

Software Development Seminar

Sound (Basic)



SPU Architecture Overview

SPU

Sound Processing Unit

Data format: ADPCM

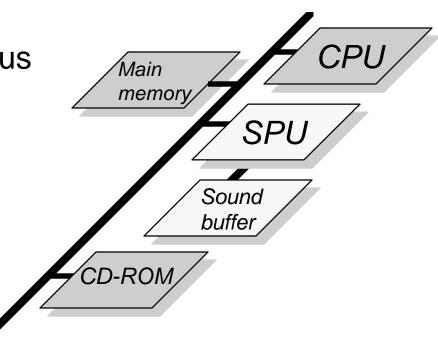
Number of simultaneous

sounds: 24

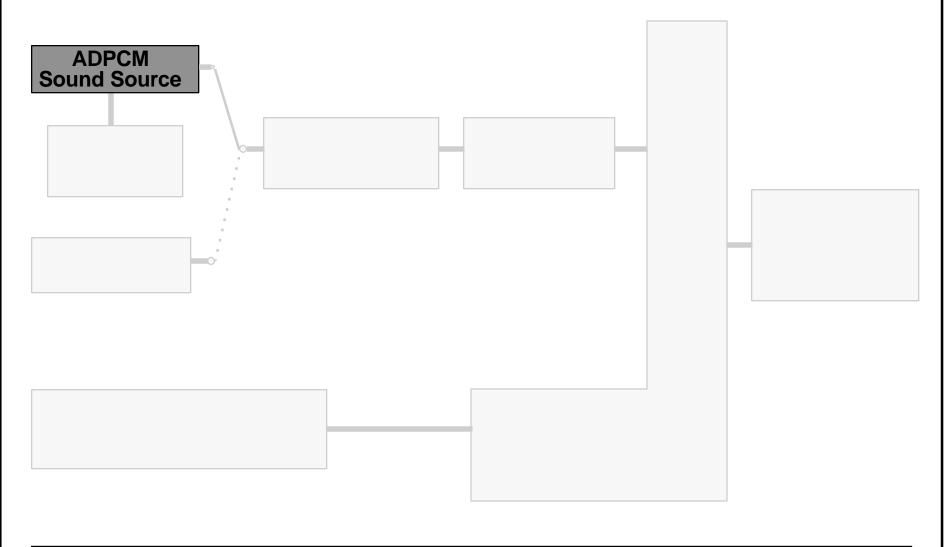
Sampling frequency:

44.1 KHz

Sound buffer:512KB (4Mb)



ADPCM sound source

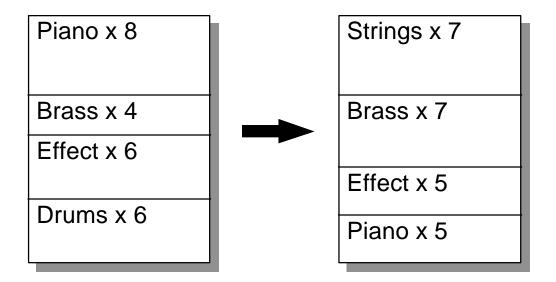




ADPCM sound source (cont)

ADPCM Sound Source

- Number of simultaneous sounds: 24
- Can set waveform data individually
- Can control attributes individually



Pitch setting

Pitch Setting

Can generate sound with varying waveform data interval

Range: -12 oct. ~ 2 oct.

Resolution: Can specify fine intervals

of a halftone or less

LFO

Pitch LFO

When two voices are adjacent, it is possible to apply modulation to the pitch of one voice even if the other voice's volume is undergoing time changes

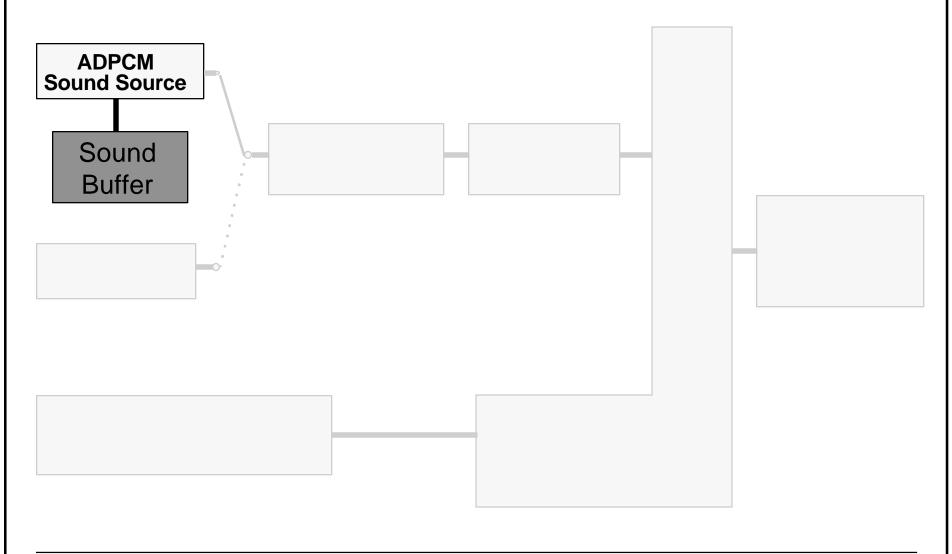
NewPitch_n = $(1 + V_{n-1})$ Pitch_n

NewPitch_n: Final pitch of voice(n)

