

AKM

AK4311

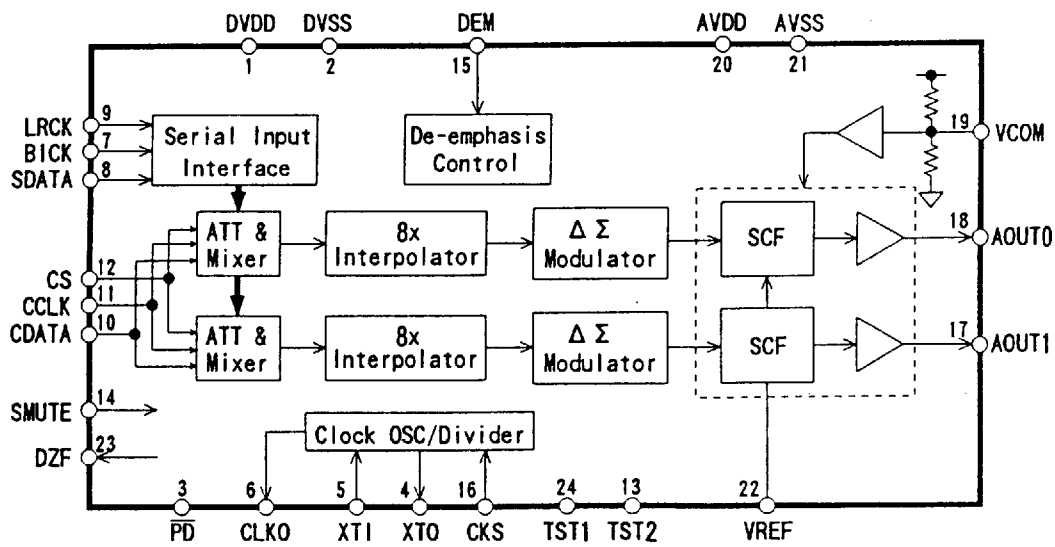
16Bit $\Delta \Sigma$ DAC with Volume & Mixing Control

General Description

The AK4311 is a 1bit stereo DAC with individually controllable channel volume for multimedia audio system. A 1bit DAC can achieve monotonicity and low distortion with no adjustment and is superior to traditional R-2R ladder based DACs. In the AK4311, the loss of accuracy from clock jitter is also improved by using SCF techniques for on-chip post filter. The AK4311 includes continuous time filter with single end output and does not need any external parts. The AK4311 also has channel mixing function and meets ATAPI CD-ROM specification.

Features

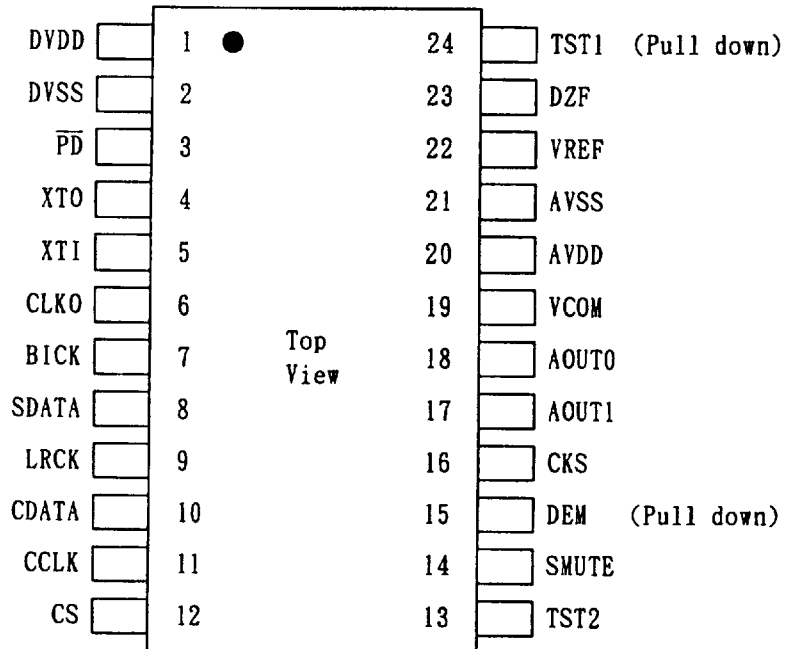
- 1bit $\Delta \Sigma$ DAC
- Sampling Rate Ranging from 10kHz to 50kHz
- On chip Perfect filtering
 - 8 times FIR Interpolator
 - 2nd order SCF
 - 2nd order CTF
 - Total Response: $\pm 0.5\text{dB}$ at 20kHz
- On chip Buffer with Single End Output
- Digital de-emphasis for 32, 44.1, 48kHz sampling
- On chip Individually controllable Channel Volume & Mixing Control
- Soft Mute
- Master Clock: 256fs or 384fs
- High Tolerance to Clock Jitter
- THD+N: -86dB
- Dynamic Range: 92dB
- Wide Voltage Operation: 3V ~ 5.5V
- Small Package: 24pin SSOP



