AK4650

AKM

= Preliminary =

16Bit $\Delta\Sigma$ CODEC with MIC/HP/SPK-AMP & TSC

GENERAL DESCRIPTION

The AK4650 targeted at PDA and other low-power, small size applications. It features a 16-bit stereo CODEC with a built-in Microphone-Amplifier, Headphone-Amplifier, Speaker-Amplifier and Touch Screen Controller. Input circuits include a Microphone-Amplifier and an ALC (Auto Level Control) circuit. The AK4650 is connected with AC'97 controller (CPU) via AC-Link. The AK4650 is available in a very small 57-BGA, utilizing less board space than competitive offerings.

FEATURES

- 1. Resolution: 16bits
- 2. Recording Function
 - Mono Input (Single-ended or Differential Input)
 - 2 to 1 Selector (Internal and External MIC)
 - MIC Power: 2 outputs (Internal and External MIC)
 - 1st MIC Amplifier: +20dB or 0dB
 - 2nd Amplifier with ALC: +27.5dB ~ -8dB, 0.5dB Step
 - ADC Performance (@MIC-Amp=+20dB): S/(N+D): 75dB, DR, S/N: 83dB
 - MIC Detection
- 3. Playback Function
 - Digital De-emphasis Filter (tc=50/15µs, fs=32kHz, 44.1kHz, 48kHz)
 - Digital Volume (0dB ~ -63dB, 0.5dB Step, Mute)
 - Bass Boost
 - Mono Output
 - Full-differential Output
 - S/(N+D): 85dB, S/N: 95dB
 - Analog Volume: +6dB ~ -15dB, 3dB Step
 - Headphone-Amp
 - Output Power: 40mW@16Ω (HVDD=3.3V)
 - S/(N+D): 60dB@10mW, S/N: 90dB
 - Headphone Jack Detection
 - Mono Speaker-Amp
 - Output Power: 300mW@8Ω (HVDD=3.3V, ALC2=OFF)
 - S/(N+D): 55dB@110mW, S/N: 90dB
 - BTL Output
 - ALC (Auto Level Control) circuit
 - Mono Beep Input
 - AUX Input
 - Full-differential Input
 - Analog Volume: +12dB ~ -34.5dB, 1.5dB Step, Mute
 - Stereo Line Input
 - Single-ended Input
 - Analog Volume: +12dB ~ -34.5dB, 1.5dB Step, Mute

- 4. Touch Screen Controller
 - Two Auxiliary Analog Inputs
 - Pen Pressure Measurement
 - Direct Battery Measurement
 - On-Chip Voltage Reference (2.5V)
 - 12 bit SAR type A/D Converter
- 5. System Clock: 24.576MHz, 12MHz, 3.6864MHz
- 6. Sampling Rate: 48kHz, 44.1kHz, 32kHz, 24kHz, 22.05kHz, 16kHz, 11.025kHz, 8kHz
- 7. Power Management
- 8. Audio, Touch Screen & Control I/F: AC-Link I/F
- 9. Ta = −30 ~ 85°C
- 10. Power Supply: 2.7V ~ 3.6V (typ. 3.3V)
- 11. Package: 57pin BGA (5mm x 5mm)

Block Diagram (Audio Block)

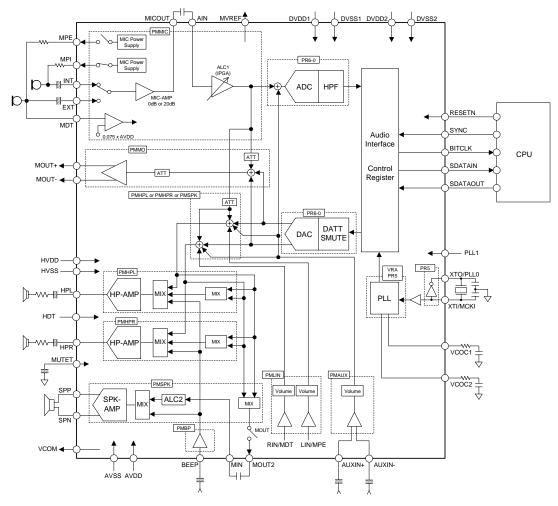
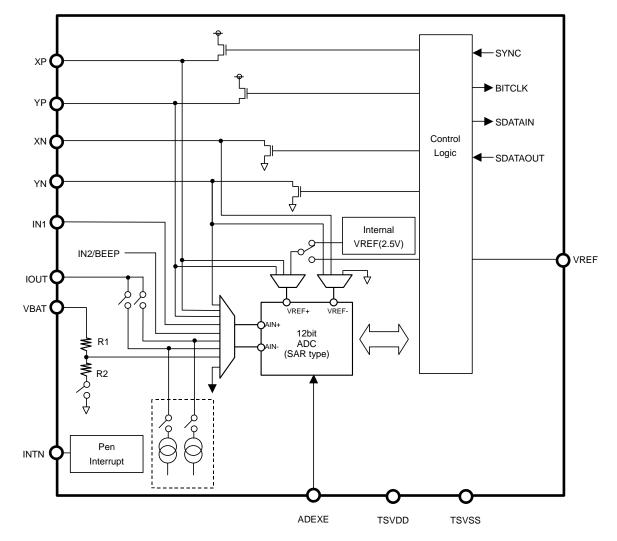


Figure 1. Block Diagram (Audio Block)



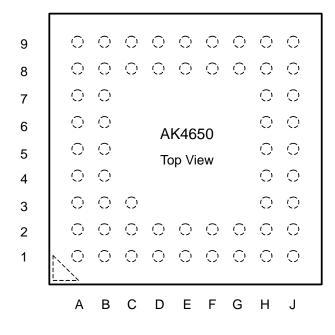
■ Block Diagram (Touch Screen Controller Block)

Figure 2. Block Diagram (TSC Blcok)

■ Ordering Guide

AK4650VG	$-30 \sim +85^{\circ}C$	57pin BGA (0.5mm pitch)
AKD4650	Evaluation board for AK4	4650

Pin Layout



9	NC	BEEP/IN2	AVDD	VCOM	AUXIN+	MPI	EXT/MIC+	MPE/LIN	NC
8	VCOC1	VCOC2	AVSS	MVREF	AUXIN-	INT/MIC-	MDT/RIN	AIN	MICOUT
7	ХР	TSVDD						MOUT-	MOUT+
6	ΥP	XN						HPL	HPR
5	YN	TSVSS			Top View			HVSS	HVDD
4	IN1	VBAT						SPP	SPN
3	VREF	IOUT	NC					MUTET	HDT
2	ADEXE	INTN	XTO/PLL0	SDATA OUT	DVSS2	SDATAIN	SYNC	MOUT2	MIN
1	NC	DVDD1	хті/мскі	DVSS1	BITCLK	DVDD2	RESETN	PLL1	TEST
	Α	В	С	D	E	F	G	н	J

■ Comparison with AK4640

Item	AK4640	AK4650
Mic Input	Single-ended	Single-ended / Differential
Mic Detection	Not Available	Available
Mic Power	Vout = 0.75 x AVDD	Vout = 2.2V
Bass Boost MID	+5.94dB@20Hz, fs=44.1kHz	+10.78dB@20Hz, fs=44.1kHz
Mono Output Volume	+6/-17dB	$+6 \sim -15$ dB, 3dB step
HP-Amp Output Power	15mW	40mW
Headphone Jack Detection	Not Available	Available
Stereo Beep Input	Available	Not Available
Stereo Line Input	Not Available	Availble
AUXIN Volume	+24 ~ -21dB, 3dB step	+12 ~ -34.5dB, 1.5dB step, Mute
Audio I/F	I^2S etc.	AC-Link
PLL Input Clock	12.288MHz, 11.2896MHz, 12MHz	24.576MHz, 12MHz, 3.6864MHz
Control I/F	4-wire/I ² C	AC-Link
Package	52QFN (7.2mm x 7.2mm) and 57BGA (5mm x 5mm)	57BGA (5mm x 5mm)

■ Comparison with AK4181

Item	AK4181	AK4650
Temperature Measurement	On-chip Sensor	Current Output for Off-chip Sensor
SYNC Signal Input for ADC	Not Available	Available
Control I/F	4-wire	AC-Link

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			PIN/FUNCTION				
		-					
No.	Pin Name	I/O	Function				
A1	NC	_	No Connect Pin				
			No internal bonding. This pin should be connected to the ground.				
B1	DVDD1	-	Digital Power Supply 1 Pin				
C1	XTI	Ι	X'tal Input Pin				
01	MCKI	Ι	External Master Clock Input Pin				
	ХТО	0	X'tal Output Pin (PLL1 pin = "L")				
C2	PLL0	Ι	PLL Input Master Clock Frequency Select 0 Pin (PLL1 pin = "H")				
	FLL0	1	"L": 3.6864MHz, "H": 12MHz				
D1	DVSS1	-	Digital Ground 1 Pin				
D2	SDATAOUT	Ι	Serial 256-bit AC'97 data stream from digital controller				
E1	BITCLK	0	12.288MHz(256fs) serial data clock				
E2	DVSS2	-	Digital Ground 2 Pin				
F2	SDATAIN	0	Serial 256-bit AC'97 data stream to digital controller				
F1	DVDD2	-	Digital Power Supply 2 Pin				
G2	SYNC	Ι	AC'97 Sync Clock, 48kHz(1fs) fixed rate sampling rate				
G1	RESETN	Ι	AC'97 Master Hardware Reset				
			PLL Input Master Clock Frequency Select 1 Pin				
771	DI I I	Ŧ	"L": 24.576 MHz (PLL0 pin = "L")				
H1	PLL1	Ι	"H": 3.6864MHz (PLL0 pin = "L") or 12MHz (PLL0 pin = "H")				
			When PLL1 pin = "H", \hat{X} tal oscillation circuit is not available.				
T 1	TEST		Test Pin				
J1	TEST	-	This pin should be connected to the ground.				
J2	MIN	Ι	ALC Input Pin				
H2	MOUT2	0	Analog Mixing Output Pin				
112	MUTET	0	Mute Time Constant Control Pin				
H3	MUTET	0	Connected to HVSS pin with a capacitor for mute time constant.				
J3	HDT	Ι	Headphone Detect Pin (Internal pull up by $100k\Omega$)				
H4	SPP	0	Speaker Amp Positive Output Pin				
J4	SPN	0	Speaker Amp Negative Output Pin				
H5	HVSS	-	Headphone & Speaker Amp Ground Pin				
J5	HVDD	-	Headphone & Speaker Amp Power Supply Pin				
J6	HPR	0	Rch Headphone Amp Output Pin				
H6	HPL	0	Lch Headphone Amp Output Pin				
H7	MOUT-	0	Mono Line Negative Output Pin				
J7	MOUT+	0	Mono Line Positive Output Pin				
H8	AIN	Ι	Analog Input Pin				
10			No Connect Pin				
J9	NC	-	No internal bonding. This pin should be connected to the ground.				
	•	•					

PIN/FUNCTION

No.	Pin Name	I/O	Function				
J8	MICOUT	0	Microphone Analog Output Pin				
G8	MDT	Ι	Microphone Detect Pin (Internal pull down by $500k\Omega$) (RNMD bit = "0")				
00	RIN	Ι	ch Line Input Pin (RNMD bit = "1")				
H9	MPE	O MIC Power Supply Pin for External Microphone (LNMP bit = "0")					
119	LIN	Ι	Lch Line Input Pin (LNMP bit = "1")				
G9	EXT	Ι	External Microphone Input Pin (Single-ended Input: MDIF bit = "0")				
09	MIC+	Ι	Microphone Positive Input Pin (Differential Input: MDIF bit = "1")				
F8	INT	Ι	Internal Microphone Input Pin (Single-ended Input: MDIF bit = "0")				
1.0	MIC-	Ι	Microphone Negative Input Pin (Differential Input: MDIF bit = "1")				
F9	MPI	0	MIC Power Supply Pin for Internal Microphone				
E8	AUXIN-	Ι	Mono AUX Negative Input Pin				
E9	AUXIN+	Ι	Mono AUX Positive Input Pin				
D8	MVREF	0	MIC Power Supply Reference Voltage Output Pin				
D9	VCOM	COM	Common Voltage Output Pin				
D9	VCOM	0	Bias voltage of ADC inputs and DAC outputs.				
C8	AVSS	-	Analog Ground Pin				
C9	AVDD	-	Analog Power Supply Pin				
B9	BEEP	I	Mono Beep Signal Input Pin (IN/BP bit = "0")				
	IN2	Ι	Auxiliary 2 Analog Input (IN/BP bit = "1")				
A9	NC	_	No Connect Pin				
			No internal bonding. This pin should be connected to the ground.				
			Output 1 Pin for Loop Filter of PLL Circuit				
A8	VCOC1	0	This pin should be connected to DVSS with a resistor ($10k\Omega$) and a capacitor				
			(4.7nF) in series.				
			Output 1 Pin for Loop Filter of PLL Circuit				
B8	VCOC2	0	This pin should be connected to DVSS with a resistor $(10k\Omega)$ and a capacitor				
			(4.7nF) in series.				

Note: All input pins except analog input pins.

No.	Pin Name	I/O	Function
B7	TSVDD	-	Touch Screen Controller Power Supply Pin
A7	ХР	I/O	 Touch Screen X+ plate Voltage Supply Pin X axis Measurement: Supplies the voltage. Y axis Measurement: This pin is used as the input for the A/D converter. Pen Pressure Measurement: This pin is the input to the A/D converter at Z₁ measurement. Temperature/VBAT/IN1/IN2 Measurement: OPEN state. Power-down State: is pulled up by an internal resistor (typ.50k ohm).
A6	ҮР	I/O	 Touch Screen Y+ plate Voltage Supply Pin Y axis Measurement: Supplies the voltage. X axis Measurement: This pin is used as the input for the A/D converter. Pen Pressure Measurement: Supplies the voltage. Temperature/VBAT/IN1/IN2 Measurement: OPEN state. Power-down State: OPEN state.
B6	XN	I/O	 Touch Screen X- plate Voltage Supply Pin X axis Measurement: Supplies the voltage. Y axis Measurement: OPEN state. Pen Pressure Measurement: Supplies the voltage. Temperature/VBAT/IN1/IN2 Measurement: OPEN state. Power-down State: OPEN state.
A5	YN	I/O	 Touch Screen Y- plate Voltage Supply Pin Y axis Measurement: Supplies the voltage. X axis Measurement: OPEN state. Pen Pressure Measurement: This pin is the input to the A/D converter at Z₂ measurement. Temperature/VBAT/IN1/IN2 Measurement: OPEN state. Power-down State: connected to GND.
B5	TSVSS	-	Touch Screen Controller Ground Pin
B4	VBAT	Ι	Analog Input Pin for Battery Monitor
A4	VREF	I/O	Voltage Reference Input/Output Outputs 2.5V at PMVREF bit = "1".
B3	IN1	Ι	Auxiliary 1 Analog Input Pin
A3	IOUT	I/O	Auxiliary 3 Analog Input Pin / Current Output Pin for Temperature Sensor
B2	INTN	0	Pen Interrupt Output Pin This pin should be pulled up via a $100k\Omega$ resistor.
A2	ADEXE	Ι	SYNC Signal Input Pin for ADC of Touch Screen Controller
C3	NC	-	No Connect Pin No internal bonding. This pin should be connected to the ground.

Handling of Unused Pin

The unused I/O pins should be processed appropriately as below.

Classification	Pin Name	Setting
Analog	MIN, MOUT2, MUTET, SPP, SPN, HPR, HPL, MOUT–, MOUT+, AIN, MICOUT, MDT/RIN, MPE/LIN, EXT/MIC+, INT/MIC–, MPI, AUXIN–, AUXIN+, BEEP/IN2, XP, YP, XN, YN, VBAT, IN1, IOUT, INTN	These pins should be open.
	HDT	This pin should be connected to AVSS.
Digital	ХТО	This pin should be open.

	ABSOLUTE MAXIMUM RATINGS							
(AVSS, DVSS (D	(AVSS, DVSS (DVSS1 & DVSS2), HVSS, TSVSS=0V; Note 1)							
Parameter	Parameter		min	max	Units			
Power Supplies:	Analog	AVDD	-0.3	6.0	V			
	Digital (DVDD1 & DVDD2)	DVDD	-0.3	6.0	V			
	Headphone-Amp / Speaker-Amp	HVDD	-0.3	6.0	V			
	Touch Screen Controller	TSVDD	-0.3	6.0	V			
	AVSS – DVSS (Note 2)	$\Delta GND1$	-	0.3	V			
	AVSS – HVSS (Note 2)	$\Delta GND2$	-	0.3	V			
	AVSS – TSVSS (Note 2)	∆GND3	-	0.3	V			
Input Current, An	y Pin Except Supplies	IIN	-	± 10	mA			
Analog Input Volt	tage	VINA	-0.3	AVDD+0.3	V			
Digital Input Volt	age	VIND	-0.3	DVDD+0.3	V			
Touch Screen Controller Input Voltage		VINTS	-0.3	TSVDD+0.3	V			
Touch Screen Drive Current		IOUTDRV	-	50	mA			
Ambient Temperature (powered applied)		Та	-30	85	°C			
Storage Temperat	ure	Tstg	-65	150	°C			

Note 1. All voltages with respect to ground.

Note 2. AVSS, DVSS (DVSS1 & DVSS2), HVSS and TSVSS must be connected to the same analog ground plane.

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 (D	(AVSS, DVSS (DVSS1 & DVSS2), HVSS, TSVSS=0V; Note 1)							
Parameter	Parameter Symbol min typ max Units					Units		
Power Supplies:	Analog	AVDD	2.7	3.3	3.6	V		
(Note 3)	Digital (DVDD1 & DVDD2)	DVDD	2.7	3.3	AVDD	V		
	HP / SPK-Amp	HVDD	2.7	3.3	3.6	V		
	Touch Screen Controller	TSVDD	2.7	3.3	3.6	V		

Note 1. All voltages with respect to ground.

Note 3. The power up sequence between AVDD, DVDD, HVDD and TSVDD is not critical. When the voltage difference among AVDD, DVDD, HVDD and TSVDD is larger than 0.3V, the power supply current at power down mode increases (see Note 22). When the power supplies are partially powered OFF, the AK4650 must be reset by bringing PDN pin "L" after these power supplies are powered ON again. DVDD1 and DVDD2 should be same voltage.