Pitch_n: Pitch of voice(n)

 V_{n-1} : Volume of voice(n-1)

Sound buffer (1)



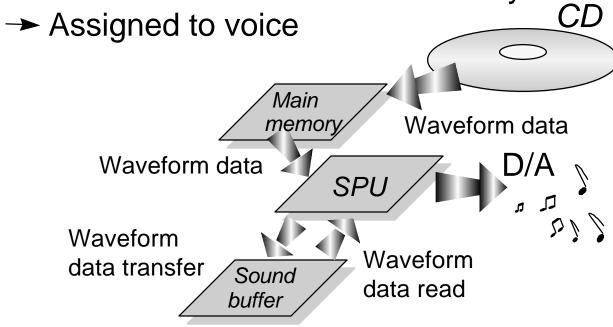


Sound buffer (2)

Sound buffer

Capacity: 512KB (4Mb)

ADPCM waveform data is transferred to the sound buffer via main memory



Sound buffer (3)

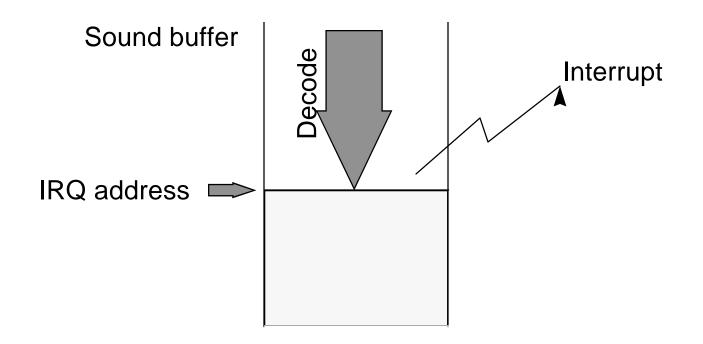
Transfer systems

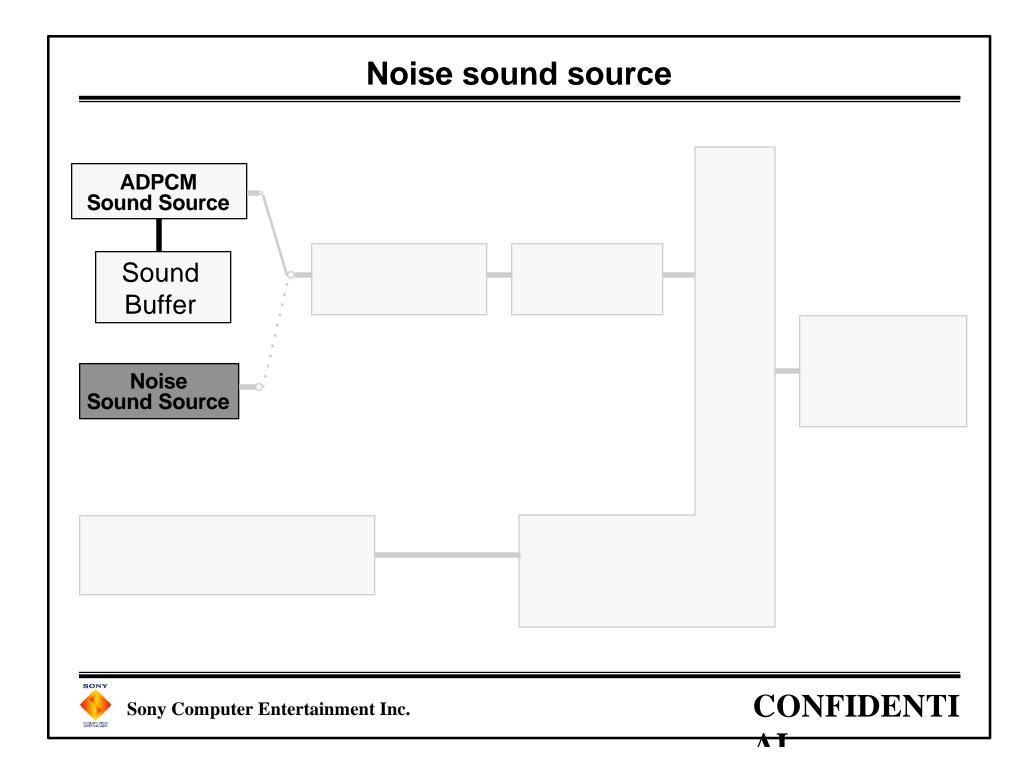
- DMA transfer
- I/O transfer
 (Main memory → sound buffer only)

