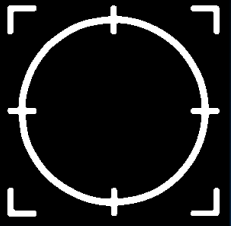


Sound and Music in Videogames



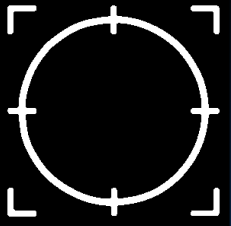
Harry Holmwood
Sony Computer Entertainment
Europe





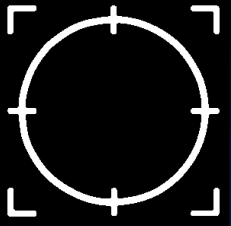
Use of Sound Effects

- To enhance atmosphere
- Provide audible 'feedback'
- To surprise players
- Give clues



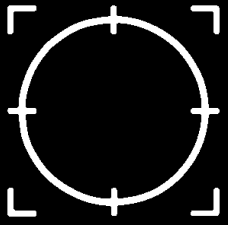
Sound effects

- Be careful with tuning when playing with music
- Provide positive feedback
- Realistic is too boring
 - Be 'bigger than life'



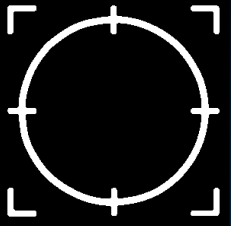
Music

- Enhance atmosphere
- Provide excitement
- Build up expectations....
- either fulfill them or not (© Fatman)



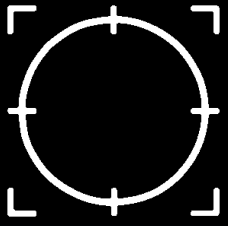
Interactive Music

- Music which reacts to in-game action
- Sudden changes
 - Dramatic effect
 - Scary
 - Easy to do
- Smooth changes
 - Less dramatic
 - More subtle
 - Harder to do ?



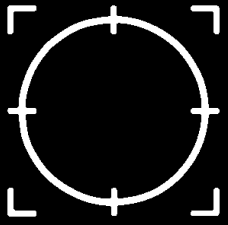
Sudden changes

- Branching MIDI
 - Separate MIDI sequences
 - Stop one, start another
 - Use an SEP file
- CD-DA
 - No processor overhead
 - Best quality
 - Long seek times
 - High storage requirements



Multi-channel CD-XA

- Interleave multiple audio channels
- Playback one at a time
- Saves disk space
- No processor overhead



Example CTI file fragment

```
XAFileAttributes Form1 Audio
XAVideoAttributes ApplicationSpecific
XAAudioAttributes ADPCM_C Stereo
XAInterleavedFile even.xa c:\gamedata\even.xa
```

```
XAChannelInterleave TimeCritical 1-2-3-4
```

```
XAChannel 1
```

```
    XAFileAttributes Form2 Audio
```

```
    Source c:\data\wav\d1.xa
```

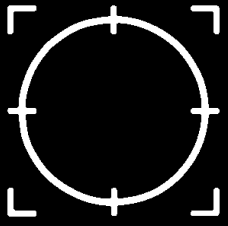
```
    MinLength 270000
```

```
        ;note this is the length of the longest of the 4
```

```
    XAEndChannel
```

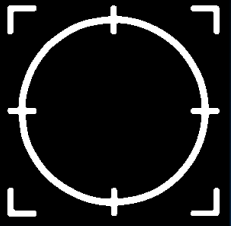
```
    ; ..... other 3 channels in here
```

```
XAEndInterleavedFile
```

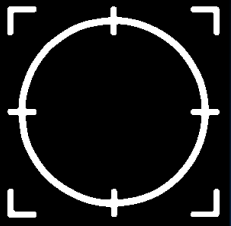
Smooth branches

- Layered MIDI sequences
 - One for each character in an adventure game
 - Use 'markers' in the MIDI sequence to synchronise branches
 - Can't jump into a sequence - running status
 - Lower quality than CD audio
- Atmosphere loops
 - Fade sample loops to alter atmosphere
 - No MIDI sequencing
 - Uses a lot of Sound RAM



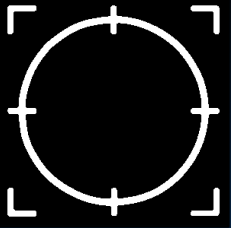
Atmosphere Loops

- Several samples playing simultaneously
- Fade them in and out to change atmosphere
- `SsUtSetVVol()` to change a channel's volume
- Dynamically start and stop loops as required to save polyphony



