

# *Sound and Music in Videogames*

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# *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
- 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*

```
XFileAttributes Form1 Audio
```

```
XVideoAttributes ApplicationSpecific
```

```
XAudioAttributes ADPCM_C Stereo
```

```
XInterleavedFile even.xa c:\gamedata\even.xa
```

```
XChannelInterleave TimeCritical 1-2-3-4
```

```
XChannel 1
```

```
    XFileAttributes 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, made 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
- ❖ Check MIDI playback on the DTL-H2000

# *MIDI commands*

- ❖ Must set 'tick mode'
  - SS\_TICK60 / SS\_TICK240
  - SS\_TICKVSYNC works for NTSC or PAL
- ❖ SsSeqOpen( addr, vab\_id ) - Must do this first
  - Give it the address of the SEQ file
  - And the ID number of the VAB
  - returns a Sequence Access Number

# *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 to animation

# *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 - `SsUtSetReverbFeedback`

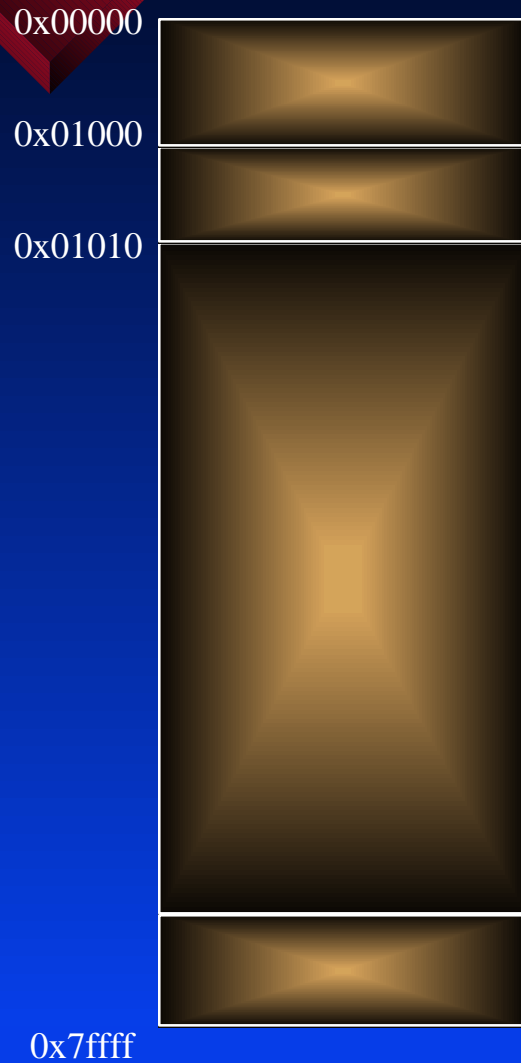


# *Effect Algorithms*

| <b>Effect</b> | <b>Memory</b> | <b>Description</b>   |
|---------------|---------------|----------------------|
| 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         | 98,368        | 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

# 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" */
```

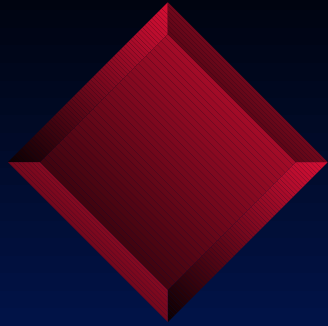
# *'Free running' problem*

- ❖ 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





*The End*

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