Interrupt from SPU

Interrupt from SPU

Can set address 1



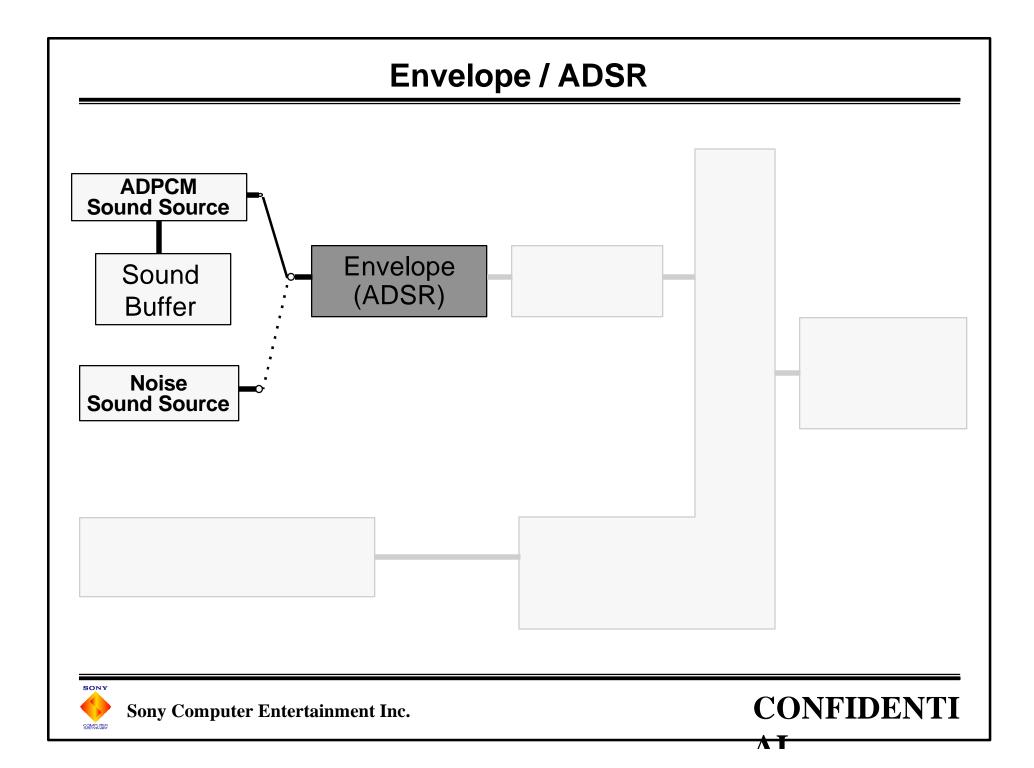


Noise sound source (cont)

Noise sound source

One noise generator is built in

- Can set instead of waveform data
- Can be assigned to more than one voice
- Only one audible pitch can be set



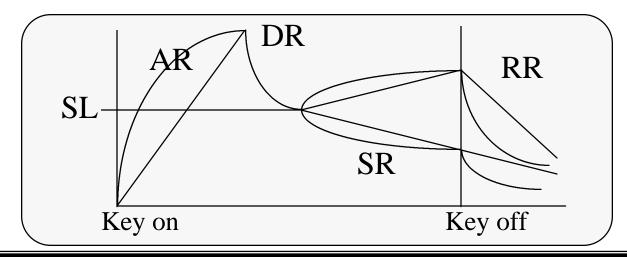
Envelope / ADSR (cont)

Envelope (ADSR)

Rate (speed) can be set for A, D, S, R respectively Level (volume) can be set for sustain

rate curves

- Linear variation
- Exponential variation



Volume of each voice **ADPCM Sound Source** Envelope Volume of Sound (ADSR) each voice Buffer Noise **Sound Source CONFIDENTI Sony Computer Entertainment Inc.**

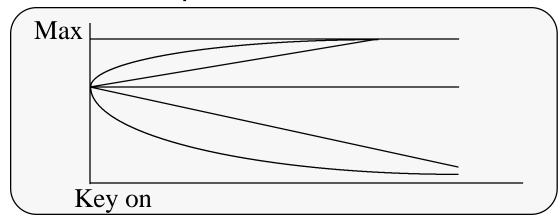
Volume of each voice (cont)

Volume of each voice

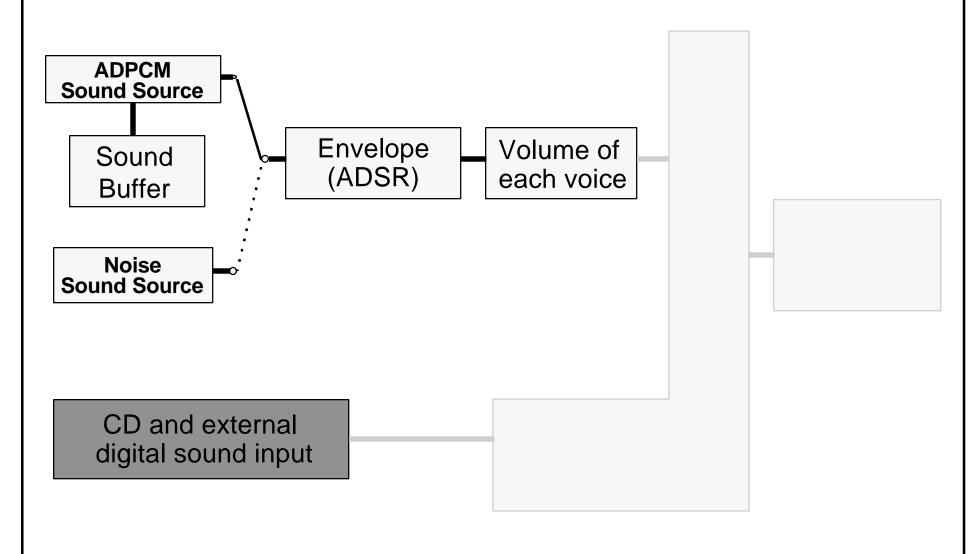
Time variation can be set independently of envelope