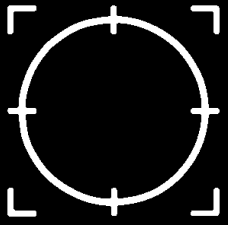
MIDI

- Very low storage space for sequences
- Must store instrument sounds in SPU RAM
- LibSnd provides MIDI playback functionality
- Can adjust tempo etc in real-time



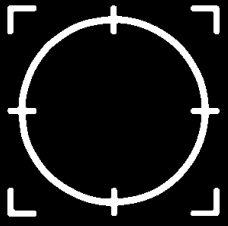
Sequencers

- Sequencing with SoundDelicatessen ?
- Don't bother
 - Use a separate sampler
 - Write your music
 - Convert at the end



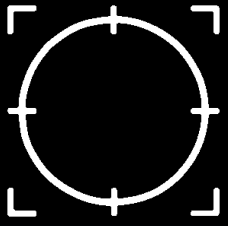
Why doesn't it work ?

- SoundDelicatessen written for Apple MIDI Manager
- Sequencers NOT written for MIDI Manager
- OMS can work



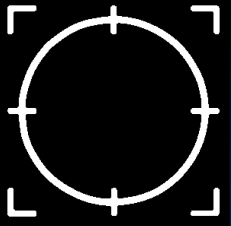
So how do you do it?

- Preferably, use two computers
 - One (Mac, PC, ST etc) for sequencing
 - One for Sound Artist Card
 - Treat card as any other MIDI device
- Otherwise (what I do)
 - Write music using a separate sampler (eg SampleCell, AKAI etc)
 - Convert samples once composition is complete



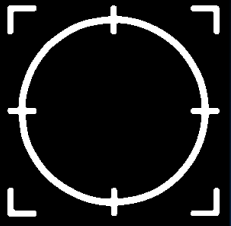
Sampling

- Equipment needed
 - Sampler
 - Mixer
 - EQ
 - Compressor
- Sample from DAT / CD
 - Can re-sample if necessary
 - Can EQ and compress sample
 - Normalise samples



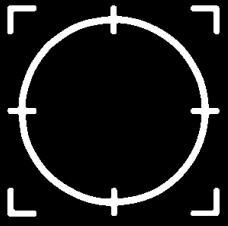
Sample Editing

- Sample must start and end cleanly
- Loop points must be on 28 sample boundaries
- Divisible by 28
- No other markers
- Use a crossfade to smooth loop



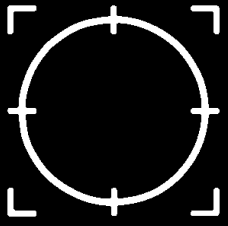
Sample Editing

- ▶ Save as AIFF mono files
- ▶ Any sample rate up to 48kHz
- ▶ Convert with AIFF2VAG



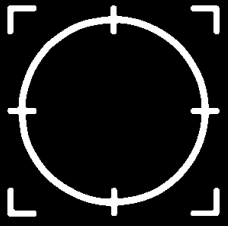
CD-DA

- Provides the highest quality
- Can be played in a standard CD player
- Drive plays at single speed
- Cannot interleave data or graphics



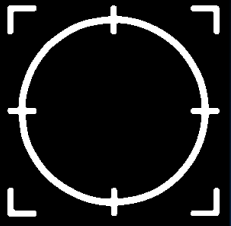
CD-XA

- Audio encoded as XA-ADPCM
 - Approx. 4 times smaller
 - 37.8 kHz or 18.9 kHz
- Use RAW2XA on the Mac
 - Takes 18.9/37.8 kHz mono or stereo
 - Sound Designer II format
 - Batching - 'Interactive' mode



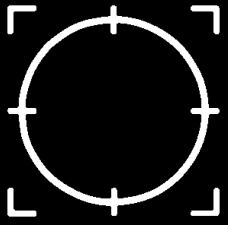
Producing audio for video

- ▶ Use XA-ADPCM audio
 - ▶ Created with RAW2XA
 - ▶ Can be interleaved with video
 - ▶ MOVCONV / BUILDCD



