TWL-System NITRO-Composer

Sound Programmer Guide

2008/11/11

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1 Introduction

This document provides programmers with fundamental information about developing sound programs on the Nintendo DS (DS).

The setup for the NITRO-Composer environment is explained. This is followed by an example that shows how to implement a sound program. The last section explains the structure of the sound library, and lists the type of provided functions.

For a detailed explanation of each function, see the function reference.

2 Sound Program Development Environment

2.1 TWL-System Build Environment

NITRO-Composer is part of the TWL-System. By setting up the TWL-System build environment, you will be able to use NITRO-Composer.

See the TWL-System documentation for details.

2.2 File Organization

2.2.1 Library File

The TWL-SDK and TWL-System Library files listed below must be linked.

libsnd.a

libnnssnd.a

2.2.2 Header File

Header files that include definitions for created functions must be placed in an include statement using the following statement format.

#include <nnsys/snd.h>

By listing an include statement for the sound label file (SADL) created by the sound designer as shown below, the sound data can be specified using the label defined by the sound designer instead of the number.

#include "../data/sound_data.sadl"

2.2.3 Sound Data

All sound data sets are stored in a single sound archive file that has the .sdat file extension. Set the configuration to ensure that this sound archive file is stored in ROM. An example of how to store this file in ROM is shown in Chapter 3 Basic Implementation Example.

2.2.4 ARM7 Component

The ARM7 component is stored in the TWL-SDK. The sound functionality must be implemented in the ARM7 component. If the ARM7 component is not specified explicitly, the component that implements the sound functionality will be used.

3 Basic Implementation Example

This chapter demonstrates a basic implementation using NITRO-Composer for a project called simple. The simple project can be found in \$TwlSystem/build/demos/snd/simple.

3.1 Development Environment

This section explains how to set up the development environment.

3.1.1 Makefile

The following is an example makefile. Some of the comments have been omitted.

Code 3-1 Makefile

The basic elements of a makefile are not discussed in this section. Refer to the TWL-SDK and TWL-System manuals for makefile information. Setting the two MAKEROM_.* variables is crucial.

3.1.2 ROM Storage File

MAKEROM_ROMROOT defines the root directory in ROM. MAKEROM_ROMFILES defines the files to be stored in the root directory. In other words, the files that have the path /data/sound_data.sdat are stored in ROM using the path directory and filename.

3.1.3 Operational Procedure

If the makefile in section 3.1.1 Makefile is used, the program builds in the following sequence.

- 1. The main.c file registered in SRCS compiles.
- 2. The compiled file is linked to the library, which creates the ARM9 component main.
- 3. The ARM9 component main combines with the ARM7 component and sound archive, which creates main.srl.
- 4. The main.srl file becomes the executable file.

3.2 NitroMain Function

First, examine the NitroMain function in src/main.c. Some of the comments have been omitted.

Code 3-2 NitroMain Function

```
void NitroMain()
   OS_Init();
   GX_Init();
   // VBlank settings
   OS_SetIrqFunction(OS_IE_V_BLANK, VBlankIntr);
   (void)OS_EnableIrqMask( OS_IE_V_BLANK );
   (void)OS_EnableIrq();
   (void)GX_VBlankIntr(TRUE);
FS_Init( MI_DMA_MAX_NUM );
   // Initialize sound
   NNS_SndInit();
   heap = NNS_SndHeapCreate( & sndHeap, sizeof( sndHeap ) );
   NNS_SndArcInit( &arc, "/sound_data.sdat", heap, FALSE );
   (void)NNS_SndArcPlayerSetup( heap );
   NNS_SndArcStrmInit( STREAM_THREAD_PRIO, heap );
   // Load sound data
   (void)NNS_SndArcLoadSeq( SEQ_MARIOKART64_TITLE, heap );
```

```
(void)NNS_SndArcLoadSeqArc( SEQ_SE, heap );
 (void)NNS_SndArcLoadBank( BANK_SE, heap );
 // Initialize sound handles
NNS_SndHandleInit( &bgmHandle );
 NNS_SndHandleInit( &seHandle );
 // dummy pad read
 Cont = PAD_Read();
 //====== Main Loop
 while(1)
    u16 ReadData;
    SVC_WaitVBlankIntr();
    ReadData = PAD_Read();
    Trg = (u16)(ReadData & (ReadData ^ Cont));
    Cont = ReadData;
    if ( Trg & PAD_BUTTON_A ) {
// start BGM
        (void)NNS_SndArcPlayerStartSeq(&bgmHandle, SEQ_MARIOKART64_TITLE );
    if ( Trg & PAD_BUTTON_B ) {
// stop BGM
       (void)NNS_SndPlayerStopSeq( &bgmHandle, 1 );
    if ( Trg & PAD_KEY_UP ) {
// start SE
       (void)NNS_SndArcPlayerStartSeqArc( &seHandle, SEQ_SE, SE_COIN );
    //--- framework
    NNS_SndMain();
```

The key points are described in subsequent sections.

