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MSM6295

4-CHANNEL MIXING ADPCM VOICE SYNTEHSIS LSI

GENERAL DESCRIPTION

The Oki MSM6295 is a 4-channel mixing ADPCM voice synthesis LSI which is fabricated using Oki's low power CMOS silicon gate technology.

The MSM6295 can access an external ROM, where speech or sound effects data is stored. The maximum external ROM size is 256K

bytes.

The MSM6295 has a 4-channel synthesis stage which allows the simultaneous playback of four different channels. It is used to have a voice with BGM (Back Ground Music) effect, instrumental sound, echo etc.

FEATURES

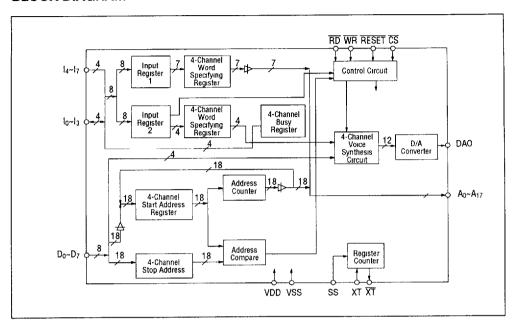
- · Oki straight ADPCM algorithm
- Number of bits/sample: 4
- 18 address lines for external ROM
- 8-bit control bus for mode setting
- External memory capacity 2Mbit
- Interface with common CPU and MPU
- Clock frequency: 1 MHz to 5 MHz.
- Sampling frequency:

6.5 kHz and 8 kHz (@1.056 MHz clock) 25.6 kHz and 32 kHz (@4.224 MHz clock)

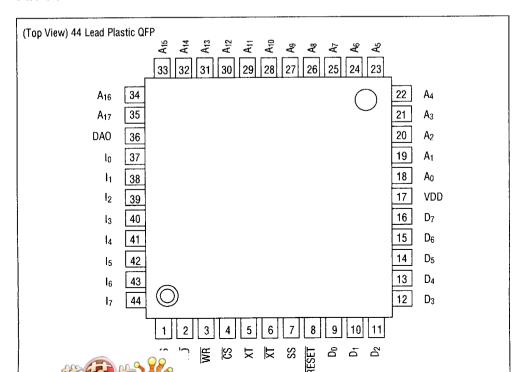
- Number of words: 127 maximum
- Vocalization time: 60 sec maximum (@8 kHz, straight)
- Built-in DA converter: 12-bit
- DA output format: A-class
- Voice level attenuation: 0dB ~ -24dB on each channel (9 steps) -3dB/step
- Low power CMOS process
- 5 V single power supply
- 44-pin plastic QFP (QFP44-P-910-V1K) (QFP44-P-910-K)
- 42-pin plastic DIP (DIP42-P-600)



BLOCK DIAGRAM



PIN CONFIGURATION



	(Top View) 42 Lead Pla	astic DIP
l ₃	1	42 l ₂
14	2	41 11
15	3	40 I ₀
16	4	39 DAO
I ₇	5	38 A ₁₆
VSS	6	37 A ₁₅
WR		36 A ₁₄
CS	8	35 A ₁₃
XT	9	34 A ₁₂
XT	10	33 A ₁₁
	11	32 A ₁₀
RESET	12	31 A ₉
D_0	13	30 A ₈
D ₁	15	29 A ₇
D_2 D_3	16	28 A ₆
	17	27 A ₅ 26 A ₄
	18	25 A ₃
	19	24 A ₂
	20	23 A ₁
	21	22 A ₀

ELECTRICAL CHARACTERISTICS

Absolute Maximum Ratings

Parameter	Symbol	Conditions	Value	Unit
Power supply voltage	VDD	Ta = 25°C	-0.3 ~ +7.0	٧
Input voltage	Vin	Ta = 25°C	-0.3 ~ VDD +0.3	٧
Storage temperature	T _{stg}	_	-55 ~ +150	°C

• Recommended Operating Conditions

Parameter	Symbol	Conditions	Value	Unit
Power supply voltage	VDD	VSS = 0V	4.5 ~ +5.5	V
Operating temperature	Top	VSS = 0V	-40 ~ +85	°C
Oscillation frequency	fosc	VSS = 0V	1 ~ 5	MHz

