#### **General overview**

The MP3 cartridge contains an AtMega644 processor together with a VS1011 MP3 decoder. All registers of the decoder are shadowed by the AtMega and can be read by the MSX without additional wait states. After the MSX reads a register from the shadow memory, the AtMega reads the (new) contents of this register from the VS1011. If the shadow would be updated continiously, VS1011 registers which have auto incremented pointers will give unpredictable data. This update after read will give one byte delay when using user code or reading memory from a VS1011E.

All these registers are as described in the VS1011 datasheet, expect for 6 additional bits. These bits are not used by the VS1011 and are now used to switch between audio chip, enable interrupts and audio buffer warning level.

&H21 Any write to this address will reset the cartridge

- &H22 This is the data port for registers read/write and audio write
- &H23 Mode port:
  - 7: When this bit it set in register mode, all written data is send to the register written in bits 0-4
  - 6: 1=Switch to register mode \* 0=Switch to audio mode \*
  - 5: When set, auto increment is disabled when writing registers
  - 4-0: Address of register to access

&H24 I2C Status, for reading back I2C errors, wait for finishing of last action

- &H25 I2C Data, used for writing slave addresses and data
- &H26 I2C Own address, not used in this application
- &H27 I2C Control, for generating start/stop conditions

\* Always read out the busy flag after an update to &H23

The I2C part is a completely separated part of the cartridge, this will be described in detail later on. There are two analog audio chips which both have their own audio registers, and of course the I2C controller itself which is directly controlled by the I/O addresses has registers as well.

To prevent any confusions, the AtMega (and VS1011) registers will be called 'MP3 registers'. The audio chip registers 'Audio registers' and the I2C controller 'I2C registers'.

Up to chapter 'Audio settings and AUX input' the document only describes the MP3 registers, IO &H22 and &H23.

							N	<u>1</u> P	<b>3</b>	R	eg	ist	er	ď	es	cr	ip	tio	on			
0	InvLeft	StVol				Chans			MP3Emph	MP3Prot												
2	Keep0	55 L	BassFrq							MP3Lay												
	' Reset	StPD1	Ba						MP3CRgt MP3Copy	MP												
e	StopWAV	StPD2							MP3CRgt	MP3ID		VolRight									ossible.	
4	Keep0								MP3Ext	MP		0/								011	liodata imp	
5	TEST	StVer	BassEnh						MP											to the VS10	erring of auc	
9	Stream		Bas						MP3Mode											er sontent :	and VS1011. Changing thee bits will make the transferring of audiodata impossible.	
7	Keep0			CIkFreg	Playtime		RAMData	RAMAddress	MP3		UserSA		UserCtrID UserCtrI1 UserCtrI2	JserCtrl3				g the regist	s will make			
00	SCIK				Play	SmpFrq			MP3Priv					Use	Use Use		MSX user		hen sendin	ing thee bit		
<b>б</b>	SDat	Buffer							MP3Pad									possible to modify by MSX user		< the bits w	011. Chang	
10	SShare	ā							MP3SRate	MP3Header								possible to	datasheet	ga will masl	a and VS10	
11	SNew								MP3			Left						full and are	ne VS1011	only. AtMeg	'een AtMeg	
12	Keep0											VolLeft						VS1011 bits which might be usefull and are	Should be kept 0, according to the VS1011 (	Additional bits used by AtMega only. AtMega will mask the bits when sending the register sontent to the VS1011	Settings for communication between AtMega	
13	Keep0									MP3BRate									ts which mi	kept 0, acc	bits used b	ir communi
14	<b>INTEnable</b>								MP3E									VS1011 bit	Should be	Additional	Settings fo	
15	SetAUX			CIKD	е			r														
Register:	0 Mode	1 Status	2 Bass	3 ClockF	4 DecodeTime	5 AuData	6 WRAM	7 WRAMAdd	8 HDATO	9 HDAT1	10 AlAddr	11 Volume	12 AICTRLD	13 AICTRL1	14 AICTRL2	15 AICTRL3						