Curve characteristics

- Fixed value
- Linear variation
- Exponential variation



CD and external digital sound input (1)



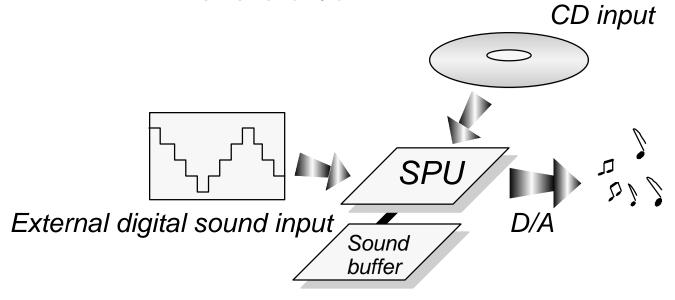


CD and external digital sound input (2)

CD and external digital sound input

Can mix via SPU

- Mixing on/off
- Mixing volume
- Reverb on/off





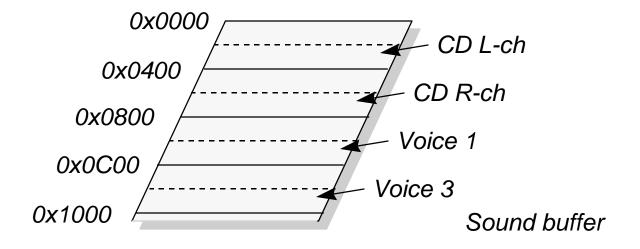
CD and external digital sound input (3)

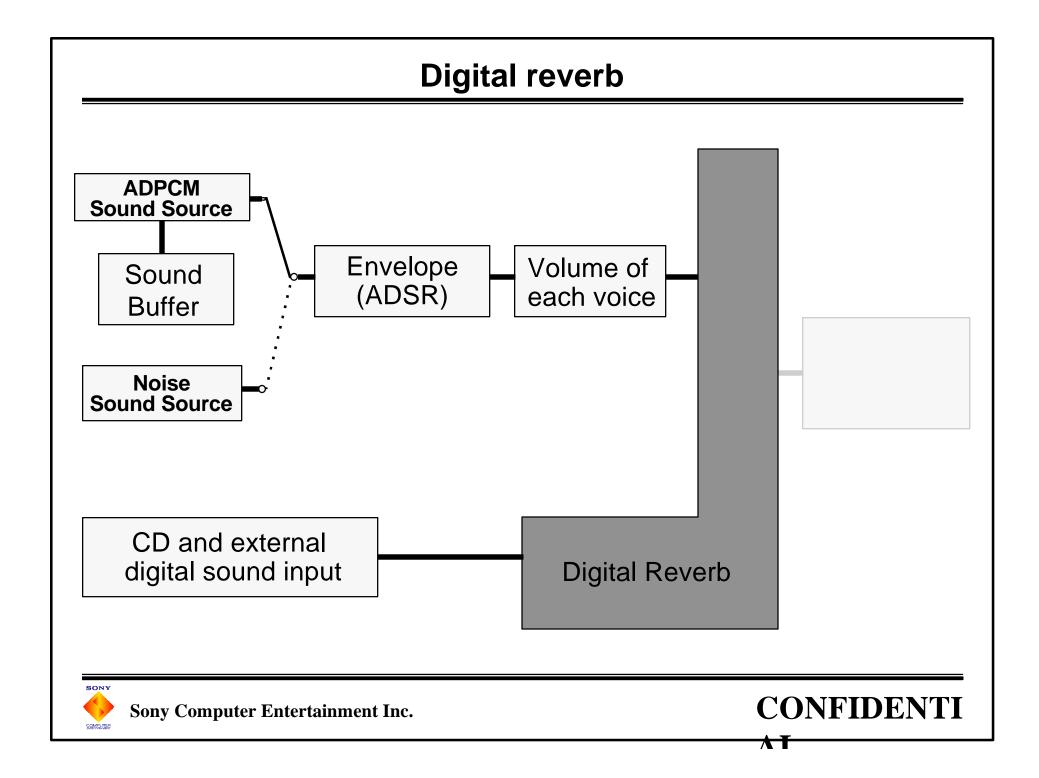
Feedback sound data to sound buffer

Can get the following data

- Voice 1 and 3 data after envelope processing
- Data before CD volume processing