■ Ordering Guide

AK4311-VM -10~+70°C 24pin SSOP(0.65mm pitch)
 AKD4311 Evaluation Board

■ Pin Layout



| PIN/FUNCTION | | | |
|--------------|----------|-----|---|
| No. | Pin Name | I/O | Function |
| 1 | DVDD | - | Digital Power Supply Pin |
| 2 | DVSS | - | Digital Ground Pin |
| 3 | PD | I | Power-Down Pin When at "L", the AK4311 is in power-down mode and is held in reset. The AK4311 should always be reset upon power-up. |
| 4 | XTO | 0 | Crystal Oscillator Output Pin When an external clock is input, this pin should be left floating. |
| 5 | XTI | I | Master Clock Input Pin A crystal can be connected between this pin and XTO, or an external CMOS clock can be input on XTI. |
| 6 | CLKO | 0 | Clock Buffer Output Pin An inverting clock of XTI is output from this pin. |
| 7 | BICK | I | Serial Bit Input Clock Pin This clock is used to latch audio data. |
| 8 | SDATA | I | Audio Data Input Pin 2's complement MSB-first data is input on this pin. |
| 9 | LRCK | I | L/R Clock Pin This input determines which audio channel is currently being input on SDATA pin. "H": Lch, "L": Rch |
| 10 | CDATA | I | Control Data Input Pin |
| 11 | CCLK | I | Control Clock Input Pin |
| 12 | CS | I | Chip Select Pin |
| 13 | TST2 | 0 | Test Pin Must be left floating. |
| 14 | SMUTE | I | Soft Mute Pin When this pin goes "H", soft mute cycle is initiated. |
| 15 | DEM | I | De-emphasis Enable Pin (Pull-down pin) When "H", De-emphasis of fs=44.1kHz is enabled. |
| 16 | CKS | I | Master Clock Select Pin "L": CLK=256fs, "H": CLK=384fs |
| 17 | AOUT1 | 0 | Ch1 Analog Output Pin |
| 18 | AOUT0 | 0 | Ch0 Analog Output Pin |
| 19 | VCOM | 0 | Common Voltage pin, AVDD/2 Normally connected to AVSS with a 0.1uF ceramic capacitor in parallel with a 10uF electrolytic cap. |
| 20 | AVDD | - | Analog Power Supply Pin |
| 21 | AVSS | - | Analog Ground Pin |
| 22 | VREF | I | Voltage Reference Input Pin The differential Voltage between this pin and AVSS set the analog output range. Normally connected to AVSS with a 0.1uF ceramic capacitor. |
| 23 | DZF | 0 | Zero Input Detect Pin When SDATA of both channels follow a total 8192 LRCK cycles with "0" input data, this pin goes "H". |
| 24 | TST1 | I | Test Pin (Pull-down pin) Must be left floating or tied to AVSS. |

ABSOLUTE MAXIMUM RATINGS

(AVSS, DVSS=0V; Note 1)

| Parameter | Symbol | min | max | Units |
|--|--------|------|----------|-------|
| Power Supplies: Analog (AVDD pin) | AVDD | -0.3 | 6.0 | V |
| Digital (DVDD pin) | DVDD | -0.3 | AVDD+0.3 | V |
| Input Current, Any Pin Except Supplies | IIN | - | ±10 | mA |
| Input Voltage | VIND | -0.3 | AVDD+0.3 | V |
| Ambient Operating Temperature | Ta | -10 | 70 | °C |
| Storage Temperature | Tstg | -65 | 150 | °C |

Note: 1. All voltages with respect to ground.

WARNING: Operation at or beyond these limits may result in permanent damage to the device.
Normal operation is not guaranteed at these extremes.

RECOMMENDED OPERATING CONDITIONS

(AVSS, DVSS=0V; Note 1)

| Parameter | Symbol | min | typ | max | Units |
|-----------------------------------|--------------|-----|-----|------|-------|
| Power Supplies: Analog (AVDD pin) | AVDD | 3.0 | 5.0 | 5.5 | V |
| Digital (DVDD pin) | DVDD | 3.0 | 5.0 | AVDD | V |
| AVDD-DVDD (Note 2) | ΔVDD | 0.0 | - | 1.0 | V |
| Voltage Reference (Note 3) | VREF | 2.5 | - | AVDD | V |

Notes: 1. All voltages with respect to ground.

2. AVDD and DVDD should be powered at the same time or AVDD should be powered earlier than DVDD.

3. Analog output voltage scales with the voltage of VREF.

$$AOUT(\text{typ. } @0\text{dB}) = 2.83V_{pp} * VREF / 5.$$

* AKM assumes no responsibility for the usage beyond the conditions in this data sheet.

ANALOG CHARACTERISTICS

($T_a=25^{\circ}\text{C}$; $\text{AVDD}, \text{DVDD}=5.0\text{V}$; $\text{VREF}=\text{AVDD}$; $f_s=44.1\text{kHz}$; Signal Frequency=1kHz;
 $R_L \geq 10\text{k}\Omega$; Measurement Bandwidth=10Hz~20kHz; unless otherwise specified)

| Parameter | min | typ | max | Units |
|--|------|------|------|-------------------------|
| Dynamic Characteristics (Note 4) | | | | |
| THD+N | -80 | -86 | | dB |
| (Note 5) | -74 | -83 | | |
| Dynamic Range (A-Weighted) | 86 | 92 | | dB |
| (Note 5) | 82 | 88 | | |
| S/N (A-Weighted) | 86 | 92 | | dB |
| (Note 5) | 82 | 88 | | dB |
| Interchannel Isolation | 80 | 90 | | dB |
| DC Accuracy | | | | |
| Interchannel Gain Mismatch | | 0.1 | 0.2 | dB |
| Gain Drift | | 60 | - | ppm/ $^{\circ}\text{C}$ |
| Analog Output | | | | |
| Output Voltage (Note 6) | 2.69 | 2.83 | 2.97 | V_{pp} |
| (Note 5) | 1.78 | 1.87 | 1.96 | V_{pp} |
| Load Resistance | 10 | | | $\text{k}\Omega$ |
| Power Supplies | | | | |
| Power Supply Current (Note 7) | | | | |
| Normal Operation ($\overline{\text{PD}}=\text{"H"}$) | | | | |
| AVDD | | 12 | 18 | mA |
| DVDD | | 6 | 9 | mA |
| Power-Down-Mode ($\overline{\text{PD}}=\text{"L"}$) | | | | |
| AVDD+DVDD (Note 8) | | 10 | 50 | μA |
| Power Dissipation | | | | |
| Normal Operation | | 90 | 135 | mW |
| Power-Down-Mode (Note 8) | | 50 | 250 | μW |
| Power Supply Rejection | | 50 | | dB |

Notes:4. Measured by AD725C(SHIBASOKU). Averaging mode. (★)

5. $\text{AVDD}, \text{DVDD}=3.3\text{V}$

6. Full-scale voltage(0dB). Output voltage scales with the voltage of VREF.
 $\text{AOUT}(\text{typ. } 0\text{dB})=2.83\text{V}_{pp} \cdot \text{VREF}/5$.

7. CLKO pin is open. The typical supply current of DVDD drops to 3mA at 3.3V supply voltage. The AVDD supply current does not change.