<mark>SetAUX</mark> INTEnable SNew SShare SDat SClk	<ul> <li>When set, audio settings written by I2C will affect the AUX input.</li> <li>When set, an interrupt will be generated when buffer level is reached.</li> <li>SDI bus. Audio data and register data are send trough the same bus.</li> <li>SDI bus. One chip select is used.</li> <li>SDI bus. When set, data should be LSB first</li> <li>SDI bus. When this bit is set, data is read at falling edge.</li> <li>The SDI and SCI settings are affecting communication between</li> <li>AtMega and VS1011. Changing these bits will make further</li> </ul>
Stream	communication impossible. Set for stream mode, playback speed is adjusted depending on buffer. It is not supported by the AtMega.
TEST	When set, some test functions of the VS1011 can be executed
<b>StopWAV</b>	Set this bit to terminate playback of a way file.
Reset	Setting this bit will reset the VS1011 player.
InvLeft	When set, audio of left channel is inverted (180deg phase
shifted).	
Buffer	In these 4 bits the buffer warning level is specified.
StVer	(Read only) VS1011 device version.
StPD2	Power down VS1011 output, can be used to avoid transient at reset.
StPD1	Power down VS1011 analog.
StVOL	Ouput 0=0dB, 1=-6dB, 3=-12dB. This is not meant for volume.
	St bits are meant for power down, not useful for MSX.
<mark>BassEnh</mark>	Bass enhancement, active when BassFrq>0.
<mark>BassFrq</mark>	Lowest frequency the audio system can reproduce.
ClkD	Clock doubler. When set, the crystal frequency is doubled.
<mark>ClkFreq</mark>	Crystal/2000 → &H1800 by default.
	For pitch bending the formula 100*&H1800 / Pitch can be used.
<u>Playtime</u>	Setting the playtime to 0 should be done by writing 0 two times.
SmpFrq	Current samplefrequency, can be changed while playing.
<mark>Chans</mark>	The number of channels used. 0=Mono, 1=Stereo.
RAMData	Data for VS1011 RAM,, written to address specified in RAMAddress
RAMAddress	VS1011 RAM address pointer for data write.
MP3***	HDAT is the MP3 package header, found by the decoder, read only.
UserSA	Activate or deactivate user code by writing its start address or 0 here.
VolLeft	Volume left channel, this value times -0.5dB attenuation.
VolRight	Volume right channel, this value times -0.5dB attenuation.
UserCtrl*	It is recommended to adjust volume by the I2C audio chips instead. These 4 registers are settings and feedback for the uploaded user code.

The highlighted bits can be useful for MSX applications.

#### **MP3** packages

Version2

Lay2 & Lay3

Free

Bad

MP3 files are divided into packages, a package is recognized by the 11 ones at the beginning. For more detailed information, refer to ISO11172-3.

MP3Header	All set to one.
MP3ID	00 – MPEG Version 2.5 (unofficial)
	01 – reserved
	10 – MPEG Version 2 (ISO/IEC 13818-3)
	11 – MPEG Version 1 (ISO/IEC 11172-3)
MP3Lay	00 – reserved
-	01 – Layer III
	10 – Layer II
	11 – Layer I
MP3Prot	When set, the file is protected by CRC

MP3BRate

#### Version1 Version1 Version1 Version2 Bits Lay1 Lay2 Lay3 Lay1 Free free free free

bad

bad

bad

Bad

MP3SRate

	bits MPEG1 MPEG2 MPEG2.5					
	00 44100 22050 11025					
	01 48000 24000 12000					
	10 32000 16000 8000					
	11 reserv. reserv.					
MP3Pad	When set, the frame is padded					
MP3Priv	Free to use bit for MP3 encoders					
MP3Mode	00 – Stereo					
	01 – Joint stereo (Stereo)					
	10 – Dual channel (2 mono channels)					
	11 – Single channel (Mono)					
MP3Ext	In joint-stereo mode, these bits indicate the real stereo subbands.					
MP3CRgt	When set, the file is copyrighted.					
MP3Copy	When set, the file is the original.					
1 2						
MP3Emph	00 – No emphasis					

01 – 50/15 microseconds emphasis 10 – Reserved 11 – CCITT J17

#### Initialisation

After power up the AtMega processor enters a bootloader program which gives the possibility of flashing the main program. The bootloader part is protected at production to prevent the boot part to be damaged.

For normal operation, first write a zero to &H22... Or any value different from &HAA. In this case the AtMega will execute the normal program.

