute. In this mode, the device continues to monitor the input signal, and, depending on the configuration, unmute either individual channels or all channels at the same time when a valid non-zero signal arrives.

7.3.3.5 Gain Compensation Biquads

The modulator and output LC filter of the Class-D amplifier can have an undesired influence on frequency response linearity causing frequency drop/peaking. To help compensate for this effect and achieve a flat response, TAS6754-Q1 offers integrated and channel-based gain compensation biquads.

The biquad is configurable by channel and is disabled by default. To enable the desired tuning, the respective coefficients need to be written to the DSP memory.

7.3.3.6 Low Latency Signal Path

For time-sensitive audio signals that require a minimal processing delay, such as active noise cancellation (ANC) or road noise cancellation (RNC), TAS6754-Q1 offers a low latency signal path. At 48kHz sampling frequency, this path reduces the signal delay by more than 70% between the input and output of the amplifier by minimizing the internal signal processing.

The low latency signal path is established in parallel to the regular audio signal path. When both signal paths are provided with input data, the two signals - audio and low latency - for each channel are internally mixed together right before the output amplification stage of the channel. Both signals are added together and the combined signal amplitude must not exceed the available gain range nor the voltage headroom to avoid distortion. Note that the low latency signals pass through the device with less delay than the regular audio path signals. The device with leading in the device can automatically set

The device confinities to monitor the input signal, and, depending on the

evice confinities for all channels at the same time when a valid non-zono signal

e C

The low latency signal path is only available at a sampling frequency of 48kHz or 96kHz.

Figure 7-14. Low Latency and Audio Signal Path

7.3.3.7 Full Feature Low Latency Path

In addition to the [Low Latency Signal Path](#page-22-0) the TAS6754-Q1 integrates a Full Feature Low Latency Path. Using the Full Feature Low Latency Path the time-sensitive audio signals, such as active noise cancellation (ANC) or road noise cancellation (RNC), can be premixed and do not need to be separated from the non-time-senstive audio data. The premixed audio data then processed through the DSP and then amplifed at the output stage with a lower group delay.

Figure 7-15. Full Feature Low Latency and Audio Signal Path

7.3.4 Class-D operation and Spread Spectrum Control

7.3.4.1 1L Modulation

TAS6754-Q1 supports 1L modulation. With 1L modulation, the OUT xP side switches from 0 volts to the supply voltage at the switching frequency (Fsw). The OUT xM side switches from 0 volts to the supply voltage at the

audio frequency. Since the OUT_xM side of the channel switches at a lower frequency, no inductor is required on this side of the BTL channel.

7.3.4.2 High-Frequency Pulse-Width Modulator (PWM)

The PWM modulator converts the input audio data into a switched signal of varying duty cycle. The PWM modulator is an advanced design with high bandwidth, low noise, low distortion, and excellent stability.

TAS6754-Q1 has configurable output PWM phase control to manage conducted and radiated emissions. This feature allows the channel output PWM phase offset to be changed relative to other channels.

7.3.4.3 Spread Spectrum Control

TAS6754-Q1 applies spread spectrum control to the modulator's clock signal. Controlling the spectrum of the clock signal translates into an optimized behavior of higher frequency signal components which are visible during EMI testing. Spread spectrum modulation is a PWM modulation technique that reduces the peaks seen in EMI measurements by varying the output PWM frequency, resulting in a wider spectrum but lower level.

7.3.4.4 Gate Drive

The gate driver accepts the low-voltage PWM signal and level shifts it to drive a high-current, full-bridge, power-FET stage.

The device uses proprietary techniques to optimize EMI and audio performance. The gate driver power supply voltage, GVDD, is internally generated and a decoupling capacitor must be connected.

The full H-bridge output stages use only NMOS transistors. Therefore, bootstrap capacitors are required for the proper operation of the high side NMOS transistors for the OUT-xP sides. A 1-µF ceramic capacitor of quality X7R or better, rated appropriately for the applied voltages (including load dump voltages), must be connected from each output to the corresponding bootstrap input. The bootstrap capacitors connected between the BST pins and the corresponding output function as a floating power supply for the high-side N-channel power MOSFET gate drive circuitry. During each high-side switching cycle, the bootstrap capacitors hold the gate-to-source voltage high, keeping the high- side MOSFETs turned on. If requency, resulting in a wider spectrum but lower level.

PWM signal and level shifts it to drive a high-current, full-bridge,
 Doptimize EMI and audio performance. The gate driver power supply

a decoupling capacit

The high-side FET gate driver of the OUT_xM side is supplied by a charge pump (CP) supply for all four channels. A 330 nF ceramic capacitor of quality X7R or better, rated appropriately for the applied voltages (including load dump voltages), must be connected from the CP pin to PVDD. Additionally, a similarly rated 100 nF ceramic capacitor must be connected from the CPC_TOP pin to the CPC_BOT pin.

7.3.4.5 Power FETs

The BTL output channel comprises four N-channel FETs for high efficiency and maximum power transfer to the load. These FETs are designed to handle the fast switching frequency and large voltage transients during operation within the *[Recommended Operating Conditions](#page-5-0)*.

7.3.5 Load Diagnostics

The device incorporates both DC and [AC](#page-26-0) load diagnostics which are used to determine the status of the load. The DC-diagnostics are turned on by default.

7.3.5.1 DC Load Diagnostics

The DC load diagnostics are used to verify if the load is connected properly.

In order to support system level start up requirements of a fast time to audio:

- The diagnostics are available as soon as the device leaves the DEEP SLEEP mode and supplies are within the recommended operating range.
- The diagnostics do not rely on external audio input signals or clock and sync frequencies to be available.

DC Diagnostics complete successfully and allow a channel to enter MUTE or PLAY mode if the following tests pass on the output pins:

- No [Short-to-Ground](#page-25-0)
- No [Short-to-Power](#page-25-0)

- No Shorted Load
- No Open Load

7.3.5.1.1 Automatic DC Load Diagnostics at Device Initialization

The TAS6754-Q1 supports automatic and autonomous DC load diagnostics at device start-up. When leaving [DEEP SLEEP state](#page-34-0) and under the condition that all power supplies are within the recommended operating range, the device transitions into [SLEEP state](#page-34-0) and automatically starts DC load diagnostics on all four channels.

Neither I²C configuration nor any audio signals are necessary for the TAS6754-Q1 to perform short-to-power (S2P), short-to-ground (S2G), open-load (OL), and shorted-load (SL) tests based on default configuration. Systems can benefit from this autonomous operation as it is possible to run the load diagnostics while bringing up the digital part of the audio chain.