3.3 Basic Setup

This section explains fundamental functions (for example, library initialization).

3.3.1 Initializing the OS and Other Processes

First, initialize the OS and other basic processes.

```
OS_Init();

GX_Init();

// VBlank settings

OS_SetIrqFunction(OS_IE_V_BLANK, VBlankIntr);

(void)OS_EnableIrqMask( OS_IE_V_BLANK );

(void)OS_EnableIrq();

(void)GX_VBlankIntr(TRUE);

FS_Init( MI_DMA_MAX_NUM );
```

3.3.2 Initializing the Sound Library

The sound library must be initialized before any NNS_Snd functions are called.

Code 3-3 Initializing the Sound Library

```
NNS_SndInit();
```

3.3.3 Creating the Sound Heap

Create the heap that is used to store sound data.

Code 3-4 Sound Heap Creation

```
heap = NNS_SndHeapCreate( &sndHeap, sizeof( sndHeap ) );
```

The first argument is the starting address in memory that is used for the sound heap. The second argument is the size of the sound heap.

The return value is the heap handle. The heap handle is used to allocate memory from the sound heap.

3.3.4 Initializing the Sound Archive

Initialize the sound archive. The sound archive structure must be allocated statically.

```
NNS_SndArcInit( &arc, "/sound_data.sdat", heap, FALSE );
```

The first argument is the sound archive structure. The second argument is the path to the sound archive on the ROM file system.

The third argument is the heap needed to allocate memory for initialization of sound archives. The sound heap handle that was just created is used. Note that if the allocated memory is released, the sound archive will no longer be usable.

The fourth argument is a flag that controls the loading of symbol data in the sound archive. If the argument is set to True, symbol data is used for debugging. Set the argument to FALSE for standard initialization.

3.3.5 Setting Up the Player

Set up the player.

```
NNS_SndArcPlayerSetup( heap );
```

The player settings in the sound archive determine the setup.

Because the player setup requires memory, enter the heap handle as an argument.

3.3.6 Stream Library Initialization

To do stream playback, the stream library must be initialized.

```
NNS_SndArcStrmInit( STREAM_THREAD_PRIO, heap );
```

For details on streams, see Chapter 5 Stream Playback.

3.3.7 Sound Frame Processing

Perform sound library frame processing. This function should be called once for each frame. The location of the function call is not important.

```
NNS_SndMain();
```

3.4 Loading Sound Data

Before playing a sound sequence, the sound data must be loaded.

```
(void)NNS_SndArcLoadSeq( SEQ_MARIOKART64_TITLE, heap );
(void)NNS_SndArcLoadSeqArc( SEQ_SE, heap );
(void)NNS_SndArcLoadBank( BANK_SE, heap );
```

NNS_SndArcLoadSeq loads the data that is required to play the sequence SEQ_MARIOKART64_TITLE. This function concurrently loads the bank and waveform data, in addition to the sequence data.

NNS_SndArcLoadSeqArc loads the SE sequence archive. Because sequence archives are associated with multiple banks, bank and waveform data are not loaded automatically. The following function, NNS_SndArcLoadBank, loads the bank data for SE. With this function, both the bank and waveform data are loaded. Therefore, it is unnecessary to load waveform data separately.

3.4.1 Loading Groups

Sound data is not normally loaded alone, unlike the example. If the sound designer defines a group, the group can be loaded as shown in the following example.

```
(void)NNS_SndArcLoadGroup( GROUP_STATIC, heap );
```

The group defines which data sets to load. By calling NNS_SndArcLoadGroup, all the data sets are loaded at once. By loading groups, the data can be loaded without changing the code in the program.

