

QSOUND / QPRINT

This upgrade adds to the good characteristics of the QL a 'real' sound generation facility and a parallel printer interface.

Fitting the board

Ensure that the QL is disconnected from the mains. If you do not you may damage the QL, the QSOUND/QPRINT board or both. Remove the rectangular plastic cover on the left hand side of your QL which covers the expansion bus slot. This may require some effort.

Now push the QSOUND/QPRINT board firmly in the expansion slot. This may take also a little effort. You should be able to feel when the board is firmly in place.

To test your new upgrade connect the QL to the mains. After the usual memory test screen, the TV/Monitor selection screen will appear with an additional copyright message from the QSOUND/QPRINT board. If you have connected an amplifier you should hear a short sound like a bell: QSOUND/QPRINT tells you that it is OK! After pressing F1 (or F2) you can use the full power of your QSOUND/QPRINT expansion.

To connect an amplifier to the QSOUND/QPRINT card, there is a 3.5 mm cinch socket on the left hand side of the card. You will also find a centronics compatible printer interface there.

Obligatory notice

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PARALLEL INTERFACEUse

After powering up and initialisation you can use a new QL device named PAR.

You can connect a centronics compatible printer to the parallel interface and use it via the device name PAR.

A procedure to list the actual SuperBASIC program to a printer connected to the PAR port could look like this:

```
30000 DEFine PROCedure llist
30001  OPEN #3,PAR : LIST #3 : CLOSE #3
30002 END DEFine
```

Buffers

You can use a part of the QL's RAM as a buffer for efficient print spooling. Its size in blocks of 512 bytes must be specified in the OPEN command as in the following examples:

```
OPEN #3, PAR_1           512 byte buffer
OPEN #3, PAR_32         16 kbyte buffer
```

The maximum buffer size is 63 kbytes.

Options

Two options are accepted as part of the device name.

PARC_<bfsz> The 'C' flag is used if the <LF> character (chr\$(10)) should be converted to a <CR> character (chr\$(13)).

PARF_<bfsz> The 'F' flag is used if a <FF> character should be sent when the channel is closed.

PARCF_<bfsz> Combination of the two flags is allowed.

SER emulation

Using the new PAR_USE command you can get the benefits of your new parallel interface without changing existing programs. This command accepts a three character device name (with or without string quotes) as a parameter.

PAR_USE SER will emulate the SER ports of the QL. All output sent to the serial port will be sent to the new PAR device.

Psion software

If you want to use the PAR interface with ABACUS, ARCHIVE EASEL and QUILL you may include the PAR_USE command in the corresponding boot program.

Alternatively, you can also change the printer driver with the

INSTALL_BAS program supplied with the Psion software package: Load and run the program and choose the Microdrive as required. Choose the PAR port option by pressing the space bar. Get the list of printer parameters by pressing F2. Change the PORT by pressing either the left or right cursor key and then type 'PAR' as the valid device name. Don't use a buffer unless you have a memory expansion as the Psion programs use the entire RAM of an unexpanded QL. Don't add the flags 'C' or 'F'. You can obtain these options specifying the END OF LINE code as CR and by including FF in the POSTAMBLE CODE, respectively.

ADDITIONAL SUPERBASIC COMMANDS

The QSOUND/QPRINT firmware consists of 30 new SuperBASIC commands which allow you to utilize the PAR interface and the full capability of the sound chip. A few useful procedures/functions are also included. The commands are described below. Parameters enclosed in < > are optional. The default window is #1.

BELL procedure
causes the sound chip to emit a short bell like sound

CONTROL procedure
creates a job which opens a window in the upper right-hand corner of the screen shows the number of the job that is currently ready to receive input (i.e. whose cursor is flashing)

CUR_FLASH f procedure
changes the flashing rate of the cursor
f = 0 normal
f = 1 medium
f = 2 high

CURDIS <#n> procedure
disables the cursor in window #n

CURSEN <#n> procedure
enables the cursor in window #n
The INKEY\$ command doesn't show a cursor. The following function also returns the character corresponding to the key pressed but shows a flashing cursor for the number of frames specified in wait:

```

DEFINE FUNCTION getkey$(chn, wait)
  CURSEN #chn
  r$=INKEY$(#chn, wait)
  CURDIS #chn
  RETURN r$
END DEFINE

```

m = D_MODE function
returns the current display mode
m = 4 high resolution (4 colours)
m = 8 low resolution (8 colours)

`t = D_TYPE` function
returns the current display type (which
determines the default windows)
t = 0 monitor
t = 1 TV

`DOWN <#n>` procedure
moves the cursor in window #n one row
down

`EXPLODE` procedure
causes the sound chip to emit an
explosion like noise

`HOLD` procedure
stops all interrupt sound lists

`HOLD n` procedure
stops the interrupt sound list n (1 .. 3)