7.3.5.1.2 Automatic DC load diagnostics during Hi-Z or PLAY

When a fault occurs while a channel is in Hi-Z or PLAY state, the device will place the channel in either [FAULT](#page-35-0) [state](#page-35-0) or [Auto Recovery State.](#page-35-0) After the fault is resolved or cleared, the device will run automatic DC load diagnostics on the affected channel(s) and recover to previous [Hi-Z](#page-34-0) respective [PLAY state](#page-34-0) unless a different state was requested through I^2C .

7.3.5.1.3 Manual start of DC load diagnostics

Manual DC load diagnostics can be enabled in any state after all power supplies are within the recommended operating range and after the device has transitioned to [SLEEP state](#page-34-0) for the first time. DC diagnostics can be enabled manually by setting the I²C control state register to [LOAD DIAG state](#page-34-0) to run on any or all channels. If either the STBY pin or a GPIO pin function set the device to [SLEEP](#page-34-0) or [DEEP SLEEP state](#page-34-0), manual DC Load diagnostics cannot be run. This doesn't apply when the device is set to [SLEEP](#page-34-0) or [DEEP SLEEP state](#page-34-0) through ²C control, in which manual DC Load diagnostics are available. When a fault occurs while a channel is in Hi-Z or PLAY state, the developed
tate or Auto Recovery State. After the fault is resolved or cleared
diagnostics on the affected channel(s) and recover to previous Hi-Z

7.3.5.1.3

7.3.5.1.4 Short-to-Ground

The Short-to-Ground (S2G) tests triggers a fault condition if there is a conductive path from output pin OUT (i)M or OUT (i)P of the tested channel (i) to GND with an impedance below that specified in the [Electrical](#page-6-0) [Characteristics](#page-6-0) section.

7.3.5.1.5 Short-to-Power

The Short-to-Power (S2P) tests triggers a fault condition if there is a conductive path from output pin OUT_(i)M or OUT (i)P of the tested channel (i) to a power rail with an impedance below that specified in the [Electrical](#page-6-0) [Characteristics](#page-6-0) section.

7.3.5.1.6 Shorted-Load and Open-Load

The Shorted-Load (SL) test triggers a fault condition if the conductive path between the OUT_(i)M pin and OUT (i)P pin of the tested channel (i) has an impedance below the threshold set. The SL test has a configurable threshold depending on the expected load to be connected. Because the speakers and cable impedance connected to each channel might be different, each channel can be assigned a unique threshold value.

The Open-Load (OL) test triggers a fault condition if the conductive path between the OUT_(i)M pin and OUT_(i)P pin of the tested channel (i) has an impedance higher than that specified in the [Electrical](#page-6-0) [Characteristics](#page-6-0) section.

Figure 7-16. DC Load Diagnostic Reporting Thresholds

7.3.5.2 Line Output Diagnostics

The device also includes an optional test to detect a line output load (LO). A line output load is a high-impedance load that is above the open load (OL) threshold such that the DC-load diagnostics report an OL condition. If the line output detection bit is set high, when an OL condition is detected during the DC Diagnostic test, the system will also test if a line output load is present.

7.3.5.3 AC Load Diagnostics

The AC load diagnostic is used to determine the proper connection of a capacitive coupled speaker or tweeter when used with a passive crossover. The AC load diagnostic is controlled through I²C. The TAS6754-Q1 provides a required signal source to determine the AC impedance and reports the tweeter detection result back to I²C registers. The I²C selected test frequency should create current flow through the desired speaker for proper detection. AC Load Diagnostics can operate without the TDM/I2S clocks being present. Constructed into output to ad (LO). A line output load is a high-impedance

Shoftshoft such that the DC-load diagnostics report an OL condition is detected during the DC Diagnostic test, the system

ADVA CHO diagnostic is

Note

If a fault occurs during AC diagnostics, the AC diagnostics is stopped. AC Diagnostics is not allowed to be performed again until the DC Diagnostics are performed. This is to make sure the fault is not a potential hazard during AC diagnostics.

7.3.5.3.1 Operating Principal

The AC Load Diagnostic circuit of TAS6754-Q1 provides an internally generated stimulus to the load; captures the response of the load; provides real and imaginary parts of the captured complex load impedance; and offers a magnitude estimator and tweeter detection comparator.

7.3.5.3.2 Stimulus

The device drives a low level, 10 mA output current through the load which does not create any significant sound pressure levels from the speaker.

7.3.5.3.3 Load Impedance

The load impedance as seen by the device is simply the ratio of the voltage across the output pins and the current flowing through the load.

Typically the load has a frequency dependent magnitude and causes current and voltage to have a phase shift. The TAS6754-Q1 internally captures the load impedance as a complex value consisting of a real and imaginary part. Expressing a load impedance in magnitude and phase or in real and imaginary part is mathematically equivalent. Both forms can be transformed into each other without loss of information. After AC load diagnostics

have finished, the real and imaginary parts of the complex impedance are available by channel for readout in ${}^{12}C$ registers as shown in .

7.3.5.3.4 Tweeter Detection

In most cases, it is sufficient to use the TAS6754-Q1 built-in magnitude estimator and tweeter detection report to perform the desired tweeter detection test. If a tweeter is properly connected in the system, the magnitude of the load impedance is close to the nominal impedance of the speaker, for example 4 $Ω$.

7.3.5.4 Real-Time Load Diagnostics

Real-Time Load Diagnostics (RTLDG) allows the detection of shorted load (SL), open load (OL), Short-to-Power (S2P) and Short-to-Ground (S2G) conditions during audio operation of the amplifier. To monitor the load impedance while in PLAY state the TAS6754-Q1 uses its integrated current sense to measure the output impedance by channel and compare it with configurable thresholds. An internally generated pilot tone ensures the continuous detection of the output impedance, regardless if an external audio input signal is present.

7.3.5.5 DC Resistance Measurement

The TAS6754-Q1 supports a DC Resistance Measurement of the loads connected to each channel that can be read back to the system processor via ${}^{12}C$. To read out the DC resistance of the load connected to each channel, DC load diagnostics must be completed.

7.3.6 Protection and Monitoring

7.3.6.1 Overcurrent Limit (Cycle-By-Cycle)

Under normal operation, during high level music playback, it is possible that dynamic load currents can rise beyond the maximum load current, I_{LIM} , of the device. In these cases, the device dynamically limits the current into the load and operation continues without disruption and prevents undesired shutdown for transient music events.