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

ANALOG CHARACTERISTICS (A	Audio Block)
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(Ta=25°C; AVDD, DVDD, HVDD, TSVDD=3.3V; AVSS=DVSS=HVSS=TSVSS=0V; fs=48kHz; Signal Frequency=1kHz; 16bit Data; Measurement frequency=20Hz ~ 20kHz; unless otherwise specified)

Parameter		min	typ	max	Units
MIC Amplifier: INT, EXT pins,	MDIF bit = "0" (Sir	ngle-ended input)			
Input Resistance		TBD	30	TBD	kΩ
Gain	MGAIN bit = "0"	-	0	-	dB
	MGAIN bit = "1"	-	+20	-	dB
MIC Amplifier: MIC+, MIC- pi	ns, MDIF bit = "1" (Full-differential in	nput), MGAIN bit =	= "1" (+20dB)	
Maximum Input Voltage (Note 4))	-	0.099	-	Vpp
MIC Power Supply: MPI, MPE	pins				
Output Voltage		TBD	2.2	TBD	V
Load Resistance		2	-	-	kΩ
Load Capacitance		-	-	30	pF
Mic Detection: MDT pin					
Comparator Voltage Level		TBD	0.18	TBD	V
Internal pull down Resistance		250	500	750	kΩ
Input PGA Characteristics: All	N pin				
Input Resistance (Note 5)		TBD	10	TBD	kΩ
Step Size		TBD	0.5	TBD	dB
Gain Control Range (ALC1 bit =	"0")				
max: IPGA	6-0 bits = "3FH"	-	+27.5	-	dB
min: IPGA	6-0 bits = "00H"	-	-8	-	dB
ADC Analog Input Characteris	tics: MIC Gain=+20	dB, IPGA=0dB, A	LC1=OFF, MIC -	\rightarrow IPGA \rightarrow ADC	
Resolution		-	-	16	Bits
Input Voltage (Note 6)		TBD	0.198	TBD	Vpp
S/(N+D) (-1dBFS)		TBD	75	-	dB
D-Range (-60dBFS, A-weigh	nted)	TBD	83	-	dB
S/N (A-weighted)		TBD	83	-	dB
DAC Characteristics:					
Resolution		-	-	16	Bits
Mono Line Output Characteris	tics: $R_L = 20 k\Omega$, DAC	$C \rightarrow MOUT + /MO$	UT– pins, MOGN2	2-0 bits = $+6dB$	
Output Voltage (Note 7)		TBD	3.96	TBD	Vpp
S/(N+D) (-3dBFS)		TBD	85	-	dB
S/N (A-weighted)		TBD	95	-	dB
Load Resistance		20	-	-	kΩ
Load Capacitance		-	-	30	pF
Step Size		TBD	3	TBD	dB
Gain Control Range					
	N2-0 bits = "111"	-	+6	-	dB
min: MOG	N2-0 bits = "000"	-	-15	-	dB

Note 4. Maximum input voltage of MIC+ and MIC- pins are proportional to AVDD voltage. Vin = $0.03 \times AVDD(typ)$. Note 5. When IPGA Gain is changed, this typical value changes between $8k\Omega$ and $11k\Omega$.

Note 6. Input voltage is proportional to AVDD voltage. Vin = $0.06 \times \text{AVDD}$.

Note 7. Output voltage is proportional to AVDD voltage. Vout = 1.2 x AVDD(typ)@MOGN2-0 bits = "111" at full-differential output. Vout = 0.6 x AVDD(typ)@MOGN2-0 bits = "111" at single-ended Output.

Parameter	min	typ	max	Units	
Headphone-Amp Characteris	stics: R _L =16Ω, DAC –	→ HPL/HPR pins, l	DATT=0dB		
Output Voltage (Note 8)	0dBFS Input	-	0.82	-	Vrms
	-6dBFS Input	TBD	0.41	TBD	Vrms
S/(N+D)	0dBFS Input	-	50	-	dB
	-6dBFS Input	TBD	60	-	dB
S/N (A-weighted)		TBD	90	-	dB
Interchannel Isolation		TBD	85	-	dB
Interchannel Gain Mismatch		-	0.1	TBD	dB
Load Resistance		16	-	-	Ω
Load Capacitance (Note 9)		-	-	300	pF
Headphone Detection: HDT p	bin				
Comparator Voltage Level (No	ote 10)	0.99	-	2.31	V
Internal pull down Resistance		TBD	100	TBD	kΩ
Speaker-Amp Characteristic	s: $R_L = 8\Omega$, BTL, DAC	\rightarrow MOUT2 pin \rightarrow	MIN pin \rightarrow SPP/S	SPN pins, ALC2=	=OFF
Output Voltage (Note 11)	-2.5dBFS Input	-	1.55	-	Vrms
	-7.5dBFS Input	TBD	0.94	TBD	Vrms
S/(N+D)	-2.5dBFS Input	-	20	-	dB
	-7.5dBFS Input	TBD	55	-	dB
S/N (A-weighted)		TBD	90	-	dB
Load Resistance		8	-	-	Ω
Load Capacitance		-	-	30	pF
Mono Output: DAC \rightarrow MIX \cdot	\rightarrow MOUT2 pin				
Output Voltage (Note 12)		-	1.98	-	Vpp
Load Resistance (Note 13)		30	-	-	kΩ
Load Capacitance (Note 9, Not	te 13)	-	-	20	pF
Mono Input: MIN pin				•	
Maximum Input Voltage (Note	: 14)	-	1.98	-	Vpp
Input Resistance (Note 15)		TBD	24	TBD	kΩ
BEEP Input: BEEP pin, Exter	mal input resistance = 2	20kΩ		1	
Maximum Input Voltage (Note	<u> </u>	-	1.98	-	Vpp
Feedback Resistance		TBD	20	TBD	kΩ
Output Voltage (0.8Vpp input)					•
$BEEP pin \rightarrow HPL/HPR$		TBD	0.09	TBD	Vpp
BEEP pin \rightarrow SPP/SPN 1		TBD	3.01	TBD	Vpp

Note 8. Output voltage is proportional to AVDD voltage. Vout = 0.12 x AVDD Vrms(typ)@-6dBFS.

Note 9. When the output pin drives a capacitive load, a resistor should be added in series between the output pin and capacitive load.

Note 10. Comparator Voltage Level is proportional to HVDD voltage. Vth = 0.3 x HVDD(min), 0.7 x HVDD(max).

Note 11. Output voltage is proportional to HVDD voltage. Vout = 0.28 x AVDD Vrms(typ)@-6dBFS at Full-differential output.

Note 12. Output Voltage is proportional to AVDD voltage. Vout = 0.6 x AVDD(typ).

- Note 13. These values do not include the input resistance or capacitance of the MIN pin.
- Note 14. Maximum Input Voltage is proportional to AVDD voltage. $Vin = 0.6 \times AVDD(typ)$.
- Note 15. When ALC2 Gain is changed, this typical value changes between $22k\Omega$ and $26k\Omega$.
- Note 16. Maximum Input Voltage depends on AVDD voltage, internal feedback resistance (Rf) and external input resistance (Ri). Vin = 0.6 x AVDD x Ri / Rf (typ).

Parameter		min	typ	max	Units
Line Input: LIN, RIN p	pins				
Maximum Input Voltage	e (Note 17)	-	1.98	-	Vpp
Input Resistance		TBD	40	TBD	kΩ
Step Size		TBD	1.5	TBD	dB
Gain Control Range					
	max: GL4-0 bits = "00H"	-	+12	-	dB
	min: GL4-0 bits = "1FH"	-	-34.5	-	dB
AUX Input: AUXIN+,	AUXIN– pins				
Maximum Input Voltage	e (Note 18)	-	1.98	-	Vpp
Input Resistance		TBD	40	TBD	kΩ
Step Size		TBD	1.5	TBD	dB
Gain Control Range					
	max: GN4-0 bits = "00H"	-	+12	-	dB
	min: GN4-0 bits = "1FH"	-	-34.5	-	dB
Power Supplies:					
Power Up (RESETN pin	n = "H")				
All Circuit Power	-up:				
AVDD+DVI		-	14	TBD	mA
HVDD: HP-A	HVDD: HP-AMP Normal Operation		2.5	TBD	mA
	No Output (Note 20)		2.5	IDD	IIIA
	HVDD: SPK-AMP Normal Operation		7	TBD	mA
	No Output (Note 21)		,	122	1117 1
TSVDD		-	0.2	TBD	mA
Power Down (RESETN	pin = "L") (Note 22)				
AVDD+DVDD+I	HVDD+TSVDD	-	-	TBD	μΑ

Note 17. Maximum Input Voltage is proportional to AVDD voltage. Vin = 0.6 x AVDD(typ).

Note 18. Maximum Input Voltage is proportional to AVDD voltage. Vin = $(AUXIN+) - (AUXIN-) = 0.6 \times AVDD(typ)$.

Note 19. PR0-6 bits = all "0", PMMIC=PMMO=PMSPK=PMHPL=PMHPR=PMBPM=PMAUX= "1".

AVDD=10mA (typ.), DVDD=4mA (typ.).

Note 20. PR0-6 bits = all "0", PMMIC=PMMO=PMHPL=PMHPR=PMBPM=PMAUX= "1", PMSPK= "0".

Note 21. PR0-6 bits = all "0", PMMIC=PMMO=PMSPK=PMBPM=PMAUX= "1", PMHPL=PMHPR= "0".

Note 22. All digital input pins are fixed to DVDD or DVSS. When the voltage difference among AVDD, DVDD, HVDD and TSVDD is larger than 0.3V, the power supply current at power down mode increases.

Parameter	min	typ	max	Units
ADC for Touch Screen				
Resolution	-	12	-	Bits
No Missing Codes	11	12	-	Bits
Integral Linearity Error	-	-	±5	LSB
DNL	-	±2	-	LSB
Analog Input Voltage Range	0	-	VREF	V
Offset Error	-	-	±6	LSB
Gain Error	-	-	±4	LSB
Touch Screen Driver				
X+, Y+, RL=300Ω	-	5	-	Ω
X-, Y-, RL=300Ω	-	5	-	Ω
PSRR (10KHz 100mVpp)	-	70	-	dB
Reference Output				
Internal Reference	2.45	2.50	2.55	V
Drift	-	30	-	ppm/°C
Load Capacitance	0.1	-	-	μF
Reference Input				
Input Voltage Range	-	-	TSVDD	V
Input Impedance	TBD	196	-	kΩ
Battery Monitor				
Input Voltage Range	-	-	5.0	V
Input Impedance (Battery Measure Mode)	5	10	-	kΩ
Accuracy (Note 23)	-	-	±2	%
Temperature Measurement				
Temperature Range	-30	-	85	°C
Resolution (Note 24)	-	1.6	-	°C
Accuracy (Note 25)	-	±3	-	°C

ANALOG CHARACTERISTICS (TSC Block)

(Ta=25°C; AVDD, DVDD, HVDD, TSVDD=3.3V; AVSS=DVSS=HVSS=TSVSS=0V; External Vref=2.5V, unless

Note 23. Accuracy is the difference between the output code when 5 volts is input to the VBAT pin and the "ideal" code at 1.25 volts.

Note 24. "Ideal" value derived from theory

Note 25. Accuracy is defined as the difference between the voltage measured by two current sources, and the ideal voltage derived from theory at specific temperatures.

FILTER CHARACTERISTICS								
(Ta=25°C; AVDD, HV	DD, DVI	DD, TSVDD	$=2.7 \sim 3.6V$; fs	=48kHz; DEM	I=OFF)			
Parameter			Symbol	min	typ	max	Units	
ADC Digital Filter (De	ecimation	n LPF):						
Passband (Note 26)		±0.1dB	PB	0	-	18.9	kHz	
		-1.0dB		-	21.8	-	kHz	
		-3.0dB		-	23.0	-	kHz	
Stopband			SB	29.4	-	-	kHz	
Passband Ripple			PR	-	-	±0.1	dB	
Stopband Attenuation			SA	65	-	-	dB	
Group Delay (Note 27)			GD	-	17.0	-	1/fs	
Group Delay Distortion			ΔGD	-	0	-	μs	
ADC Digital Filter (H	PF):							
Frequency Response (N	lote 26)	-3.0dB	FR	-	1.0	-	Hz	
		-0.1dB		-	6.5	-	Hz	
DAC Digital Filter:								
Passband (Note 27)		±0.1dB	PB	0	-	21.3	kHz	
		-6.0dB		-	24.0	-	kHz	
Stopband			SB	25.2	-	-	kHz	
Passband Ripple			PR	-	-	±0.01	dB	
Stopband Attenuation			SA	59	-	-	dB	
Group Delay (Note 27)			GD	-	16.8	-	1/fs	
DAC Digital Filter + S	CF:							
Frequency Response: 0	~ 20.0kH	łz	FR	-	±1.0	-	dB	
BOOST Filter: (Note 2	28)							
Frequency Response	MIN	20Hz	FR	-	5.80	-	dB	
		100Hz		-	3.17	-	dB	
		1kHz		-	0.03	-	dB	
	MID	20Hz	FR	-	10.85	-	dB	
		100Hz		-	7.23	-	dB	
		1kHz		-	0.18	-	dB	
	MAX	20Hz	FR	-	16.14	-	dB	
		100Hz		-	11.05	-	dB	
		1kHz		-	0.47	-	dB	

Note 26. The passband and stopband frequencies scale with fs (system sampling rate).

For example, ADC is PB= $0.454 \times \text{fs}$ (@-1.0dB). The reference frequency of these responses is 1kHz.

Note 27. The calculated delay time caused by digital filtering. This time is from the input of analog signal to setting of the 16-bit data of both channels from the input register to the output register of the ADC. This time includes the group delay of the HPF. For the DAC, this time is from setting the 16-bit data of both channels from the input register to the output of analog signal.

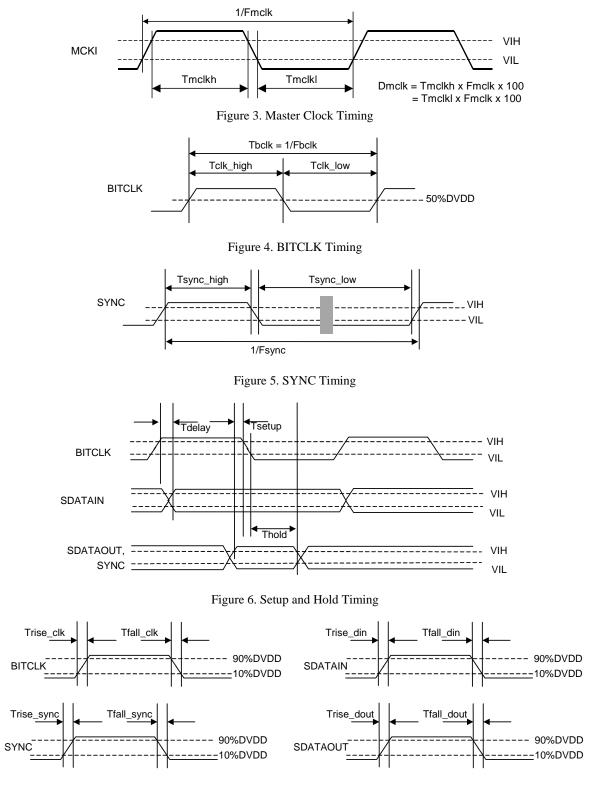
Note 28. These frequency responses scale with fs. If a high-level and low frequency signal is input, the analog output clips to the full-scale.

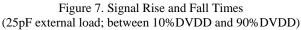
	DC CHARACTERISTICS							
(Ta=25°C; AVDD, HVDD, DVDD, TSVDD=2.7 ~ 3.6V)								
Parameter		Symbol	min	typ	max	Units		
High-Level Input Voltage		VIH	80%DVDD	-	-	V		
Low-Level Input Voltage		VIL	-	-	20%DVDD	V		
Input Voltage at AC Coupling (Note 29)		VAC	50%DVDD	-	-	V		
High-Level Output Voltage (Iout=-400µA)		VOH	DVDD-0.4	-	-	V		
Low-Level Output Voltage	(Iout=400µA)	VOL	-	-	0.4	V		
Input Leakage Current		Iin	-	-	±10	μA		
Tri-state Leakage Current (TSC	Tri-state Leakage Current (TSC block)							
All pins except for XP, YP, XN, YN pins		IOLK	-10	-	+10	μΑ		
XP, YP, XN, YN pins		IOLK	-50	-	+50	μΑ		
INTN "L" level output voltage (100kΩ Pull-Up)	VOLP	-	-	0.8	V		

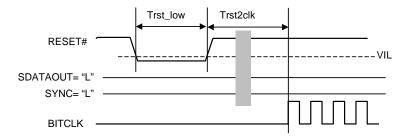
Note 29. The external clock is input to MCKI pin via AC coupled capacitor.

SWITCHING CHARACTERISTICS							
(Ta=25°C; AVDD, HVDD, DVDD, TSVDD=2.7	$\sim 3.6V; C_L=25p$	oF)					
Parameter	Symbol	min	typ	max	Units		
Master Clock Timing							
Frequency							
PLL1 pin = "L", PLL0 pin = "L"	Fmclk	-	24.576	-	MHz		
PLL1 $pin = "H"$, PLL0 $pin = "L"$	Fmclk	-	3.6864	-	MHz		
PLL1 pin = "H", PLL0 pin= "H"	Fmclk	-	12	-	MHz		
Duty Cycle	Dmclk	40	-	60	%		
AC link Interface Timing							
BITCLK frequency	Fbclk	-	12.288	-	MHz		
BITCLK clock Period (Tbclk=1/Fbclk)	Tbclk	-	81.38	-	ns		
BIT_BLK low pulse width	Tclk_low	36	40.7	45	ns		
BIT_BLK low pulse width	Tclk_high	36	40.7	45	ns		
BITCLK rise time	Trise_clk	-	-	6	ns		
BITCLK fall time	Tfall_clk	-	-	6	ns		
SYNC frequency	Fsync	-	48	-	kHz		
SYNC low pulse width	Tsync_low	-	19.5	-	μs		
			(240 cycle)		(Tbclk)		
SYNC high pulse width	Tsync_high	-	1.3	-	μs		
			(16 cycle)		(Tbclk)		
SYNC rise time	Trise_sync	-	-	6	ns		
SYNC fall time	Tfall_sync	-	-	6	ns		
Setup time (SYNC, SDATAOUT)	Tsetup	10	-	-	ns		
Hold time (SYNC, SDATAOUT)	Thold	25	-	-	ns		
SDATAIN delay time from BITCLK rising	Tdelay	-	-	15	ns		
edge	-						
SDATAIN rise time	Trise_din	-	-	6	ns		
SDATAIN fall time	Tfall_din	-	-	6	ns		
SDATAOUT rise time	Trise_dout	-	-	6	ns		
SDATAOUT fall time	Tfall_dout	-	-	6	ns		
Cold Reset (SDATAOUT = "L", SYNC = "L")							
RESETN active low pulse width	Trst_low	1.0	-	-	μs		
RESETN inactive to BITCLK delay							
PLL1 pin = "L" (External clock)	Trst2clk	-	42	-	μs		
PLL1 pin = "L" (X'tal oscillator)	Trst2clk	-	0.5	-	ms		
PLL1 pin = "H", PLL0 pin = "L"	Trst2clk	-	9.5	-	ms		
PLL1 pin = "H", PLL0 pin = "H"	Trst2clk	-	3.2	-	ms		
Warm Reset Timing							
SYNC active high pulse width	Tsync_high	1.0	1.3	-	μs		
			(16 cycle)		(Tbclk)		
SYNC inactive to BITCLK delay			-				
PLL1 pin = "L" (External clock)	Trst2clk	-	42	-	μs		
PLL1 $pin = "L"$ (X'tal oscillator)	Tsync2clk	-	0.5	-	ms		
PLL1 pin = "H", PLL0 pin = "L"	Tsync2clk	-	9.5	-	ms		
PLL1 pin = "H", PLL0 pin = "H"	Tsync2clk	-	3.2	-	ms		
AC-link Low Power Mode Timing							
End of Slot 2 to BITCLK, SDATAIN Low	Ts2_pdwn	-	-	1.0	μs		
Activate Test Mode Timing							
Setup to trailing edge of RESETN	Tsetup2rst	15.0	-	-	ns		
Hold from RESETN rising edge	Thold2rst	100	_	-	ns		
Rising edge of RESETN to Hi-Z	Toff	-	_	50	ns		
Falling edge of RESETN to "L"	Tlow	-		50	ns		

Timing Diagram









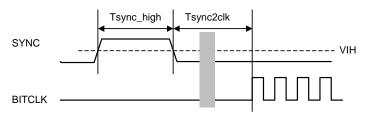
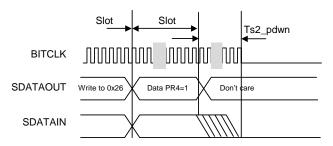
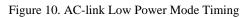


Figure 9. Warm Reset Timing





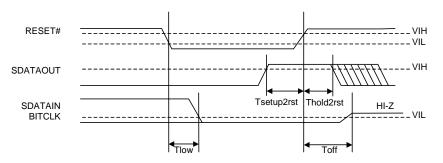


Figure 11. Activate Test Mode Timing

OPERATION OVERVIEW

■ Master Clock Source

The AK4650 requires a master clock (MCLK). This master clock is input to the AK4650 by the following three methods:

- (1) Connect a X'tal oscillator between XTI and XTO pins.
- (2) Input an external CMOS-level clock to the XTI pin.
- (3) Input an external clock whose amplitude is greater than 50% DVDD to the XTI pin with AC coupling.