3.5 Sequence Operation

3.5.1 Sound Handles

3.5.1.1 Using Sound Handles

A sound handle is required to work with a sequence.

```
NNSSndHandle bgmHandle;
NNSSndHandle seHandle;
```

Temporarily allocate a sound handle statically. Before using a sound handle, initialize it with the following function.

```
void NNS_SndHandleInit( NNSSndHandle* handle );
```

3.5.1.2 What Is a Sound Handle?

A sound handle is an object that controls the sequence after playback. A sound handle can control one sequence. If a sequence playback is successful, that sequence will be linked to a sound handle. From that point and until that link is disconnected, operations for that sound handle will operate the sequence.

3.5.1.3 Disconnecting the Sequence

Sometimes a sequence is manually or automatically disconnected. A sequence can be automatically disconnected if a second sequence attempts to start when a player can play only one sequence. The sequence played back is forcibly stopped. Under these circumstances, the sound handle is involuntarily disconnected from the sequence and disabled. Even if operations are performed on a disabled sound handle, no processing will occur.

This means that the programmer does not need to check to see if the sequence played back is still playing. Even if the same process is executed while the sequence is being played or if the sequence is stopped, there will be no problems such as a separate sequence being operated by error.

3.5.1.4 Tips for Creating Sound Handles

When sound is played back without any pauses, as with one-shot sound effects, the same sound handle can be used to play repeated sounds. All sounds can be played simultaneously as long as the number does not exceed the maximum number of simultaneous sequences that are allowed. Each parameter can be changed separately, immediately after the playback occurs.

Because continuous sounds such as background music or engine noises must be stopped, each of these sound effects requires a separate sound handle.

3.5.2 Sequence Playback

The following function plays back a sequence.

```
BOOL NNS_SndArcPlayerStartSeq( NNSSndHandle* handle, int seqNo );
```

seqNo is the sequence number, and the sounds are ordered as they appear in the sound archive.

If the function executes, the sequence links to a sound handle that is passed in as an argument. From this point, this sound handle can be used to carry out processes (for example, stopping the sequence).

If the sound handle is already linked to a sequence, the connection to the original sequence is disconnected and the sound handle connects to the new sequence. The sequence disconnected from the sound handle can no longer be controlled, so be careful. However, when a one-shot sound effect is generated, or when there will be no need to control the sequence later, there is no need for concern.

```
if ( Trg & PAD_BUTTON_A ) {
  (void)NNS_SndArcPlayerStartSeq(&bgmHandle, SEQ_MARIOKART64_TITLE );
}
```

3.5.3 Sequence Archive Playback

The following function plays back the sequence from the sequence archive.

```
BOOL NNS_SndArcPlayerStartSeqArc(
NNSSndHandle* handle, int seqArcNo, int index );
```

seqArcNo is the sequence archive number, which is the order of the sequences in the sound archive. index is the index number of the sequence in the sequence archive. The rest is the same as the playback of the sequence.

```
if ( Trg & PAD_KEY_UP ) {
  (void)NNS_SndArcPlayerStartSeqArc( &seHandle, SEQ_SE, SE_COIN );
}
```

3.5.4 Stopping the Sequence

The following function stops the sequence.

```
void NNS_SndPlayerStopSeq( NNSSndHandle* handle, int fadeFrame );
```

Enter the sound handle passed into NNSSndHandle when played back to handle. fadeFrame is the fadeout frame. The volume level gradually decreases over the specified number of frames.

```
if ( Trg & PAD_BUTTON_B ) {
  (void)NNS_SndPlayerStopSeq( &bgmHandle, 1 );
}
```

3.6 Other Demos

No descriptions for the functions that are used with the simple demo are described. This section contains an overview of other demos. The demo programs for NITRO-Composer are all stored under Twlsystem/build/demos/snd.

3.6.1 stream

Plays back streams. Stream playback is explained in Chapter 5 Stream Playback.

3.6.2 stream-2

Combines multiple stream data in real time and plays back. It registers the callback function that performs the combining process by using the NNS_SndArcStrmStartEx2 function.

3.6.3 stream-3

Applies effects to the stream and plays back. It registers the callback function that processes effects by using the NNS_SndArcStrmStartEx2 function.

3.6.4 moveVolume

Changes the volume of a sequence over a period of time. This demo uses the NNS_SndPlayerMoveVolumefunction to change the volume. It includes code for a fade-in playback.