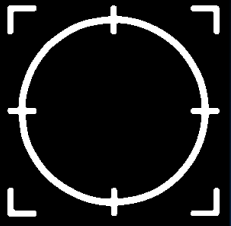
Ideally

- Have video on BetaCam tape
 - MIDI interface with SMPTE
 - Synchronise sequencer to video
- More likely
 - Write down the frame numbers
 - Guess



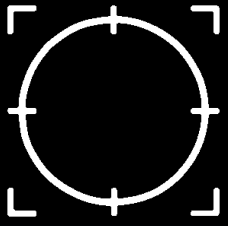
Interleaving using BUILDCD

- SN Systems provide BUILDCD with the CD-Emulator
- For 37.8 kHz stereo, can interleave one sector of audio for every seven of video (double speed).
- Output a .CCS file, which can be used with the CD-Generator software
- Use XA files created by RAW2XA - remove subheaders



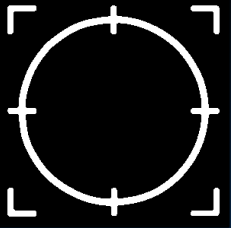
Interleaving with MOVCONV

- ▶ Can convert WAVs to XA files (ie no need for RAW2XA on the Mac)
- ▶ Use 16-bit wavs at either 37.8 or 18.9 Khz
- ▶ No need to remove subheaders



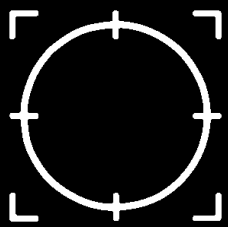
PlayStation Sound Specs

- ▶ PlayStation uses ADPCM sound compression
 - ▶ Approx 4:1 compression ratio
 - ▶ 16-bit compressed down to 4
 - ▶ Similar to SNES sound chip
- ▶ 24 simultaneous sounds
- ▶ 512kB sound RAM to store samples



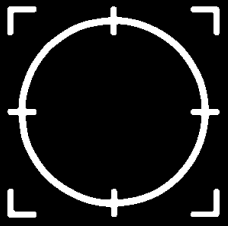
Pitch setting

- ▶ Raise samples by up to 2 octaves
- ▶ Lower samples by up to 12 octaves
- ▶ Can specify fine intervals of a semitone or less



Getting samples into sound RAM

- Samples built into VAB files - a sound bank
- VAB split into .VH (Header) and .VB (Body) files using VABSPLIT.EXE
- Open the VAB header - gives you a VAB id
- Load the VB into memory
- Transfer it into sound RAM



Getting samples into sound RAM

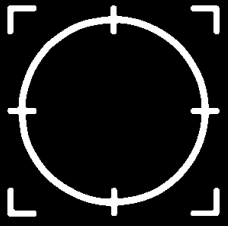
```
#define VH_ADDR    0x80025000
#define VB_ADDR    0x80030000

short gVAB;
/* open VAB header */
gVAB = SsVabOpenHead (VH_ADDR, -1);

if (gVAB < 0)
    printf ("SsVabOpenHead : failed\n");

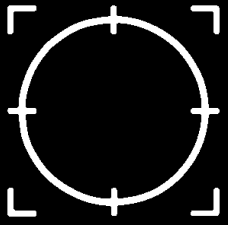
if (SsVabTransBody (VB_ADDR, gVAB) != gVAB)
    printf ("SsVabTransBody : failed!\n");

    SsVabTransCompleted (SS_WAIT_COMPLETED);
```



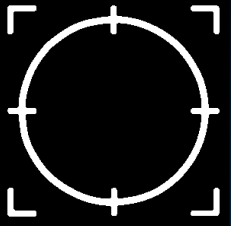
Saving Main RAM

- ▶ Not enough RAM to store entire VB file?
 - ▶ Use `SsVABTransBodyPartly`
 - ▶ Uses a small buffer to store parts of the VB file
 - ▶ Fill the RAM buffer, transfer it to SPU RAM, and start again
 - ▶ Until whole VB is transferred



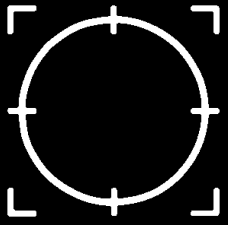
Transfer in background

- ▶ `SsVabTransCompleted` (`SS_WAIT_COMPLETED`) blocks until the transfer is complete
- ▶ `SsVabTransCompleted` (`SS_IMMEDIATE`) returns immediately.
- ▶ Returns 1 if the transfer is completed, 0 if it is ongoing