When using a X'tal oscillator, there should be capacitors between XTI/XTO pins and DVSS (Figure 12).

Master Clock		Status	PR5 bit	MCKPD bit
X'tal Oscillator	(Figure 12)	Oscillator ON	0	0
(PLL1 pin = "L")	-	Oscillator OFF	1	1
External Clock Direct Input	(Figure 13)	Clock is input to MCKI pin.	0	0
(PLL1 pin = "L")		MCKI pin is fixed to "L".	1	0/1
		MCKI pin is fixed to "H".	1	0
		MCKI pin is Hi-Z	1	1
External Clock Direct Input	(Figure 14)	Clock is input to MCKI pin.	0	0
(PLL1 pin = "H")		MCKI pin is fixed to "L".	0	0/1
		MCKI pin is fixed to "H".	0	0
		MCKI pin is Hi-Z	0	1
AC Coupling Input	(Figure 15)	Clock is input to MCKI pin.	0	0
(PLL1 pin = "L")		Clock isn't input to MCKI pin.	1	1

Table 1. Master Clock Status by PR5 bit and MCKPD bit

(1) X'tal Oscillator (PLL1 pin = "L")

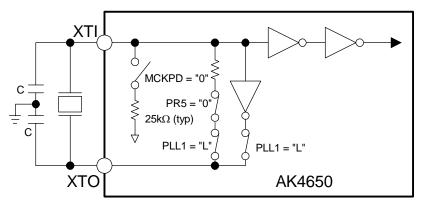


Figure 12. X'tal mode Note 30. The capacitor values depend on the X'tal oscillator used. (C : typ. $10 \sim 30 \text{pF}$)

(2) External Clock Direct Input

(2-1) PLL1 pin = "L"

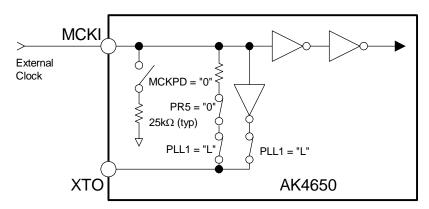


Figure 13. External Clock mode (PLL1 pin = "L", Input : CMOS Level) Note 31. This clock level must not exceed DVDD level.

(2-2) PLL1 pin = "H"

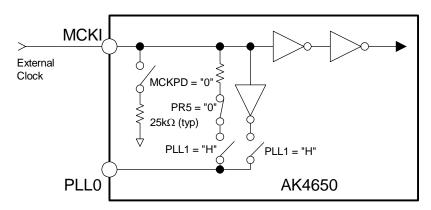


Figure 14. External Clock mode (PLL1 pin = "H", Input : CMOS Level) Note 32. This clock level must not exceed DVDD level.

(3) AC Coupling Input (PLL1 pin = "L")

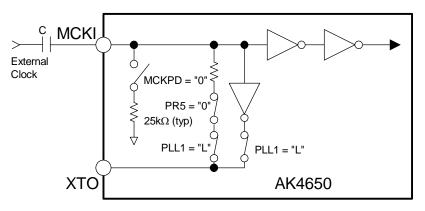


Figure 15. External Clock mode (Input : \geq 50%DVDD) Note 33. This clock level must not exceed DVDD level. (C : 0.1µF)

System Clock

A fully integrated analog phase locked loop (PLL) generates a clock that is selected by the PLL1-0 pins (Table 2).

When the external master clock is used, the PLL should be powered-up after the external master clock is input. It takes 0.5 ms(typ) for X'tal oscillator to be stable after PR5 bit = "0" which depends on the X'tal. The PLL needs 9ms lock time, whenever the sampling frequency changes or the PLL is powered-up (VRA bit = "0" \rightarrow "1").

When the clock input to MCKI pin stops during normal operation of AC-Link (PR4 = PR5 bits = "0"), the internal PLL continues to oscillate (a few MHz), and BITCLK output goes to "L" (see Table 3).

MCLK and SYNC must be present whenever the ADC or DAC is operating (PR0 = PR1 = PR3 = PR4 = PR5 bits = "0"). If these clocks are not provided, the AK4650 may draw excess current due to its use of internal dynamically refreshed logic. If the external clocks are not present, the ADC and DAC must be placed in the power-down mode by setting PR0-6 bits.

PLL1 pin	PLL0 pin	MCKI	
T	L	24.576MHz	
L	Н	Reserved	
п	L	3.6864MHz	
П	Н	12MHz	

	Power up	Power down	PLL Unlock
MCVI nin	Frequency set by PLL1-0	Refer to Table 1	Frequency set by PLL1-0 pins
MCKI pin	pins (Refer to Table 2)	Refer to Table 1	(Refer to Table 2)
BITCLK pin	12.288MHz Output	"L"	"L"
			Input
SYNC pin	Input	Fixed to "L" or "H" externally	or
			Fixed to "L" or "H" externally

Table 2. MCKI Input Frequency

Table 3. Clock Operation

■ Audio Sample Rate

Sample Rate for DAC and ADC is controlled by register 2CH and 32H, respectively. 16bit data in D15(MSB) to D0 show unsigned value from 0 to 65535, representing the exact sampling frequency in Hz. These sample rate setting is done at VRA bit = "1" of Extended Audio Status and Control Register(2Ah).

Sample Rate (kHz)	Data in D15 – D0
8.0	1F40H
11.025	2B11H
16.0	3E80H
22.05	5622H
24.0	5DC0H
32.0	7D00H
44.1	AC44H
48.0	BB80H

Table 4. Audio Sample Rate

The AK4650 supports these discrete frequencies. When any other codes is written in this register, the AK4650 operates at the sampling rate rounded to the closest one in Table 4 by decoding only D15-12 bits. If D15-12 = 5H, the AK4650 operates at 22.05kHz or 24kHz when D11 = "0" or "1", respectively (Table 5).

D15 – D12	D11	Sample Rate (kHz)
0H,1H	Х	8.0
2H	Х	11.025
3H	Х	16.0
4H	Х	22.05
5H	0	22.05
Л	1	24.0
6H	Х	24.0
7H,8H	Х	32.0
9H,AH	Х	44.1
BH-FH	х	48.0

 Table 5. Audio Sample Rate (x: Don't care)

At VRA bit = "0", 2CH and 32H are fixed to "BB80H" and cannot be written. When VRA bit is set to "0", 2CH and 32H register are set to "BB80H" automatically.

And the change of sample rate will be executed on the fly.

Power Management

Power management of each block is controlled via 26H and 60H register.

	PR0 = "1"	PR1 = "1"	PR2 = "1"	PR3 = "1"	PR4 = "1"	PR5 = "1"	PR6 = "1"
ADC	PD	PU	PU	PD	PD	PD	PU
DAC	PU	PD	PU	PD	PD	PD	PU
VCM	PU	PU	PU	PD	PU	PU	PU
XTL	PU	PU	PU	PU	PU	PD	PU
PLL	VRA	VRA	VRA	PD	VRA	PD	VRA
AC-Link	PU	PU	PU	PU(Note 34)	PD	PD	PU
HP	PMHPL/R	PMHPL/R	PD	PD	PMHPL/R	PMHPL/R	PD
SPK	PMSPK	PMSPK	PD	PD	PMSPK	PMSPK	PMSPK
MIC	PMMIC	PMMIC	PD	PD	PMMIC	PMMIC	PMMIC
Line In	PMLIN	PMLIN	PD	PD	PMLIN	PMLIN	PMLIN
AUXIN	PMAUX	PMAUX	PD	PD	PMAUX	PMAUX	PMAUX
Mono Out	PMMO	PMMO	PD	PD	PMMO	PMMO	PMMO
BEEP	PMBPM	PMBPM	PD	PD	PMBPM	PMBPM	PMBPM
TSC	PM12AD	PM12AD	PM12AD	PD	PM12AD	PM12AD	PM12AD

Table 6. Power Management

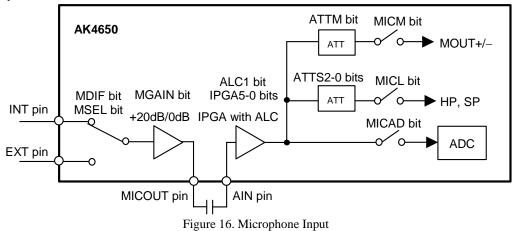
PD: Power Down PU: Power Up PM***: depends on PM*** bit. VRA VRA bit = "1": PLL Power Up VRA bit = "0": PLL Power Down

Note 34. When PLL1 pin = "H"(MCKI=3.6864MHz or 12MHz), AC-Link is powered-down by PR3 bit = "1".

■ Digital High Pass Filter

The ADC has a digital high pass filter for DC offset cancellation. The cut-off frequency of the HPF is 1.0Hz (@fs=48kHz) and scales with sampling rate (fs).

■ MIC Input



The AK4650 has the following functions for Mic Input.

- (1) 2 Inputs Selector. The switch configure is controlled by MDIF and MSEL bits (Table 9).
- (2) 1st MIC Amplifier with +20dB gain, The gain can be selected ON/OFF by MGAIN bit (Table 10).
- (3) IPGA with ALC. This volume is controlled by IPGA5-0 bits as Table 14.
- (4) Attenuator for stereo mixer. The volume is controlled by ATTS2-0 bits as Table 7.
- (5) Attenuator for mono mixer. The attenuator level is 4dB and the ON/OFF is controlled by ATTM bit (Table 8).

ATTS2-0	Attenuation	STEP	
0H	-6dB		
1H	-9dB		
2H	-12dB		Default
3H	-15dB	3dB	
4H	-18dB	Jub	
5H	-21dB		
6H	-24dB		
7H	-27dB		

Table 7. Attenuator Table (IPGA \rightarrow Stereo Mixer)

ATTM	Attenuation	
0	0dB	Default
1	-4dB	

Table 8. Attenuator Table (IPGA \rightarrow Mono Mixer)

MIC Input Selector

AK4650 has mic input selector in front of mic amp. MSEL bit selects internal or external mic (Figure 17). When MDIF bit = "1", INT and EXT pins become MIC– and MIC+ pins, respectively, and differential input is available (Figure 18).

MDIF bit	MSEL bit	Selector	
0	0	INT	Default
0	1	EXT	
1	Х	Differential	

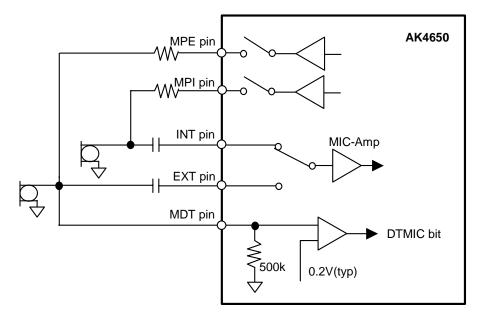


Table 9. MIC Input Selector (x: Don't care)

Figure 17. MIC Input (MDIF bit = "0": Single-ended Input)

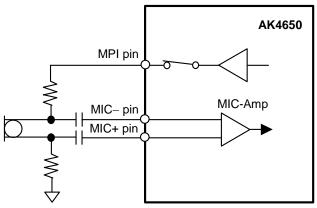


Figure 18. MIC Input (MDIF bit = "1": Differential Input)

Note 35. In case of differential input, MGAIN bit should be set to "1". Maximum input voltage of each input pin is | (MIC+) – (MIC-) | = 0.198Vpp(typ)@AVDD=3.3V.

■ MIC Gain Amplifier

The AK4650 has a Gain Amplifier for Microphone input. The gain is 0dB or +20dB, selected by the MGAIN bit. The typical input impedance is $30k\Omega$.

MGAIN bit	Input Gain				
0	0dB	Default			
1	+20dB				
Table 10. MIC Input Gain					

■ MIC Power

The MPI and MPE pins supply power for the Microphone. These output voltages are 2.2V (typ) and load resistance is $2k\Omega$ (min). MPWRI and MPWRE bits control output from MPI and MPE pins, respectively. When LNMP bit = "1", MPE pin becomes LIN pin.

PMMIC bit	MPWRI bit	MPI pin	
0	Х	Hi-Z	Default
1	0	Hi-Z	
1	1	Output	

Table 11. Internal Microphone Power Supply (x: Don't care)

PMMIC bit	MPWRE bit	MPE pin	
0	Х	Hi-Z	Default
1	0	Hi-Z	
1	1	Output	

Table 12. External Microphone Power Supply (x: Don't care)

■ MIC Detection Function

The AK4650 includes the detection function of microphone.

Example of the detection of external microphone.

- (1) MPWRE bit = "1".
- (2) MPE drives external microphone.
- (3) DTMIC bit is set by Table 13.

Input Level of MDT	DTMIC bit	External microphone
> 0.247V	1	Connect
< 0.165V	0	Disconnect

Table 13. Microphone detection result

When RNMD bit = "1", MDT pin becomes RIN pin.

Manual Mode

The AK4650 becomes a manual mode at ALC1 bit = "0". The mode is used in the case shown below.

- (1) After exiting reset state, set up the registers for the ALC1 operation (ZTM1-0, LMTH and etc)
- (2) When the registers for the ALC1 operation (Limiter period, Recovery period and etc) are changed. For example; When the change of the sampling frequency.
- (3) When IPGA is used as a manual volume.

When writing to the IPGA5-0 bits continually, the control register should be written by an interval more than zero crossing timeout.

MICMT	IPGA5-0	GAIN (dB)	STEP]
	3FH	+27.5		
	3EH	+27.0		
	:	:	0.5dB	
0	09H	+0.5		
	08H	+0.0		Default
	07H	-1.0		
	06H	-2.0	-	
	:	:	1.0dB	
	01H	-7.0		
	00H	-8.0		
1	Х	MUTE		

Table 14. IPGA Volume (x: Don't care)

■ MIC-ALC Operation

The ALC (Automatic Level Control) of MIC input is done by ALC1 block when ALC1 bit is "1".

(1) ALC1 Limiter Operation

When the ALC1 limiter is enabled, and IPGA output exceeds the ALC1 limiter detection level (LMTH bit: Table 15), the IPGA value is attenuated by the amount defined in the ALC1 limiter ATT step (LMAT1-0 bits: Table 16) automatically.

When the ZELMN bit = "1", the timeout period is set by the LTM1-0 bits (Table 17). The operation for attenuation is done continuously until the IPGA output signal level becomes LMTH or less. If the ALC1 bit does not change into "0" after completing the attenuation, the attenuation operation repeats while the IPGA output signal level equals or exceeds LMTH.

When the ZELMN bit = "0", the timeout period is set by the ZTM1-0 bits (Table 18). This enables the zero-crossing attenuation function so that the IPGA value is attenuated at the zero-detect points of the waveform.

LMTH	ALC1 Limiter Detection Level	ALC1 Recovery Waiting Counter Reset Level	
0	ADC Input ≥ −6.0dBFS	-6.0dBFS > ADC Input ≥ -8.0dBFS	Default
1	ADC Input ≥ -4.0 dBFS	-4.0dBFS > ADC Input ≥ -6.0dBFS	

Table 15. ALC1 Limiter Detection Level / Recovery Waiting Counter Reset Level

LMAT1	LMAT0	ATT STEP	
0	0	0.5dB	Default
0	1	1.0dB	
1	0	1.5dB	
1	1	2.0dB	

Table 16. ALC1 Limiter ATT Step Setting

Note: When IPGA gain is 0dB or less, ALC1 limiter ATT step is fixed to 1 regardless as LMAT1-0 bits.

LTM1	LTM0		ALC1 Limiter	Operation Perio	d]
LIMI	LINIO		8kHz	16kHz	44.1kHz	
0	0	0.5/fs	63µs	31µs	11µs	Default
0	1	1/fs	125µs	63µs	23µs	
1	0	2/fs	250µs	125µs	45µs	
1	1	4/fs	500µs	250µs	91µs	

Table 17. ALC1 Limiter Operation Period at zero crossing disable (ZELMN bit = "1")

ZTM1	ZTM0		Zero Crossing Timeout Period			
ZIMI	ZIMU		8kHz	16kHz	44.1kHz	
0	0	128/fs	16ms	8ms	2.9ms	Default
0	1	256/fs	32ms	16ms	5.8ms	
1	0	512/fs	64ms	32ms	11.6ms	
1	1	1024/fs	128ms	64ms	23.2ms	

Table 18. Zero Crossing Timeout Period

(2) ALC1 Recovery Operation

The ALC1 recovery refers to the amount of time that the AK4650 will allow a signal to exceed a predetermined limiting value prior to enabling the limiting function. The ALC1 recovery operation uses the WTM1-0 bits to define the wait period used after completing an ALC1 limiter operation (Table 19). If the input signal does not exceed the "ALC1 Recovery Waiting Counter Reset Level" (LMTH: Table 15), the ALC1 recovery operation starts. The IPGA value increases automatically by the recovery gain step (RGAIN bit: Table 20) with zero crossing operation (timeout is set by ZTM1-0: Table 18) during this operation up to the reference level (REF5-0 bit: Table 21). The ALC1 recovery operation is done at a period set by the WTM1-0 bits. Zero crossing is detected during WTM1-0 period, the ALC1 recovery operation waits WTM1-0 period and the next recovery operation starts.

During the ALC1 recovery operation, when input signal level exceeds the ALC1 limiter detection level (LMTH), the ALC1 recovery operation changes immediately into an ALC1 limiter operation.

In the case of

(Recovery waiting counter reset level) \leq (IPGA Output Level) < (Limiter detection level) during the ALC1 recovery operation, the wait timer for the ALC1 recovery operation is reset. Therefore, in the case of

(Recovery waiting counter reset level) > (IPGA Output Level),

the wait timer for the ALC1 recovery operation starts.

The ALC1 operation corresponds to the impulse noise. When the impulse noise is input, the ALC1 recovery operation becomes faster than a normal recovery operation.

WTM1	WTM0	ALC	C1 Recovery Op	eration Waiting	Period	
W 1 W11	W INIO		8kHz	16kHz	44.1kHz	
0	0	128/fs	16ms	8ms	2.9ms	Default
0	1	256/fs	32ms	16ms	5.8ms	
1	0	512/fs	64ms	32ms	11.6ms	
1	1	1024/fs	128ms	64ms	23.2ms	J

Table 19. ALC1 Recovery Operation Waiting Period

RGAIN	GAIN STEP	
0	0.5dB	Default
1	1.0dB	

Table 20. ALC1 Recovery Gain Step Setting

REF5-0	GAIN (dB)	STEP	1
3DH	+26.5		1
3CH	+26.0		
:	:		
2DH	+19.0	0.5dB	Default
:	:		
05H	+0.5		
04H	+0.0		
03H	-1.0		
02H	-2.0		
:	:	1.0dB	
01H	-7.0		
00H	-8.0		

Table 21. Setting Reference Value at ALC1 Recovery Operation

(3) Example of ALC1 Operation

Table 22 shows the examples of the ALC1 setting. In case of this examples, ALC1 operation starts from 0dB.