8. Power Dissipation in the power-down mode is applied with no external clocks (XTI, BICK, LRCK held "H" or "L").

When using the internal oscillation in the power-down mode, the AK4311 may draw the current of about 3mA.

| |
|-------------------------------|
| FILTER CHARACTERISTICS |
|-------------------------------|

(Ta=25°C; AVDD, DVDD=3.0~5.5V; fs=44.1kHz)

| Parameter | Symbol | min | typ | max | Units |
|---------------------------------------|---------------------------|-----|------|-------|-------|
| Digital Filter | | | | | |
| Passband | -0.1dB (Note 9) -6.0dB | PB | 0 | 22.05 | 20.0 |
| | | | - | | - |
| Stopband | (Note 9) | SB | 24.1 | | kHz |
| Passband Ripple | | PR | | ±0.06 | dB |
| Stopband Attenuation | | SA | 43 | | dB |
| Group Delay | (Note 10) | GD | - | 14.7 | 1/fs |
| Digital Filter + Analog Filter | | | | | |
| Frequency Response | 0~20.0kHz | | - | ±0.5 | dB |

Note: 9. The passband and stopband frequencies scale with fs.

For example, PB=0.4535*fs(@-0.1dB), SB=0.546*fs(@-43dB).

10. The calculating delay time which occurred by digital filtering. This time is from setting the 16bit data of both channels to input register to the output of analog signal.

| |
|--------------------------------|
| DIGITAL CHARACTERISTICS |
|--------------------------------|

(Ta=25°C; AVDD, DVDD=3~5.5V)

| Parameter | Symbol | min | typ | max | Units |
|--|--------|----------|-----|---------|-------|
| High-Level Input Voltage | VIH | 70%DVDD | - | - | V |
| Low-Level Input Voltage | VIL | - | - | 30%DVDD | V |
| Input Voltage at AC coupling (XTI pin) | VAC | 1 | - | - | Vpp |
| High-Level Output Voltage Iout=-20uA | VOH | DVDD-0.1 | - | - | V |
| Low-Level Output Voltage Iout=20uA | VOL | - | - | 0.1 | V |
| Input Leakage Current (Note 11) | Iin | - | - | ±10 | uA |

Note: 11. DEM, TSTI pins have internal pull-down devices, nominally 90kΩ.

SWITCHING CHARACTERISTICS

(Ta=25°C; AVDD, DVDD=3~5.5V; CL=20pF)

| Parameter | Symbol | min | typ | max | Unit | |
|---|------------------|-----------|------|---------|----------|-----|
| Master Clock Frequency | | | | | | |
| Crystal Resonator | 256fs: | fCLK | 7.1 | 11.2896 | 12.8 | MHz |
| | 384fs: | fCLK | 10.7 | 16.9344 | 19.2 | MHz |
| External Clock | 256fs: | fCLK | 2.56 | 11.2896 | 12.8 | MHz |
| | Pulse Width Low | tCLKL | 28 | | | ns |
| | Pulse Width High | tCLKH | 28 | | | ns |
| | 384fs: | fCLK | 3.84 | 16.9344 | 19.2 | MHz |
| | Pulse Width Low | tCLKL | 23 | | | ns |
| | Pulse Width High | tCLKH | 23 | | | ns |
| LRCK Frequency | (Note 12) | fs | 10 | 44.1 | 50 | kHz |
| Audio Interface Timing (Note 13) | | | | | | |
| BICK Period | tBCK | 312.5 | | | | ns |
| BICK Pulse Width Low | tBCKL | 100 | | | | ns |
| | tBCKH | 100 | | | tBCKL-50 | ns |
| LRCK Edge to BICK falling | tLRB | -tBCKH+50 | | | | ns |
| SDATA Hold Time | tSDH | 50 | | | | ns |
| SDATA Setup Time | tSDS | 50 | | | | ns |
| Control Interface Timing (Note 15) | | | | | | |
| CCLK Pulse Width Low | tCCKL | 100 | | | | ns |
| | tCCKH | 100 | | | | ns |
| CDATA Latch Hold Time | tCDS | 50 | | | | ns |
| CDATA Latch Setup Time | tCDH | 50 | | | | ns |
| CS Pulse Width Low | tCSW | 100 | | | | ns |
| CCLK to CS falling | tCSS | 50 | | | | ns |
| CS rising to CCLK | tCSH | 50 | | | | ns |
| Reset Timing | | | | | | |
| $\overline{\text{PD}}$ Pulse Width | (Note 16) | tRST | 100 | | | ns |

Notes:12. If the duty of LRCK changes larger than $\pm 1/8$ from 50%, the AK4311 is reset by the internal phase circuit automatically.

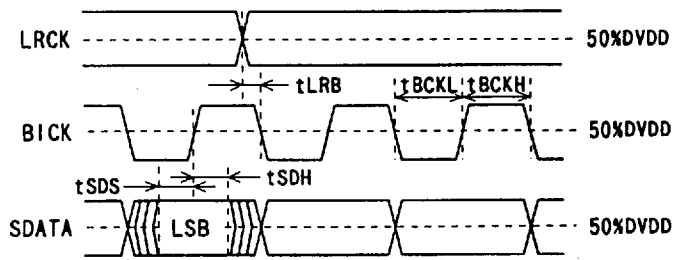
13. Refer to the operating overview section "Audio Data Interface".

14. BICK rising edge must not occur at the same time as LRCK edge.

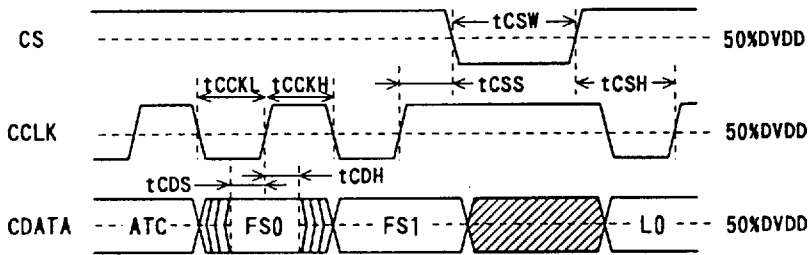
15. Refer to the operating overview section "Serial Mode Control".