7.3.6.2 Overcurrent Shutdown

If the output load current reaches I_{SD}, such as during an output short to GND, then an Overcurrent Shutdown (OCSD) event occurs, limiting the peak current and shutting down the affected channel. The time to shutdown the channel varies depending on the severity of the short condition.

The channel is placed into the [FAULT state](#page-35-0) with the output stage in Hi-Z.

Based on the configuration, a fault signal is generated, which by default generates an active low signal at the FAULT pin.

7.3.6.3 Current Sense

TAS6754-Q1 can measure the output current of each channel simultaneously. This functionality is completely integrated and requires no external components.

The channel output current measurement is performed at the rate of the sampling frequency F_s . The measured current amplitude is provided through SDOUT. For more details on the data transmission configuration see [SDOUT.](#page-20-0) Note that the current measurement and the data transmission are two separate functions and both require proper setup before the data can be made available. **Production of the output impedance, regardless if an external and the United That [TAS6754-Q1](https://www.ti.com/product/tas6754-q1?qgpn=tas6754-q1) supports an DC Resistance Measurement of the load cand black to the system processor via i²C. To read out the DC resistance M**

7.3.6.4 DC Detect

This circuit detects the DC offset continuously of the amplifier during normal operation. If the DC offset exceeds the DC_{FAULT} threshold, that channel triggers a DC Fault Event and is placed in the [FAULT state](#page-35-0) and the output stage is set to high impedance.

Based on the configuration, a fault signal is generated, which by default generates an active low signal at the FAULT Pin.

It is highly recommend to leave DC Detect enabled at all times to prevent speaker damage from excessive DC bias output.

7.3.6.5 Digital Clip Detect

The DSP monitors the audio signal of each channel individually and compares the magnitude of the audio signal at the input to the interpolation filters to a configurable threshold. If the audio signal exceeds the threshold, a [Clip](#page-36-0) [Detect Warning Event](#page-36-0) is triggered.

TAS6754-Q1 supports a second Clip Detect implementation, further referred to as Pseudo-Analog Clip Detect (PACD). With PACD the DSP processing mimics an analog Clip Detect approach by taking the analog gain and supply voltage into account when determining if the output is clipping against a set clipping threshold. PACD is disabled by default and can be enabled to replace the default Clip Detect mechanism.

7.3.6.6 Charge Pump

The TAS6754-Q1 has built-in protections on the charge pump pins (CP, CPC_TOP, and CPC_BOT). When the device detects a fault condition on the charge pins the charge pump will skip a clock cycle. This is considered one fault cycle. The device will then try again on the next incoming clock cycle. If a fault condition is detected two times consecutively or three times in total, the charge pump will start to ignore the incoming clock and go into a [Charge Pump Fault](#page-35-0) condition with the output stage in Hi-Z and charge pump shutdown.

Based on the configuration, a fault signal is generated, which by default generates an active low signal at the FAULT pin.

7.3.6.7 Temperature Protection and Monitoring

The device monitors temperature with five temperature sensors. Every output channel has a temperature sensor in close proximity to the center of its output stage to monitor the temperature of each channel separately. An additional temperature sensor is located in a global position on the die, so it better represents the actual die junction temperature. Based on these sensors, warning and fault signals can be generated. A [Thermal Gain](#page-29-0) [Foldback](#page-29-0) scheme is available that autonomously regulates audio gain and consequently limits die temperature.

7.3.6.7.1 Overtemperature Shutdown

The temperature thresholds for global OTSD and for overtemperature shutdowns generated by the output channels, OTSD(i) are set to fixed values. Refer to [Section 5.5](#page-6-0) for the nominal temperature and recovery hysteresis values.

Global OTSD: If the global junction temperature rises above the OTSD threshold, all channels are placed into a protective shutdown state and an [Overtemperature Shutdown \(OTSD\) Event](#page-36-0) is created.

Channel-specific OTSD: If the junction temperature of a channel rises above the OTSD threshold, the affected channel is put into a protective state and an [Overtemperature Shutdown \(OTSD\) Event](#page-36-0) is created.

The tolerance of the warning levels and OTSD temperatures track each other.

By default, a fault signal generates an active low signal on the FAULT Pin when an OTSD event occurs.

7.3.6.7.2 Overtemperature Warning

The temperature threshold for global Overtemperature Warning (OTW) is fixed to the level shown in the *[Electrical Characteristics](#page-6-0)*. Each output channel has an independent temperature sensor.

During operation, if the device heats up and crosses the threshold, a global [Overtemperature Warning Event](#page-36-0) is generated. Similarly, if the temperature at a channel raises above the threshold an Overtemperature Warning Event for that channel is generated. While the device continues to operate, the OTW information enables higher level software to make decisions to optimize thermal system performance.

As described in the [Overtemperature Warning Event](#page-36-0), the report can either be polled via I²C register or a hardware signal can be generated by assigning a GPIO Pin for Warning Signals and enabling the OTW report routing.

7.3.6.7.3 Thermal Gain Foldback

Thermal Gain Foldback (TGFB) is a power limiting feature to protect the TAS6754-Q1 from excessive die temperature while maintaining audio output.

The main purpose of foldback power limiting is to keep the output stage within its safe power dissipation limit to avoid unexpected [Overtemperature Shutdown](#page-28-0). The feature provides a smooth audio response and allows for uninterrupted music playback when temperature limits are crossed. That means the TAS6754-Q1 does not simply shut down, but continues to operate with considerable music output power while avoiding the trigger of OTSD.

The DSP of TAS6754-Q1 monitors the die temperature continuously in real-time for safe operation. The device can warn the host if the die temperature is approaching the OTW limits. TAS6754-Q1 still functions until the temperature reaches the OTSD threshold, at which either individual channels or the amplifier is shut down.

If the channel die temperature rises above the configured foldback level, the thermal gain foldback circuit initially activates. The device starts to reduce the gain in steps of 0.25dB per sample and thereby output power. This attack rate can be configured. The configured max attenuations are individual to the level and do not stack between levels.

When the temperature decreases below the foldback level, the attenuation will be held for a configurable number of samples before the attenuation will begin releasing at the gain step rate of 0.1dB per sample. This release rate of the TGFB can be programmed.

Figure 7-18. Thermal Foldback Attack and Release

7.3.6.8 Power Failures

The power supplies VBAT, PVDD, DVDD, and charge pump are monitored for undervoltage and overvoltage events as described in [Section 7.4.3.1.](#page-35-0) This automatically engages shutdown and protects the device. VBAT and PVDD safe operating voltage ranges can be found in the *[Recommended Operating Conditions](#page-6-0)* table.