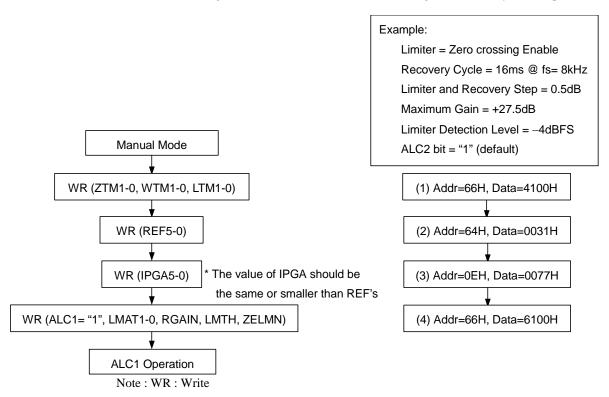
Register	Comment	fs=8kHz		fs=16kHz		fs=44.1kHz	
Name	Comment	Data	Operation	Data	Operation	Data	Operation
LMTH	Limiter detection Level	1	-4dBFS	1	-4dBFS	1	-4dBFS
LTM1-0	Limiter operation period at ZELMN bit = "1"	00	Don't use	00	Don't use	00	Don't use
ZELMN	Limiter zero crossing detection	0	Enable	0	Enable	0	Enable
ZTM1-0	Zero crossing timeout period	00	16ms	01	16ms	10	11.6ms
WTM1-0	Recovery waiting period *WTM1-0 bits should be the same data as ZTM1-0 bits	00	16ms	01	16ms	10	11.6ms
REF5-0	Maximum gain at recovery operation	00H	+27.5dB	00H	+27.5dB	00H	+27.5dB
IPGA5-0	Gain of IPGA at ALC1 operation start	37H	0dB	37H	0dB	37H	0dB
LMAT1-0	Limiter ATT Step	00	0.5dB	00	0.5dB	00	0.5dB
RGAIN	Recovery GAIN Step	0	0.5dB	0	0.5dB	0	0.5dB
ALC1	ALC1 Enable bit	1	Enable	1	Enable	1	Enable

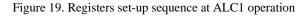
Table 22. Example of the ALC1 setting

The following registers should not be changed during the ALC1 operation. These bits should be changed, after the ALC1 operation is finished by ALC1 bit = "0" or PMMIC bit = "0".

• LTM1-0, LMTH, LMAT1-0, WTM1-0, ZTM1-0, RGAIN, REF5-0, ZELMN bits

IPGA gain at ALC1 operation start can be changed from the default value of IPGA5-0 bits while PMMIC bit is "1" and ALC1 bit is "0". When ALC1 bit is changed from "1" to "0", IPGA holds the last gain value set by ALC1 operation.





■ De-emphasis Filter

The AK4650 includes the digital de-emphasis filter (tc = $50/15\mu$ s) by IIR filter. Setting the DEM1-0 bits enables the de-emphasis filter (Table 23).

DEM1	DEM0	Mode	
0	0	44.1kHz	
0	1	OFF	Default
1	0	48kHz	
1	1	32kHz	
	T 11 00 D	1	

Table 23. De-emphasis Control

Bass Boost Function

The BST1-0 bits control the amount of low frequency boost applied to the DAC output signal (Table 24). If the BST1-0 bits are set to "10" (MID Level), AC-coupling capacitor can be sized down to 47μ F. If the boosted signal exceeds the full scale, the analog output clips to the full scale. Figure 20 shows the boost frequency response at –20dB signal input.

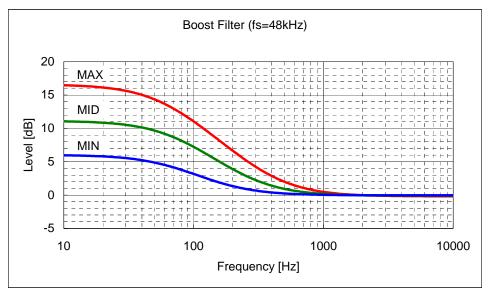


Figure 20. Bass Boost Frequency (fs=48kHz)

BST1	BST0	Mode	
0	0	OFF	Default
0	1	MIN	
1	0	MID	
1	1	MAX	

Table 24. Bass Boost Control

Digital Attenuator

The AK4650 has a channel-independent digital attenuator (128 levels, 0.5dB step, Mute). The attenuation level of each channel can be set by the ATTL/R6-0 bits (Table 25). When the DATTC bit = "1", the ATTL6-0 bits control both Lch and Rch attenuation levels. When the DATTC bit = "0", the ATTL6-0 bits control Lch level and ATTR6-0 bits control Rch level.

ATTL/R6-0	Attenuation	STEP			
00H	0dB		Default		
01H	-0.5dB				
02H	-1.0dB	0.5dB			
:	:	0.50D			
7DH	-62.5dB				
7EH	-63.0dB				
7FH	MUTE (-∞) -				
Table 25. DATT Attenuation Table					

The ATS bit sets the transition time between set values of ATTL/R6-0 bits as either 531/fs or 128/fs (Table 26). When ATS bit = "0", a soft transition between the set values occurs (531 levels). It takes 531/fs (11ms@fs=48kHz) from 00H(0dB) to 7FH(MUTE).

ATS	ATT		
AIS	0dB to MUTE	1 step	
0	531/fs	4/fs	Default
1	128/fs	29/fs	

Table 26. Transition time between set values of ATTL/R6-0 bits

Soft Mute

Soft mute operation is performed in the digital domain. When the SMUTE bit goes to "1", the output signal is attenuated by $-\infty$ ("0") during the cycle set by the TM1-0 bits (Table 27). When the SMUTE bit is returned to "0", the mute is cancelled and the output attenuation gradually changes to 0dB during the cycle set of the TM1-0 bits. If the soft mute is cancelled within the cycle set by the TM1-0 bits 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 (Figure 21).

The soft mute function is independent of output volume and cascade connected between both functions.

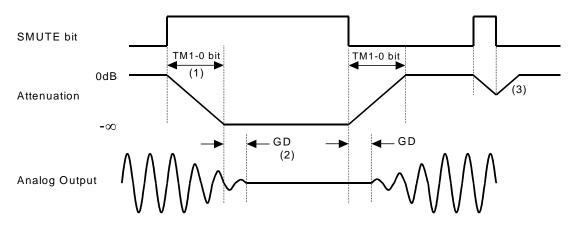


Figure 21. Soft Mute Function

Note:

- (1) The output signal is attenuated until $-\infty$ ("0") by the cycle set by the TM1-0 bits.
- (2) Analog output corresponding to digital input has the group delay (GD).
- (3) If the soft mute is cancelled within the cycle of setting the TM1-0 bits, the attenuation is discounted and returned to 0dB(the set value).

TM1	TM0	Cycle	
0	0	1024/fs	Default
0	1	512/fs	
1	0	256/fs	
1	1	128/fs	

Table 27.	Soft Mute	Time Setting
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AUX Input

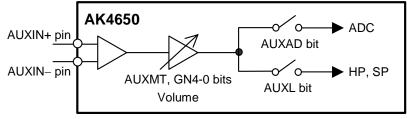


Figure 22. AUX Input

AUX input is a differential input. The AK4650 has a volume for AUX Input. This Volume is controlled by GN3-0 bits as shown in Table 28. The AK4650 register control for GN3-0 does not have function to reduce any pop noise at volume setting change.

AUXMT	GN4-0	GAIN (dB)	STEP]
	00H	+12.0		
0	01H	+10.5		
	02H	+9.0		Default
	:	:	1.5dB	
	08H	+0.0		
	:	:		
	1EH	-33.0		
	1FH	-34.5		
1	Х	MUTE]

Table 28. AUX Input Gain Setting (x: Don't care)

Stereo Line Input

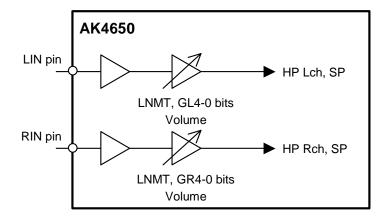


Figure 23. Stereo Line Input

When LNMP bit is "1", MPE pin becomes LIN pin. When RNMD bit is "1", MDT pin becomes RIN pin. LIN/RIN is single-ended input. The AK4650 has a volume for Stereo Line Input. This Volume is controlled by GL4-0 and GR4-0 bits as shown in Table 28. The AK4650 register control for GL4-0 and GR4-0 does not have function to reduce any pop noise at volume setting change.

LNMT	GL/GR4-0	GAIN (dB)	STEP]
	00H	+12.0		
	01H	+10.5		
	02H	+9.0		
0	:	:	1.5dB	
0	08H	+0.0		Default
	:	:		
	1EH	-33.0		
	1FH	-34.5		
1	Х	MUTE]

 Table 29. Stereo Line Input Volume Setting (x: Don't care)

BEEP Input

When the PMBPM bit is set to "1", mono beep input is powered up. And when the BPMHP bit is set to "1", the signal from the BEEP pin is input to Headphone-amp. When the BPMSP bit is set to "1", the signal from the BEEP pin is input to Speaker output. The external resisters Ri adjust the signal level of each BEEP input that are mixed to Headphone and Speaker outputs.

The signal from the BEEP pin is mixed to the Headphone-amp through a –20dB gain stage. The signal from the BEEP pin is mixed to the Speaker-amp without gain. The internal feedback resistance is $20k\Omega \pm 30\%$. When BPMT bit is "1", BEEP input is muted.

When INBP bit is "1", BEEP pin becomes IN2 pin.

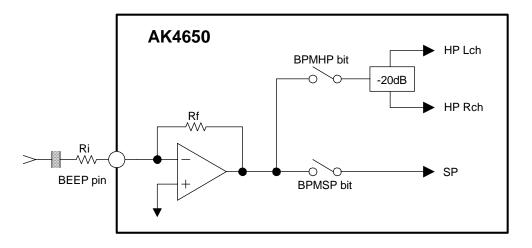


Figure 24. Block Diagram of BEEP pins $(Rf=20k\Omega\pm30\%)$

■ MONO LINE OUTPUT (MOUT+ and MOUT- pins)

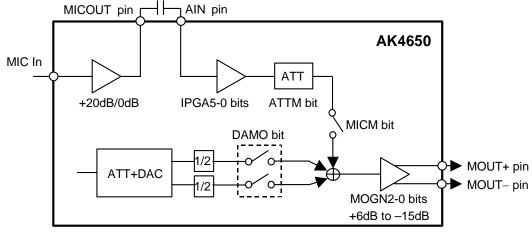


Figure 25. Mono Output

Mono mixer mixes signals from MIC In, DAC Lch and Rch. This mixed signal is output from the MOUT+ and MOUT– pins, creating a differential output. Either the MOUT+ or MOUT– pin can be also used as single-ended output. Load resistance is $20k\Omega(min)$. When PMMO bit is "0", mono output is powered-down and MOUT+/– pins become Hi-Z.

PMMO	MOMT	Mode	MOUT+/MOUT- pin	
0	Х	Power-down	Hi-Z	Default
1	1	Mute	VCOM	
1	0	Normal operation	Normal operation]

Table 30. Mono Output Setting

Volume of path from DAC is controlled by ATTL7-0 and ATTR7-0 bits (Table 25). Volume of path from IPGA is controlled by ATTM bit (Table 8). Mono output amp has +6dB to -15dB gain that are set by the MOGN2-0 bits (Table 31).

		GTED	
MOGN2-0	GAIN (dB)	STEP	
0H	+6.0		Default
1H	+3.0		
2H	+0.0		
3H	-3.0	3dB	
4H	-6.0	SUD	
5H	-9.0		
6H	-12.0		
7H	-15.0		

Table 31. Mono Output Gain Control

Headphone Output

Power supply voltage for the Headphone-amp is supplied from the HVDD pin and centered on the MUTET voltage. The Headphone-amp output load resistance is min.16 Ω . When the HPMT bit is "1" at PMHPL=PMHPR= "1", the common voltage rises to HVDD/2. When the HPMT bit is "1", the common voltage of Headphone-amp falls and the outputs (HPL and HPR pins) go to HVSS.

A capacitor between the MUTET pin and ground reduces pop noise at power-up/down. It is recommended that the capacitor with small variation of capacitance and low ESR (Equivalent Series Resistance) over all temperature range, since the rise and fall time in Table 32 depend on the capacitance and ESR of the external capacitor at MUTET pin.

t _r : Rise Time up to HVDD/4	100k x C (typ)
t _f : Fall Time down to 0V	200k x C (typ)

Table 32. Headphone-Amp Rise/Fall Time

[Example]: A capacitor between the MUTET pin and ground = 1.0μ F: Rise Time up to HVDD/4: $t_r = 100k\Omega \times 1\mu$ F = 100ms(typ)

Fall Time down to 0V: $t_f = 200k\Omega \times 1\mu F = 200ms(typ)$

When PMHPL and PMHPR bits are "0", the Headphone-amp is powered-down, and the outputs (HPL and HPR pins) go to HVSS.

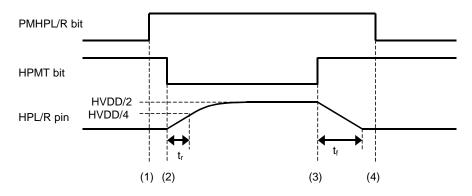
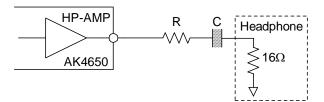


Figure 26. Power-up/Power-down Timing for Headphone-amp

- (1) Headphone-amp power-up (PMHPL and PMHPR bits = "1"). The outputs are still HVSS.
- (2) Headphone-amp common voltage rises up (HPMT bit = "0"). Common voltage of Headphone-amp is rising. This rise time depends on the capacitor value connected with the MUTET pin. The rise time up to HVDD/4 is t_r = 100k x C(typ) when the capacitor value on MUTET pin is "C".
- (3) Headphone-amp common voltage falls down (HPMT bit = "1"). Common voltage of Headphone-amp is falling to HVSS. This fall time depends on the capacitor value connected with the MUTET pin. The fall time down to 0V is $t_f = 200k \times C(typ)$ when the capacitor value on MUTET pin is "C".
- (4) Headphone-amp power-down (PMHPL, PMHPR bits = "0"). The outputs are HVSS. If the power supply is switched off or Headphone-amp is powered-down before the common voltage goes to HVSS, some pop noise occurs.

The cut-off frequency of Headphone-amp output depends on the external resistor and capacitor used. Table 33 shows the cut off frequency and the output power for various resistor/capacitor combinations. The Headphone impedance R_L is 16 Ω . Output powers are shown at HVDD = 2.7, 3.0 and 3.3V. The output voltage of Headphone is 0.6 x AVDD (Vpp).



R [Ω]	C [µF]	fc [Hz]	[z] fc [Hz]		tput Power [m	iW]
IX [52]	υլμη	BOOST=OFF	BOOST=MIN	2.7V	3.0V	3.3V
0	220	45.2	17	27.9	34.5	41.7
0	100	99.5	42	21.9	54.5	41.7
6.8	100	69.8	28	13.7	17.0	20.5
0.8	47	148.5	74	13.7	17.0	20.3
16	100	49.7	19	7.0	8.6	10.4
10	47	105.8	46	7.0	7.0 8.0	10.4

Figure 27. External Circuit Example of Headphone

Table 33. Relationship of external circuit, output power and frequency response

Headphone Jack Detection

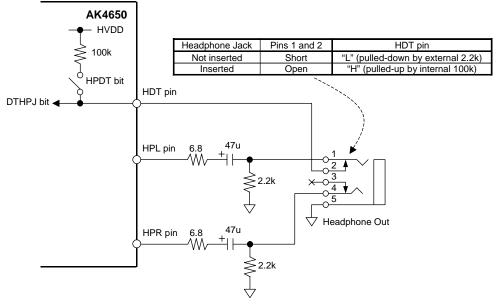


Figure 28. Headphone Jack Detection

Headphone jack detection sequence example:

- (1) HPDT bit = "1".
- (2) HDT pin is pulled-up to HVDD by $100k\Omega$.
- (3) DTHPJ bit indicates whether headphone jack is inserted or not.

Headphone jack detection result is reported to DTHPJ bit (Table 34). If HPINT bit is "1", INTN pin becomes "L" when headphone jack is detected.

Input Level of HDT	DTHPJ bit	Headphone Jack		
< 0.3 x HVDD	0	Not inserted		
> 0.7 x HVDD	1	Inserted		

When ATSW bit is "1" at PMHPL=PMHPR=PMSPK= "1" and HPMT=SPPS= "0", Headphone-amp and Speaker-amp are automatically powered-up/down according to headphone jack detection result (Table 35, Table 36).

DTHPJ	PMHPL PMHPR	HPMT	HP-Amp
0	Х	Х	Power Down
1	0	1	Power Down
1	1	0	Power UP

Table 35. Headphone-amp automatic power-down (ATSW bit = "1")

DTHPJ	PMSPK	SPPS	SPK-Amp
	0	Х	Power Down
0	1	1	Power Save
	1	0	Power UP
1	0	Х	Power Down
1	1	Х	Power Save

Table 36. Speaker-amp automatic power-save (ATSW bit = "1")

Speaker Output

Mono signal [(L+R)/2] converted from stereo DAC output and BEEP input signal can be output via Speaker-amp which output is BTL. DAC output signal can be input to the Speaker-amp via the ALC2 circuit. This Speaker-amp can output a maximum of 300mW@ALC2 bit = "0" and 200mW@ALC2 bit = "1".

ALC2	Ро	
0	300mW	Default
1	200mW	
		_

Table 37. Speaker-Amp Output Power

Speaker blocks (MOUT2, ALC2 and Speaker-amp) can be powered-up/down by controlling the PMSPK bit. When the PMSPK bit is "0", the MOUT2, SPP and SPN pins are placed in a Hi-Z state.

When the SPPS bit is "1", the Speaker-amp is power-save-mode. In this mode, the SPP pin is placed in a Hi-Z state and the SPN pin goes to HVDD/2 voltage. And then the Speaker output gradually changes to the HVDD/2 voltage and this mode can reduce pop noise at power-up. When the AK4650 is powered-down, pop noise can be also reduced in power-save-mode.

PMSPK	SPPS	Mode	SPP pin	SPN pin	
0	Х	Power-down	Hi-Z	Hi-Z	Default
1	1	Power-save	Hi-Z	HVDD/2	
1	0	Normal operation	Normal operation	Normal operation	

Table 38. Speaker Output Setting

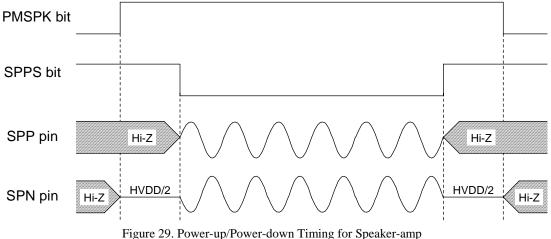


Figure 29. Power-up/Power-down Timing for Speaker-a

■ Mono Output (MOUT2 pin)

The mixed Lch/Rch signal of DAC is output from the MOUT2 pin. When the MO2 bit is "0", this output is OFF and the MOUT2 pin is forced to VCOM voltage. The load impedance is $10k\Omega$ (min.). When the PMSPK bit is "0", the Speaker-amp enters power-down-mode and the output is placed in a Hi-Z state.

■ ALC2 Operation

Input resistance of the ALC2 (MIN pin) is $24k\Omega$ (typ) and centered around VCOM voltage. Figure 30 shows input-output relationship at ALC2 operation (0dBV=1Vrms =2.828Vpp).

The limiter detection level is proportional to HVDD. The output level is limited by the ALC2 circuit when the Speaker-amp output level exceeds +2dBV@HVDD=3.3V. When a continuous signal of +2dBV or greater is input to the ALC2 circuit, the output level is attenuated by ALC2 operation. The change period of the ALC2 limiter operation is set by the ROTM bit and the attenuation level is 0.5dB/step (Table 39).

When the Speaker-amp output level is equal to or lower than 0dBV@HVDD=3.3V, the ALC2 recovery operation starts. The ALC2 recovery operation uses zero crossings and gains of 1dB/step. The ALC2 recovery operation is done until the output level of the Speaker-amp goes to 0dBV@HVDD=3.3V. The ALC2 maximum gain is +18dB. The ROTM bit sets the ALC2 recovery operation period (Table 39).

When the output signal is between +2dBV and 0dBV, the ALC2 limiter or recovery operations are not done.