DC Characteristics

(VDD = 4.5~5 5V, VSS = 0V, Ta = -40 ~ +85°C)

_		0		Limits		Unit
Parameter	Symbol	Conditions	Min.	Тур.	Max.	Onit
"L" input current	11L	V _{IL} = VSS	-10			µА
"H" input current	Ин	V _{IH} = VDD	T —		10	μA
"L" input voltage	VIL		-		0.8	ν
"H" input voltage	V _{IH}		2.4		_	
"L" output voltage	VoL	$I_{OL} = 0.8 \text{ mA}$			0.45	ν
"H" output voltage	VoH	I _{OH} = -40 μA	3.7	_	-	\
Output leakage current	ILO	VSS ≤ V _{OUT} ≤ VDD	-10	-	10	μА
Operating current	I _{DD}	fosc = 5.0MHz	-	5	10	mA
DA output relative error	VDAE	No load	_	_	20	m∨
DA output impedance	R _{DAOUT}		_	15		kΩ

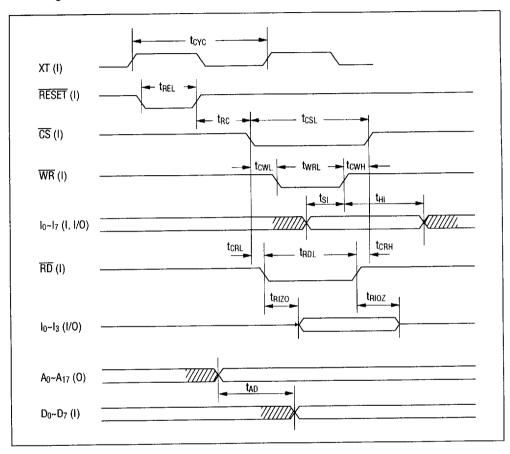


AC Characteristics

 $(VDD = 4.5 \sim 5.5V, VSS = 0V, Ta = -40 \sim +85 °C)$

Parameter	Symbol	Min.	Тур.	Max.	Unit
Clock cycle	tcyc	200	_	-	ns
Clock duty cycle	f _{DUTY}	40	50	60	%
RESET pulse width	t _{REL}	100	_	_	ns
CS pulse width	t _{CSL}	250	_	-	ns
WR pulse width	twar	200	_	_	ns
RD pulse width	trdl	300	_		ns
RESET fall to CS fall	t _{RC}	250	-	_	пѕ
CS fall to WR fall	tcwL	50	_	-	ns
WR raise to CS raise	tcwn	0	_	_	ns
Data set up time of I ₀ -I ₇ in respect to WR raise	t _{SI}	80	-	_	ns
Data hold time of I ₀ -I ₇ in respect to WR raise	t _{HI}	80	_	-	ns
RD fall to stable output of I ₀ -I ₃	t _{RIZO}		_	120	ns
RD raise to flow status output of I ₀ -I ₃	t _{RIOZ}	0	_	120	ns
CS fall to RD fall	tcrl	20	_	_	ns
RD raise to CS raise	tcRH	0	_	_	ns
Address stable (A ₀ -A ₁₇) to data input of D ₀ -D ₇	t _{AD}	_	_	5•t _{CYC} +90	ns