The device will shutdown if DVDD supply falls below V_{POR OFF. The DVDD POR fault event is described in [Section 7.4.3.1](#page-35-0).

7.3.7 Hardware Control Pins

7.3.7.1 FAULT Pin

By default the FAULT pin reports fault events and is active low under any of the following conditions:

- Overtemperature shutdown (OTSD) Latching and non-latching
- Overcurrent Limit and Shutdown events Latching
- DC Detect Latching

Register bits are available to mask fault categories from reporting to the FAULT pin. These bits only mask the setting of the pin and do not affect the register reporting or protection of the device. Additional fault events can be assigned to be reported by the FAULT pin. These include:

- Power Faults Latching and non-latching
- DC Load Diagnostic faults
- Real-time Load Diagnostic reports Latching and non-latching
- Clock Errors Latching
- Charge Pump faults Latching and non-latching
- Warning events

This pin is an open-drain output with an internal 110 k Ω pull-up resistor to DVDD.

7.3.7.2 PD Pin

The PD pin is active low. When asserted the device goes into shutdown and current draw is limited to a minimum. During shutdown all internal blocks are powered off and registers initialize to their default values during the next start-up. ategories from reporting to the FAULT pin. These bits only mask the
jistic reporting or protection of the device. Additional fault events can
in. These include:
The simulation of the device and alternating
and non-latchin

This pin has a 110 k Ω internal pull-down resistor.

7.3.7.3 STBY Pin

The STBY pin is active low. When asserted it sets the device into DEEP SLEEP state. In this mode the device has a reduced current while the output pins are placed into a Hi-Z state. All internal analog bias are disabled. In DEEP SLEEP and while DVDD is present, the I²C bus is active and the internal registers are active.

This pin has a 110 k Ω internal pull-down resistor.

7.3.7.4 GPIO Pins

TAS6754-Q1 offers two configurable GPIO pins. They can be configured as input or outputs. The pins must be configured through I2C before becoming operational after [Device Initialization and Power-On-Reset \(POR\).](#page-15-0)

7.3.7.4.1 General Purpose Input

A General Purpose Input (GPI) pin can be configured by assigning a function to the pin through I2C in the related register.

7.3.7.4.2 General Purpose Output

A General Purpose Output (GPO) pin can be configured by writing the value of the intended output function to the GPO pin configuration register through l^2C . lists the GPO configuration register address for all GPIO pins.

7.3.7.5 Advanced GPIO functions

7.3.7.5.1 Clock Synchronization

TAS6754-Q1 supports multiple options for clock synchronization to improve system EMI behavior and control supply peak current conditions.

7.3.7.5.1.1 External SYNC signal (GPIO sync)

Multiple TAS6754-Q1 synchronize their clocks by using an externally provided sync signal.

Figure 7-19. External SYNC signal architecture

Figure 7-20. GPIO Sync source signal

7.3.7.5.1.2 Synchronization through the audio serial clock (SCLK)

Multiple TAS6754-Q1 synchronize their clocks through the Audio Serial Clock (SCLK).

7.3.7.5.1.3 TAS6754-Q1 as clock source for external devices

This synchronization option allows the TAS6754-Q1 to share its clock with external system components such as a DC-DC regulator. In this mode the device shares its internal ramp clock through the selected GPIO pin. If Spread Spectrum is enabled, this clock output will be affected and share the spread signal frequency with the

connected component. Refer to the technical documentation of the connected system components to ensure correct sequencing for clock synchronization and avoiding unexpected system behavior.

Figure 7-22. Clock source for external devices architecture

7.4 Device Functional Modes

7.4.1 Internal Reporting Signals

To support software driver development, the TAS6754-Q1 allows the flexible configuration of internal fault and warning signals. These signals, where applicable, can be configured based on current device status registers or events stored in memory registers. These signals can be configured and routed to the available GPIO pins for signaling purposes.

7.4.1.1 Fault Signal

Automotive systems have a high demand on gathering device information in case of unexpected conditions. of TAS6754-Q1 allow for a flexible configuration of fault information necessary for higher level system software to effectively control the system.

The Fault Signal can be configured to be active in response to the following [Fault Events](#page-35-0):

- Power Fault (latched or non-latched)
- Overtemperature Shutdown (latched or non-latched)
- DC Load Diagnostic events
- Overcurrent limiting and shutdown (latched)
- DC Detect (latched)
- Channels entering the FAULT state
- Real-Time Load Diagnostics faults (latched)
- Clock error (latched)
- Charge Pump Fault (latched)

The Fault Signal, by default, gets routed to the [FAULT](#page-30-0) to create a HW signal. The Fault Signal can optionally be routed to additional [GPIO pins.](#page-30-0)

There are two report bits for Fault signals:

- GLOBAL FAULT Reports any active fault in device, regardless of fault signal configuration
- FAULT SIGNAL Reports active fault signals that are configured accordingly in the fault signal configuration registers

7.4.1.2 Warning Signal

The Warning Signal can be configured to be active in response to the following [Warning Events](#page-36-0):

- Power Fault (latched or non-latched)
- Overtemperature Shutdown (latched or non-latched)
- Overtemperature Warning (latched or non-latched)
- DC Load Diagnostic events
- Overcurrent limiting (latched or non-latched)
- Clip Detect (latched or non-latched)
- Real-Time Load Diagnostics faults (latched)
- Clock error (latched)

The Warning Signal is by default not routed to a pin. TAS6754-Q1 can be configured to route warning signals to [GPIO pins](#page-30-0) to create a HW signal. • Power Fault (latched or non-latched)

• Overtemperature Shutdown (latched or non-latched)

• Overcurent limiting and shutdown (latched)

• Overcurent limiting and shutdown (latched)

• Channels entering the FAULT state

- GLOBAL WARNING Reports any active warning in device, regardless of warning signal configuration with the exception of Clock Fault events.
- WARNING SIGNAL Reports active warning signals that are configured accordingly in the warning signal configuration registers

7.4.2 Device States and Flags

7.4.2.1 Audio Channel States

Every audio channel has its set of states that carefully control the set up and shut down procedure of an audio path from source to load. These states are listed in [Table 7-2](#page-34-0).

Table 7-2. Audio Channel States

7.4.2.1.1 SHUTDOWN State

The device remains in shutdown when the \overline{PD} pin is pulled low. All internal regulators are disabled for minimal power consumption.

Releasing the PD pin starts the device and resets all registers to their default value. SHUTDOWN is maintained on a device level with no deviation for individual channels.