When the PMSPK bit changes from "0" to "1", the initilization cycle (2048/fs = 46.4ms @fs=44.1kHz at ROTM bit = "0", 512/fs = 11.6ms @fs=44.1kHz at the ROTM bit = "1") starts. This fs value is set by Addr=32H (ADC sampling frequence). The ALC2 is disabled during the initilization cycle and the ALC2 starts after completing the initilization cycle.

Parameter		ALC2 Limiter operation	ALC2 Recovery operation
Operation Start Level		+2dBV	0dBV
ROTM bit = "0		$2/fs = 45 \mu s@fs = 44.1 kHz$	2048/fs = 46.4ms@fs=44.1kHz
Period	ROTM bit = "1"	$2/fs = 181 \mu s@fs = 11.025 kHz$	512/fs = 46.4ms@fs=11.025kHz
Zero-crossing Detection		Diabled	Enabled (Timeout = $2048/fs$)
ATT/GAIN		0.5dB step	1dB step

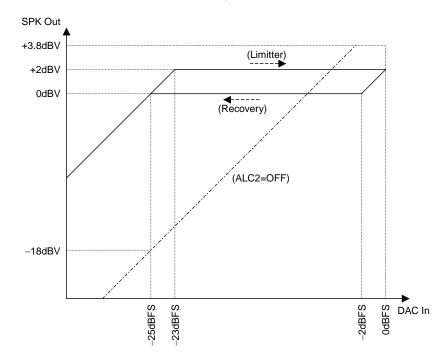


Table 39. Limiter /Recovery of ALC2 at HVDD=3.3V

Figure 30. DAC input – Speaker output relationship (HVDD=3.3V, ALC2 bit = "1")

Example of Path

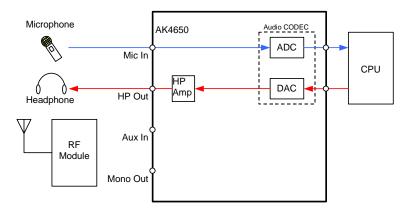


Figure 31. MIC recording & Headphone playback

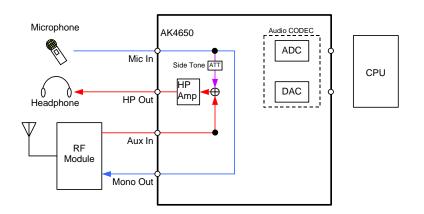


Figure 32. Phone

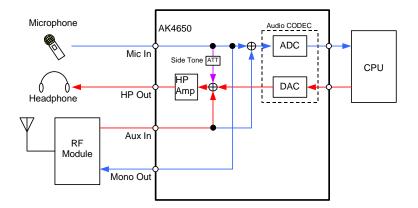


Figure 33. Recording/Playback & Phone

■ A/D Converter for Touch Screen

The AK4650 incorporates a 12-bit successive approximation resistor A/D converter for position measurement, temperature, and battery voltage. The architecture is based on capacitive redistribution algorithm, and an internal capacitor array functions as the sample/hold circuit.

Input Voltage	Output Code					
$(\Delta VREF-1.5LSB) \sim \Delta VREF$	FFF					
$(\Delta VREF-2.5LSB) \sim (\Delta VREF-1.5LSB)$	FFE					
:	:					
0.5LSB ~ 1.5LSB	001					
0 ~ 0.5LSB	000					
Δ VREF: (VREF+) – (VREF–)						

The A/D converter output is a straight binary format as shown in Table 40:

Table 40. Output Code

The A/D converter's full scale measurements depend on the input mode.

Analog Inputs

Analog input is selected via the A2-0 and SER bits in the control register. If the analog inputs are the X or Y-axis and the SER bit is set to "0"(differential mode), the full scale (Δ VREF) is the differential voltage between the non-inverting terminal and the inverting terminal of the measured axis (e.g. (XP) – (XN)). Analog non-inverting input to A/D converter (Δ AIN) is the non-inverting terminal of the non-measured axis while the inverting input is the inverting terminal of the measured axis. If the SER bit is set to "1" (single-ended mode), the full scale of A/D converter (Δ VREF) is the internal reference voltage, or external reference voltage. Note that the SER bit should be set to "0" if the IN2 pin is selected as analog input; nevertheless, the IN2 pin is actually measured by single-ended mode.

SER bit	Analog Input Type	ΔVREF	ΔΑΙΝ
0	Differential	X-axis measurement: XP – XN Y-axis measurement: YP – YN	X-axis measurement: YP – XN Y-axis measurement: XP – YN
1	Single-ended	VREF – TSVSS	AIN – TSVSS

Table 41. Analog Input Type, Δ VREF and Δ AIN

Pen Position Detection of Touch Screen

The selected touch screen position is detected by the voltage measurement of one axis when the voltage is supplied between the two terminals of the other axis. At least two A/D conversions are needed to get the two-dimensional (X/Y axis) position.

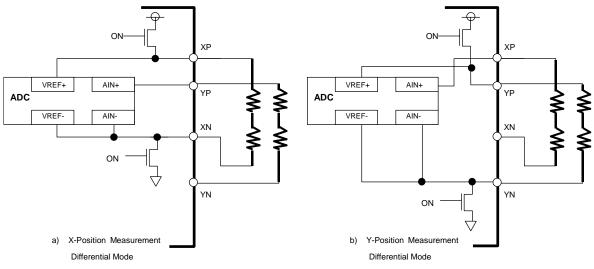


Figure 34. Axis Measurement

The differential mode is generally more accurate than the single-ended mode for the position detection. Since the full scale of single-ended mode is fixed to the internal (or external) reference voltage, the input voltage may exceed the full-scale reference voltage. This problem does not occur in differential mode. In addition to this, the differential mode is less influenced by power supply voltage variation by the ratio-metric measurement.

However, note that the touch screen driver switch is still ON and the current flows even for the A/D conversion time.

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Pen Pressure Measurement

The touch screen pen pressure can be derived from the measurement of the contact resistance between two plates. The contact resistance depends on the size of the depressed area and the pressure. The area of the spot is proportional to the contact resistance. This resistance (R_{touch}) can be calculated using two different methods.

The first method uses the total resistance of the X-plate sheet. The resistance, Rtouch, is calculated from the results of three conversions, X-position, Z1-Position, and Z2-Position, using the following formula:

1) Total resistance of the X-plate sheet (R_{xplate}) is known

The resistance, R_{touch} , is calculated from the results of three conversions, X-position, Z_1 -Position, and Z_2 -Position, using the following formula:

$$R_{\text{touch}} = R_{\text{xplate}} \times \frac{X_{\text{position}}}{4096} \times \left(\frac{Z_2}{Z_1} - 1\right)$$

The second method uses the resistances of both the X-plate and Y-plate. The resistance, Rtouch, is calculated from the results of three conversions, X-position, Y-Position, and Z1-Position, using the following formula:

2) The resistances of both the X-plate and Y-plate (R_{xplate} , R_{yplate}) are known

The resistance, R_{touch} , is calculated from the results of three conversions, X-position, Y-Position, and Z₁-Position, using the following formula:

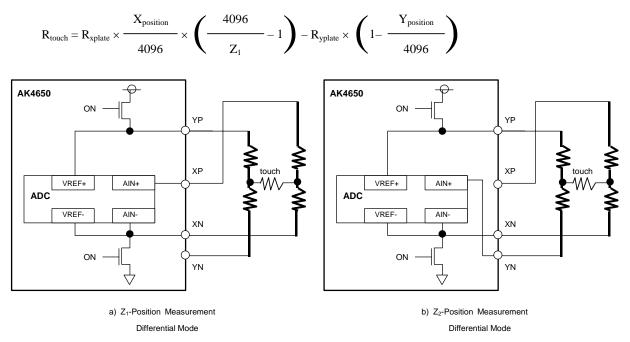


Figure 35. Pen Pressure Measurement

■ Voltage Reference (VREF)

The AK4650 has an internal 2.5V voltage reference. This reference can be turned ON when PMVREF = "1", and OFF when PMVREF = "0". This reference is used in the single-ended mode for the battery monitoring, temperature measurement, or for auxiliary input. A 0.1μ F or larger capacitor should be connected for stable operation of the VREF circuit. Settling time depends on this external capacitance, but 400 μ s or longer time is required if the external capacitance is 0.1μ F.

If an external voltage reference is used, PMVREF bit should be set to "0".

■ Battery Measurement

The AK4650 can measure the battery voltage of 5V or less directly while the AK4650 operates at 2.7V to 3.6V. The input voltage is internally divided down by four.

When the tracking time is adjusted at battery measurement, the method using ADEXE pin is available. ADEXE1-0 bits select the tracking edge (Table 52).

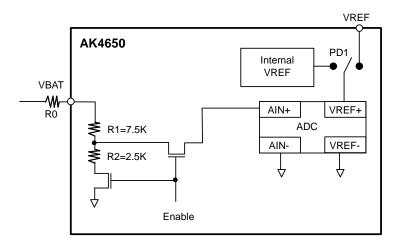


Figure 36. Battery Monitoring

■ Temperature Measurement

Equation <1> describes the forward characteristics of the diode.

$$\begin{split} i_D = & I_0 exp(v_D/V_T) \quad (V_T = kT/q) \qquad <1 > \\ & I_0: \mbox{ reverse saturation current} \\ & q: 1.602189 \times 10^{-19} \mbox{ (electron charge)} \\ & k: 1.38054 \times 10^{-23} \mbox{ (Boltzmann's constant)} \\ & v_D: \mbox{ voltage across diode} \\ & T: \mbox{ absolute temperature } K \end{split}$$

Since a diode junction voltage is theoretically proportional to the temperature, the ambient temperature can be predicted by measuring this voltage.

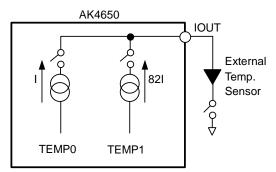


Figure 37. Temperature Measurement

Since the AK4650 has two different fixed current circuits (Figure 37), the temperature can be measured by using two different methods.

The first method needs two conversions, but can derive the temperature directly without knowing the voltage at a specific temperature.

From equation <1>,

 $(i_{D2} / i_{D1}) = exp\{(v_{(NI)} - v_{(I)})/V_T\}, N = (i_{D2} / i_{D1}) = 82 \text{ (ratio of the current)}$

$$\begin{split} T[^{\circ}C] &= \left(\Delta V_{be} \times q\right) / \left(k \times ln \; N\right) - 273 \\ \Delta V_{be} &= V(I82) - V(I) \end{split}$$

 $T[^{\circ}C] = 2.63 \times 10^3 \times \Delta V_{be} - 273$

The second method needs only one conversion as the following equation, but requires knowing the junction voltage at the specific temperature.

$$T[K] = (k/q) \times v_D / \ln(i_D/I_0)$$
 <2>

Pen Interrupt

The AK4650 has pen interrupt function to detect the pen touch. Pen interrupt function is enabled at power-down state. YN pin is connected to GND at the PEN interrupt enabled state. And XP pin is pulled up via an internal resistor (Ri), typically $50k\Omega$. INTN pulled up via an external resistor, $100k\Omega$, is also connected to XP pin. If the two plates are touched, the current flows via $\langle TSVDD \rangle - \langle Ri \rangle - \langle XP \rangle - \langle the plates \rangle - \langle YN \rangle$. The resistance of the plate is generally $1k\Omega$ or less, INTN pin is forced to "L". If the pen is released, INTN pin returns "H" because two plates are disconnected, and the current does not flow via two plates.

The operation of INTN pin depends on A2-0 and PINTE bits when PM12AD bit is "1". If PINTE bit is "1", INTN pin goes to "L" when the plate is touched with pen or finger, and goes to "H" when the plate is not touched regardless of the selection of the analog input channel. If PINTE bit is "0" or 12bit ADC is executed, the operation of INTN pin depends on the selection of the analog input channel regardless of the touched/non-touched state. If the X-axis or Y-axis is selected as analog input, INTN pin is forced to "L" regardless of the touched/non-touched state. If the temperature, VBAT, or auxiliary inputs is selected, INTN pin is forced to "H" regardless of the touched/non-touched state.

When PM12AD bit is "0", XP, YP and XN pins go to Hi-Z state at PINTE bit = "0". At that state, no current flows on touch screen even if pen is touched. INTN pin is forced to "H" regardless of the touched/non-touched state. If PINTE bit = "1", INTN pin goes to "L" when the plate is touched.

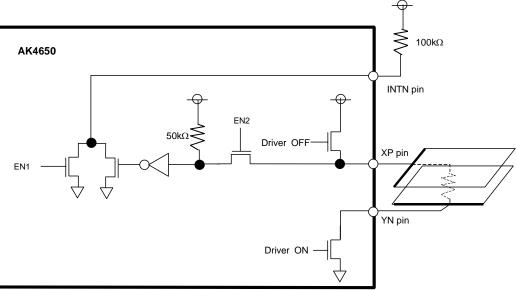
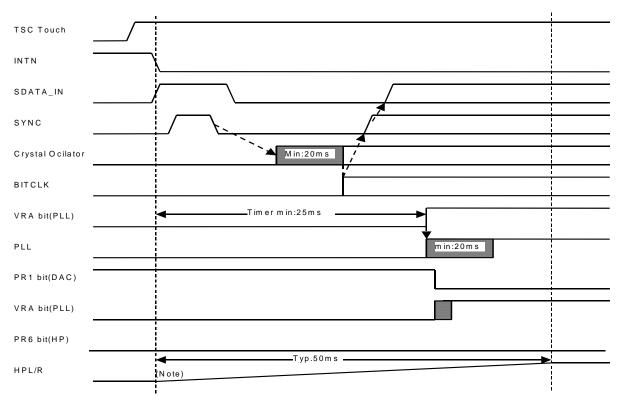


Figure 38. Pen Interrupt Block Diagram

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■ Sequence from AC-Link Power-up to Headphone Output



Note 36. When PR2 = PR3 = PR6 = "0", auto power-up function is selected (ATPU bit = "1") and pen touch is detected, then the AK4650 automatically starts to power-up the headphone-amp by setting PMHPL bit = "1", PMHPR bit = "1" and HPMT bit = "0".

Figure 39. Sequence from AC-Link power-up to headphone output

■ Waking up AC-Link

1. AC-Link Power-down

The AK4650 controls the AC-link power-up/down by PR4 and PR5 bits. When PR4 bit is "1", BITCLK and SDATAIN go to "L", but X'tal oscillator still operates. When PR5 bit is "1", BITCLK and SDATAIN go to "L", and X'tal oscillator is powered-down. PLL power-up/down is controlled by VRA bit.

	BITCLK/SDATAIN output	X'tal oscillator	PLL
PR4 bit = "1"	Stop	Normal operation	Power Down
PR5 bit = "1"	Stop	Stop	Power Down
VRAbit = "0"	Output	Normal operation	Power Down
			

Table 42. AC-Link Power-down

2. AC-Link Power-up

The AK4650 supports the AC-link power-up sequence using SDATAIN output in addition to COLD RESET and WARM RESET. In this mode the operation is same as WARM RESET when the AK4650 receives the SYNC signal.

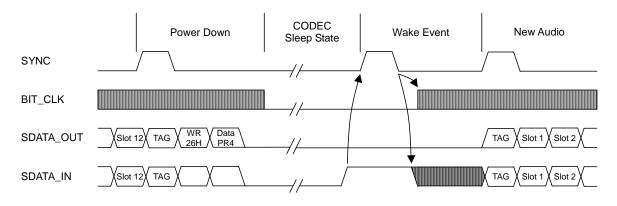


Figure 40. AC-Link Wake Up Function by SDATAIN

3. SDATAIN Output

SDATAIN pin goes to "H" when pen interrupt is enable, AC-link is powered-down and pen touch is detected. SDATAIN pin goes to "L" after SYNC signal is changed to "L" regardless of the pen touch detection. The AK4650 operates same as WARM RESET by receiving this SYNC signal.

4. Wake Up Time

The AK4650 can measures the touch screen position by 12bit ADC at arround 50ms from power-down state.

(1) Wake Up from PR4 bit = "1" and PR5 bit = "1"

The time from pen touch to position measurement available dependes on CPU, X'tal oscillator and the driver in addition to the AK4650. T5 and T7 depend on the AK4650. T7 is less than 1µs. T5 (X'tal oscillation stable and BITCLK is output) is 20ms.

T1 : From pen touch to SDATAIN output => depends on touch screen

T2 : The AK4650 outputs SDATAIN and CPU outputs SYNC => depends on CPU

T3 : SYNC "H" time => depends on CPU, more than 1μ s

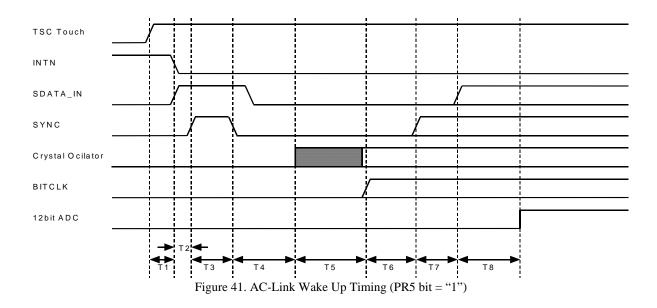
T4 : From SYNC falling edge to X'tal oscillation start => depends on CPU

T5 : X'tal oscillation stable and BITCLK is output => depends on X'tal

T6 : From CPU receives BITCLK to CPU outputs SYNC => depends on CPU

T7 : From the AK4650 receives SYNC to the AK4650 outputs CODEC Ready => 162.8ns

T8 : From CPU receives CODEC Ready to CPU reads ADC data => depends on CPU



5. Pen Interrupt

The AK4650 outputs pen interrupt signal when pen interrupt is enabled and pen touch is detected.

(1) When PENINT bit is "1", INTN pin goes to "L"

- (2) When AC-Link operates, GINT bit is "1" and SLOT bit is "1", bit 0 of slot 12 on SDATAIN goes to "1".
- (3) When PINTR bit is "1", DTPEN bit (Addr=6AH: D2) output the pen touch detection result.

GPIO pin of the controller is not needed in case of (2) or (3).

■ 12bit ADC Data Reception Method

1. Selection of data reception method

The AK4650 supports the following two method for 12bit ADC data reception which is selected by SLOT bit.

SLOT bit	12bit ADC data reception method	Ι
0	"Register"	Default
	The AK4650 reports the measurement result (12bit ADC data) to the control register, then	
	controller gets the data by reading the register of 74H, 76H and 78H.	
1	"Slot"	Ĭ
	The AK4650 reports the measurement result (12bit ADC data) to slot 5, slot 6 or slot 12,	
	then controller gets the data from the asigned slot.	

Table 43. 12bit ADC data reception method

2. Measurement Mode Setting

A2-0 and SER bits set the measurement mode (Table 44). PM12AD and PMVREF bits control the 12bit ADC power-up/down.

DLY3-0 bits set the wait time from mode setting to measurement (Table 46). ADFLT bit sets the averaging of four data or not (Table 47). If these modes are changed during wait time, 12bit ADC is reset and starts with latest mode setting.