#### Accessing the cartridge

Data is send and read through port &H22, to distinguish between data and MP3 registers access to &H23 is used.

After a write action to &H23, the programmer should wait a while before sending new data. In case of changing register read pointer the wait time will be around 300us roughly. When updating registers with data send to &H22 previously, the wait time can be up to 8 milliseconds. The AtMega is not allowed to send MP3 register data in between audio data.

### Writing MP3 registers

Writing only register 3:

LD A, &H40 ;Go to register mode OUT (&H23),A CALL ATWAIT OUT (&H22),A ;High byte OUT (&H22),A ;Low byte LD A, &HC3 ;Update registers, offset is register 3 OUT (&H23),A CALL ATWAIT ;This wait can be milliseconds ATWAIT: IN A, (&H23) AND A, 16 JR NZ,ATWAIT RET

Writing **9** registers, starting from **0**:

	LD OUT CALL	A,&H40 (&H23),A ATWAIT	;Go to register mode
	LD LD OTIR	hl , regtab BC , &h <mark>12</mark> 22	;9 registers = 18 bytes
	LD OUT CALL	A,&HC <mark>O</mark> (&H23),A ATWAIT	;Update registers, offset is register 0 ;This wait can be milliseconds
ATWAIT:	IN AND JR RET	A,(&H23) A,16 NZ,ATWAIT	

# Reading MP3 registers

Reading register 4:

LD OUT CALL	A,&HO <mark>4</mark> (&H23),A ATWAIT	;Can stay in audio mode, only update pointer
IN	A,(&H22)	;High byte
IN	A,(&H22)	;Low byte

Reading 16 registers, starting from 0:

LD OUT CALL	A,&H0 <mark>3</mark> (&H23),A ATWAIT	;Can stay in audio mode, only update pointer
LD LD INIR	hl , regtab bc , &h <mark>20</mark> 22	;16 registers = 32 bytes

#### Sending audio data

Audio data is send as raw MP3 of WAV data.

While playing audio, the buffer level will decrease, the cartridge sets bit 7 of IO &H23 when the audio buffer reaches its minimum level programmed by bits 8-11 of MP3 register 1. This bit will be cleared when the audio buffer is filled again to an amount of the minimum level plus 512 bytes.

It is possible to let the cartridge generate an interrupt by setting bit 14 of MP3 register 0, this interrupt will be generated once until the hardware detects an interrupt acknowledge from the Z80, or after a write to &H22. The interrupt service routine can recognize the interrupt also by checking bit 7 of IO &H23.

When the minimum buffer level at bits 8-11 of register 1 is set to 0, the warning bit is never set. When set to 1, the warning bit will be set when the buffer level is smaller than 256 bytes and reset when the buffer level is higher than 768.

The total buffer size is 3584 bytes. To choose the right buffer warning level, two values are important. The first one is the *number of bytes* that will be send each time and the second is the *period between two transfers*.

One clear rule is the buffer may not be overfilled because data will be lost. So when sending 1024 bytes each time, the highest possible buffer warning level will be 3584-1024=2560 bytes. When the warning level is set to a higher value e.g. 2816 and 1024 bytes will be send when bit 7 of &H23 is set, the buffer will overflow, 2816+1024 is more than the buffer size.

Second, if the time between transmitting data is high, the buffer warning should also be high. The exact time of the interval allowed depends on the bitrate.

In general, a higher warning level gives more safety. If the MSX for some reason needs more time between two transmissions, there is enough data left in the buffer to continue playing. It is recommended to send packages with a fixed time interval to achieve a known time shift between currently sent package and actual played part of the sound.

The cartridge always starts playing the audio data immediately after receiving the first complete MP3 package.

#### Status register

The information of currently read register, busy and data request can be checked by reading back I/O &H23.

&H23 Mode port:

- 7: When this bit it set, the programmed minimum buffer level is reached Note: This bit will be reset some milliseconds after filling the buffer, don't check again too fast.
- 6: Not used, floating internally, might be 1 or 0
- 5: Not used, floating internally, might be 1 or 0
- 4: Busy. This should be checked after updating registers.
- 3-0: Address of register which data is available at I/O &H22 for reading

#### Writing VS1011 memory

For uploading and controlling user code, access to memory is needed. Writing is simply done by register 6 and 7. The first RAM address is written to register 7 and the data to 6. See next chapter.