7.4.2.1.2 DEEP SLEEP State

DEEP SLEEP puts the device in a standby state. In DEEP SLEEP, the I²C communication and registers as well as the 1.5V LDO for the digital core are active. All other regulators remain deactivated to save energy.

This state is ideally suited to configure the device through l^2C before powering up. Unlike SHUTDOWN state, entering or exiting DEEP SLEEP state maintains the register map and DSP memory.

7.4.2.1.3 LOAD DIAG State

Diagnostic mode engages the DC Diagnostic circuitry to test for Short to Power, Short to Ground, Shorted Load and Open Load without activating the output power stage. These tests must be completed without fault before the Output FETs can be activated. For a more detailed description see [DC Load Diagnostics.](#page-24-0)

The DC diagnostics are available as soon as the device supplies are within the *[Recommended Operating](#page-5-0) [Conditions](#page-5-0)*. The DC diagnostics do not rely on external audio input signals or clock and sync frequencies to be available. **PD** pin is pulled low. All internal regulators are disabled for minimal
directes all registers to their default value. SHUTDOWN is maintained
didual channels.
 State. In DEEP SLEEP, the ^pC communication and registers

LOAD DIAG can be set on a channel level. Channels transition to SLEEP state mode after successfully passing the diagnostic tests.

7.4.2.1.4 SLEEP State

SLEEP state activates further functional blocks in comparison to the DEEP SLEEP state, including the internal LDO for analog circuitry and gate driver. The supply for the Digital-to-PWM conversion remains deactivated.

SLEEP can be set on a channel level. Each channel transitions to Hi-Z state by setting the State Control register to either Hi-Z or PLAY under the condition that no clock error is present.

7.4.2.1.5 Hi-Z State

In Hi-Z state the output driver is set to a high impedance state while all other blocks are fully functional.

The channel transitions to PLAY state by setting the State Control register to PLAY.

7.4.2.1.6 PLAY State

In PLAY state the device is fully operational. The output stages are active, switching and amplify the input signal.

[Real-Time Load Diagnostics](#page-27-0) can be activated to monitor the connected load for shorts or open conditions.

7.4.2.1.7 FAULT State

FAULT state is a device-internally generated mode that cannot be manually set by the user.

If one or more channels of the device are in PLAY state and encounter a fault, the device may need to take protective actions and shutdown one or more audio channels. The output FETs of only the affected channels are turned off and the output pins become high impedance. The reported state for affected channels is 'FAULT'.

Possible reasons for individual channels to enter this state are:

- Overcurrent Shutdown
- Load Current fault
- DC Diag fault
- Real-Time Load Diagnostic fault
- Channel over temperature shutdown, if configured to no auto-recovery

Possible reason for all channels to enter this state is:

- Global over temperature shutdown, if configured to no auto-recovery
- Charge Pump fault

7.4.2.1.8 Auto Recovery (AUTOREC) State

AUTOREC is a device-internally generated state that cannot be manually set by the user.

If one or more channels of the device are in PLAY state and encounter a fault, the device may need to take protective actions and shutdown one or more audio channels. The output FETs of the affected channels are turned off and the output pins become high impedance. Once the cause for the protective shutdown is no longer present, the device auto-recovers and resumes back to PLAY. The reported state for affected channels is 'AUTOREC'. Global over temperature shutdown, if configured to no auto-recover

7.42.18 *Auto Recovery (AUTOREC)* State

AUTOREC is a device-internally generated state that cannot be manual

from cor more channels of the device are

Possible reason for individual channels to enter this state is:

• Channel over temperature shutdown, OTSD(i), if configured to auto-recovery

Possible reasons for all channels to enter this state are:

- Power failures
- Clock Error
- Global over temperature shutdown, OTSD, if configured to auto-recovery

7.4.2.2 Status and Memory Registers

7.4.3 Fault Events

7.4.3.1 Power Fault Events

Power fault events are by default masked from pin reporting. This can be enabled. See [FAULT](#page-30-0) for more details.

7.4.3.1.1 DVDD Power-On-Reset (POR)

When DVDD falls below V_{POR _{OFF}, the device shuts down. All channels are set to SLEEP State, the DSP is disabled and I²C communication terminates. When DVDD rises above V_{POR SET} or when the device is first powered and DVDD rises above V_{POR SET} the device initiates a Power-On-Reset routine. During this routine all registers and device states are set to default values.

7.4.3.1.2 DVDD Undervoltage Fault

The DVDD undervoltage (UV) protection detects low voltages on the DVDD pin. In the event of a UV condition, the device will move all channels from PLAY/HI-Z to Auto Recovery (AUTOREC) state, disable the DSP, and update the $1²C$ report registers.

7.4.3.1.3 VBAT Undervoltage Fault

When the VBAT supply rail falls below nominal range, a VBAT Underoltage Fault event is created and the device enters into Auto Recovery (AUTOREC) state. Once VBAT rises back up into nominal range, the fault event is cleared.

7.4.3.1.4 PVDD Overvoltage Fault

When the PVDD supply rail rises above nominal range, a PVDD Overvoltage Fault event is created and the device enters into [Auto Recovery \(AUTOREC\) state.](#page-35-0) Once PVDD falls back down into nominal range, the fault event is cleared.

7.4.3.1.5 PVDD Undervoltage Fault

When the PVDD supply rail falls below nominal range, a PVDD Undervoltage Fault event is created and the device enters into [Auto Recovery \(AUTOREC\) state](#page-35-0). Once PVDD rises back up into nominal range, the fault event is cleared.

7.4.3.2 Overtemperature Shutdown (OTSD) Event

Section [Overtemperature Shutdown](#page-27-0) describes the circumstances under which the device creates an OTSD event as well as the configurable recovery behavior.

7.4.3.3 Overcurrent Limit Fault Event

Section [Overcurrent Limit \(Cycle-By-Cycle\)](#page-27-0) describes the circumstances under which the device creates an Overcurrent Limit Fault event. This is a transient event that only lasts for a limited time.

7.4.3.4 Overcurrent Shutdown Event

Section [Overcurrent Protection](#page-27-0) describes the circumstances under which the device creates an OCSD event.

As Overrcurrent Shutdown (OCSD) Event is a transient event it is not reported in a status register. The latched OCSD events are reported in Channel Overcurrent and DC Detection Fault Memory Register. Affected channels are placed in [FAULT state.](#page-35-0)

7.4.3.5 DC Fault Event

Section [DC Detect](#page-27-0) describes the circumstances under which the device creates an DC Fault event.