• Timing Chart



PIN DESCRIPTION

Pin Name	Pin No.	1/0	Function	n			
l ₀	37	1/0	Instruction bus and condition output	S			
l ₁	38	1/0	These pins are inputs for phrase specification Maximum number				
12	39	1/0	of phrases is 127 I ₀ ~I ₃ pins are also outputs of the operating				
13	40	1/0	state, busy state, for channels 1~4 ar	state, busy state, for channels 1~4 and are further used to select the			
14	41	1	channel attenuation rate.				
l ₅	42	1					
16	43	1					
17	44	1					
WR	3	1	Write enable input		* +**		
			Data is written on the data bus of I ₀ ~	l ₇ The data is writ	ten when WR		
			goes low				
RD	2		Read enable input	*******			
			The output busy state of channels 1~	4 on the data bus	of I ₀ ~I ₃ Can		
			be read using this input. A high level	indicates busy			
CS	4	l	Chip select input		=		
			Input "L" level either when WR signal	is input or when F	D signal is		
			input				
RESET	8	I	Reset input				
			Reset condition is available by inputt	ing "L" level			
			All functions are suspended during re	eset			
A ₀	18	1	Address outputs		,		
1	1	t	These pins are to address the externa	al ROM in which vo	oice data		
A ₁₇	35		ıs stored.				
D_0	9	ı	Voice data inputs				
ì	1	t					
D ₇	16	1					
SS	7		Sampling frequency select input				
			When oscillation frequency is 1.056	MHz or 4 224 MHz	, the following		
			choices are available by inputting "H"	level or "L" level to	SS.		
				SS = "H"	SS = "L"		
			Oscillation frequency 1.056 MHz	8 kHz	6 5 kHz		
			Oscillation frequency 4.224 MHz	32 kHz	25 6 kHz		
DAO	36	0	Voice synthesis output				
			Voice synthesized analog signal is or	tput from this pin			
XT	5	ı	Crystal oscillator pin				
XT	6	0	Crystal oscillator pin				
VDD	17	_	Power supply pin				

FUNCTION EXPLANATION

1. Phrase Selection

Phrases are specified and read into the 2 byte data which consists of $I_0 \sim I_7$ data bus.

The phrase selection data is latched when \overline{WR} goes high while \overline{CS} is low (L).

The format of the phrase specification input is as follows.

	I ₇	l _a	l ₅	I ₄	l ₃	l ₂	I ₄	I _o
1st Byte	1			Phrase sele	ection data			
2nd Byte		Channel sp	ecification			Reduction	specification	

As shown in the above chart, I_7 of the first 1 data byte is always 1. $I_0 \sim I_6$ of the first 1 data byte specifies the phrase. Phrase selection data has a selection of 127 phrases which corresponds to $0000001 \sim 11111111$. The

phrase selection data is used for to $A_3 \sim A_9$ address outputs, and they specify both start and stop address which are stored in the external ROM.

Relation between Phrase Selection Data and ROM Address

Phrase selection data	-	I _e	l _s	I ₄	l ₃	I ₂	I ₁	l _o	-	-	-
External ROM address	A ₁₇ ~A ₁₀	A ₉	A ₈	A ₇	A ₆	A ₅	A ₄	A ₃	A ₂	A,	A _o
Selection Not valid Phrase 1	0~0 0~0	0	0	0	0 0	0	0	0	0	0	0
Phrase 2 Phrase 3	0~0 0~0	0 0	0	0	0 0	0	1	0	0	0	0
Phrase 127	0~	1	1	1	1	1	1	1	0	0	0

^{*} Phrases cannot be specified with all inputs = "0"

The second byte of data specifies the synthesis operation channel as well as specific channel reduction of the synthesized playback. The channel selection format is shown below.

It is not possible to specify multiple channels at the same time. For example, it is not possible to specify channel 1 and channel 3 simultaneously.

Channel Specification

Channel	l ₇	l _e	l ₅	l ₄
1	0	0	0	1
2	0	0	1	0
3	0	1	0	0
4	1	0	0	0

REDUCTION SELECTION

All 0's is considered as 0 dB of the analyzed sound itself. The reduction is made through

9 levels from about 0 dB to - 24 dB with the steps of about - 3 dB. Reduction format is shown below.

Reduction Specification

Attenuation level	I ₃	l ₂	I,	l _o
0 dB	0	0	0	0
-3.2 dB	0	0	0	1
-6.0 dB	0	0	1	0
−9.2 dB	0	0	1	1
-12.0 dB	0	1	0	0
-14.5 dB	0	11	0	1
-18 0 dB	0	1	1	0
-20 5 dB	0	1	1	1
-24.0 dB	1	0	0	0



2. Voice Synthesis Channel Suspension

Voice synthesis operation of any channel can be suspended. Channel suspension is controlled by bits $I_3 \sim I_6$ of data bytes $I_0 \sim I_7$. To suspend a channel, make $I_7 = 0$, while $I_3 \sim I_6$ represent the channels which should be sus-

pended.

Channel suspension occurs even if multiple channels are selected. For example, if $I_3 \sim I_6$ are all 1 and $I_7 = 0$, then channels 1~4 are suspended simultaneously.