16. The AK4311 can be reset by bringing $\overline{\text{PD}}$ "L" to "H" only upon power up.

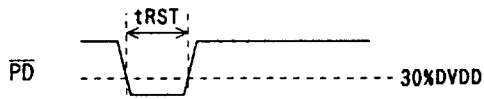
■ Timing Diagram



Audio Data Input Timing



Mode Control Timing



Reset Timing

OPERATION OVERVIEW

■ System Clock Input

The external clocks which are required to operate the AK4311 are XTI(256fs/384fs), LRCK(fs), BICK(32fs~). The master clock (XTI) should be synchronized with LRCK but the phase is free of care. The XTI is used to operate the digital interpolation filter and the delta-sigma modulator. The frequency of XTI is determined by the sampling rate(LRCK), and the setting of the Clock Select, CKS pin. Setting CKS "L" selects an XTI frequency of 256fs while setting CKS "H" selects 384fs. When the 384fs is selected, the internal master clock becomes 256fs(=384fs*2/3). *fs is audio word rate.

The master clock can be either a crystal resonator placed across the XTI and XTO pin, or external clock input to the XTI pin with the XTO pin left floating. Not only CMOS clock but sine wave signal with 1Vpp can be input to the XTI pin by AC coupling. Table 1 illustrates standard audio word rates and corresponding frequencies used in the DAC. When using internal oscillation, CLKO can not be used by external circuit at the power-down mode.

As the AK4311 includes the phase detect circuit for LRCK, the AK4311 is reset automatically when the synchronization is out of phase by changing the clock frequencies. Therefore, the reset is not needed except only upon power-up. (Please refer to the "System Reset" section.)

All external clocks(XTI, BICK, LRCK) should always be present whenever the AK4311 is in normal operation mode(PD="H"). If these clocks are not provided, the AK4311 may draw excess current and do not possibly operate properly because the device utilizes dynamic refreshed logic internally. If the external clocks are not present, the AK4311 should be in the power-down mode(PD="L").

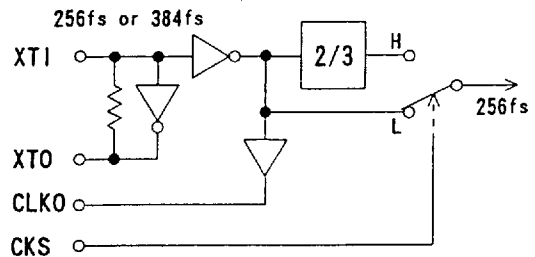


Figure 1. Internal Clock Circuit

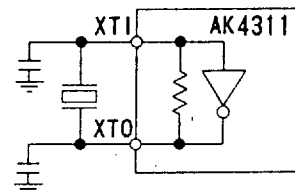


Figure 2. Crystal resonator connection

| LRCK (fs) (kHz) | CKS | XTI (MHz) |
|--------------------|-----|--------------|
| 32.0 | L | 8.1920 |
| | H | 12.2880 |
| 44.1 | L | 11.2896 |
| | H | 16.9344 |
| 48.0 | L | 12.2880 |
| | H | 18.4320 |

Table 1. Examples of System Clock

■ Audio Data Interface

The AK4311 has three serial input pins(SDATA, BICK, LRCK). Data bits is clocked into the AK4311 via SDATA pin and is latched by LRCK. The data format is MSB-first, 2's complement and LSB justified.

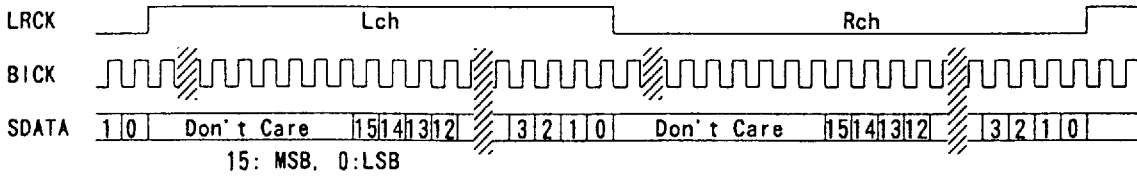


Figure 3. Audio Data Input Format

■ De-emphasis filter

The AK4311 includes the digital de-emphasis filter(tc=50/15us) by IIR filter. This filter corresponds to three sampling frequencies(32kHz, 44.1kHz, 48kHz). De-emphasis is enabled by the following two ways.

1. Way of using DEM pin

Only one de-emphasis(fs=44.1kHz) set initially can be controlled by DEM pin at resetting. The de-emphasis is enabled by setting DEM pin "H". When the frequency of de-emphasis is set by FS0, FS1 of serial mode control bits, the corresponding de-emphasis can be enabled. In this case, DEM bit in the serial mode control should be "0".

2. Way of using serial mode control

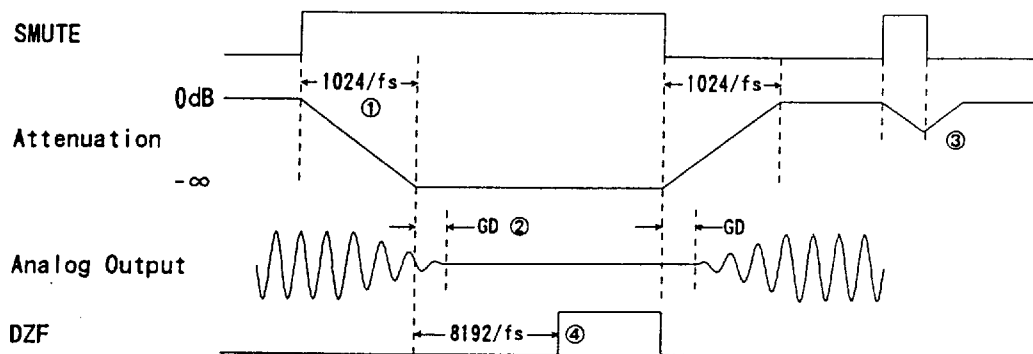
DEM pin should be open or "L". In this case, The de-emphasis corresponding to fs=32kHz, 44.1kHz, 48kHz can be controlled by DEM, FS0 and FS1 in the serial mode control bits.

■ Zero detection

When the input data at both channels are continuously zeros for 8192 LRCK cycles, DZF goes to "H". DZF immediately goes "L" if input data are not zero after going DZF "H".