	A2-0,	SER bi	t	D	river sw	itch statu	15	ADC Input R (ΔAIN)		Reference voltage $(\Delta VREF)$		
A2	A1	A0	SER	XP	XN	YP	YN	AIN+	AIN-	VREF+	VREF-	Measurement item
0	0	0	1	OFF	OFF	OFF	OFF	TEMP0	GND	VREF	GND	TEMP0 temperature (Default)
0	0	1	1	OFF	OFF	ON	ON	XP	GND	VREF	GND	Y-axis
0	1	0	1	OFF	OFF	OFF	OFF	VBAT	GND	VREF	GND	Battery voltage
0	1	1	1	OFF	ON	ON	OFF	XP(Z1)	GND	VREF	GND	Pen pressure (Z_1)
1	0	0	1	OFF	ON	ON	OFF	YN(Z2)	GND	VREF	GND	Pen pressure (Z_2)
1	0	1	1	ON	ON	OFF	OFF	YP	GND	VREF	GND	X-axis
1	1	0	1	OFF	OFF	OFF	OFF	IN1	GND	VREF	GND	IN1 input
1	1	1	1	OFF	OFF	OFF	OFF	TEMP1	GND	VREF	GND	TEMP1 temperature
0	0	0	0									N/A
0	0	1	0	OFF	OFF	ON	ON	XP	YN	YP	YN	Y-axis
0	1	0	0									N/A
0	1	1	0	OFF	ON	ON	OFF	XP(Z1)	XN	YP	XN	Pen pressure (Z_1)
1	0	0	0	OFF	ON	ON	OFF	YN(Z2)	XN	YP	XN	Pen pressure (Z ₂)
1	0	1	0	ON	ON	OFF	OFF	YP	XN	XP	XN	X-axis
1	1	0	0	OFF	OFF	OFF	OFF	IN2	GND	VREF	GND	IN2 input
1	1	1	0									NA

Table 44. 12bit ADC measurement mode

PM12AD	PMVREF	PINTE	INTN	VREF	Function
0	Х	0	Disable	Х	12bit ADC is powered down. Touch screen driver switch is
					open.
0	х	1	Enable	Х	The AK4650 waits for pen touch. 12bit ADC is powered down.
					Touch screen driver switch is open (though XP pin is TSVDD
					and YN pin is TSVSS).
1	Х	0	Disable	Х	12bit ADC is always powered up.
1	х	1	Auto	Х	After the measurement of XY or XYZ, 12bit ADC is
					automatically powered down and the AK4650 waits for pen
					touch.
х	0	Х	Х	OFF	PMVREF bit controls the VREF power-up/down.
х	1	Х	Х	ON	

Table 45. 12bit ADC Power-up/down control (x: Don't care)

DLY3-0 bit	Wait time	
0H	1TS	Default
1H	2TS	
2H	4TS	
3Н	8TS	
4H	12TS	
5H	16TS	
6H	24TS	
7H	32TS	
8H	48TS	
9H	64TS	
AH	96TS	
BH	128TS	
СН	160TS	
DH	192TS	
EH	224TS	
FH	256TS	

Table 46. Wait time from mode setting to measurement (1TS = 1SYNC period = $20.83\mu s@fs=48kHz$)

ADFLT bit	Filter	
0	Data is not averaged. The AK4650	Default
	outputs one measurement result.	
1	Four data are averaged.	

Table 47. 12bit ADC data filtering (SLOT bit = "0")

3. Register Read

The AK4650 supports the following three measurement methods which are selected by ADMODE1-0 bits.

- (1) 74H Register Read
- (2) XY Measurement Mode
- (3) XYZ Measurement Mode

ADMODE 1-0 bits	Measurement Method	
00	74H Register Read	Default
01	XY measurement mode	
10	XYZ measurement mode	
11	Reserved	

Table 48. 12bit ADC measurement mode setting

3-1. 74H Register Read Method

The AK4650 reports the measurement result set by A2-0 and SER bits to the control register address 74H. The controller gets the measurement result by reading the control register address 74H. 74H register is cleared by PM12AD bit = "0". 74H consists of the followings:

D15(ADE)	: 0: Measurement is not completed.
	1: Measurement result is reported to D11-0.
D14-12(A2-0)	: Measurement mode (A2-0 bits)

D11-0 : Measurement results (12bit)

3-1-1.ADFLT bit = "0" : No average

- (1) 70H(ADFLT, DLY3-0, ADEXE1-0) is set. A2-0 and SER bits set the measurement mode at the same time, and 12bit ADC is powered-up by setting PM12AD bit = "1".
- (2) After the wait time set by DLY3-0 bits, the measurement is available.
- (3) When 74H is read, then 12bit ADC is executed, data is reported to 74H, and the measurement mode is reported at next SYNC.

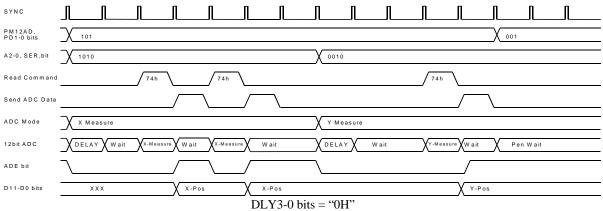


Figure 42. X and Y position measurement sequence (ADFLT bit = "0")

3-1-2.ADFLT bit = "1" : Averaging of four data

- (1) 70H (ADFLT, DLY3-0 and ADEXE1-0) is set. A2-0 and SER bits set the measurement mode at the same time, and 12bit ADC is powered-up by setting PM12AD bit = "1".
- (2) After the wait time set by DLY3-0 bits, the measurement is executed for four times and the averaged result is reported to 74H.
- (3) When 74H is read, the data of 74H is output.

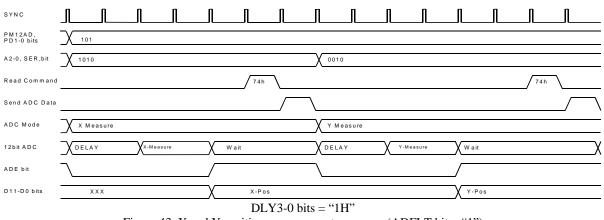


Figure 43. X and Y position measurement sequence (ADFLT bit = "1")

3-2. X-Y Measurement Mode

When ADMODE1-0 bits are set to "01" and PM12AD bit is "1", the AK4650 measures X-position at first, Y-position at second, and reports X-position to 74H, Y-position to 76H, respectively, regardless of A2-0 and SER bits. 74H and 76H registers are cleared by PM12AD bit = "0".

- (1) 70H(ADFLT, DLY3-0, MSR1-0, ADEXE1-0) is set. ADMODE1-0 bits should be set to "01" at the same time, and 12bit ADC is powered-up by setting PM12AD bit = "1".
- (2) After the wait time set by DLY3-0 bits, the AK4650 measures X-position and reports the result to 74H. When ADFLT bit is "1", averaged data of four measurement results is reported to 74H.
- (3) After X-position measurement and the wait time set by DLY3-0 bits, the AK4650 measures Y-position and reports the result to 76H. When ADFLT bit is "1", averaged data of four measurement results is reported to 76H.
- (4) After the measurement, ADMODE1-0 bits are automatically reset to "00".
- (5) When PINTE bit = "1", 12bit ADC is powered-down and the AK4650 waits for pen touch.

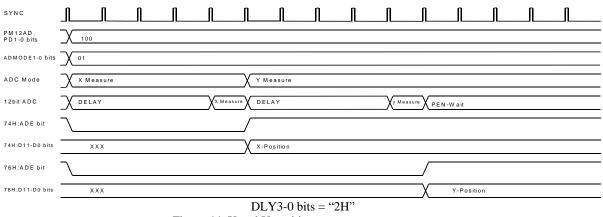


Figure 44. X and Y position measurement sequence

When ADMODE1-0 bit is set to "01" again, then the AK4650 set ADE bit as "0" and measurement is executed again.

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3-3. XYZ Measurement Mode

When ADMODE1-0 bits are set to "10" and PM12AD bit is "1", the AK4650 measures X-position at first, Y-position at second, the item set by A2-0 and SER bits at third, and reports X-position to 74H, Y-position to 76H, third result to 78H, respectively. 74H, 76H and 78H registers are cleared by PM12AD bit = "0".

- (1) 70H(ADFLT, DLY3-0, MSR1-0, ADEXE1-0) is set. Measurement mode should be set by A2-0 and SER bits, and ADMODE1-0 bits should be set to "10" at the same time. 12bit ADC is powered-up by setting PM12AD bit = "1".
- (2) After the wait time set by DLY3-0 bits, the AK4650 measures X-position and reports the result to 74H. When ADFLT bit is "1", averaged data of four measurement results is reported to 74H.
- (3) After X-position measurement and the wait time set by DLY3-0 bits, the AK4650 measures Y-position and reports the result to 76H. When ADFLT bit is "1", averaged data of four measurement results is reported to 76H.
- (4) After Y-position measurement and the wait time set by DLY3-0 bits, the AK4650 measures the item set by A2-0 and SER bits, and reports the result to 78H. When ADFLT bit is "1", averaged data of four measurement results is reported to 78H.
- (5) After the measurement, ADMODE1-0 bits are automatically reset to "00".
- (6) When PINTE bit = "1", 12bit ADC is powered-down and the AK4650 waits for pen touch.

SYNC	⅃∟														
PM 12AD, PM VREF, PINTE bits	X	101													
ADMODE1-0 bits		10													
ADC Mode	\sum	K Measure			Х м	easure			X Z1 N	leasure					
12bit ADC	\mathcal{T}	DELAY	X Meas	ure	X DELA	Υ	y Measu	Ire	X DELA	λY	Z1 Mea	sure	_X	PEN	-Wait
74H:ADE bit															
74H:D11-D0 bits		xxx			Х-Р	osition									
76H:ADE bit															
76H:D11-D0 bits		xxx							χ	Y - Position					
78H:ADE bit															
78H:D11-D0 bits		ххх											X	Z1-Positi	on

DLY3-0 bits = "1H"

Figure 45. X, Y and Z position measurement sequence (ADFLT bit = "1", A2-0 bits = "011", SER bit = "0")

When ADMODE1-0 bit is set to "10" again, then the AK4650 set ADE bit as "1" and measurement is executed again.

4. Method using SDATAIN Slot

When SLOT bit is "1", 12bit ADC data is output via slot 5, 6 or 12 of SDATAIN. When SLOTNO1-0 bits are "00", "01" or "10", slot 5, 6 or 12 is used, respectively. Output data is 16bit. The measurement result is output only when 12bit ADC is executed. When data is output, tag bit for slot 5, 6 or 12 (bit 10, 9 or 3) is set to "1". When ADC is not executed, tag bit is set to "0" and data of slot 5, 6 or 12 are set to all "0".

SLOT bit	SLOTNO1-0 bits	Slot number	I
0	Х	Not used	Default
	00	Slot 5	
1	01	Slot 6	
	10	Slot 12	
	11	Reserved	

Table 49. Slot number setting for 12bit ADC measurement result output

Bit 19 : "1" when data is valid.

= Tag bit in slot 0 is also set to "1" when data is output.

- Bit 18-16 : Measurement mode (A2-0 bit)
- Bit 15-4 : 12bit ADC data

In this mode, ADFLT bit is ignored. Even if ADFLT bit is set to "1", four data average is not executed. The AK4650 supports the following three measurement modes:

- (1) Sequential measurement of set item
- (2) XY measurement mode
- (3) XYZ measurement mode

ADMODE1-0 bits	Measurement method	
00	Sequential measurement of set item	Default
01	XY measurement mode	
10	XYZ measurement mode	
11	Reserved	

Table 50. 12bit ADC measurement mode setting

4-1. Sequential measurement of set item

The AK4650 measures the item set by A2-0 and SER bits sequentially, and outputs the results via slot. When 12bit ADC is powered-down, bit 19 is "0".

- (1) When SLOT bit = "1", pen touch waiting state (PM12AD bit = "0") should set.
- (2) When the controller receives PEN Interrupt, the controller should set PM12AD bit = "1". Then data is output via slot.
- (3) If 12bit ADC is powered-up by setting PM12AD bit = "1" and measurement item is set by A2-0 and SER bits, then the set item is sequentially measured after the wait time set by DLY3-0 bits.

If the measurement is not completed in one SYNC period, bit 19 (ADE) is "0".

4-2. X-Y Measurement Mode

When PM12AD bit is "1" and then ADMODE1-0 bits are set to "01", the AK4650 measures X-position for times set by MSR1-0 bits at first, Y-position for times set by MSR1-0 bits at second, regardless of A2-0 and SER bits. Measurement results are output via slot after wait time set by DLY3-0 bits.

- (1) When SLOT bit = "1", pen touch waiting state (PM12AD bit = "0" and PINTE bit = "1") should be set.
- (2) When the controller receives PEN Interrupt, the controller should set PM12AD bit = "1". Then data is output via slot.
- (3) 12bit ADC is powered-up by setting PM12AD bit = "1". ADMODE1-0 bits should be set to "01" at the same time.
- (4) After the wait time set by DLY3-0 bits, the AK4650 measures X-position for times set by MSR1-0 bits and outputs the result via slot.
- (5) After X-position measurement and the wait time set by DLY3-0 bits, the AK4650 measures Y-position for times set by MSR1-0 bits and outputs the result via slot.
- (6) After Y-position measurement, 12bit ADC is powered-down and the AK4650 waits for pen touch.

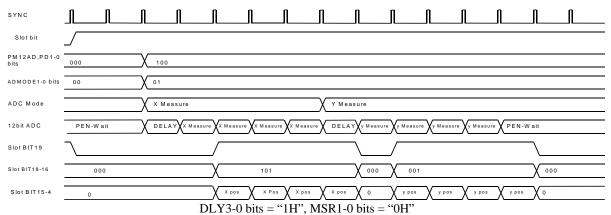


Figure 46. X and Y position measurement sequence

MSR1-0 bit	Measurement times	
00	4 times	Default
01	8 times	
10	16 times	
11	32 times	

Table 51. Measurement times (SLOT bit = "1")

4-3. X-Y-Z Measurement Mode

When PM12AD bit is "1" and then ADMODE1-0 bits are set to "10", the AK4650 measures X-position for times set by MSR1-0 bits at first, Y-position for times set by MSR1-0 bits at second, the item set by A2-0 and SER bits for times set by MSR1-0 bits at third. Measurement results are output via slot after wait time set by DLY3-0 bits.

- (1) When SLOT bit = "1", pen touch waiting state (PM12AD bit = "1" and PINTE bit = "1") should be set.
- (2) When the controller receives PEN Interrupt, the controller should set SLOT bit = "1". Then data is output via slot.
- (3) 12bit ADC is powered-up by setting PM12AD bit = "1". ADMODE1-0 bits should be set to "10" at the same time.
- (4) After the wait time set by DLY3-0 bits, the AK4650 measures X-position for times set by MSR1-0 bits and outputs the result via slot.
- (5) After X-position measurement and the wait time set by DLY3-0 bits, the AK4650 measures Y-position for times set by MSR1-0 bits and outputs the result via slot.
- (6) After Y-position measurement and the wait time set by DLY3-0 bits, the AK4650 measures the item set by A2-0 and SER bits for times set by MSR1-0 bits and outputs the result via slot.
- (7) After the measurement, 12bit ADC is powered-down and the AK4650 waits for pen touch.

5. ADEXE pin

When the AK4650 measures the position during the data is sent to LCD, the position measurement result may have some error caused by the noise from LCD. The AK4650 prepares the measurement mode by using sync signal from LCD after data transferring is completed. The sync signal is input to ADEXE pin of the AK4650.

ADEXE1-0 bit	Measurement mode			
00	ADC is executed regardless of ADEXE pin.			
01	ADC is executed at ADEXE pin = " \uparrow ".			
10	ADC is executed at ADEXE pin = " \downarrow ".			
11	Reserved			

Table 52. ADEXE mode

Method using Slot 12 of SDATAIN

When SLOT bit is "1", pen interrupt and headphone jack detection results are output via slot 12 of SDATAIN.

Bit 1: Headphone jack detection result Bit 0: Pen interrupt

■ Connection with Digital AC '97 Controller

The AK4650 communicates with its companion AC '97 controller via a digital serial link, "AC-link". All digital audio streams, and command/status information are communicated over this point to point serial interconnect. A breakout of the signals connecting the two is shown in the following figure.

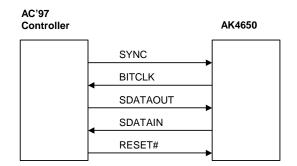


Figure 47. Connection between AK4650 and AC '97 controller

RESETN	(Input)	: Control signal to reset the AK4650
BITCLK	(Output)	: 12.288MHz clock output from the AK4650
SYNC	(Input)	: Control signal to synchronize the AK4650 with AC'97 controller
SDATAIN	(Output)	: Data signal input to the controller (output from the AK4650)
SDATAOUT	(Input)	: Data signal output to the controller (input from the AK4650)

Digital Interface

The AK4650 incorporates a 5 pin digital serial interface that links it to the AC '97 controller. AC-link is a bi-directional, fixed rate(48kHz), serial PCM digital stream. It handles input/output audio streams and 12bit ADC results, as well as control register accesses employing a time division multiplexed (TDM) scheme. The AC-link architecture divides each audio frame into 12 outgoing and 12 incoming data streams, each with 20-bit sample resolution. DAC and ADC resolution of the AK4650 is 16 bit resolution. The data streams currently defined by the AC '97 specification include:

- PCM Playback 2 output slots
 - 2 channel composite PCM output stream
- PCM Record data 2 input slots 1 channel composite PCM input stream
- Control 2 output slots
- Control register write port
- Status 2 input slots
- Control register read port • 12bit ADC data 3 input slots
- 12bit ADC data input stream

SYNC, fixed at 48kHz, is derived by dividing down the serial bit clock (BITCLK) output from the AK4650. BITCLK, fixed at 12.288 MHz, provides the necessary clocking granularity to support 12, 20-bit outgoing and incoming time slots. AC-link serial data is transitioned on each rising edge of BITCLK. The receiver of AC-link data, the AK4650 for outgoing data and AC '97 controller for incoming data, samples each serial bit on the falling edges of BITCLK.

The AC-link protocol provides for a special 16-bit slot (Slot 0) wherein each bit conveys a valid tag for its corresponding time slot within the current audio frame. A "1" in a given bit position of slot 0 indicates that the corresponding time slot within the current audio frame has been assigned to a data stream, and contains valid data. If a slot is "Tagged" invalid, it is the responsibility of the source of the data (the AK4650 for the input stream, AC '97 controller for the output stream), to stuff all bit positions with 0's during that slot's active time.

SYNC remains high for a total duration of 16 BITCLKs at the beginning of each audio frame. The portion of the audio frame where SYNC is high is defined as the "Tag Phase". The remainder of the audio frame where SYNC is low is defined as the "Data Phase".

Note that SDATAOUT and SDATAIN data is delayed one BITCLK because AC'97 controller causes SYNC signal high at a rising edge of BITCLK which initiates a frame.

"Output" stream means the direction from AC'97 controller to the AK4650, and "Input" stream means the direction from the AK4650 to AC'97 controller.

AC-Link Protocol

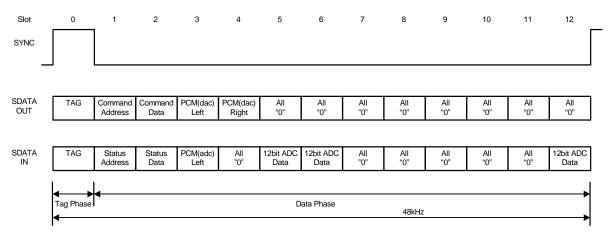
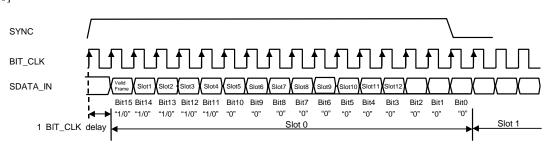


Figure 48. AC-Link protocol

AC-link protocol identifies 13 slots of data per frame. The frequency of SYNC is fixed to 48kHz. Only Slot 0, which is the Tag phase, is 16bits, all other slots are 20bits in length. These slots are explained in later sections.

1) AC-Link Audio Output Frame (SDATAOUT)







Slot 0 consists of sixteen bits (bit 15-0). Bit 15-11 are available in the AK4650. Each bit means valid by "1" and invalid by "0".