Suspended channel	l,	l _e	I _s	I ₄	l ₃	l ₂	l,	I _o
1	0	0	0	0	1	Х	Х	Х
2	0	0	0	1	0	Х	Х	Х
3	0	0	1	0	0	Χ	Х	Х
4	0	1	0	0	0	Х	Χ	Х

3. Data ROM

1) ADDRESS DATA

This specifies start and stop address of ADPCM speech data. One phrase start and end address consists of 8 bytes. The first 3

bytes show start address while the last 3 bytes show stop address. The other 2 bytes are empty.

By selecting the first address in which the start address is stored, the selected speech data is played back.

Address 0	SA,
1	SA ₂
2	SA ₃
3	EA,
4	EA ₂
5	EA ₃
6	EMPTY
7	EMPTY

Start addresses $(SA_1 \sim SA_3)$ and stop addresses $(EA_1 \sim EA_3)$ are stored according to the chart

shown below.

SA ₁ /EA ₁
SA ₂ /EA ₂
SA ₃ /EA ₃

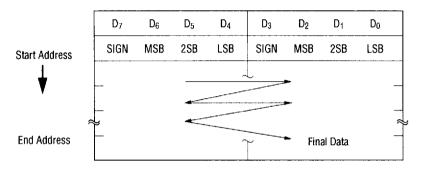
D_7	D_6	D_5	D_4	D_3	D_2	D_1	D _o
0	0	0	0	0	0	A ₁₇	A ₁₆
A ₁₅	A ₁₄	A ₁₃	A ₁₂	A ₁₁	A ₁₀	A ₉	A ₈
A ₇	A ₆	A ₅	A ₄	A ₃	A ₂	Α,	A _o

2) ADPCM SPEECH DATA

ADPCM speech data consists of (2) 4-bit samples. So, 1 byte stores 2 samples. Data arrangement proceeds from higher rank bits $(D_4 \sim D_7)$ to lower rank bits $(D_0 \sim D_3)$. The storage of speech data should always be

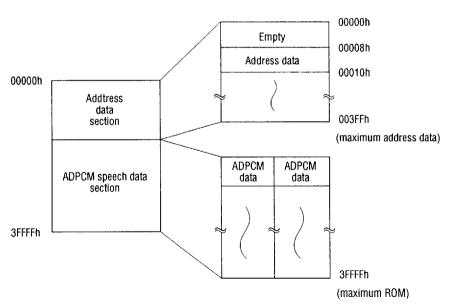
ended with the lower rank bit. So, always store an even number of samples.

Speech data is produced by Speech Development Tool AR76-202.



3) DATA ROM STRUCTURE

The following chart shows the memory map of the source data ROM.



When the maximum 127 phrases are selected in address data section, the data is written up to ROM address 003FFh.

When 1 phrase is selected, address data is written from ROM address 00008h to 0000Fh, and the rest is used as the ADPCM data section.



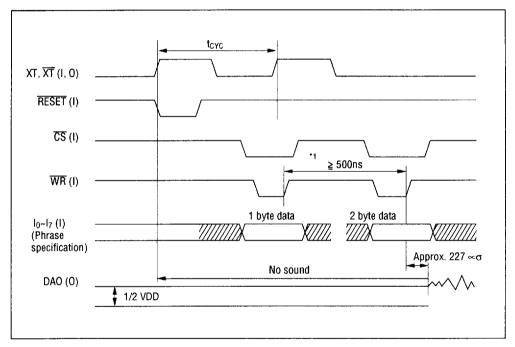
FUNCTIONAL DESCRIPTION

1. Phrase Selection Input

This procedure is to input phrase selection data onto the data bus inputs $I_0 \sim I_7$. The data is latched internally when \overline{WR} rises from "L"

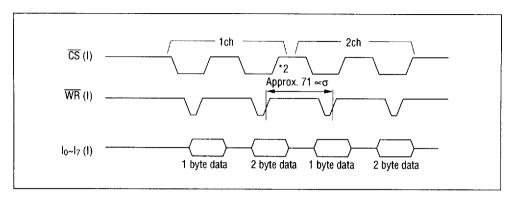
to "H", while \overline{CS} remains "L".

Voice synthesis operation does not start till the second byte is fully latched.