■ Soft mute operation

Soft mute operation is performed at digital domain. When SMUTE goes "H", the output signal is attenuated by $-\infty$ during 1024 LRCK cycles. When SMUTE is returned to "L", the mute is cancelled and the output attenuation gradually changes to 0dB during 1024 LRCK cycles. If the soft mute is cancelled within 1024 LRCK cycles after starting the operation, the attenuation is discontinued and returned to 0dB. The soft mute is effective for changing the signal source without stopping the signal transmission.



Notes:

- ① The output signal is attenuated by $-\infty$ during 1024 LRCK cycles ($1024/f_s$).
- ② Analog output corresponding to digital input have the group delay (GD).
- ③ If the soft mute is cancelled within 1024 LRCK cycles, the attenuation is discontinued and returned to 0dB.
- ④ When the input data at both channels are continuously zeros for 8192 LRCK cycles, DZF goes to "H". DZF immediately goes "L" if input data are not zero after going DZF "H".

Figure 4. Soft mute and zero detection

■ Serial Mode Control

The AK4311 can control output attenuation level, output mode, de-emphasis type and attenuation mode via the serial interface. The serial data consists of two 8 bits for setting the attenuation level of each channel and 8 bits for the mode control.

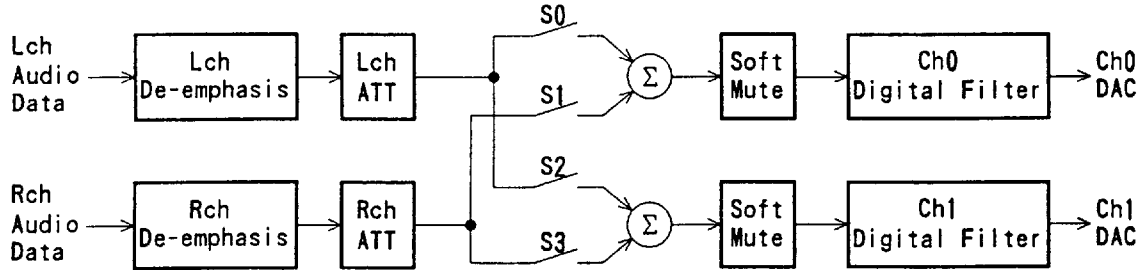
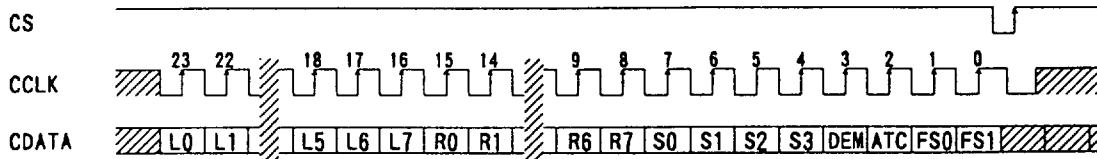


Figure 5. Configuration of attenuator and mixer



L0-L7: Lch ATT data, L0=LSB, L7=MSB :FFH at RESET
 R0-R7: Rch ATT data, R0=LSB, R7=MSB :FFH at RESET
 S0-S3: Output mode Control :1001 at RESET
 DEM: De-Emphasis Control :0 at RESET
 ATC: ATT mode Control :0 at RESET
 FS0,FS1: fs Control for de-emphasis Filter :00 at RESET

Note: CCLK should be held "H" or "L" except writing to ATT & mode registers in order to avoid the performance degradation.

Figure 6. Serial mode control timing

1. Attenuator Operation

The AK4311 has individually controllable attenuator with linear scale and 256 levels for each channel.

$$\text{Equation of attenuation level: } \text{ATT} = 20 \times \text{Log}_{10}(\text{Binary level}/255)$$

FFH: 0dB
 |
 01H: -48.1dB
 00H: Mute (Infinity zero:-∞)

The transition between ATT values is same as soft mute operation. When current value is ATT1 and new value is set as ATT2, ATT1 gradually becomes ATT2 with same operation as soft mute. If new value is set as ATT3 before reaching ATT2, ATT value gradually becomes ATT3 from the way of transition.

$$\text{Cycle time of soft mute: } T_s = 1024/fs$$

When resetting, ATT value is set 00H(Infinity zero). ATT value gradually changes from 00H to FFH(0dB) during T_s after exiting reset.

2. Output mode

The AK4311 supports the following output modes.

- Normal stereo output
- L/R Reverse output
- Monaural mixing output:(L+R)/2
- Output muting with soft mute operation

When resetting, ATT values of both channels are FFH and the attenuation levels are set 0dB. The output mode is also set normal stereo output.

| S0 | S1 | S2 | S3 | AOUT0 | AOUT1 | Mode |
|----|----|----|----|---------|---------|---------|
| 0 | 0 | 0 | 0 | MUTE | MUTE | MUTE |
| 0 | 0 | 0 | 1 | MUTE | R | |
| 0 | 0 | 1 | 0 | MUTE | L | |
| 0 | 0 | 1 | 1 | MUTE | (L+R)/2 | |
| 0 | 1 | 0 | 0 | R | MUTE | |
| 0 | 1 | 0 | 1 | R | R | |
| 0 | 1 | 1 | 0 | R | L | Reverse |
| 0 | 1 | 1 | 1 | R | (L+R)/2 | |
| 1 | 0 | 0 | 0 | L | MUTE | |
| 1 | 0 | 0 | 1 | L | R | Stereo |
| 1 | 0 | 1 | 0 | L | L | |
| 1 | 0 | 1 | 1 | L | (L+R)/2 | |
| 1 | 1 | 0 | 0 | (L+R)/2 | MUTE | |
| 1 | 1 | 0 | 1 | (L+R)/2 | R | |
| 1 | 1 | 1 | 0 | (L+R)/2 | L | |
| 1 | 1 | 1 | 1 | (L+R)/2 | (L+R)/2 | MONO |

* at RESET

Table 2. Output mode

3. De-emphasis control

DEM bit and DEM pin are ORed internally. The de-emphasis(tc=50/15us) corresponding to fs(sampling frequency) selected by FS0 and FS1 is enabled by setting DEM bit "1" or DEM pin "H". When DEM bit is "0" and DEM pin is "L", the de-emphasis is disabled and the setting of FS0 and FS1 is invalid. The de-emphasis is also disabled at FS0="1" and FS1="0". When resetting, DEM bit is set "0".