Data format: 16 bit straight PCM





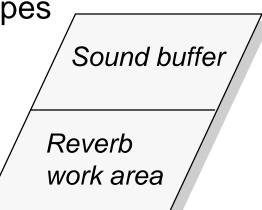
Digital reverb (cont)

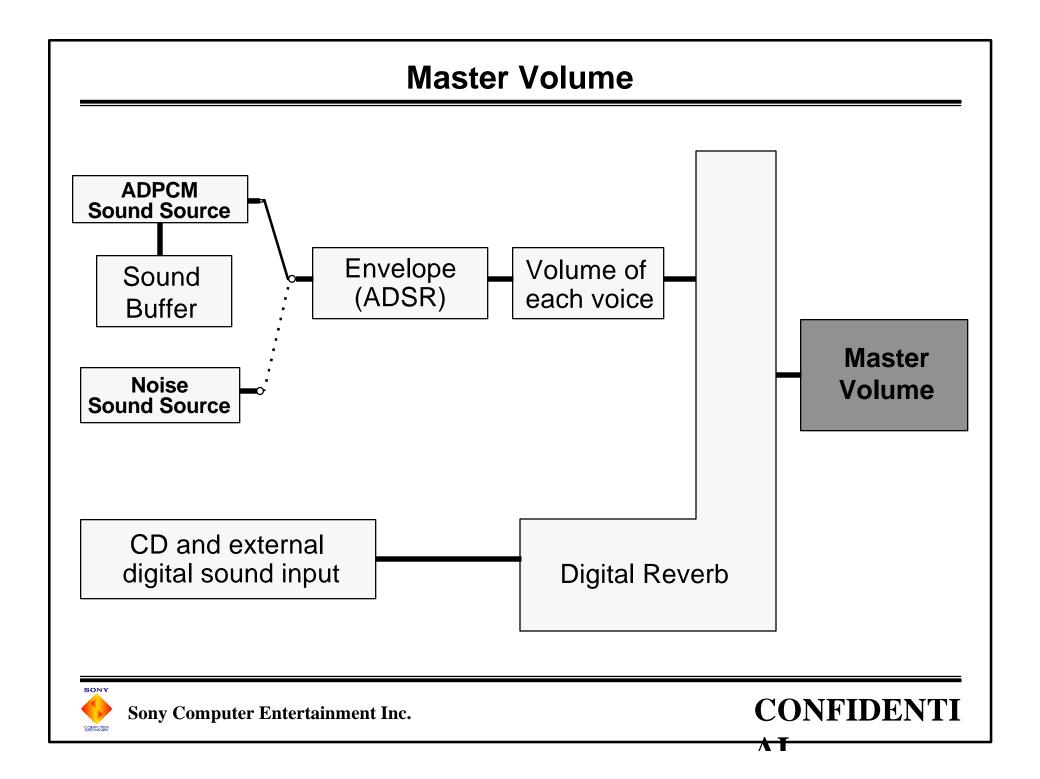
Digital reverb

- The work area can be inside the sound buffer
 - Share with waveform data
- Parameters can use multiple templates at the library side

→ Currently, nine types

Waveform data region :



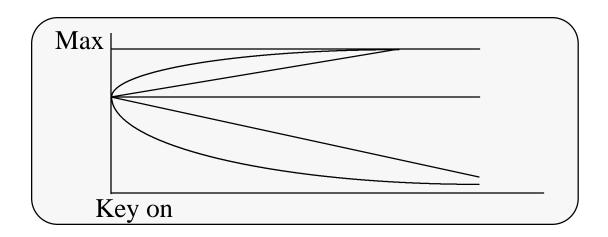


Master volume (cont)

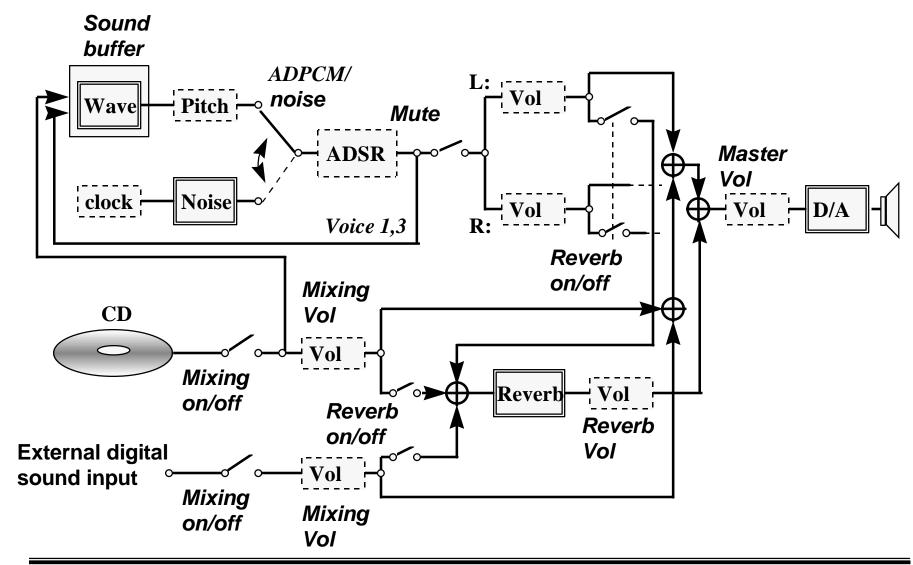
Master volume

Can set time variation in the same way for each voice Curve characteristics:

- Fixed value
- Linear variation
- Exponential variation



Signal flow





Basic Sound Library (Libspu) Overview

libspu

libspu

Provides the PlayStation's basic sound functions

- Sets each voice's attributes, generates sound, extinguishes sound
- Sets common attributes such as master volume, reverb, etc.
- Sound buffer and main memory data transfer
- When there's an SPU interrupt, supports callback function when DMA transfer ends

Functions not supported by libspu

Functions not supported by libspu

- Sequence processing
- Effects accompanying time operations
 - 1. Vibrato, pitch bend, portamento
 - 2. Fade in, fade out, etc.

- Strategies for solving this
 - 1. Use expanded sound library (libsnd)
 - 2. Use and create kernel library (libapi)

Initialization/end

Initialization/end

Spulnit() Initialize SPU

[SpuStart() Start SPU processing]

[SpuQuit() End SPU processing]

Common attributes

Common attributes

SpuSetCommonAttr() Set common attributes for all voices

SpuGetCommonAttr() Get common attributes for all voices

SpuSetMute() Sound mute on/off

SpuGetMute() Get status of sound mute

Each voice (1)

Each voice

Each attribute

SpuSetVoiceAttr() Set attribute of each voice

SpuGetVoiceAttr() Get attribute of each voice

SpuSetKey() Set key on/key off for each voice

SpuSetKeyOnWithAttr() Set key on with attribute for each voice

SpuSetReverbVoice() Set reverb on/off for each voice

SpuGetReverbVoice() Get reverb on/off for each voice



Each voice (2)

Noise Sound Source

SpuSetNoiseVoice() Set noise sound source on/off for

each voice

SpuGetNoiseVoice() Get noise sound source on/off for

each voice

SpuSetNoiseClock() Set noise sound source clock

SpuGetNoiseClock() Get noise sound source clock

Pitch LFO

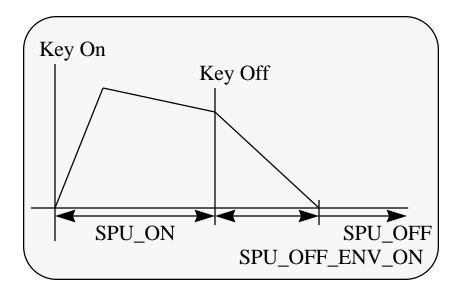
SpuSetPitchLFOVoice() Set pitch LFO on/off for each voice

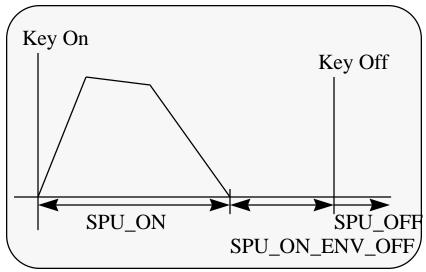
SpuGetPitchLFOVoice() Get pitch LFO on/off for each voice



Each voice (3)

Key on/key off/envelope status





SpuGetKeyStatus()
SpuGetAllKeysStatus()

Get key on/key off for specified voice Get key on/key off for all voices

Data transfer (1)

Data transfer

Transfer

SpuWrite() Transfer from main memory

to sound buffer

SpuWrite0() Clear sound buffer

SpuWritePartly() Transfer from main memory

to sound buffer (assuming

partially divided transfer)

SpuRead() Transfer from sound buffer to

main memory

SpuReadDecodeData() Transfer SPU decoded sound

data from sound buffer to

main memory



Data transfer (2)

Transfer Address

SpuSetTransferStartAddr() Set sound buffer's transfer

destination/transfer source

starting address

SpuGetTransferStartAddr() Get sound buffer's transfer

destination/transfer source

starting address

Transfer Mode

SpuSetTransferMode() Set transfer mode to sound buffer

SpuGetTransferMode() Get transfer mode to sound buffer



Data transfer (3)

Transfer synchronization

SpulsTransferCompleted() Get completion of transfer to

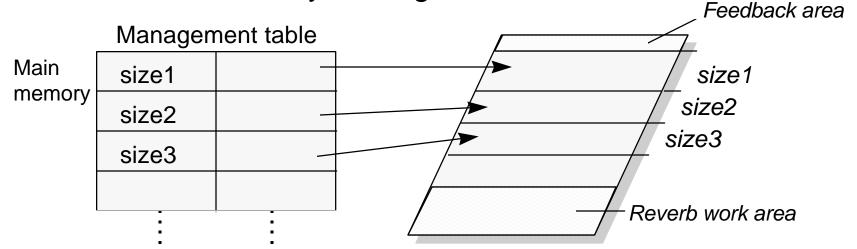
sound buffer

SpuSetTransferCallback() Set callback function when

DMA transfer completed

Sound buffer memory management

Sound buffer memory management



SpulnitMalloc()