Bit 15 (Valid Frame bit): Validity of the frame

- "1" = At least one of bit 14-11 (slot 1-4) must be valid. Bit 10-0 are ignored.
- "0" = The AK4650 ignores all following information in the frame.
- Bit 14 (Slot 1 valid bit): Validity of slot 1 (command address input)
- Bit 13 (Slot 2 valid bit): Validity of slot 2 (command data input)
- Bit 12 (Slot 3 valid bit): Validity of slot 3 (DAC Left data input)
- Bit 11 (Slot 4 valid bit): Validity of slot 4 (DAC Right data input)

If each bit is "0", the AK4650 ignores the slot indicated by "0". On the other hand, if each bit is "1", the slot is valid. Bit 10-0 should be "0".

A new audio output frame begins with a low to high transition of SYNC. SYNC is synchronous to the rising edge of BITCLK. On the immediately following falling edge of BITCLK, the AK4650 samples the assertion of SYNC. This falling edge marks the time when both sides of AC-link are aware of the start of a new audio frame. On the next rising of BITCLK, the AC '97 controller transitions SDATAOUT into the first bit position of slot 0 (Valid Frame bit). Each new bit position is presented to AC-link on a rising edge of BITCLK, and subsequently sampled by the AK4650 on the following falling edge of BITCLK. This sequence ensures that data transitions, and subsequent sample points for both incoming and outgoing data streams are time aligned.

Data should be sent to the AK4650 with MSB first through the SDATAOUT.

Table 53 shows the relationship of bit 14&13	and the Read/Write operation.
--	-------------------------------

Bit 15 Valid Frame	Bit 14: Slot1 Valid Bit (Command Address)	Bit 13: Slot 2 Valid Bit (Command Data)	Read/Write Operation
1	1	1	Read/Write (Normal Operation)
1	0	1	Ignore
1	1	0	Read: Normal Operation
			Write: Ignore
1	0	0	Ignore

Table 53. AK4650 Addressing: Slot 0 Tag Bits

[Slot 1]: Command Address Port

Slot1 gives the address of the command data, which is given in the slot 2. The AK4650 has 30 valid registers of 16bit data. See "Mixer Registers".

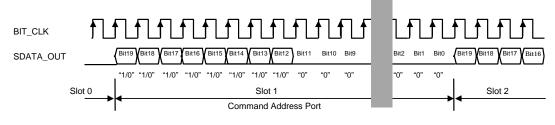


Figure 50. Slot 1

- Bit 19: Read/Write command (1bit; "1"=read, "0"=write)
- Bit 18-12: Control Register Index (7bit; see "Mixer Registers" for the detail)
- Bit 11-0: Reserved (12bit; "0")

Bit 18 of this slot 1 is equivalent to the most significant bit of the index register address.

The AK4650 ignores bit 11-0. These bits will be reserved for future enhancement and must be staffed with 0's by the AC'97 controller.

[Slot 2]: Command Data Port

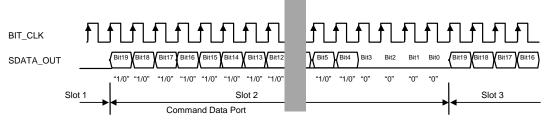


Figure 51. Slot 2

Bit19-4:	Control Register Write Data (16bit)
	(If bit 19 of slot 1 is "1", all bit19-4 should be "0".)
Bit3-0:	Reserved (4bit; "0")

If bit 19 in slot 1 is "0", the AC'97 controller must output Command Data Port data in slot 2 **of the same frame**. If the bit 19 in slot 1 is "1", the AK4650 will ignore any Command Data Port data in slot 2.

Bit19 of this slot 2 is equivalent to D15 bit of mixer register value.

[Slot 3]: PCM Playback Left Channel (16bit)

The AK4650 uses the playback (DAC) data format in slot 3 for left channel. Playback data format is MSB first. Data format is 16bits 2's complement. AC'97 controller should stuff bit 3-0 with "0". If valid bit (slot 3) in the slot 0 is invalid ("0"), the AK4650 interprets the data as all "0".

Bit 19-4: Playback data (16bit)

Bit 3-0: "0" (4bit)

[Slot 4]: PCM Playback Right Channel (16bit)

The AK4650 uses the playback (DAC) data format in slot 4 for right channel. Playback data format is MSB first. Data format is 16bits 2's complement. AC'97 controller should stuff bit 3-0 with "0". If valid bit (slot 4) in the slot 0 is invalid ("0"), the AK4650 interprets the data as all "0".

Bit 19-4: Playback data (16bit) Bit 3-0: "0" (4bit)

[Slot 5-12]: Not implemented in the AK4650

2) AC-Link Input Frame (SDATAIN)

Each AC-link frame consists of one 16bit tag phase and twelve 20bit slots used for data and control.

[Slot 0]

Slot 0 is a special time frame, and consists of 16bits. Slot 0 is also named the Tag phase. The AK4650 supports bits 15-9 and bit 3. Each bit indicates "1"=valid(normal operation) or ready, "0"=invalid (abnormal operation) or not ready. If the first bit in the slot 0 (Bit15 = "Codec Ready") is valid, the AK4650 is ready for normal operation. If the "Codec Ready" bit is invalid, the following bits and remaining slots are all "0". AC'97 controller should ignore the following bits in the slot 0 (and all other slots. When the ADC sampling rate is set for less than 48kHz, then bits 12 and 11 in slot 0 (corresponds to slot 3 and slot 4 respectively) will be 1's when valid data is transferred in SDATAIN, and will be 0's when no data is transmitted.

< "On-demand" base data transaction>

For variable sample rate input, the tag bit for each input slot indicates whether valid data is present or not. Thus, even in variable sample rate mode, the AK4650 is always the master. For SDATAIN (AK4650 to Controller), the AK4650 sets the TAG bit. For SDATAOUT (Controller to AK4650), the AK4650 sets the SLOTREQ bit and then checks for the TAG bit in the next frame. AK4650 expects Controller will reply TAG bit in the next frame correctly.

Bit 14 means that Slot 1 (Status Address) output is valid or invalid. And Bit 13 means that Slot 2 (Status Data) is valid or invalid. Table 54 shows the relationship between bit 14,13 and each Status of the AK4650.

Bit 15 (Codec Ready)	Bit 14 (Status Address)	Bit 13 (Status Data)	Status
1	1	1	There is a Read Command in the previous frame. Then both Slot 1 and Slot 2 output normal data. If the access to non-implemented register or odd register is requested, the AK4650 returns "valid" 7-bit register address in slot 1 and returns "valid" 0000h data in slot 2 on the next AC-link frame.
1	1	0	Prohibited or non-existing
1	0	1	Prohibited or non-existing
1	0	0	There is no Read Command in the previous frame. Bits 19-12 and 9-0 in Slot 1 are set to "0". And Slot 2 outputs all "0".

Table 54. SDATAIN Slot0

Note 37. The above Read sequence is done as response for previous frames read command. That is, if the previous frame is the Write Command, AK4650 outputs bit14 ="0", bit13 ="0" and slot 1&2 = All"0", if there is no SLOTREQ.

Note 38. The Bits 14 and 13 in Slot 0 is independent of the SLOTREQ Bits 11 and 10 in Slot 1 which the AK4650 supports.

Bit12 means the output of Slot 3 (PCM(ADC) Left) is valid or invalid. Bit 11 is same as bit 12. Slot 4 is all "0" regardless of bit 11. When ADEXE1-0 bits are not "00" and SLOT bit is "1", Tag bit corresponding to the slot set by SLOTNO1-0 bits (Slot 5=Bit 10, Slot 6=Bit 9, Slot 12=Bit 3) are fixed to "1" (valid). Bits 8-4 and 2-0 are occupied with "0".

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A new audio input frame begins with a low to high transition of SYNC. SYNC is synchronous to the rising edge of BITCLK. On the immediately following falling edge of BITCLK, the AK4650 samples the assertion of SYNC. This falling edge marks the time when both sides of AC-link are aware of the start of a new audio frame. On the next rising of BITCLK, the AK4650 transitions SDATAIN into the first bit position of slot 0 ("Codec Ready" bit). Each new bit position is presented to AC-link on a rising edge of BITCLK, and subsequently sampled by the AC '97 controller on the following falling edge of BITCLK. This sequence ensures that data transitions, and subsequent sample points for both incoming and outgoing data streams are time aligned.

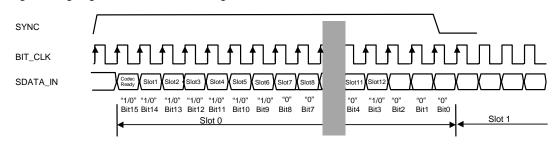


Figure 52. Slot 0

[Slot 1]: Status Address Port

Audio input frame slot 1's stream echoes the control register index, for historical reference, for the data to be returned in slot 2. (Assuming that slot 1 valid bit and slot 2 valid bit in the slot 0 had been tagged "valid" by the AK4650.)

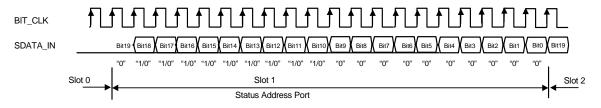


Figure 53. Slot 1

This address shows register index for which data is being returned in the slot 2. This address port is the copy of slot 1 of the output frame, and index address input to SDATAOUT is looped back to the AC'97 controller through SDATAIN even for non-supported register.

For "On Demand" base data transaction, when the DAC sampling rate is set less than 48kHz, then AK4650 will request new audio data as required by setting the SLOTREQ bits 11 and 10 in slot 1 to 0's. When no data is required to support the selected sampling rate, these bits will be 1's. When SLOTREQ bits are asserted as "send data request" during the current frame on SDATAIN, AC'97 digital controller should send data onto the corresponding slot in the next frame on SDATAOUT. If VRA bit is set to "0", SLOTREQ bits always show "0" and sample rate is forced to 48kHz.

Description			
Reserved (Set to "0")			
Control Register Index (7bit; Set to "0" if tagged invalid)			
Slot 3 Request: PCM Lch			
"0": send data request, "1": do not send			
Slot 4 Request: PCM Rch			
"0": send data request, "1": do not send			
Reserved (10bit; Set to "0")			

Table 55. SLOTREQ bit

[Slot 2]: Status Data Port

Status data addressed by command address port of Output Stream is output through SDATAIN pin.

Bit 19-4 Control Register Read Data (16bit; the contents of indexed address in the slot 1)

Bit 3-0 "0" (4bit)

Note that the address of Status Data Port data are consistent with Status Address Port data of the slot 1 in the same frame. If the read operation is issued in the frame N by AC'97 controller, Status Data Port data is output through SDATAIN in the frame N+1. Note that data is output in only this frame, only one time and that the following frames are invalid if the next read operation is not issued.

[Slot 3]: PCM Record Left Channel

Record (ADC) data format is MSB first. Data format is 2's complement. As the resolution of the AK4650 is 16bit, lower 4 bits are ignored. If ADC block is powered down, slot 3 valid bit in the slot 0 is invalid ("0"), and data is output as all "0".

Bit 19-4: Audio ADC left channel output (16bit)

Bit 3-0: "0" (4bit)

[Slot 4]: Reserved for future enhancement Bit 19-0 "0"

Dit 17-0 0

[Slot 5]: 12bit ADC data for TSC

When SLOT bit = "1" and SLOTNO1-0 bits = "00", 12bit ADC data is output.

Bit 19: ADE (1bit; "0"=Invalid, "1"=Valid)

Bit 18-16: A2-0 (3bit; Measurement mode)

- Bit 15-4: D11-0 (12bit; 12bit ADC data)
- Bit 3-0: "0" (4bit)

[Slot 6]: 12bit ADC data for TSC

When SLOT bit = "1" and SLOTNO1-0 bits = "00", 12bit ADC data is output.

Bit 19: ADE (1bit; "0"=Invalid, "1"=Valid)

Bit 18-16: A2-0 (3bit; Measurement mode)

Bit 15-4: D11-0 (12bit; 12bit ADC data)

Bit 3-0: "0" (4bit)

[Slot 7-11]: Reserved for future enhancement Bit 19-0 "0"

[Slot 12]: 12bit ADC data for TSC, pen touch and headphone jack detection results

When SLOT bit = "1" and SLOTNO1-0 bits = "00", 12bit ADC data is output. When SLOT bit = "1", pen touch and headphone jack detection results are output regardless of SLOTNO1-0 bits.

Bit 19: ADE (1bit; "0"=Invalid, "1"=Valid)

- Bit 18-16: A2-0 (3bit; Measurement mode)
- Bit 15-4: D11-0 (12bit; 12bit ADC data)

Bit 3-2: "0" (2bit)

- Bit 1: DTHPJ (1bit; "0"=Not inserted, "1"=Inserted)
- Bit 0: DTPEN (1bit; "0"=Not touched, "1"=Touched)

Power On

Note that AK4650 must be in cold reset at power on and RESETN must be "L" until master crystal clock becomes stable, or cold reset must be done once after master clock is stable.

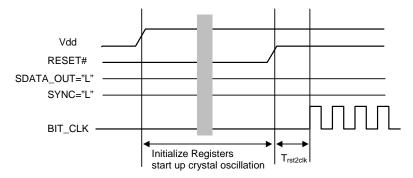


Figure 54. Power On Timing

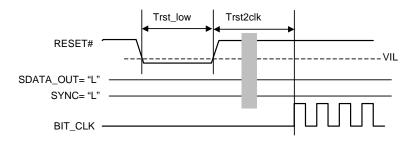
Cold Reset

Note that both SDATAOUT and SYNC must be "L" at the rising edge of RESETN for cold reset.

The AK4650 initializes all registers including the Power-down Control Registers, BIT-CLK is reactivated and each analog output except for HP-Amp is in Hi-Z state while RESETN pin is "L".

At the rising edge of RESETN, the AK4650 starts the initialization of ADC and DAC, which takes 1028TS cycles. After that, the AK4650 is ready for normal operation. At that time, VRA bit is its default value ("0"). Therefore, fs=48kHz and TS= $1/fs=20.83\mu s$.

Status bit in the slot 0 is "0" (not ready) when the AK4650 is in RESET period ("L") or in initialization process. After initialization cycles, the status bit goes to "1" (ready).





Warm Reset

The AK4650 initiates warm reset process by receiving a single pulse on the SYNC. The AK4650 clears PR4 bit and PR5 bit in the Power-down Control Register. However, warm reset does not influence PR0-3, 6 and 7 bits in Power-down Control Register.

Note 39. SYNC signal should synchronize with BITCLK after AK4650 starts to output BITCLK clock.

Note 40. If an external clock is used, external clocks should be supplied before issuing a sync pulse for warm reset. ADC and DAC require 1028TS for the initialization.

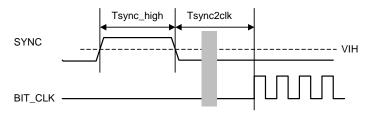


Figure 56. Warm Reset Timing

■ Active Test Mode

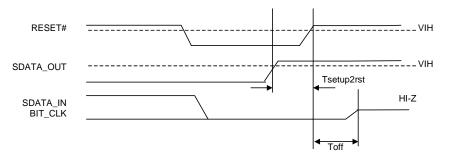


Figure 57. Activate Test Mode Timing

- Note 41. All AC-link signals are normally low through the trailing edge of RESETN. Bringing RESETN high for the rising edge of SDATAOUT causes the AK4650 AC-link outputs to go high impedance which is suitable for ATE in circuit testing. Note that the AK4650 enters in the ATE test mode regardless SYNC is high or low.
- Note 42. Once test modes have been entered, the only way to return to the normal operating state is to issue "cold reset" which issues RESETN with both SYNC and SDATAOUT "L".

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[AK4650]

Register Map

Reg Num	Name	D15	D14	D13	D12	D11	D10	D9	D8	D7	D6	D5	D4	D3	D2	D1	D0	Default
00H	Reset	0	0	0	0	0	0	0	0	0	0	1	1	0	0	0	0	0030H
02H	Speaker Output	SPPS	Х	Х	Х	Х	Х	Х	Х	Х	Х	Х	Х	Х	Х	Х	Х	8000H
04H	Headphone Output	HPMT	Х	Х	Х	Х	Х	Х	Х	Х	Х	Х	Х	Х	Х	Х	Х	8000H
06H	Mono Output	MOMT	Х	Х	Х	Х	Х	Х	Х	Х	Х	Х	Х	Х	MOGN2	MOGN1	MOGN0	8000H
08H	Boost Control	Х	Х	Х	Х	BST1	BST0	Х	Х	Х	Х	Х	Х	Х	Х	Х	Х	0000H
0AH	PC_BEEP Volume	BPMT	Х	Х	Х	Х	Х	Х	Х	Х	Х	Х	Х	Х	Х	Х	Х	8000H
0CH	Phone Volume	AUXMT	Х	Х	Х	Х	Х	Х	Х	Х	Х	Х	GN4	GN3	GN2	GN1	GN0	8008H
0EH	Mic Volume	MICMT	Х	Х	Х	Х	Х	Х	Х	Х	MGAIN	IPGA5	IPGA4	IPGA3	IPGA2	IPGA1	IPGA0	8008H
10H	Line In Volume	LNMT	Х	Х	GL4	GL3	GL2	GL1	GL0	Х	Х	Х	GR4	GR3	GR2	GR1	GR0	8808H
18H	PCM Out Volume	SMUTE	ATTL6	ATTL5	ATTL4	ATTL3	ATTL2	ATTL1	ATTL0	Х	ATTR6	ATTR5	ATTR4	ATTR3	ATTR2	ATTR1	ATTR0	8000H
1AH	Record Select	Х	Х	Х	Х	Х	Х	MICAD	AUXAD	Х	Х	Х	Х	Х	Х	Х	Х	0200H
20H	General Purpose	Х	Х	Х	Х	Х	Х	MDIF	MSEL	LOOP	Х	Х	Х	Х	Х	Х	Х	0000H
26H	Powerdown Ctrl/Stat	Х	PR6	PR5	PR4	PR3	PR2	PR1	PR0	Х	Х	Х	Х	0	ANL	DAC	ADC	030CH
28H	Extended Audio ID	0	0	Х	Х	0	0	0	0	0	0	0	0	0	0	0	1	0001H
2AH	Ext'd audio Stat/Ctrl	Х	Х	Х	Х	Х	Х	Х	Х	Х	Х	Х	Х	Х	Х	Х	VRA	0000H
2CH	PCM Front DAC Rate	SR15	SR14	SR13	SR12	SR11	SR10	SR9	SR8	SR7	SR6	SR5	SR4	SR3	SR2	SR1	SR0	BB80H
32H	PCM LR ADC Rate	SR15	SR14	SR13	SR12	SR11	SR10	SR9	SR8	SR7	SR6	SR5	SR4	SR3	SR2	SR1	SR0	BB80H
60H	Power Management	MPWRE	MPWRI	Х	Х	Х	Х	Х	MCKPD	PMSPK	PMHPR	PMHPL	PMBPM	PMLIN	PMMO	PMAUX	PMMIC	0000H
62H	Signal Select	HPM	Х	Х	Х	INBP	RNMD	LNMP	DAHS	AUXL	MICL	MICM	DAMO	BPMHP	BPMSP	ALCS	MO2	0103H
64H	ALC/DAC Control	Х	HPINT	REF5	REF4	REF3	REF2	REF1	REF0	DATTC	ATPU	ATSW	HPDT	TM1	TM0	DEM1	DEM0	2D21H
66H	ALC Mode Control	Х	ALC2	ALC1	ZELMN	LMAT1	LMAT0	RGAIN	LMTH	Х	ROTM	ZTM1	ZTM0	WTM1	WTM0	LTM1	LTM0	0000H
68H	Volume Cotrol	Х	Х	Х	Х	Х	Х	Х	Х	Х	Х	Х	ATS	ATTM	ATTS2	ATTS1	ATTS0	0002H
6AH	Detect Result	Х	Х	Х	Х	Х	Х	Х	Х	Х	Х	Х	Х	Х	DTPEN	DTHPJ	DTMIC	X
70H	12bit ADC Control 1	PM12AD	ADFLT	MSR1	MSR0	DLY3	DLY2	DLY1	DLY0	ADEXE 1	ADEXE 0	ADMO DE1	ADMO DE0	A2	A1	A0	SER	0001H
72H	12bit ADC Control 2	1	GINT	PENINT	PINTR	PMVRE F	PINTE	Х	Х	Х	Х	Х	Х	Х	SLOTN O1	SLOTN O0	SLOT	9400H
74H	ADC Data 1	ADE	A2	A1	A0	D11	D10	D9	D8	D7	D6	D5	D4	D3	D2	D1	D0	х
76H	ADC Data 2	ADE	A2	A1	A0	D11	D10	D9	D8	D7	D6	D5	D4	D3	D2	D1	D0	х
78H	ADC Data 3	ADE	A2	A1	A0	D11	D10	D9	D8	D7	D6	D5	D4	D3	D2	D1	D0	х
7CH	Vendor ID1	0	1	0	0	0	0	0	1	0	1	0	0	1	0	1	1	414BH
7EH	Vendor ID2	0	1	0	0	1	1	0	1	0	0	0	1	0	0	0	0	4D10H