For example, when the de-emphasis is controlled by only DEM pin at fs=44.1kHz, DEM, FS0, FS1 bits should be "0". This condition is also set at resetting.

| FS0 | FS1 | Mode |
|-----|-----|---------|
| 0 | 0 | 44.1kHz |
| 1 | 0 | OFF |
| 0 | 1 | 48kHz |
| 1 | 1 | 32kHz |

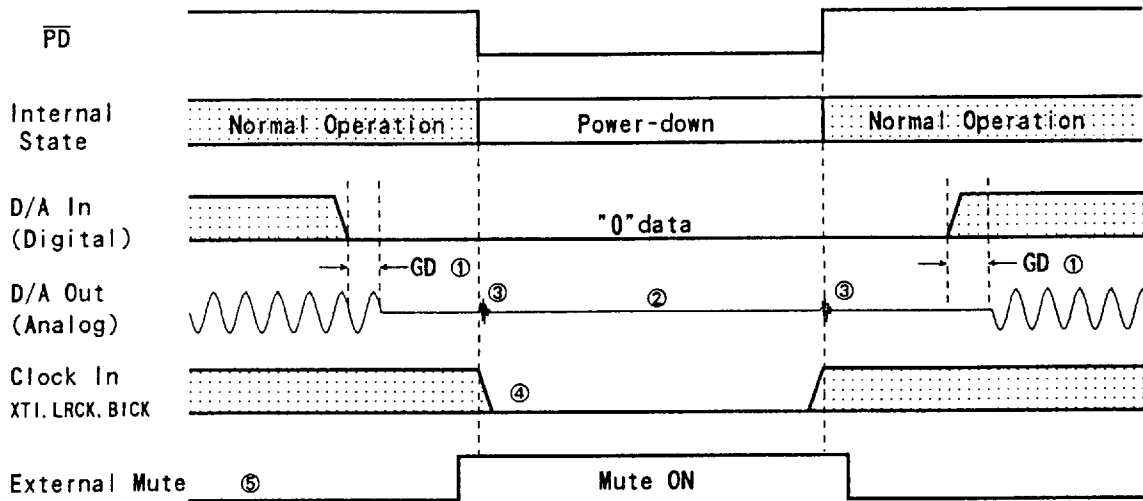
Table 3. De-emphasis filter setting
(Valid at DEM bit="1" or DEM pin="H")

4. Attenuation control

ATT values of both channels are set Lch ATT data by setting ATC bit "1". In this case, Rch ATT data is ignored. When resetting, ATC bit is set "0" (individually control).

■ Power-Down

The AK4311 is placed in the power-down mode by bringing \overline{PD} pin "L" and the analog outputs are floating(Hi-Z). ATT value is set 00H at the power-down mode. When exiting the power-down mode, ATT value returns from 00H to FFH(0dB) with soft transition during T_s . Figure 7 shows an example of the system timing at the power-down and power-up.



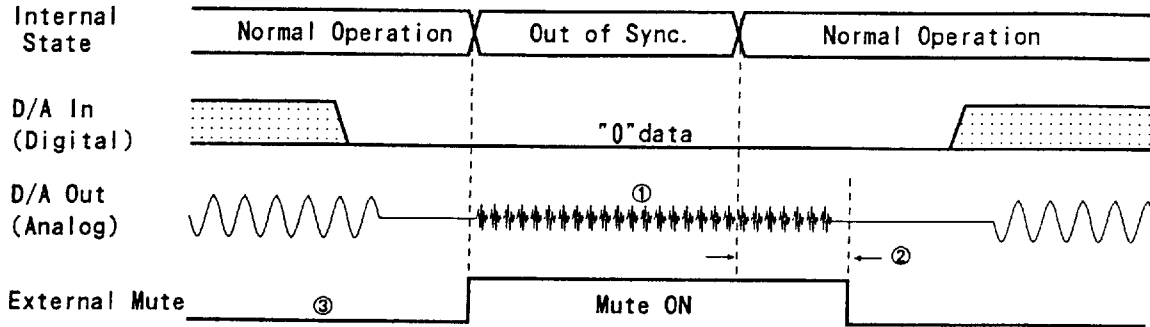
Notes:

- ① Analog output corresponding to digital input have the group delay(GD).
- ② Analog outputs are floating(Hi-Z) at the power-down mode. The output noise level is about -110dB.
- ③ Click noise about -50dB occurs at the edges("↑↓") of \overline{PD} signal.
- ④ When the external clocks(XTI, BICK, LRCK) are stopped, the AK4311 should be in the power-down mode.
- ⑤ Please mute the analog output externally if the click noise(③) influences system application. The timing example is shown in this figure.

Figure 7. Power-down/up sequence example

■ System Reset

The AK4311 should be reset once by bringing \overline{PD} "L" upon power-up. The internal timing starts clocking by LRCK "↑" upon exiting reset. If the phase difference between LRCK and internal control signals is larger than $+1/16 \sim -1/16$ of word period($1/f_s$), the synchronization of internal control signals with LRCK is done automatically at the first rising edge of LRCK. Since RAM address shifts during this synchronization, correct data would not be output until 14 sampled data are input even if the AK4311 returns to the normal operation. Refer to Figure 8.



When cycle ratio between LRCK and XTI can be not kept 1:256(1:384 at 384fs) by changing LRCK frequency etc., internal reset by out-of-synchronization may occur. Some noise occurs at resetting and after returning to normal operation. This noise also occurs even if "0" data is being input to the AK4311.

- ① Click noise is output continuously when out-of-synchronization occurs continuously.
- ② Some noise occurs until $14 \times \text{LRCK}$ cycles after LRCK returns to normal condition.
- ③ Please mute the analog output externally if there is possibility of out of synchronization in the application. The timing example is shown in this figure.