SpuMalloc()
SpuMallocWithStartAddr()

SpuFree()

Initialize sound buffer memory management mechanism
Reserve area in sound buffer
Reserve area starting from specified address in sound buffer
Release reserved area in sound buffer



Reverb

Reverb

Types

Room, studio (small, medium, large), hall, space echo, echo, delay, half echo

Attributes

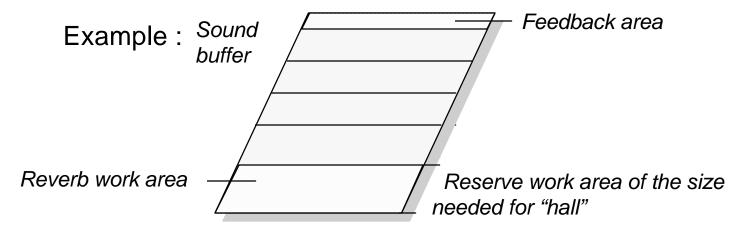
SpuSetReverb()
SpuGetReverb()
SpuSetReverbModeParam()
SpuGetReverbModeParam()
SpuGetReverbModeParam()
SpuSetReverbDepth()
Set reverb on/off



Reverb (cont)

Reverb work area

- In order to avoid conflicts with data, the size of the work area can be reserved in the sound buffer according to the type of reverb



SpuReserveReverbWorkArea()
SpulsReverbWorkAreaReserved()

SpuClearReverbWorkArea()

Reserve/release reverb work area Check if reverb work area is reserved, get whether or not reverb work area can be reserved Area of reverb work area



SPU interrupt

SPU interrupt

SpuSetIRQ()

SpuGetIRQ()

SpuSetIRQAddr()

SpuGetIRQAddr()

SpuSetIRQCallback()

Interrupt on/off

Get status of interrupt on/off

Set interrupt address

Get interrupt address

Set callback function when

interrupt

Expanded Sound Library (Libsnd) Overview

libsnd

libsnd

Structure based on libspu

- Can co-exist with libspu
- Use libapi for time control
 - Can also add changes specified by user
- Transfer of wave data to sound buffer
- Automatic performance of music data

Sound data format (1)

Score data format

- SEQ format
 - Convert from SMF (Standard MIDI File)
 - Has attributes and notation data needed for playing
 - Resident in main memory
- SEP format
 - SEQ group (up to 16 units can be integrated)
 - Created by SEQ2SEP.EXE in PC

Sound data format (2)

SMF ••• realtime processing using NRPN

When SMF is created, the following realtime processing can be done by combining Control Change-NRPN data

- Change VAB attributes data (Reverb, ADSR)
- Set repeat during music
- Marking function
 - → Can set callback function at program side

Sound data format (3)

Waveform data format

VAG format

 Convert from AIFF (Audio Interchanged File Format)

VAB format

- VAG group
- libsnd uses VAB as waveform data
- Divide into header part and body part using VABSPLIT.EXE in PC
- → Header part : each VAG's attributes (VH file)
- → Body part : Transfer to sound buffer (VB file)

libsnd controls

Voice manager

- Can explicitly set priority sequence in VAG units in VAB header
- Monitor key on/key off/envelope information
- SEQ/SEP can separate voice used by score data and single sound generated sound (SE)

Initialization / completion

SsInit, SsInitHot Initialize sound system, hot reset

SsSetTableSize Specify the area of a SEQ/SEP data

attribute table

SsSetTickMode Set tick

SsSetTickCallback Set tick (Tick Callback) callback

SsStart, SsStart Start the sound system

SsEnd Stop the sound system

SsQuit Quit sound system



VAB access functions

SsVabOpenHead Recognize a sound source header list

SsVabTransBody Transfer sound source data

SsVabTransCompleted Get VAB data transfer state

SsVabTransPartly Partly transfer sound source data

SsVabOpenHeadSticky Recognize a sound source header list

SsVabFakeHead Recognize a sound source header list

(re-recognition)

SsVabFakeBody Set up a library using the sound

source data in the sound buffer as the

given VAB ID (no transfer)

SsVabClose Close VAB data file



SEQ/SEP access functions

SsSeqCalledTbyT Interpret SEQ data and carry out

playing, called at each 1 tick

SsSeq(SsSep)Open Open SEQ/SEP data

SsSeq(SsSep)Close Close SEQ/SEP data

SsSeq(SsSep)Play Read (play) SEQ/SEP data

SsSeq(SsSep)Pause Pause the reading of SEQ/SEP data

SsSeq(SsSep)Replay Resume (replay) the reading of SEQ/SEP

data

SsSeq(SsSep)Stop Pause the reading of SEQ/SEP data

SsSeq(SsSep)SetCrescendo Crescendo

SsSeq(SsSep)SetDecrescendo Decrescendo

SsSeq(SsSep)SetRitardando Slow the tempo

SsSeq(SsSep)SetAccelerando Accelerate the tempo



SEQ/SEP access functions (cont)