The implementation of the AtMega for writing memory is only passing on data. One advantage, an amount of data smaller than 256 bytes meant for one register (&H06), can be written to &H22 at one time. Then send &HE6 to I/O register &H23. to send all data in one burst.

### **Reading VS1011 memory**

For there is a difference between VS1011E and VS1011B devices. For VS1011E it is always possible to read memory at any time, the VS1011B does not support reading by default. The amplitude information of the spectrum analyzer requires memory read. Therefore the spectrum analyzer code is slightly different for these versions, in the VS1011B code the software creates a read possibility through register &H0F while the VS1011E uses the default register &H06.

The AtMega will read out 64 bytes of the VS1011 memory, starting at the last address written to register &H07. By default this should be part of the memory where the spectrum analyzer data is available because this is the most common application, for other future applications this value can be changed to another address.

#### VS1011 User code

The VS1011 has the possibility to execute small programs (plug-ins) written for a VS\_DSP core. There are a number of plugins available at <u>www.vlsi.fi</u>, also a development kit with a C compiler is available. It is a small program which stays available in the VS1011 memory until reset or powerdown.

After compiling several files are created, the .bin files are not just binary but do also contain labels. A .cmd file can be used to upload to the VS1011. This is an example of a .cmd file:

The 3<sup>rd</sup> column is the MP3 register to write to, the last column the data, high byte first.

Register 7 is the address in RAM to write to, register 6 is the data and will de auto incremented by the VS1011. After uploading the code can be executed by writing the start address to register 10. Writing a 0 to register 10 will disable to code execution.

After a hard reset or write to I/O &H21, the spectrum analyzer is uploaded to the VS1011 by the AtMega as default user code.

To optimize the uploading (other) user code on MSX, after writing register 7 all data for register 6 can be send to the cartridge in packages of 128 words. When sending the 'update registers', set bit 5 to disable auto increment of the AtMega. In this case all data in the AtMega buffer will be sent to register 6 only.

Uploading the existing spectrum analyzer code by MSX, which has a size of 987 words, takes about 0.2 seconds in Z80 mode without this optimization.

#### Spectrum analyzer

At the moment of developing the cartridge there are two plug-ins available. A loudness and the spectrum analyzer. The loudness function works without any exceptions. For the spectrum analyzer reading back of registers is required, when using the VS1011B all spectrum data is available at register 15, this is covered by the AtMega. Since this is the only plug-in with data feedback, the AtMega software is tuned for this plug-in. The spectrum analyzer is automatically uploaded to the VS1011 at start up, any other user code tools can be written by MSX and will simply overwrite the spectrum analyzer.

To activate the spectrum analyzer, write its start address &H0050 to register &H0A.

When the plug in is active, the AtMega won't read out register 10 to update it's shadow memory. It seems to disturb the VS1011. All spectrum data of the VS1011 is read by the AtMega and stored in a different shadow memory behind the register shadow as described in chapter 'Read VS1011 memory'. Register 7 should be &H1384, also the startup value, because this is the first address where the spectrum data is available. The spectrum data can be read out by simply reading further than register 15 using an INIR or separate IN instructions.

All odd bytes are peak values, even bytes are actual values, the first pair of bytes is the lowest frequency and the last pair is the highest frequency. By default 14 frequency bands are selected, the mirror is updated every 20ms.

Reading  $\frac{16}{16}$  registers, starting from  $\frac{0}{16}$  including spectrum data:

LD OUT CALL	A,&HO <mark>O</mark> (&H23),A ATWAIT	;Can stay in audio mode, only update pointer (XOR A)
LD LD INIR	hl , regtab bc , &h <mark>40</mark> 22	;16 registers = 32 bytes, 32 bytes of spectrum

Reading only spectrum data, start at register 15 and read e.g. 28 bytes of spectrum:

LD OUT CALL	A,&HO <mark>F</mark> (&H23),A ATWAIT	;Can stay in audio mode, only update pointer (XOR A)
LD LD INIR	HL, REGTAB BC, &H <mark>1E</mark> 22	;16 registers = 32 bytes, 32 bytes of spectrum

### Audio settings and AUX input

Audio settings can be adjusted by two TDA8425 devices, they are controlled through a I2C controller PDA9564D. Audio settings are:

- Volume R\*
- Volume L\*
- Bass
- Treble
- Pseudo stereo
- Spatial stereo
- Forced mono / one channel

\*Example code is included to create balance by software

The first audio chip is used for controlling the MP3 music, the second audio chip is connected to an auxiliary input and is mixed with the MP3 audio. Since both TDA8425 devices have the same I2C slave addresses, switching between the chips is done by bit 15 of MP3 register 0. This is one of the unused VS1011 bits. When this bit is set, the AUX audio chip is enabled.