As DC Fault Event is a transient event it is not reported in a status register. The latched DC Fault events are reported in Channel Overcurrent and DC Detection Fault Memory Register. Affected channels are placed in [FAULT state.](#page-35-0)

7.4.3.6 Clock Error Event

The [Clock Rates](#page-20-0) section describes the supported Audio Data Formats, Bit Depths, and Clock Rates. If these conditions are violated or the clock is halted, the device reports a Clock Error Fault event and the device gracefully transitions to the [AUTOREC state.](#page-35-0) After audio clock recovery, the device automatically returns to the previous state. e) describes the circumstances under which the device creates an polisient event that only lasts for a limited time.

the circumstances under which the device creates an OCSD event.

Is a transient event it is not reported

A Clock Error Event is a transient event it is not reported in a status register.

7.4.3.7 Charge Pump Fault Event

[Section 7.3.6.6](#page-28-0) describes the circumstances under which the device creates an charge pump fault event as well as the recovery behavior.

7.4.4 Warning Events

7.4.4.1 Overtemperature Warning Event

Section [Overtemperature Warning](#page-29-0) describes the circumstances under which the device creates a Overtemperature Warning event.

7.4.4.2 Overcurrent Limit Warning Event

Section [Overcurrent Limit \(Cycle -By-Cycle\)](#page-27-0) describes the circumstances under which the device creates an Overcurrent Limit Warning event. This is a transient event that only lasts for a limited time.

7.4.4.3 Clip Detect Warning Event

Section [Clip Detect](#page-28-0) describes the circumstances under which the device creates a Clip Detect Warning event.

7.5 Programming

7.5.1 I ²C Serial Communication Bus

The device communicates with the system processor through the $12C$ serial communication bus as an $12C$ target-only device and supports 100-kHz and 400-kHz data transfer rates for random and sequential write and read operation. The processor can poll the device through $1²C$ to determine the operating status, configure settings, or run diagnostics.

The TAS6754-Q1 register map and DSP memory span multiple pages and books. The user should change from page to page before writing individual registers or DSP memory. Changing from page to page is accomplished via register 0 on each page. This register value selects the page address, from 0 to 255. All registers listed in the TAS6754-Q1 Data sheet belong to Page 0.

For a complete list and description of all I²C controls, see the Register Maps section.

7.5.2 I ²C Address Selection

TAS6754-Q1 supports eight I²C addresses, thus up to eight devices can be used together in a system with no additional bus switching hardware.

The pull-up or pull-down resistor connected between the device I2C_ADDR pin and the DVDD-rail (pull-up) or GND (pull-down) determines the I²C address during power up. The I2C address latches after a POR event and is locked until the next POR event.

Table 7-3. I2C Addresses

7.5.3 I ²C Bus Protocol

The I²C bus uses two signals, SDA (data) and SCL (clock), to communicate between integrated circuits in a system. Data is transferred on the bus serially, one bit at a time. The address and data are transferred in byte (8- bit) format with the most-significant bit (MSB) transferred first. In addition, each byte transferred on the bus is acknowledged by the receiving device with an acknowledge bit. Each transfer operation begins with the controller device driving a start condition on the bus and ends with the controller device driving a stop condition on the bus. The bus uses transitions on the data terminal (SDA) while the clock is HIGH to indicate a start and stop conditions. A HIGH-to-LOW transition on SDA indicates a start, and a LOW-to-HIGH transition indicates a stop. Normal data bit transitions must occur within the low time of the clock period. The controller generates the 7-bit target address and the read/write (R/W) bit to open communication with another device and then wait for an acknowledge condition. The device holds SDA LOW during the acknowledge-clock period to indicate an acknowledgment. When this occurs, the controller transmits the next byte of the sequence. Each device is addressed by a unique 7-bit target address plus a R/W bit (1 byte). All compatible devices share the same signals via a bidirectional bus using a wired-AND connection. An external pull-up resistor must be used for the SDA and SCL signals to set the HIGH level for the bus. The number of bytes that can be transmitted between start and stop conditions is unlimited. When the last word transfers, the controller generates a stop condition to release the bus. **PACE THE PRODUCTS SHERIFOR CONTROL**

TASF754-Q1 supports eight i²C addresses, thus up to eight devices

The pull-up or pull-down resistor connected between the device I2C

The pull-up or pull-down in eister connected

7.5.4 Random Write

As shown in Figure 7-25 , a single-byte data-write transfer begins with the controller device transmitting a start condition, followed by the I2C device address, and then read/write bit. The read/write bit determines the direction of the data transfer. For a write data transfer, the read/write bit is a 0. After receiving the correct ${}^{12}C$ device address and the read/write bit, the device responds with an acknowledge bit. Next, the controller transmits the address byte or bytes corresponding to the internal memory address being accessed. After receiving the address byte, the device again responds with an acknowledge bit. Next, the controller device transmits the data byte to be written to the memory address being accessed. After receiving the data byte, the device again responds with an acknowledge bit. Finally, the controller device transmits a stop condition to complete the single-byte data-write transfer.

7.5.5 Sequential Write

A sequential data-write transfer is identical to a single-byte data-write transfer except that multiple data bytes are transmitted by the controller to the device as shown in [Figure 7-26.](#page-39-0) After receiving each data byte, the device responds with an acknowledge bit and the I^2C subaddress is automatically incremented by one.

Figure 7-26. Sequential Write Transfer

7.5.6 Random Read

As shown in Figure 7-27, a single-byte data-read transfer begins with the controller device transmitting a start condition followed by the I2C device address and the read/write bit. For the data-read transfer, both a write followed by a read are done. Initially, a write is done to transfer the address byte or bytes of the internal memory address to be read. As a result, the read/write bit is a 0. After receiving the address and the read/write bit, the device responds with an acknowledge bit. In addition, after sending the internal memory address byte or bytes, the controller device transmits another start condition followed by the address and the read/write bit again. This time the read/write bit is a 1, indicating a read transfer. After receiving the address and the read/write bit, the device again responds with an acknowledge bit. Next, the device transmits the data byte from the memory address being read. After receiving the data byte, the controller device transmits a not-acknowledge followed by a stop condition to complete the single-byte data-read transfer. As shown in Figure 7-27, a single-byle data-read transfer begins with
condition followed by a read Are done. Initially, a write is done to transfer the add
address to be read. As a result, the readwhite bt is a 0. After r

Figure 7-27. Random Read Transfer

7.5.7 Sequential Read

A sequential data-read transfer is identical to a single-byte data-read transfer except that multiple data bytes are transmitted by the device to the controller device as shown in [Figure 7-28.](#page-40-0) Except for the last data byte, the controller device responds with an acknowledge bit after receiving each data byte and automatically increments the $1²C$ subaddress by one. After receiving the last data byte, the controller device transmits a not-acknowledge followed by a stop condition to complete the transfer.