SsSeq(SsSep)SetVol Set SEQ/SEP volume

SsSeqGetVol Obtain SEQ /SEP volume

SsSeqSetNext, SsSetNext Specify subsequent SEQ data

SsPlayBack Read SEQ/SEP data

SsSetTempo Set a tempo

SsSetLoop Set a song repetition count

SsIsEos Determine whether or not a song is

being played

SsSetMarkCallback Register a function to be called when a

mark is detected

SsSetReservedVoice Declare the number of voices to be

allocated by libsnd library



Independent controls, common attributes / acquisition functions

SsVoKeyOn(Off) Key on (off)

SsSet(Get)MVol Set (get) main volume

SsSet(Get)RVol Set (get) reverb volume

SsSet(Get)SerialAttr Set (get) serial input line attribute

SsSet(Get)SerialVol Set (get) serial input line volume

SsSet(Get)Nck Set (get) noise clock value

SsSetMute Set a mute

SsSetNoiseOn(Off) Set noise on (off)

SsSetMono Set monaural mode

SsSetStereo Set stereo mode

SsSetAutoKeyOffMode Set automatic KeyOff mode



Utility functions

SsUtKeyOn(Off) Key on (off) voice

SsUtKeyOnV(OffV) Key on (off) voice specified by voice

number

SsUtPitchBend Apply a pitch bend

SsUtChangePitch Change pitch

SsUtChangeADSR Change ADSR

SsUtSet(Get)VabHdr Set (get) VAB attribute header

SsUtSet(Get)ProgAtr Set (get) program attribute table

SsUtSet(Get)VagAtr Set (get) tone attribute table

SsUtSet(Get)VVol Set (get) voice volume

SsUtSet(Get)DetVVol Set (get) detailed value of voice

volume

SsUtAutoVol (not supported) Automatically change voice volume

SsUtAutoPan (not supported) Automatically change panning

Utility functions (cont)

SsUtReverbOn(Off) Turn on (off) reverb

SsUtSet(Get)ReverbType Set (get) reverb type

SsUtSetReverbDepth Set reverb depth

SsUtSetReverbFeedback Set feedback volume

SsUtSetReverbDelay Set delay volume

SsUtAllKeyOff Key off all voices

SsUtKeyOnV(OffV) Key on (off) voice specified by voice

number

SsUtFlush Execute KeyOn/KeyOff requests that

have been queued

SsUtGetVBaddrInSB Return address inside sound buffer

to which VAB data specified by VAB

id has been transferred

SsUtGetVagAddr Return a tone attribute table

SsUtGetVagAddrFromToneAuto Return SPU buffer address where

VAG data is stored



Sound Artist Tool Overview

Sound artist tool

- Tool structure
- Hardware
- Software
- Operating environment
- Sound data creation procedure
- Playing



Sound artist structure

Hardware

Sound Artist Board - DTL-H700

Software

Sound Artist Tool Version 1.0 - DTL-S710

DTL-H700 hardware

Sound Artist Board

- PlayStation sound processor: SPU
- Waveform memory: 4 Mbit DRAM
- Sampling frequency (fs): 44.1 KHz
- Optical disk output

DTL-S710 software

Sound Artist Tool Version 1.0

- Sequence data converter SMF2SEQ
- Waveform data converter AIFF2VAG
- Sound source bank editor,
 sound board driver SoundDelicatessen
- DA and XA data converter
 - RAW2DA
 - RAW2XA



Operating environment

PowerMac / Mac Quadra / Mac II series (NuBus)

Applicable OS: KanjiTalk7 or higher

Required memory: 8 MB or more (2 MB empty)

D/A converter with optical input (DAT, etc.)

Playing equipment with amp and speaker, etc.

Sequence data converter - SMF2SEQ

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Applicable format
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Standard MIDI file (Format 1)

Applicable statuses

Note On / Note Off / Program Change /

Pitch Bend

Applicable controls

Bank Change / Modulation / Panpot /

Main Volume / Expression / Damper Pedal /

External Effect Depth / NRPN Data /

RPN Data / Reset All Controllers

Output

SEQ file



Waveform data converter - AIFF2VAG

Applicable data format

AIFF file

Optional sampling rate

Resolution fixed at 16 bits

Monaural

Loop (marker) responsive

Output

VAG file



Sound source bank editor - SoundDelicatessen

Applicable format

VAG files and SEQ files

Output

VAB files

Program (max. 128 types)

Tone (max. 16 types per one program)

VAG (max. 254 types

Sound board driver function

VAG single-sound play (can specify pitch)

SEQ play



DA,XA data converter - RAW2DA,RAW2XA

Applicable data format

16bit straight PCM

(SoundDesignerII format)

<u>DA</u>

44.1KHz Stereo

XA

37.8KHz Stereo

Monaural

18.9KHz Stereo

Monaural

Output

DA file

XA file -> to MovPack (movie tool)

Sound data creation schedule

- Create SEQ
- Create VAG
- Create sound source set
- Check using sound source and SEQ playback
- Combine programs in development environment (PC/AT)