Note: Phrase selection is from channel 1 to channel 4 continuously.

*1 An interval of 75 T_{CYC} (max.) is needed between phrases.

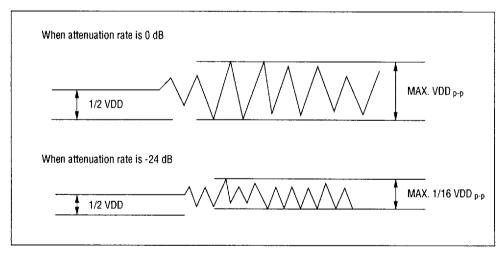


Note: *2 Oscillation frequency = 1 056 MHz SS = "L"

Voice synthesis playback can be started from any channel, 1 to 4. The arrangement of each channel can be in any order.

The second byte of the phrase selection data contains the phrase attenuation data in bits D0 - D3. Synthesized data is attenuated in -3 dB steps from 0 dB to -24 dB.

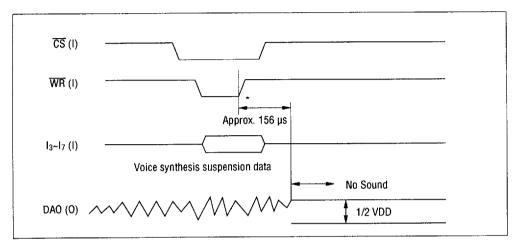
2. Attenuation of Synthesized Speech



3. Speech Synthesis Channel Suspension

This is accomplished by writting synthesis channel suspension data onto data bus inputs $I_3\sim I_7$. The data is latched internally when $\overline{\rm WR}$ goes from "L" to "H" while $\overline{\rm CS}$

remains active (L). Since synthesis suspension data is 1 byte data, synthesis operation is suspended right after the rising edge of WR. Multiple channels can be specified, making it possible to suspend channels 1~4 simultaneously.

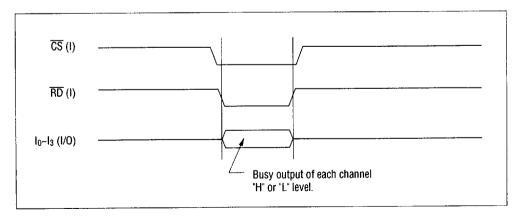


Note: * Oscillation frequency = 1.056 MHz SS = "L"

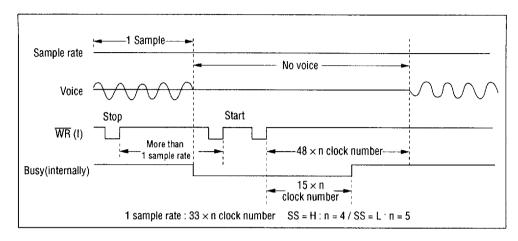
4. Reading the Busy Status

While \overline{CS} is "L" and \overline{RD} is "L", each operation

state, the busy state of channels 1~4 is output on $I_0~I_3$. "H" is output during synthesized playback.



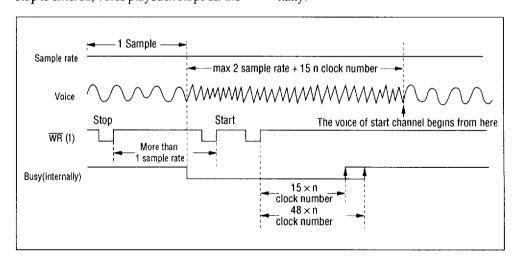
5. Start and Stop of 1 Channel



Start and Stop of Signal Channel

When a single channel (either of channels 1-4) starts again after it has stopped, the first write for start must be input with a delay of more than one sample rate from the stop write as shown in the figure above. When stop is entered, voice playback stops all the next sample and BUSY becomes "L".

When start is entered again, voice is output after 48 x n clock from the second byte write. BUSY becomes "H" after 15 x n clock internally.



Start and Stop in Plural Channels

When channels are operating, the first byte write for start must be input with a delay of more than one sample rate from stop write.

and the same

every sample.

The channel where stop was input, stops at

input is output after a maximum 2 samples = 15 x n clocks from the preceding sample point.

BUSY becomes "H" during the 48 x n clock time.

6. Application Circuit