`LEFT <#n>` procedure
moves the cursor in window #n one column
to the left

`LIST_AY r0 .. r13` procedure
sends the value of r0 .. r13 to the
registers 0 .. 13 of the sound chip

`nno = NET_NR` function
returns the network station number of
your machine which was assigned by the
SuperBASIC command NET

`NEW_FONT <#n>, add` procedure
assigns the new character font at address
add to window #n

`OLD_FONT <#n>` procedure
assigns the standard QL font to window #n

`PAR_STOP` procedure
clears the PAR device buffer and stops
printing via the PAR port

`PAR_USE ddd` procedure
renames the PAR device to the new three
character name ddd. Useful for SER
emulation.
PAR_USE SER emulates the serial port 1.

`v = PEEK_AY (r)` function
returns the contents of the register r of
the sound chip

PLAY n, sound\$ procedure
 puts the string sound\$ into the interrupt list of the sound channel n (1 .. 3).

sound\$ may contain various characters (case is not distinctive) to denote

notes: C D E F G A H
 (H corresponding to B, HB to Bflat)

sharps: #

flats: b

rests: p (one length unit)

change octave: o0 o1 .. o7
 (default: o2)

change volume: v0 v1 .. v15
 (default: v0)
 v16 switches to wrap control

duration of note in 1/50 sec: 10 .. 1255
 (default: 15)

change noise frequency: n0 n1 .. n31
 (default: n0)

determine wrap curve: w0 w1 .. w15
 (default: w0)

change length of wrap: x0 x1 .. x32767
 (default is x0)

synchronisation stop: S
 causes a sound channel to wait

activate a waiting channel: r1 r2 r3

Sound example (try it):

```
PLAY 1, 'pr15o4sCDEFGAHo5CDEFGAhp'
PLAY 2, 'pv15o2r1cDEFGAHo3CDEFGAhp'
```

POKE_AY r,v. procedure
 sets one of the registers (0 .. 13) of the sound processor to value v (0 .. 255)

vno\$ = QDOS\$ function
 returns the version number of your QDOS operating system

RELEASE procedure
 causes all interrupt sound lists to be played

RELEASE n procedure
 causes the sound list n (1 .. 3) to be played

RIGHT <#n> procedure
 moves the cursor in window #n one column
 to the right

SHOOT procedure
 causes the sound chip to emit a noise
 like a shot

SOUND procedure
 clears all sounds played by the sound
 chip. All tunes in the sound interrupt
 lists created with the PLAY command are
 cleared.

SOUND n procedure
 clears the sound channel n (1 ..3) and
 the corresponding interrupt list

SOUND n,f,v procedure
 sets the sound output to sound channel n
 to the frequency f (400 ... 5000 Hz) and
 the volume v (0 .. 15)

UP <#n> procedure
 moves the cursor in window #n one row up

MACHINE CODE PROGRAMMING WITH THE AY-3-8910 SOUND PROCESSOR

The QSOUND/QPRINT card could theoretical occupy any of the 16 expansion slots recognized by QDOS. To find the actual base address, read the system variable SV.AYBAS. The address to jump to is in the system variable SV.AYJMP. The sound processor control routines are called with a code in the register D0 of the MC68000 (like the QDOS trap routines).

A call to the routine AY.RDREG (which reads a register of the AY-3-8910) should look like this:

```
MOVEQ      #AY.RDREG,D0      code for AY-routine
MOVEQ      #0,D2             read register $00
MOVE.L     SV.AYJMP,A0      get address to jump to
JSR        (A0)             do it
```

A QDOS error code will be returned in D0.

ROUTINE AND ADDRESS SUMMARY**ADDRESS DEFINITIONS**

Name	Address	Description
SV.AYBAS	\$28160 long	Base address of the firmware
SV.AYJMP	\$28164 long	Start address for machine code routines

ROUTINES FOR SOUND CONTROL

D0	Name	Description
\$00	AY.RESET	clears sound, stops noise
\$01	AY.WRREG	writes one AY-3-8910 register
\$02	AY.RDREG	reads one AY-3-8910 register
\$03	AY.WRALL	writes all (0 .. 13) registers
\$04	AY.RDALL	reads all (0 .. 15) registers
\$05	AY.PLAY	plays a tune
\$06	AY.TSTPL	status query
\$07	AY.HOLD	causes a sound channel to wait
\$08	AY.RELSE	releases a waiting channel
\$09	AY.NOISE	emits a predefined noise
\$0A	AY.SOUND	emits a user-defined sound

HARDWARE KEY

Name	Address	Description
AY.PORTA	\$8000	PIA-Dataport A
AY.CTRLA	\$8001	Control port A
AY.PORTB	\$8002	PIA-Dataport B
AY.CTRLB	\$8003	Control port B

DO = \$00

AY. RESET

Clears the sound.