Figure 7-28. Sequential Read Transfer

8 Application Information Disclaimer

Note

Information in the following applications sections is not part of the TI component specification, and TI does not warrant its accuracy or completeness. TI's customers are responsible for determining suitability of components for their purposes, as well as validating and testing their design implementation to confirm system functionality.

8.1 Application Information

The TAS6754-Q1 is a four-channel digital input Class-D audio-amplifier design with integrated real time Current Feedback and DSP for use in automotive head units and external amplifier modules. The TAS6754-Q1 incorporates the necessary functionality to perform in demanding automotive OEM applications.

8.1.1 Reconstruction Filter Design

The amplifier outputs are driven by high-current LDMOS transistors in an H-bridge configuration. The OUTP transistors are either fully off or fully on. The result is a square-wave output signal with a duty cycle that is proportional to the amplitude of the audio signal. An LC demodulation filter is used to recover the audio signal. The filter attenuates the high-frequency components of the output signals that are out of the audio band. The design of the demodulation filter significantly affects the audio performance of the power amplifier. Therefore, to meet the system THD+N requirements, the selection of the inductors used in the output filter should be carefully considered. 8.1.1 Reconstruction Filter Design

The amplifier outputs are driven by high-current LDMOS transistors

transistors are either fully off or fully on. The result is a square-way

The filter attenuates the high-frequency co

[TAS6754-Q1](https://www.ti.com/product/TAS6754-Q1) [SLOSE88](https://www.ti.com/lit/pdf/SLOSE88) – DECEMBER 2024

8.2 Typical Application

8.2.1 BTL Application

Figure 8-1 shows the schematic for a typical 4 channel solution.

8.3 Power Supply Recommendations

The TAS6754-Q1 requires three power supplies. The PVDD supply is the high-current supply in the recommended supply range. The VBAT supply is lower current supply that must be in the recommended supply range. The PVDD and VBAT pins can be connected to the same supply. When using a higher voltage, it is recommended to use automotive battery voltage for VBAT for improved efficiency. The DVDD supply is the 1.8Vdc or 3.3Vdc logic supply and must be maintained in the tolerance as shown in the [Section 5.2](#page-5-0).

8.4 Layout

8.4.1 Layout Guidelines

The TAS6754-Q1 has two output channels on each side of the device, to achieve the best thermal performance. See [Typical Application](#page-42-0) for a typical application schematic. [Figure 8-2](#page-44-0) shows a reference low EMI layout.

8.4.1.1 Electrical Connection of Thermal pad and Heat Sink

For the DKQ package, the heat sink connected to the thermal pad of the device should be connected to GND. The heat slug must not be connected to any other electrical node.

8.4.1.2 EMI Considerations

Automotive-level EMI performance depends on both careful integrated circuit design and good system-level design. Controlling sources of electromagnetic interference (EMI) was a major consideration in all aspects of the design. The design has minimal parasitic inductances because of the short leads on the package which reduces the EMI that results from current passing from the die to the system PCB. Each channel also operates at a different phase. The design also incorporates circuitry that optimizes output transitions that cause EMI. For the DKO package, the heat sink connected to the thermal pad of

The heat slug must not be connected to any other electrical node.
 **ALMO model with performance depends on both careful integral

ALMO model EMI performan**

For optimizing the EMI a solid ground layer plane is recommended. While the TAS6754-Q1 EVM layout is a good starting point, it is not recommended for EMC testing. Further board level optimizations are suggested to ensure that the EMI requirements are passed. A first reference is seen in [Figure 8-2](#page-44-0)

8.4.1.3 General Guidelines

- The decoupling capacitors on PVDD, **A**, are very close to the device with the ground return close to the ground pins. 8 100nF capacitors are recommended as [Figure 8-2](#page-44-0) shows. Another 4 1uF capacitors can be placed on the back of the PCB.
- Traces that carry large currents should incorporate multiple vias, **B**, to reduce the series impedance of these traces.
- A ground plane, **C**, on the same side as the device pins, helps reduce EMI by providing a very-low loop impedance for the high-frequency switching current. This plane should have many vias between the ground planes on other layers.
- The ground connections for the capacitors in the LC filter, **D**, have a direct path back to the device and also the ground return for each channel is the shared. This direct path allows for improved common mode EMI rejection. This should be on the same layer of the PCB as the TAS6754-Q1.
- OUT_xP inductor, OUT_xP to OUT_xM capacitor, and the OUT_xM to GND capacitor, **E**, need to have minimum loop size, starting from the device's OUT pin to GND pins. These are the switching related PCB traces. The loop size directly influences the electric field coupling.
- Heat-sink mounting screws, **F**, should be close to the device to keep the loop short from the package to ground, providing a low impedance trace for the high frequency noise coupled into the heatsink back to the PCB.

8.4.2 Layout Example

Figure 8-2. TAS6754-Q1 Recommended Layout

8.4.3 Thermal Considerations

The thermally enhanced PowerPAD package has an exposed pad up for connection to a heat sink. The output power of any amplifier is determined by the thermal performance of the amplifier as well as limitations placed on it by the system, such as the ambient operating temperature. The heat sink absorbs heat from the TAS6754-Q1 and transfers it to the air. With proper thermal management this process can reach equilibrium and heat can be continually transferred from the device. Heat sinks can be smaller than that of classic linear amplifier design because of the excellent efficiency of class-D amplifiers. This device is intended for use with a heat sink, therefore, $R_{\theta JC}$ is used as the thermal resistance from junction to the exposed metal package. This resistance dominates the thermal management, so other thermal transfers is not considered. The thermal resistance of R_{θJA} (junction to ambient) is required to determine the full thermal solution. The thermal resistance is comprised of the following components:

- $R_{\theta\text{JC}}$ of the TAS6754-Q1
- Thermal resistance of the thermal interface material
- Thermal resistance of the heat sink

9 Device and Documentation Support

TI offers an extensive line of development tools. Tools and software to evaluate the performance of the device, generate code, and develop solutions are listed below.

9.1 Documentation Support

9.1.1 Related Documentation

For related oceumatnion see the following:

- [TAS6754Q1EVM Evaluation Module User's Guide](https://www.ti.com/lit/pdf/slou585)
- [PurePath™ Console 3](https://www.ti.com/tool/PUREPATHCONSOLE) Graphical Development Suite

9.2 Receiving Notification of Documentation Updates

To receive notification of documentation updates, navigate to the device product folder on [ti.com.](https://www.ti.com) Click on *Notifications* to register and receive a weekly digest of any product information that has changed. For change details, review the revision history included in any revised document.