Sending all settings for one audio chip using I2C takes 1.5ms for Z80 and 0.8ms in R800 mode. Changing only one register of an audio chip will take 0.4ms for an MSX2.

I2C

The IO addresses &H24 to &H27 controlled by the MSX are the I2C registers of the PCA9564D. The audio chips are accessed through this controller, according to the I2C protocol, a number of actions must be taken to change one single audio register.

Writing data over the two wire I2C bus is done by writing the data to I2C register &H25, poll I2C register &H24 to check for errors or finishing data transfer. Initiating and closing a data transmission is done via I2C register &H27.

TDAINI: WAIT:	OUT LD		;After power-up initialize oscillator ; communication is 88kHz
	LD OUT LD CALL JP LD	C,I2CERR A,&H82	<pre>;Initiate start condition ; For I2C, all communication starts with a start ; condition. The initiator (master) keeps SDA (Data) ; low, just before it keeps SCL (Clock) low. ;Wait for code 8 in I2C status register ;In case of collisions or shorted lines, exit ;Load audio chip's slave address in buffer</pre>
	OUT LD OUT LD CALL JP	(&H25),A A,&H44 (&H27),A C,&H18 STAT C,I2CERR	;Send buffer data. ; All devices on a I2C bus have their own address. ; just like the address bus in the MSX, peripherals ; can see for which one the data is meant. ;In case of no acknowledge, exit
	XOR OUT LD OUT LD CALL JP	A (&H25),A A,&H44 (&H27),A C,&H28 STAT C,I2CERR	;Load audio register number in buffer (number 0) ;Send buffer data ; After the of the device address, the address of ; the register of the device to write to is send. ;In case of no acknowledge, exit
	LD INC OUT	A,(HL) HL (&H25),A	;Load data for audio register <mark>0</mark> in buffer
	LD OUT LD CALL	A,&H44 (&H27),A C,&H28 STAT	;Send buffer data
	JP	C,I2CERR	;In case of no acknowledge, exit
	LD INC OUT LD OUT LD CALL	A, (HL) HL (&H25),A A,&H44 (&H27),A C,&H28 STAT	;Load data for audio register 1 in buffer ; I2C devices are usually auto incrementing the register ; number
	JP	C,I2CERR	;In case of no acknowledge, exit
	LD INC OUT	A,(HL) HL (&H25),A	;Load data for audio register 2 in buffer
	LD OUT LD CALL	A,&H44 (&H27),A C,&H28 STAT	;Send buffer data
	JP	C,I2CERR	;In case of no acknowledge, exit

	LD INC	A, (HL) HL	;Load data for audio register 3 in buffer
	OUT LD OUT LD	(&H25),A A,&H44 (&H27),A C,&H28 STAT	;Send buffer data
	CALL JP	C,I2CERR	;In case of no acknowledge, exit
	LD OUT LD CALL JP	A,&H64 (&H27),A C,&H10 STAT C,I2CERR	;Generate (repeated) start condition ; After reinitializing the bus, slave address ; and register must be send again. Now can we ; only send 2 bytes to reach register 8 ;In case of no acknowledge, exit
	LD OUT	A,&H82 (&H25),A	;Load audio chip's slave address in buffer
	LD OUT LD CALL	A,&H44 (&H27),A C,&H18 STAT	;Send buffer data
	JP	C,I2CERR	;In case of no acknowledge, exit
	LD OUT	A,&H08 (&H25),A	;Load audio register number in buffer (number <mark>8</mark> )
	LD OUT LD CALL	A,&H44 (&H27),A C,&H28 STAT	;Send buffer data
	JP	C,I2CERR	;In case of no acknowledge, exit
	LD INC OUT LD	A,(HL) HL (&H25),A A,&H44	;Load data for audio register <mark>8</mark> in buffer ;Send buffer data
	OUT LD CALL	(&H27),A C,&H28 STAT	
	JP	C,I2CERR	;In case of no acknowledge, exit
	LD OUT LD CALL	A,&H54 (&H27),A C,&HF8 STAT	;Generate stop condition ; Other I2C masters now know they can send their ; data.
	JR	C,I2CERR	;In case of any bus error, exit
	LD OUT	A,&H44 (&H27),A	
	RET		
STAT: STAL:	LD IN AND JR DJNZ SCF RET	B,0 A,(&H27) 8 NZ,STAC STAL	
STAC:	IN CP JR SCF	A, (&H24) C Z,STAOK	
STAOK:	RET		