Call parameters

Return parameters

D1
D2
D3

D1 undefined
D2 undefined
D3 preserved

A0
A1

A0 preserved
A1 undefined
A5 undefined

ERROR RETURNS

None

NOTES

Also clears all sound interrupt lists.

DO = \$01

AY.WRREG

Writes a value to AY-3-8910 register.

Call parameters

D1.B value
D2.B register
D3

A0
A1

Return parameters

D1 preserved
D2 preserved
D3 preserved

A0 preserved
A1 preserved
A5 undefined

ERROR RETURNS

ERR.OR Invalid register number (>13)

NOTES

Please note, that only registers 0 .. 13 are used.
The other ports are used for the PAR interface.
Bit 6 and 7 of register 7 remain unchanged. They
are also used for port managing.

DO = \$02

AY.RDREG

Reads a AY-3-8910 register.

Call parameters

Return parameters

D1
D2.B register
D3

D1 value read
D2 preserved
D3 preserved

A0
A1

A0 preserved
A1 preserved
A5 undefined

ERROR RETURNS

ERR.OR Invalid register number (>15)

DO = \$03**AY.WRALL**

Writes all registers (0 .. 13) of the AY-3-8910.

Call parameters

D1
D2
D3

A0
A1.L pointer to datablock

Return parameters

D1 undefined
D2 undefined
D3 preserved

A0 preserved
A1 undefined
A5 undefined

ERROR RETURNS

None

NOTES

The datablock must contain 14 bytes with the values for the registers in ascending order (\$00 = r0 .. \$0C = r13).

DO = \$04**AY.RDALL**

Reads all registers (0 .. 15) of the AY-3-8910.

Call parameters

D1
D2
D3

A0
A1.L pointer to buffer

Return parameters

D1 undefined
D2 undefined
D3 preserved

A0 preserved
A1 undefined
A5 undefined

ERROR RETURNS

None

NOTES

The 16 byte buffer contains the values of the registers in ascending order (\$00 = r0 .. \$10 = r15).

DO = \$05

AY.PLAY

Puts a string to the interrupt sound list.

Call parameters

Return parameters

D1.B AY-channel
D2
D3

D1 undefined
D2 preserved
D3 preserved

A0.L pointer to the string
A1

A0 undefined
A1 preserved
A5 undefined

ERROR RETURNS

ERR.OR AY-channel was not 1, 2 or 3.
ERR.BP String contains undefined sound items.

NOTES

The string must be preceded by a word containing the string length.

D0 = \$06

AY.TSTPL

Returns the status of a AY-3-8910 channel buffer.

Call parameters

Return parameters

D1.B AY-channel

D1 status

D2

D2 preserved

D3

D3 preserved

A0

A0 preserved

A1

A1 preserved

A5 undefined

ERROR RETURNS

ERR.OR

AY-channel was not 1, 2 or 3.

ERR.NO

Sound list doesn't exist.

NOTES

Status return in D0: 0 waiting
1 playing

DO = \$07

AY.HOLD

Suspends playing a sound list.

Call parameters

Return parameters

D1.B AY-channel
D2
D3

D1 undefined
D2 undefined
D3 undefined

A0
A1

A0 undefined
A1 undefined
A5 undefined

ERROR RETURNS

ERR.OR AY-channel was not 0,1,2 or 3.
ERR.NO Sound list doesn't exist.

NOTES

AY-channel number 0 stops playing on all channels.

DO = \$08

AY.RELSE

Releases a suspended sound list.

Call parameters

Return parameters

D1.B AY-channel
D2
D3

D1 undefined
D2 undefined
D3 undefined

A0
A1

A0 undefined
A1 undefined
A5 undefined

ERROR RETURNS

ERR.OR AY-channel was not 0,1,2 or 3.
ERR.NO Sound list doesn't exist.

NOTES

AY-channel number 0 causes all channels to continue.

DO = \$09

AY.NOISE

Causes the sound processor to emit predefined noises.

Call parameters

Return parameters

D1.B noise
D2
D3

D1 undefined
D2 undefined
D3 preserved

A0
A1

A0 preserved
A1 undefined
A5 undefined

ERROR RETURNS

ERR.BP noise >2

NOTES

Values of noise:
0 explosion
1 gunshot
2 bell

DO = \$OA**AY.SOUND**

Emits a sound with a specific frequency and volume on the chosen channel.

Call parameters

D1.B AY-channel
D2.W frequency
D3.B volume
A0
A1

Return parameters

D1 undefined
D2 undefined
D3 undefined
A0 preserved
A1 preserved
A5 undefined

ERROR RETURNS

ERR.BP AY-channel was not 1, 2 or 3.
ERR.OR Frequency was out of range.

NOTES

The valid frequency is 400 .. 5000 Hz.
The valid volume code is 0 .. 15.