Figure 8. Out-of-synchronization timing example

1. Internal State at resetting by out-of-synchronization(★)

When the AK4311 is reset automatically by out-of-synchronization, the contents of the serial mode control register are kept during out-of-synchronization. Therefore, it is not necessary to reset the serial mode control register. After returning to normal operation, ATT value gradually changes from 00H to the kept value (Figure 9). Some noise occurs at out-of-synchronization and this noise can not be muted perfectly by enabling the SMUTE pin or MUTE bit in the serial mode control register. Please mute the analog outputs by the timing in Figure 8.

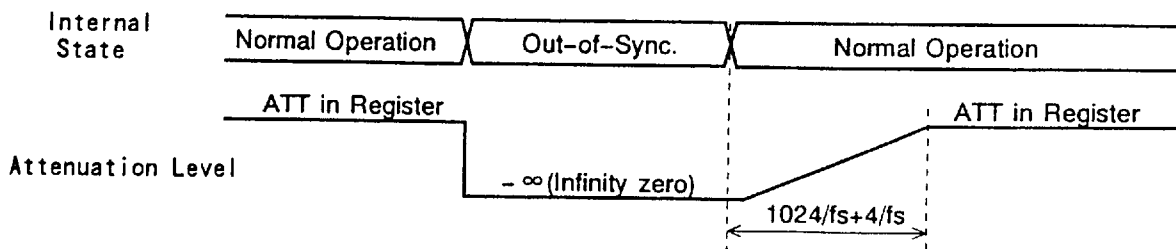


Figure 9. ATT operation at out-of-synchronization

When enabling the SMUTE pin or MUTE bit in the serial mode control register by the timing of Figure 9. The analog output corresponding the input data outputs for an instant because the ATT initial transition and the soft muting operate at the same time. There are 3 ways to avoid this phenomenon.

- a. Not use the soft muting. When using the soft muting, the soft muting is disabled before exiting out-of-synchronization.
- b. Set the input audio data to "0".
- c. Set the ATT register to 00H.

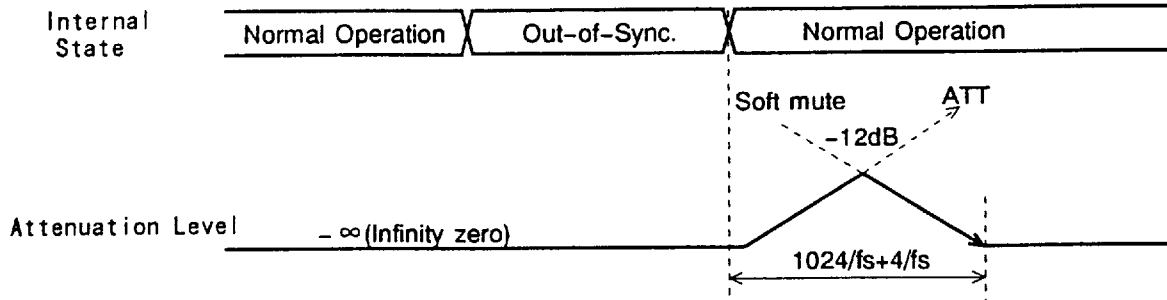


Figure 10. Out-of-synchronization timing with soft muting(@ATT=FFH)

*The solid line shows the actual attenuation level.

2. Cases where out-of-synchronization occurs

- L/R clock with 2/4 times speed is input to AK4311 in CD-ROM application.
- The clock frequency is not changed smoothly in the PLL system of Satellite Broadcasting application.
- The clock frequency is not changed smoothly between VCO and X' tal in the digital recording application.

SYSTEM DESIGN

Figure 11 shows the system connection diagram. An evaluation board[AKD4311] is available which demonstrates the optimum layout, power supply arrangements and measurement results.

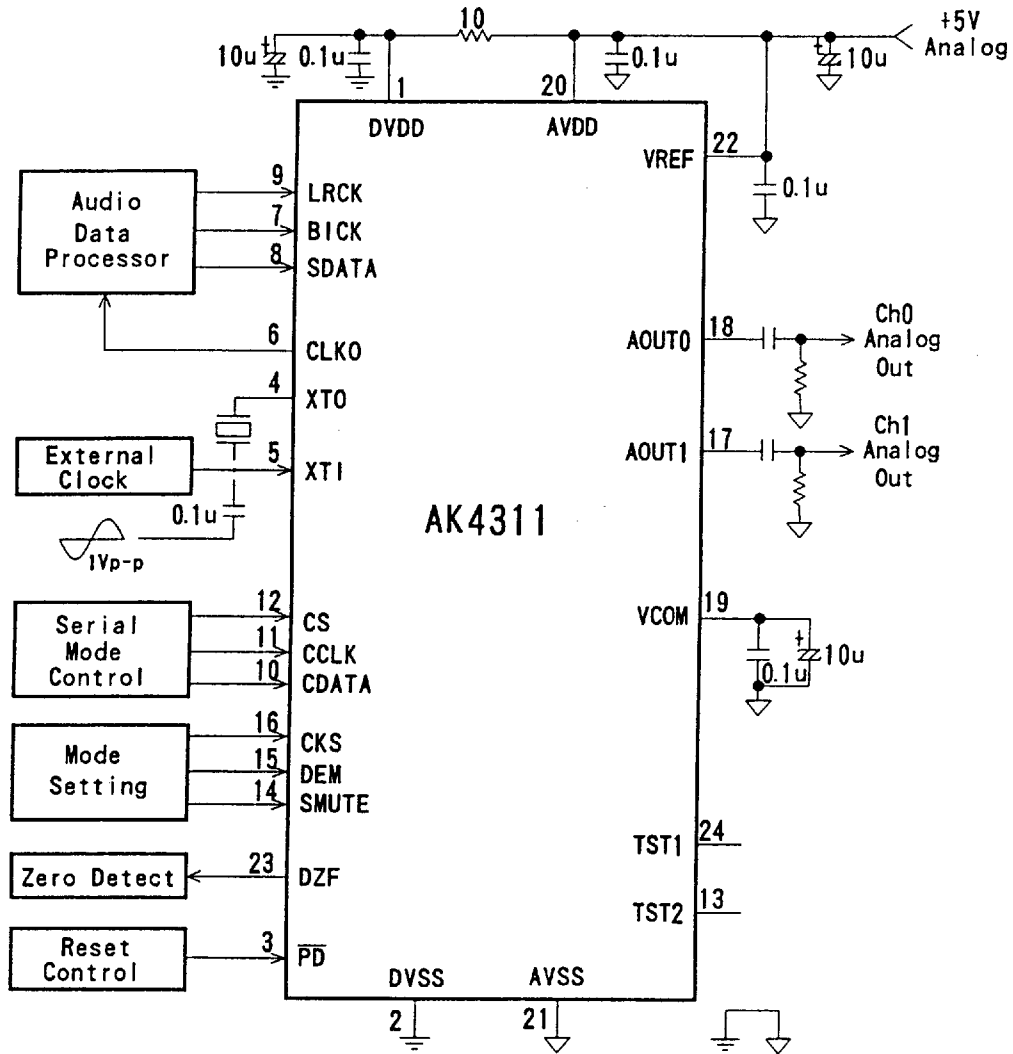


Figure 11. Typical Connection Diagram

Notes:

- LRCK=fs, BICK \geq 32fs, XTI=256fs at CKS="L", XTI=384fs at CKS="H".
- CCLK should be held "H" or "L" except writing to ATT & mode registers.
- When AOUT drives some capacitive load, some resistor should be added in series between AOUT and capacitive load.

■ System design consideration

1. Grounding and Power Supply Decoupling

To minimize coupling by digital noise, decoupling capacitors should be connected to AVDD and DVDD, respectively. AVDD is supplied from analog supply in system and DVDD is supplied from AVDD via 10Ω resistor. Alternatively if AVDD and DVDD are supplied separately, AVDD and DVDD should be powered at the same time or AVDD should be powered earlier than DVDD. Analog ground and digital ground should be connected together near to where the supplies are brought onto the printed circuit board. Decoupling capacitors for high frequency should be as near to the AK4311 device as possible, with the low value ceramic capacitor of VREF being the nearest.

2. Voltage reference

The differential Voltage between VREF and AVSS set the analog output range. VREF pin is normally connected to AVDD with a 0.1μF ceramic capacitor. VCOM is a signal ground of this chip. An electrolytic capacitor less than 10μF in parallel with a 0.1μF ceramic capacitor attached to these pins eliminates the effects of high frequency noise. No load current may be drawn from VCOM pin. All signals, especially clocks, should be kept away from the VREF and VCOM pins in order to avoid unwanted coupling into the AK4311.

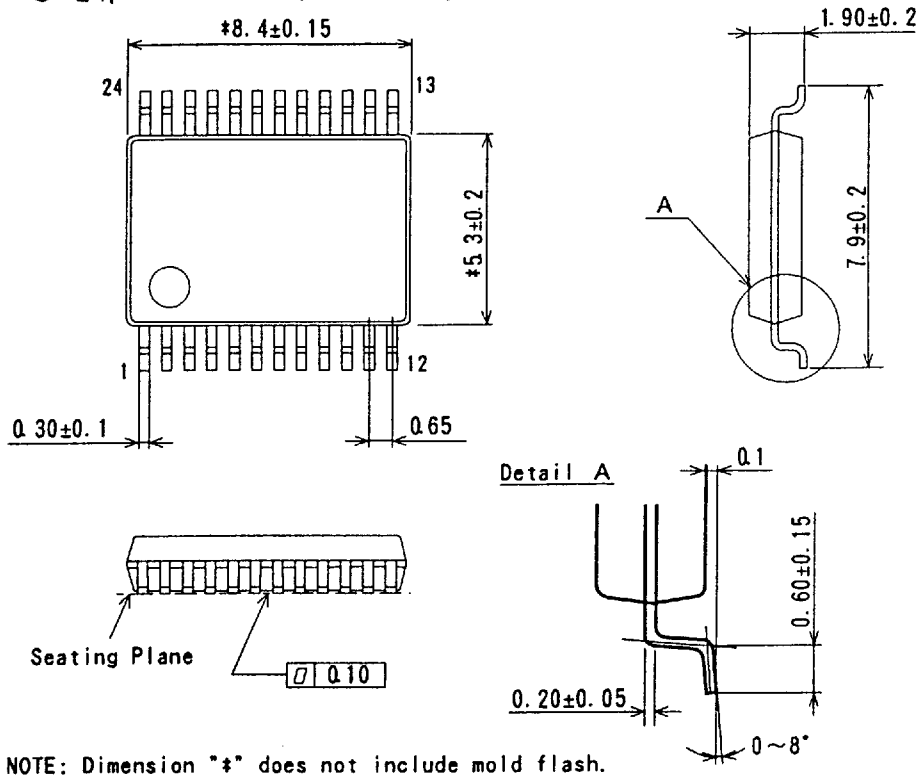
3. Analog Outputs

The analog outputs are also single-ended and centered around the VCOM voltage. The output signal range is typically 2.83Vpp. AC coupling capacitors of larger than 1μF are recommended. The internal switched-capacitor filter and continuous-time filter attenuate the noise generated by the delta-sigma modulator beyond the audio passband. Therefore, any external filters are not required for typical application. The output voltage is a positive full scale for 7FFFH(@16bit) and a negative full scale for 8000H(@16bit). The ideal output is VCOM voltage for 0000H(@16bit).

DC offsets on analog outputs are eliminated by AC coupling since analog outputs have DC offsets of a few mV.

PACKAGE

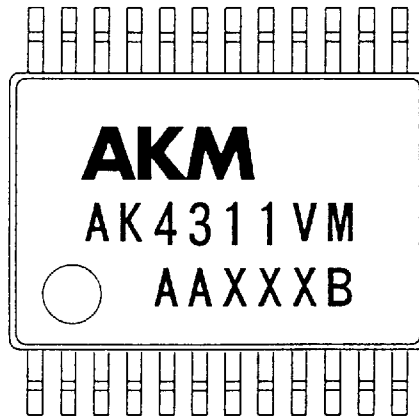
● 24pin SSOP (Unit: mm)



■ Package & Lead frame material

| | |
|--------------------------------|--------------|
| Package molding compound : | Epoxy |
| Lead frame material : | Cu |
| Lead frame surface treatment : | Solder plate |

MARKING



Contents of A A X X X B

A A : Lot# (alphabet)

X X X B : Date Code (X : numbers, B : alphabet)