#### **Balance**

TDAINI: LD A,&H64 (&H27),A OUT С,&Н08 LD CALL STAT JP C,I2CERR LD A,&H82 OUT (&H25),A LD A,&H44 OUT (&H27),A LD C,&H18 CALL STAT JP C,I2CERR XOR А (&H25),A OUT LD A,&H44 (&H27),A OUT LD C,&H28 CALL STAT C,I2CERR JP LD E,(HL) ;Volume INC HL LD A,(HL) ;Balance PUSH HL 128 CP JR C,BALL LD A,E NOBALL JR BALL: SLA Е ;Vol\*2 CALL MUL8 ;HL=A\*E LD A,H ;A=HL/256 (&H25),A NOBALL: OUT A,&H44 LD (&H27),A OUT LD C,&H28 CALL STAT POP HL C,I2CERR JP DEC HLE,(HL) ;Volume LD INC HL LD A,(HL) ;Balance PUSH HL CPL CP 127 JR C,BALR LD A,E NOBALR JR ;Vol\*2 BALR: SLA E CALL MUL8 ;HL=A\*E LD A,H ;A=HL/256 NOBALR: OUT (&H25),A A,&H44 LD (&H27),A OUT C,&H28 LD CALL STAT POP HL C,I2CERR JP INC ΗL LD A,(HL) INC HL (&H25)**,**A OUT LD A,&H44 OUT (&H27),A LD C,&H28 CALL STAT C,I2CERR JP LD A,(HL) INC HL

	OUT LD OUT LD CALL JP	(&H25),A A,&H44 (&H27),A C,&H28 STAT C,I2CERR
	LD OUT LD CALL JP	A,&H64 (&H27),A C,&H10 STAT C,I2CERR
	LD OUT LD OUT LD CALL JP	A, &H82 (&H25),A A, &H44 (&H27),A C, &H18 STAT C, I2CERR
	LD OUT LD OUT LD CALL JP	A, &H08 (&H25),A A, &H44 (&H27),A C, &H28 STAT C, I2CERR
	LD INC OUT LD OUT LD CALL JP	A,(HL) HL (&H25),A A,&H44 (&H27),A C,&H28 STAT C,I2CERR
	LD OUT LD CALL JR	A,&H54 (&H27),A C,&HF8 STAT C,I2CERR
	RET	
STAT: STAL:	LD IN AND JR DJNZ SCF RET	B,0 A,(&H27) 8 NZ,STAC STAL
STAC: STAOK:	IN CP JR SCF RET	A, (&H24) C Z,STAOK

#### Updating firmware

1 Generate a hardware reset by writing to I/O address &H21

2 Write to I/O address &H22:

&HAA &H55 &HAA Size of block to program, high byte Size of block to program, low byte Program data byte by byte

After each transmitted byte, the value read back on I/O &H22 will de incremented by one when the cartridge is ready to receive a new byte. Bytes which are written when the cartridge is not ready will be lost.

The FLASH consists of pages of 256 bytes, so after each 256 data bytes the AtMega starts flashing one block. This takes more processing time than other bytes.

The data can be verified by reading back the program data as described in the next chapter.

#### **Reading program memory**

1 Generate a hardware reset by writing to I/O address &H21

2 Write to I/O address &H22:

&HAA &H55 &H55

All data can be read sequentially at I/O address &H22.

## Known issues

Hardware interrupt acknowledge does not work in R800 mode. The interrupt request is also ended after a write to &H22, the interrupt service routine must take care of that.