REGISTERS OF THE AY-3-8910 SOUND PROCESSOR

The sound processor AY-3-8910 works with a 0.75 MHz time frequency on the QL. It uses 16 read/write registers. The use of the registers is described below.

reg		bit	7	6	5	4	3	2	1	0
\$00	chn A tone	LSB	7	6	5	4	3	2	1	0
\$01	chn A	MSB	x	x	x	x	B	A	9	8
\$02	chn B tone	LSB	7	6	5	4	3	2	1	0
\$03	chn B	MSB	x	x	x	x	B	A	9	8
\$04	chn C tone	LSB	7	6	5	4	3	2	1	0
\$05	chn C	MSB	x	x	x	x	B	A	9	8
\$06	noise period		x	x	x	x	3	2	1	0
\$07	release		ioB	ioA	nC	nB	nA	sC	sB	sA
\$08	chn A amplitude		x	x	x	w	3	2	1	0
\$09	chn B amplitude		x	x	x	w	3	2	1	0
\$0A	chn C amplitude		x	x	x	w	3	2	1	0
\$0B	wrap period	LSB	7	6	5	4	3	2	1	0
\$0C	wrap period	MSB	F	E	D	C	B	A	9	8
\$0D	wrap curve		x	x	x	x	w3	w2	w1	w0
\$0E	I/O port A		7	6	5	4	3	2	1	0
\$0F	I/O port B		7	6	5	4	3	2	1	0

NOTES

x: bit not used

ioA: If bit is set then port A is input channel else it is output channel.

ioB: If bit is set then port B is input channel else it is output channel.

nA .. nC: If bit is reset channel is emitting noise.

sA .. sC: If bit is reset channel is emitting sound.

w0 .. w3: Wrap curve (cf. page 22)

w: Bit activates wrap control.

Registers \$00 .. \$05 define the pitch of the channel. Two registers define a note. The main time is divided by 16. By counting down the 12-bit-counter the output frequency is generated.

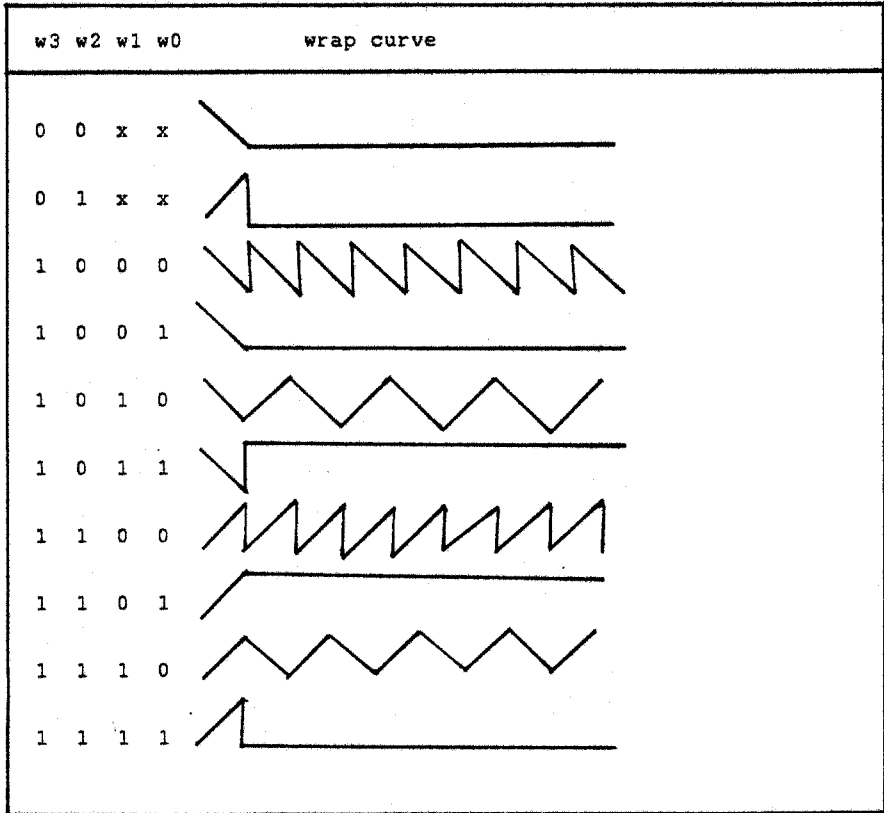
Register \$06 defines the noise frequency. This works like the pitch control but with 5 bits only.

Register \$07 releases the sources. It selects silence, sound, noise or sound and noise for each channel.

Registers \$08...\$0A define the volume. The four LSBits denote the volume in logarithmic steps. The w-bit activates the wrap control.

Registers \$0B and \$0C define a 16 bit wrap period.

Register \$0D selects the wrap curve as shown below.



Registers \$0E and \$0F describe the state of port A and port B.