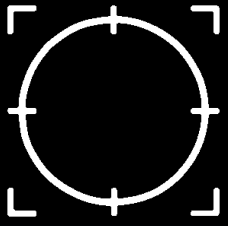
Initialising the sound system

- Set NTSC/PAL with SetVideoMode
- SsInit() to start the sound system
- Set the 'tick mode' - use SS_TICKVSYNC
- Use SsStart() to begin sound processing
- Use SsEnd() to stop it
- Finish with SsQuit()



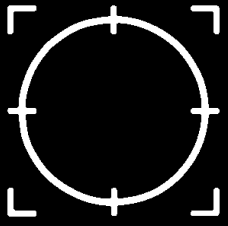
Sound Initialisation

```
/* Initialise sound system (libSnd) */  
SsInit();  
  
/* Set 'tick mode' to work regardless of NTSC/PAL  
settings */  
SsSetTickMode(SS_TICKVSYNC);  
  
/* Begin Sound Processing */  
SsStart();  
  
/* Set main volume */  
SsSetMVol(leftVOL, rightVOL);
```



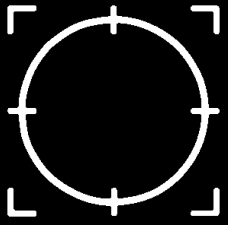
VAB file

- Bank file for sample data
- Contains VAG files which are ADPCM samples
- Has a header and a body (just sample data)
- Body goes into sound RAM
- VAB contains programs, which are made up of tones (samples) and ADSR attributes



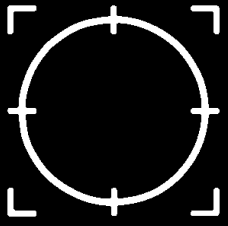
Playing a sample (libsnd)

- ▶ Easiest way is with `SsUtKeyOn`
- ▶ Specify program number
- ▶ Tone number
- ▶ Specify Note - 0 to 127
- ▶ Specify fine tuning - 0 to 127
- ▶ Returns a channel number



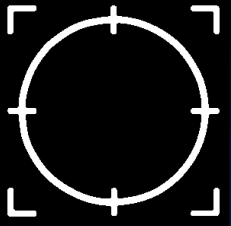
Playing MIDI files

- ▶ LibSnd provides MIDI playback functionality
- ▶ Use SMF2SEQ to convert your MIDI files
- ▶ No aftertouch - minimise continuous controllers
- ▶ Sequence playback in SoundDeli did not work
- ▶ Check MIDI playback on the DTL-H2000 instead



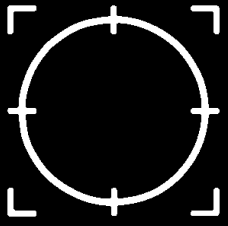
MIDI commands

- ▶ Must set 'tick mode'
 - ▶ SS_TICK60 / SS_TICK240 - change resolution
 - ▶ SS_TICKVSYNC should work for NTSC or PAL
- ▶ SsSeqOpen(addr, vab_id) - Must do this first
 - ▶ Give it the address of the SEQ file in memory
 - ▶ And the ID number of the VAB it is associated with



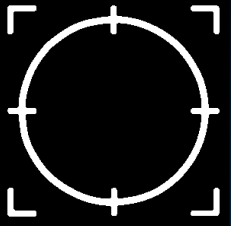
Sequence Playing

- ▶ SsSeqPlay
 - ▶ Can tell it to play the sequence once, or repeatedly
 - ▶ Can set sequence playback to 'pause'
- ▶ SsSeqStop(seq_access_num)
- ▶ SsSeqPause / SsSeqReplay



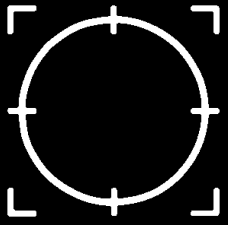
Sequence Tempo

- ▶ `SsSeqSetAccelerando / SsSeqSetRitardando`
- ▶ Allow slowing / speeding up of a sequence
- ▶ Specify a new 1/4 note resolution
- ▶ Specify a delta time - in ticks
- ▶ Basically the same function



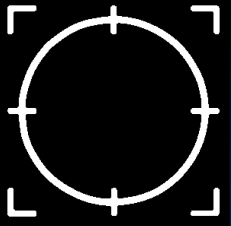
Sequence Volume

- ▶ SsSeqSetCrescendo / SsSeqSetDecrescendo
- ▶ Raise / lower volume over a period of time
- ▶ Volume is added (crescendo) or subtracted (dec.) from current volume
- ▶ Takes a delta time - in ticks