9.3 Support Resources

TI E2E™ [support forums](https://e2e.ti.com) are an engineer's go-to source for fast, verified answers and design help — straight from the experts. Search existing answers or ask your own question to get the quick design help you need. From the text of the square of the text of the experiment For the experiment Section of Links control (Links control of the experiment Sec

Linked content is provided "AS IS" by the respective contributors. They do not constitute TI specifications and do not necessarily reflect TI's views; see TI's [Terms of Use.](https://www.ti.com/corp/docs/legal/termsofuse.shtml)

9.4 Trademarks

TI E2E™ is a trademark of Texas Instruments.

All trademarks are the property of their respective owners.

9.5 Electrostatic Discharge Caution

This integrated circuit can be damaged by ESD. Texas Instruments recommends that all integrated circuits be handled with appropriate precautions. Failure to observe proper handling and installation procedures can cause damage.

ESD damage can range from subtle performance degradation to complete device failure. Precision integrated circuits may be more susceptible to damage because very small parametric changes could cause the device not to meet its published specifications.

9.6 Glossary

[TI Glossary](https://www.ti.com/lit/pdf/SLYZ022) This glossary lists and explains terms, acronyms, and definitions.

10 Revision History

NOTE: Page numbers for previous revisions may differ from page numbers in the current version.

11 Mechanical, Packaging, and Orderable Information

The following pages include mechanical, packaging, and orderable information. This information is the most current data available for the designated devices. This data is subject to change without notice and revision of this document. For browser-based versions of this data sheet, refer to the left-hand navigation.

PACKAGING INFORMATION

(1) The marketing status values are defined as follows:

ACTIVE: Product device recommended for new designs.

LIFEBUY: TI has announced that the device will be discontinued, and a lifetime-buy period is in effect.

NRND: Not recommended for new designs. Device is in production to support existing customers, but TI does not recommend using this part in a new design.

PREVIEW: Device has been announced but is not in production. Samples may or may not be available.

OBSOLETE: TI has discontinued the production of the device.

⁽²⁾ RoHS: TI defines "RoHS" to mean semiconductor products that are compliant with the current EU RoHS requirements for all 10 RoHS substances, including the requirement that RoHS substance do not exceed 0.1% by weight in homogeneous materials. Where designed to be soldered at high temperatures, "RoHS" products are suitable for use in specified lead-free processes. TI may reference these types of products as "Pb-Free".

RoHS Exempt: TI defines "RoHS Exempt" to mean products that contain lead but are compliant with EU RoHS pursuant to a specific EU RoHS exemption.

Green: TI defines "Green" to mean the content of Chlorine (CI) and Bromine (Br) based flame retardants meet JS709B low halogen requirements of <=1000ppm threshold. Antimony trioxide based flame retardants must also meet the <=1000ppm threshold requirement.

(3) MSL, Peak Temp. - The Moisture Sensitivity Level rating according to the JEDEC industry standard classifications, and peak solder temperature.

(4) There may be additional marking, which relates to the logo, the lot trace code information, or the environmental category on the device.

(5) Multiple Device Markings will be inside parentheses. Only one Device Marking contained in parentheses and separated by a "~" will appear on a device. If a line is indented then it is a continuation of the previous line and the two combined represent the entire Device Marking for that device.

(6) Lead finish/Ball material - Orderable Devices may have multiple material finish options. Finish options are separated by a vertical ruled line. Lead finish/Ball material values may wrap to two lines if the finish value exceeds the maximum column width.

Important Information and Disclaimer:The information provided on this page represents TI's knowledge and belief as of the date that it is provided. TI bases its knowledge and belief on information provided by third parties, and makes no representation or warranty as to the accuracy of such information. Efforts are underway to better integrate information from third parties. TI has taken and continues to take reasonable steps to provide representative and accurate information but may not have conducted destructive testing or chemical analysis on incoming materials and chemicals. TI and TI suppliers consider certain information to be proprietary, and thus CAS numbers and other limited information may not be available for release.

In no event shall TI's liability arising out of such information exceed the total purchase price of the TI part(s) at issue in this document sold by TI to Customer on an annual basis.

GENERIC PACKAGE VIEW

DKQ 56

PowerPAD[™] SSOP - 2.34 mm max height

PLASTIC SMALL OUTLINE

Images above are just a representation of the package family, actual package may vary. Refer to the product data sheet for package details.

IMPORTANT NOTICE AND DISCLAIMER

TI PROVIDES TECHNICAL AND RELIABILITY DATA (INCLUDING DATA SHEETS), DESIGN RESOURCES (INCLUDING REFERENCE DESIGNS), APPLICATION OR OTHER DESIGN ADVICE, WEB TOOLS, SAFETY INFORMATION, AND OTHER RESOURCES "AS IS" AND WITH ALL FAULTS, AND DISCLAIMS ALL WARRANTIES, EXPRESS AND IMPLIED, INCLUDING WITHOUT LIMITATION ANY IMPLIED WARRANTIES OF MERCHANTABILITY, FITNESS FOR A PARTICULAR PURPOSE OR NON-INFRINGEMENT OF THIRD PARTY INTELLECTUAL PROPERTY RIGHTS.

These resources are intended for skilled developers designing with TI products. You are solely responsible for (1) selecting the appropriate TI products for your application, (2) designing, validating and testing your application, and (3) ensuring your application meets applicable standards, and any other safety, security, regulatory or other requirements.

These resources are subject to change without notice. TI grants you permission to use these resources only for development of an application that uses the TI products described in the resource. Other reproduction and display of these resources is prohibited. No license is granted to any other TI intellectual property right or to any third party intellectual property right. TI disclaims responsibility for, and you will fully indemnify TI and its representatives against, any claims, damages, costs, losses, and liabilities arising out of your use of these resources.

TI's products are provided subject to [TI's Terms of Sale](https://www.ti.com/legal/terms-conditions/terms-of-sale.html) or other applicable terms available either on [ti.com](https://www.ti.com) or provided in conjunction with such TI products. TI's provision of these resources does not expand or otherwise alter TI's applicable warranties or warranty disclaimers for TI products.

TI objects to and rejects any additional or different terms you may have proposed.

Mailing Address: Texas Instruments, Post Office Box 655303, Dallas, Texas 75265 Copyright © 2025, Texas Instruments Incorporated