Table 56. Register Map

Speaker Output (02H)

SPPS: Speaker-amp Power-Save-Mode (Table 38)
0: Normal Operation
1: Power Save Mode (Default)
When the SPPS bit = "1", the Speaker-amp is in power-save-mode and the SPP pin becomes Hi-Z and SPN pin is set to HVDD/2 voltage. When the PMSPK bit = "1", this bit is valid. After the RESETN pin changes from "L" to "H", the PMSPK bit is "0", which powers down Speaker-amp

■ Headphone Output (04H)

HPMT: Headphone Amp Mute Control (Figure 26) 0: Mute OFF 1: Mute ON (Default)

■ Mono Output (06H)

MOMT: Mono Output Mute Control (Table 30) 0: Mute OFF 1: Mute ON (Default)

MOGN2-0: MOUT Volume Control (Table 31) Default: "000" (+6dB)

■ Boost Control (08H)

BST1-0: Bass Boost Control (Table 24) Default: "00" (OFF)

■ BEEP Volume (0AH)

BPMT: BEEP Input Mute Control 0: Mute OFF 1: Mute ON (Default)

■ Phone Volume (0CH)

AUXMT: AUX Input Mute Control (Table 28) 0: Mute OFF 1: Mute ON (Default)

GN4-0: AUX Input Volume Control (Table 28) Default: "08H" (0dB)

■ MIC Volume (0EH)

MICMT: Mic Input Mute Control (Table 14) 0: Mute OFF 1: Mute ON (Default)

MGAIN: MIC-Amp Gain Control (Table 10) 0: 0dB (Default) 1: +20dB

IPGA5-0: IPGA Control (Table 14) Default: "08H" (0dB)

Line In Volume (10H)

LNMT: Line Input Mute Control (Table 29) 0: Mute OFF 1: Mute ON (Default)

GL4-0: Lch Line Input Volume Control (Table 29) Default: "08H" (0dB)

GR4-0: Rch Line Input Volume Control (Table 29) Default: "08H" (0dB)

■ PCM Volume (18H)

SMUTE: Soft Mute Control (Figure 21)
0: Normal Operation (Default)
1: DAC outputs soft-muted
Soft mute operation is independent of digital attenuator and is performed in the digital domain.

ATTL/R6-0: Digital ATT Control (Table 25) Default: "00H"(0dB)

Record Select Control Register (1AH)

AUXAD: AUXIN to ADC enable 0: OFF (Default) 1: ON

MICAD: IPGA to ADC enable 0: OFF 1: ON (Default)

■ General Purpose (20H)

LOOP: Internal Digital Loopback0: OFF (Default)1: ONWhen LOOP bit is "1", VRA bit should be "0".

MSEL: Internal/External MIC Select (Table 9 at MDIF bit = "0") 0: Internal MIC (Default) 1: External MIC

MDIF: Differential MIC Input Select (Table 9) 0: Single-ended Input (Default)

1: Differential Input

Power Management (26H)

PR6-0: Power Management (Table 6) Default: "0000011" (ADC, DAC Power down)

- REF: Internal VREF Power-up (Read only) 0: NOT Ready 1: Ready
- ANL: Analog Mixer Power-up (Read only) 0: NOT Ready 1: Ready
- DAC: DAC ready to accept data (Read only) 0: NOT Ready 1: Ready
- ADC: ADC ready to transmit data (Read only) 0: NOT Ready 1: Ready

Extended Audio Status & Control (2AH)

- VRA: Enables Variable Rate Audio mode in conjunction with Audio Sample Rate Control Registers and tag-bit/SLOTREQ signaling.
 - 0: OFF(Default). PLL is powered-down.

1: ON

■ Audio Sample Rate control Registers (2CH, 32H)

SR15-0: Sample Rate Control for DAC (2CH) and ADC (32H) (Table 4, Table 5) Default: "BB80H"(48kHz) These Sample Rate setting is done at VRA bit = "1".

Power Management (60H)

PMMIC: MIC Block (MIC-Amp and ALC1) Power Management 0: Power down (Default) 1: Power up PMAUX: AUX Input Power Management 0: Power down (Default) 1: Power up PMMO: Mono Line Output Power Management 0: Power down (Default) 1: Power up PMLIN: Stereo Line Input Power Management 0: Power down (Default) 1: Power up PMBPM: Mono Beep Input Power Management 0: Power down (Default) 1: Power up Even if PMBPM= "0", the path is still connected between BEEP pin and HP/SPK-Amp. BPMHP and BPMSP bits should be set to "0" to disconnect these paths, respectively. PMHPR: Headphone-Amp Rch Power Management 0: Power down (Default) 1: Power up PMHPL: Headphone-Amp Lch Power Management 0: Power down (Default) 1: Power up PMSPK: Speaker-Amp Power Management 0: Power down (Default) 1: Power up MCKPD: XTI pin pull down control 0: Master Clock input enable (Default) 1: XTI pin is internally pulled-down MPWRI: Internal MIC Power Supply Control (Table 11) 0: OFF (Default) 1: MIC Power is ON for Internal MIC. MPWRI bit is enabled when PMMIC bit = "1". MPWRE: External MIC Power Supply Control (Table 12) 0: OFF (Default) 1: MIC Power is ON for External MIC. MPWRE bit is enabled when PMMIC bit = "1".

Signal Select (62H)

MO2: Mono Output (MOUT2 pin) Enable 0: OFF 1: ON (Default) When MO2 bit = "0", MOUT2 pin outputs VCOM voltage. MOUT2 pin outputs signal when MO2 bit = "1" and PMSPK bit = "1". MOUT2 pin goes to Hi-Z state when PMSPK bit = "0". ALCS: ALC2 to Speaker-Amp Enable 0: OFF 1: ON (Default) **BPMSP: BEEP to Speaker-Amp Enable** 0: OFF (Default) 1: ON BPMHP: BEEP to Headphone-Amp Enable 0: OFF (Default) 1: ON DAMO: DAC to Mono Line Output Enable 0: OFF (Default) 1: ON MICM: IPGA to Mono Line Output Enable 0: OFF (Default) 1: ON MICL: IPGA to Headphone/Speaker-Amp Enable 0: OFF (Default) 1: ON AUXL: AUXIN to Headphone/Speaker-Amp Enable 0: OFF (Default) 1: ON DAHS: DAC to Headphone/Speaker-Amp Enable 0: OFF 1: ON (Default) LNMP: LIN/MPE pin Selection 0: MPE pin (Default) 1: LIN pin RNMD: RIN/MDT pin Selection 0: MDT pin (Default) 1: RIN pin INBP: IN2/BEEP pin Selection 0: BEEP pin (Default) 1: IN2 pin HPM: Mono Output Select of Headphone 0: Stereo (Default) 1: Mono [(L+R)/2]

■ ALC/DAC Control (64H)

- DEM1-0: De-emphases response (Table 23) Default: "01" (OFF)
- TM1-0: Soft Mute Time Select (Table 27) Default: "00" (1024/fs)
- HPDT: Headphone Jack Insertion Detection Function Enable 0: OFF (Default) 1: ON

1. UN

ATSW: Headphone/Speaker Automatic Switch Function Enable by Headphone Jack Insertion (Table 35, Table 36) 0: OFF

1: ON (Default)

ATPU: Headphone-Amp Automatic Power-up Function Enable by Pen Touch (Figure 39) 0: OFF (Default)

1: ON

- DATTC: DAC Digital Attenuator Control Mode Select
 - 0: Independent
 - 1: Dependent (Default)

When DATTC="1", ATTL6-0 bits control both Lch and Rch at the same time. ATTR6-0 bits are not changed when the ATTL6-0 bits are written.

- REF5-0: Maximum IPGA value at ALC1 Recovery Operation (Table 21)
 - Default: "2DH" (+19dB)

During the ALC1 recovery operation, if the IPGA value exceeds the setting maximum value (REF5-0 bits) by gain operation, then the IPGA does not become larger than the maximum value.

HPINT: INTN pin Output Enable for Headphone Jack Detection

Default: "0" (OFF)

When HPINT bit = "1", INTN pin is enabled to output the interrupt signal of headphone jack detection.

■ ALC Control (66H)

LTM1-0: ALC1 limiter operation period at zero crossing disable (ZELMN bit = "1") (Table 17)

Default: "00" (0.5/fs)

The IPGA value is changed immediately when zero crossing is disabled (ZELMN bit = "1"). When the IPGA value is changed continuously, the change is done by the period specified by the LTM1-0 bits.

WTM1-0: ALC1 Recovery Waiting Period (Table 19)

Default: "00" (128/fs)

WTM1-0 bits set a period of recovery operation when any limiter operation does not occur during the ALC1 operation.

ZTM1-0: ALC1 zero crossing timeout selection (Table 18)

Default: "00" (128/fs)

When the IPGA performs zero crossing or timeout, the IPGA value is changed by the μ P WRITE operation, ALC1 recovery operation or ALC1 limiter operation (ZELMN bit = "0").

ROTM: ALC2 Recovery Waiting Period (Table 39) 0: 2048/fs (Default)

1: 512/fs

LMTH: ALC1 Limiter Detection Level / Recovery Waiting Counter Reset Level (Table 15) Default: "0" (-6dB/-8dB)

The ALC1 limiter detection level and the ALC1 recovery counter reset level may be offset by about $\pm 2dB$.

RGAIN: ALC1 Recovery GAIN Step (Table 20)

Default: "0" (0.5dB)

During the ALC1 recovery operation, RGAIN bit sets the number of steps changed from the current IPGA value. For example, when the current IPGA value is "30H" and RGAIN bit is "1", the IPGA changes to "32H" by the ALC1 recovery operation and the output signal level is gained up by 1dB (=0.5dB x 2). When the IPGA value exceeds the maximum level (REF6-0 bits), the IPGA value does not increase.

LMAT1-0: ALC1 Limiter ATT Step (Table 16)

Default: "00" (0.5dB)

During the ALC1 limiter operation, when IPGA value exceeds the ALC1 limiter detection level set by LMTH bit, LMAT1-0 bits set the number of steps attenuated from the current IPGA value. For example, when the current IPGA value is "47H" and LMAT1-0 bits is "11", the IPGA value decreases to "43H" when the ALC1 limiter operation starts, resulting in the input signal level being attenuated by 2dB (=0.5dB x 4). When the attenuation value exceeds IPGA = "00" (-8dB), it clips to "00".

ZELMN: Zero crossing detection enable at ALC1 Limiter operation

0: Enable (Default)

1: Disable

When the ZELMN bit = "0", the IPGA performs a zero crossing or timeout and the IPGA value is changed by the ALC1 operation. The zero crossing timeout is the same as the ALC1 recovery operation. When the ZELMN bit = "1", the IPGA value is changed immediately.

ALC1: ALC1 enable

0: ALC1 Disable (Default) 1: ALC1 Enable

ALC2: ALC2 enable 0: ALC2 Disable (Default) 1: ALC2 Enable

■ Volume Control (68H)

ATTS2-0: Volume control of signal from IPGA to Headphone/Speaker-Amp (Table 7) Default: "2H" (-12dB)

ATTM: Volume control of signal from IPGA to Mono Line Output (Table 8) 0: 0dB (Default)

1: -4dB

ATS: Digital attenuator transition time setting (Table 26) Default: "0" (531/fs)

Detect Result (6AH)

DTMIC: MIC detection result (Read only, Table 13)

- 0: Microphone is not detected.
- 1: Microphone is detected

DTHPJ: Headphone jack insertion detection result (Read only, Table 34)

0: Headphone jack is not inserted.

1: Headphone jack is inserted.

DTPEN: Pen touch detection result when PM12AD bit = "0", PENINT bit = "1" and PINTR bit = "1" (Read only) 0: Pen is not touched.

1: Pen is touched.

■ 12bit ADC Control 1 (70H)

- A2-0, SER: 12bit ADC measurement item select (Table 44) Default: "0001" (TEMP0 temperature)
- ADMODE1-0: 12bit ADC measurement mode select (Table 48, Table 50) Default: "00" (74H Register Read)
- ADEXE1-0: ADEXE mode select (Table 52) Default: "00" (Mode 0)
- DLY3-0: Wait time select from mode setting to ADC execution (Table 46) Default: "0000" (1TS)
- MSR1-0: Measurement times select (Table 51) Default: "00" (4 times)
- ADFLT: 12bit ADC data filtering select (Table 47) 0: OFF (Default) 1: 4 data averaging
- PM12AD: 12bit ADC for touch screen controller power management (Table 6)0: Power Down (Default)1: Power Up

■ 12bit ADC Control 2 (72H)

SLOT: 12bit ADC data reception method select (Table 43)0: Register (Default)1: Slot
SLOTNO1-0: 12bit ADC measurement result output slot select (Table 49) Default: "00" (Slot 5)
PINTE: 12bit ADC for touch screen controller power management (Table 45) 0: Power Down (Default) 1: Power Up
PMVREF: Internal reference voltage for touch screen controller power management (Table 45)0: Power Down (Default)1: Power Up
PINTR: DTPEN bit output enable for pen interrupt Default: "1" (ON) When PINTR bit is "1", pen interrupt is output at DTPEN bit.
PENINT: INTN pin output enable for pen interrupt Default: "0" (OFF) When PENINT bit is "1", pen interrupt is output via INTN pin.
GINT: Slot output enable for pen interruptDefault: "0" (OFF)When GINT bit is "1" and AC-link operates, pen interrupt is output at bit 0 of slot 12 on SDATAIN.
SIWU: Warm reset by SDATAIN Enable 0: Disable 1: Enable (Default)
12bit ADC Data (744 764 784)

■ 12bit ADC Data (74H, 76H, 78H)

ADE: Measurement data valid flag (Read only) 0: Invalid

1: Valid

A2-0: Measurement item (Read only, Table 44)

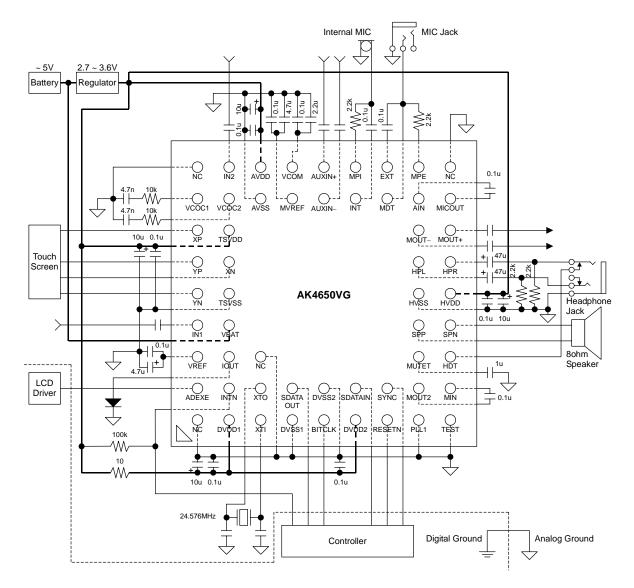
D11-0: 12bit ADC measurement data (Read only)

■ Vendor ID (7CH, 7EH)

"A(41H), K(4BH), M(4DH), 16(10H)" (Read only)

SYSTEM DESIGN

Figure 58 shows the system connection diagram for the AK4650. An evaluation board [AKD4650] is available which demonstrates the optimum layout, power supply arrangements and measurement results.



Note 43. AVSS, DVSS, HVSS and TSVSS of the AK4650 should be distributed separately from the ground of external controllers.

Note 44. Values of R and C in Figure 58 depend on each system.

Note 45. All input pins except for internal pull-down pins should not be left floating.

Figure 58. Typical Connection Diagram

1. Grounding and Power Supply Decoupling

The AK4650 requires careful attention to power supply and grounding arrangements. AVDD, DVDD, HVDD and TSVDD are usually supplied from the system's analog supply. If AVDD, DVDD and HVDD are supplied separately, the correct power up sequence should be observed. AVSS, DVSS, HVSS and TSVSS of the AK4650 should be connected to the analog ground plane. System analog ground and digital ground should be connected together near to where the supplies are brought onto the printed circuit board. Decoupling capacitors should be as near to the AK4538 as possible, with the small value ceramic capacitor being the nearest.

2. Voltage Reference

VCOM is a signal ground of this chip. A 2.2μ F electrolytic capacitor in parallel with a 0.1μ F ceramic capacitor attached to the VCOM pin eliminates the effects of high frequency noise. No load current may be drawn from the VCOM pin. All signals, especially clocks, should be kept away from the VREF and VCOM pins in order to avoid unwanted coupling into the AK4650.

3. Analog Inputs

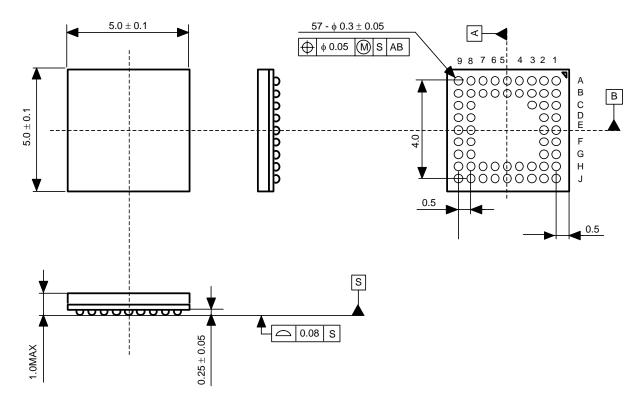
The Mic, Beep and stereo line inputs are single-ended. AUX input is differential. The input signal range scales with nominally at 0.06 x AVDD Vpp for the Mic input, 0.6 x AVDD Vpp for the Beep input, stereo line input and AUX input, centered around the internal common voltage (0.45 x AVDD). Usually the input signal is AC coupled using a capacitor. The cut-off frequency is $fc = (1/2\pi RC)$. The AK4650 can accept input voltages from AVSS to AVDD.

4. Analog Outputs

The input data format for the DAC is 2's complement. The output voltage is a positive full scale for 7FFFH(@16bit) and a negative full scale for 8000H(@16bit). Mono output from the MOUT2 pin and Mono Line Output from the MOUT+/MOUT- pins are centered at 0.45 x AVDD. Headphone-Amp and Speaker-Amp output are centered at HVDD/2.

PACKAGE

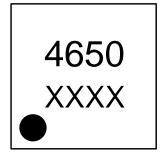
57pin BGA (Unit: mm)



■ Material & Lead finish

Package molding compound:	Epoxy
Interposer material:	BT resin
Solder ball material:	SnAgCu

MARKING



XXXX: Date code (4 digit) Pin #1 indication

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