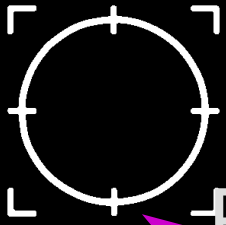
Other MIDI functionality

- SsSetLoop - Sets the number of repetitions of the song
- SsSeqSetNext - Specify the next sequence to be played
- SsSetTempo - Sets tempo explicitly
- SsSetMarkCallback - put markers in sequences
 - Perhaps synchronise animation to a beat



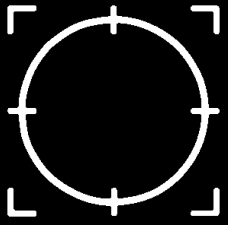
Manipulating a sample

- ▶ Adjust Volume - `SsUtSetVVol`
- ▶ Adjust pitch - `SsUtPitchBend`,
`SsUtChangePitch`
- ▶ Adjust ADSR - `SsUtChangeADSR`
- ▶ Adjust other attributes -
`SsUtSetProgAttr`
- ▶ Autopanning / autovolume
 - ▶ `SsUtAutoPan`
 - ▶ `SsUtAutoVol`



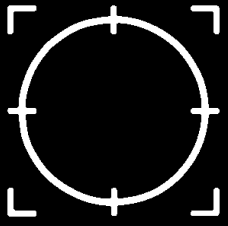
Using Reverb

- ▶ PlayStation DSP offers several reverb algorithms
- ▶ Reverb uses SPU RAM as a buffer
- ▶ The longer the reverb, the more memory is required
- ▶ You set:
 - ▶ Algorithm - `SsUtSetReverbType`
 - ▶ Depth - `SsUtSetReverbDepth`
 - ▶ Delay (for delay / echo) - `SsUtSetReverbDelay`
 - ▶ Feedback (for delay) -

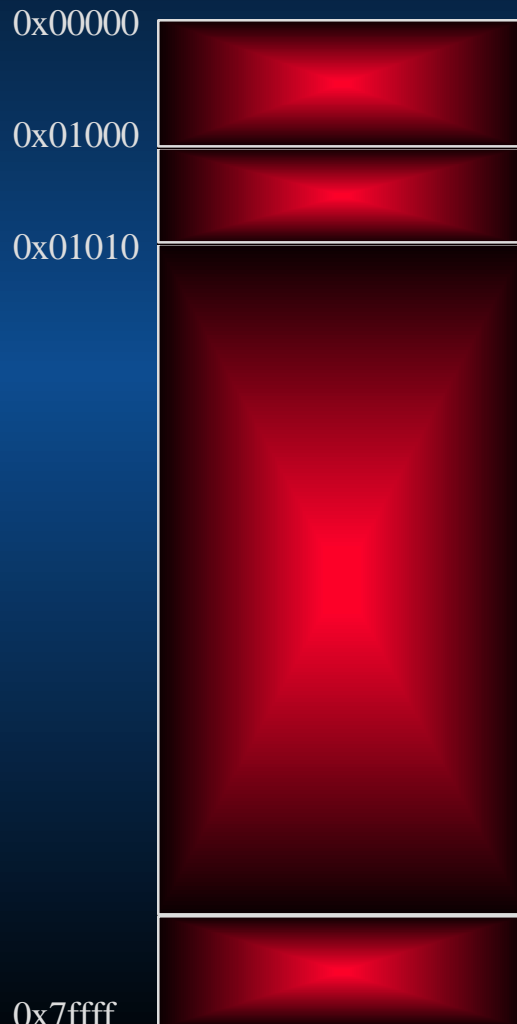


Effect Algorithms

Effect	Memory	Description
Room	9,920	Short Reverb
Studio A	8,000	Small Studio Reverb
Studio B	18,496	Medium Studio Reverb
Studio C	28,640	Large Studio Reverb
Hall	44,512	Large Hall reverb
Space	63,368	Huge spatial reverb
Echo	98,368	Single echo
Delay	90,368	Repeating delay
Pipe	7,072	Metallic pipe



SPU RAM layout

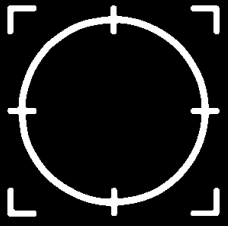


SPU Decode Data Region

Additional Loop information

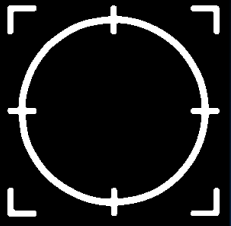
Waveform Data

Transferrable region



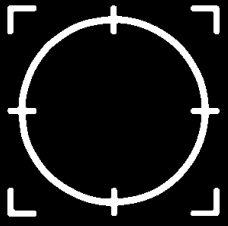
Using multiple VAB files

- Use SsVabOpenHeadSticky
- Must specify the VB address in sound RAM
- Load it above 0x01010
- Careful not to move into reverb work area



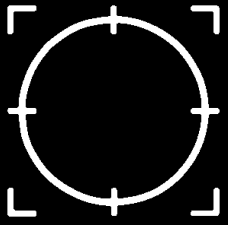
Using Sound Artist Tools

- Nubus card for Apple Macintosh
- Provides sound functionality of PlayStation
- Has optical digital out
 - Requires external DAC or DAT
- Software converters and bank builders



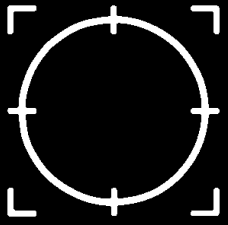
AIFF2VAG

- Converts AIFF files into VAG (PlayStation sample) format
- Loops on 28 sample boundaries
- Can batch process them in 'interactive' mode



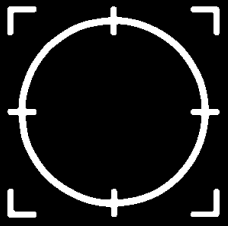
VAG Compression modes

- Standard - for general sound sources
- High band - for sound sources with high-band components
- Low band - For sound sources with low-band components
- 4-bit straight - Four bit straight compression



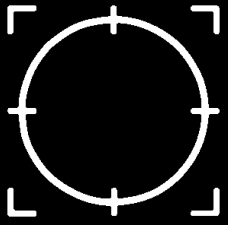
RAW2DA

- Converts raw sample data into DA format
- Use when you need CD-DA audio
- Requires 44.1kHz RAW stereo data
 - Sound Designer II format
- Has an 'auto' mode for batch processing



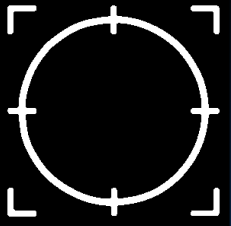
RAW2XA

- Converts raw sample data into XA-ADPCM
- For interleaving with video, other sound or any data
- Requires 18.9kHz or 37.8 kHz mono or stereo RAW sample data
 - Sound Designer II format
- Also has an 'auto' mode
- Extract subheaders for BUILDCD



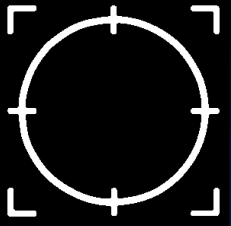
SMF2SEQ

- ▶ Converts MIDI file data into PlayStation SMF format
- ▶ Don't use Aftertouch
 - ▶ Lots of continuous controller data ruins playback
- ▶ Can use SEQ2SEP on PC to build SEQs into a bank of sequence files



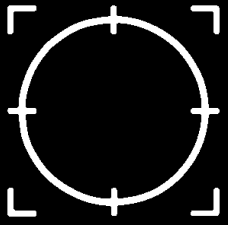
Vag Player

- ▶ Allows playback of VAGs
- ▶ Test sample is looping smoothly
- ▶ Check it fits in sound RAM



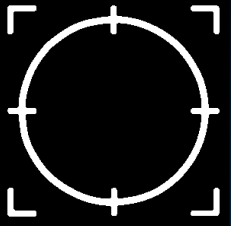
Sample looping

- Demo sample looping on Mac
- Show looping points at 28 sample boundaries
- Do a crossfade
- Show playing on VAG player



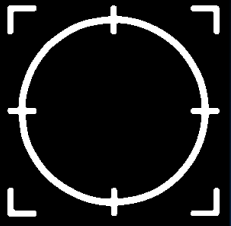
3D sound

- Make sounds appear behind player
- Can give a cinematic feel
- Pre-recorded
 - Done in a studio
 - Specialist hardware encoders
- Real-time
 - Allows movement of samples in 3D space
 - More interactive and suited to games
- Don't overdo it



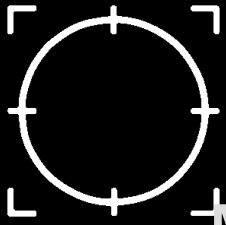
QSound

- Get 3D sound from stereo speakers
- Uses 'image file panning'
- Relys on phase shifting to fool ear
- Requires 2 copies of a sample to be stored
- Costs around £2000 per game in Europe



Dolby ProLogic

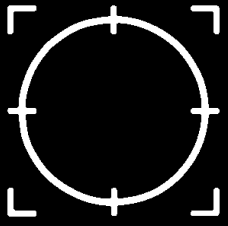
- Requires additional speakers
- Plus a Dolby ProLogic Decoder
- Many hi-fi systems now come with ProLogic
- No license fee (?)
- Here's how.....



Dolby ProLogic (contd)

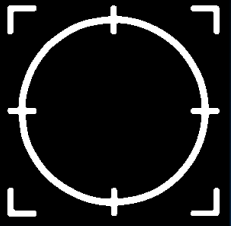
▶ Must vary amplitude and phase relationships in the left/right channels

Encoded Channel	Left Output	Right Output
Left	0dB	off
Center	-3dB	-3dB
Right	off	0dB
Surround	-3dB	-3dB



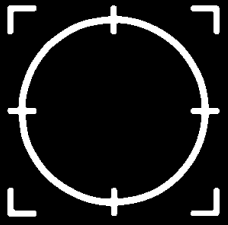
SPU streaming

- ▶ New feature of libspu
- ▶ Play VAG files of any length
- ▶ Can stream VAG files from main RAM into SPU RAM
- ▶ Unlike interleaved XA-ADPCM, can overlay VAGs



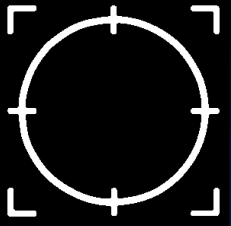
Interactive streaming

- Allow music to behave interactively
- But keep the production values high
- More cinematic
- More relevant



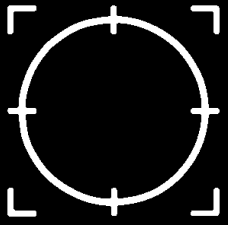
Saving SPU RAM space

- Use low sample rates for low-frequency sounds
- Use MIDI and pitch shifting to allow samples to sound longer
 - Demo Total NBA crowd
- Stream sound effects/speech from CD
- Use Multiple VAB files, loading in as necessary



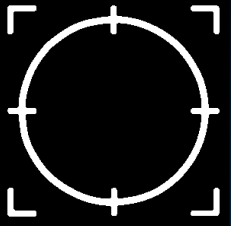
LibSPU or LibSND ?

- LibSnd good enough for most purposes
- Has reasonable MIDI functionality
- Does not allow for much manipulation of sounds
- LibSpu has no MIDI functionality
- Provides lower level functionality



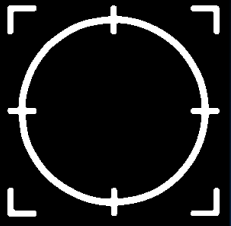
Sound RAM Interrupt

- Can set a callback function to play when an area of sound RAM is 'hit'
- Can use this to detect end of samples
- SpuSetIRQ
- SpuSetIRQAddr
- SpuSetIRQCallback
- VABSPLIT -v
 - Outputs a VAG address table



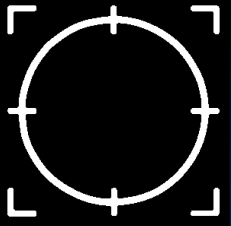
Example VAG address table

```
#define VAGS_engine 1
unsigned long engine[] =
{
    0x0,
    0x5540,
}; /*vag table from engine.vab" */
```



'Free running' problem

- Interrupts were getting triggered for loops after the loop had been stopped
- Due to internal 'virtual pointers' continuing to loop after loop end
- AIFF2VAG version 1.6 onwards should cure this
- Use latest libraries



Finally.....

- DON'T leave the sound until the end of a project
- DO include sound technologies in your technical designs
- Be as INNOVATIVE with sound as you are with graphics
- IF IT SOUNDS GOOD, IT IS GOOD