3.6.5 onMemory

The demo uses the entire sound archives that are loaded into memory. For that purpose, the sound archives are initialized with NNS_SndArcInitOnMemory function.

3.6.6 reverb

A reverb demo that uses the capture feature. This demo uses ${\tt NNS_SndCaptureStartReverb}$ and ${\tt NNS_SndCaptureStopReverb}$.

3.6.7 effect

An effects demo that uses the sound capture feature. It passes the output through a simple low-pass filter (moving average).

This demo uses NNS_SndCaptureStartEffect and NNS_SndCaptureStopEffect.

3.6.8 outputEffect

Effects demo that uses the sound capture feature. Switches between the surround mode and the headphones mode for output.

It uses the NNS_SndCaptureStartOutputEffect and NNS_SndCaptureChangeOutputEffect functions.

3.6.9 sampling

Sampling demo that uses the sound capture feature. Calculates output levels using sampling data for the display.

It uses NNS_SndCaptureStartSampling and other functions.

3.6.10 waveout

Plays waveform data directly instead of using sequence playback. It plays back sounds recorded with a microphone. Uses NNS_SndWaveOutStart for the playback of waveform data.

3.6.11 micThrough

Uses the low-level stream library ${\tt NNS_SndStrm}$. Plays back real-time input from the microphone and applies effects to output sounds.

This demo uses NNS_SndStrmSetup and NNS_SndStrmStart.

3.6.12 driverInfo

Displays on-screen sound driver information.

The sound driver information is updated with NNS_SndUpdateDriverInfo, and the player information in the sound driver can be obtained with NNS_SndPlayerReadDriverPlayerInfo.

4 Heap Operations

4.1 Overview

In the simple demo, memory management functions are not used. In the simple demo, only the sound heap is created using NNS_SndheapCreate during initialization, and data is loaded into the memory.

The heap operations are explained in subsequent sections.

4.2 Memory Management Basics

For information about the basics of memory management, see the *NITRO-Composer Sound System Manual*. The following is a brief explanation about memory management.

4.2.1 Sound Heap and Player Heap

There are two heaps, the sound heap and the player heap. The sound heap is a stack-based heap that programmers use for loading and deleting data.

The player heap is used for loading data automatically during the sequence playback. Programmers do not need to work directly with the player heap.

4.2.2 Appropriate Usage of the Two Heaps

The sound heap loads relatively large blocks of data at system startup and during scene changes. The player heap loads relatively small blocks of data (for example, BGM data) during sequence playback.

Even though sound and player heap are generally used in this way, everything can be managed in the sound heap to improve the load efficiency.

4.3 Sound Heap Operations

Because the sound heap is stack-based, memory is allocated from the top to the bottom and is released from the bottom to the top. Memory is allocated automatically from the heap when sound data loads. To delete unwanted sound data, the memory regions must be released. There are two ways to release memory regions.

- · Clearing the heap
- · Restoring the heap to the previous state

4.3.1 Clearing the Heap

All of the sound data can be cleared from the heap. This process is very simple, but when the used function is executed, all sounds playing will stop. If the memory region that is used for the initialization of the sound archive is released, the sound archive will no longer be available.

To clear all sound data from the heap, call the NNS_SndHeapClear function.

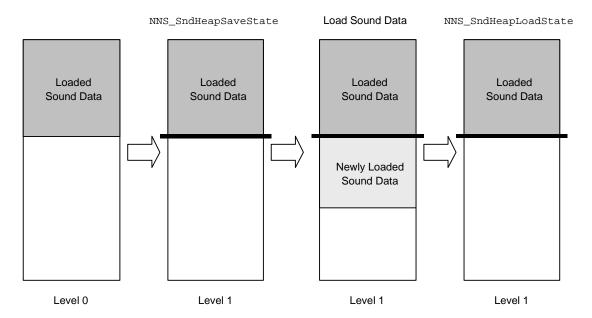
4.3.2 Restoring the Heap to the Previous State

Restoring the heap to the previous state is used more frequently than clearing the heap.

NNS_SndHeapSaveState saves the current state. After saving, the return value is the hierarchy level of the heap. The hierarchy level indicates the value for the state that was saved. The hierarchy level value can be used to restore the sound heap to the saved state.

After loading several blocks of sound data, calling NNS_SndHeapLoadState by passing the hierarchy level value will return the heap to the state that it was in immediately after the call was made to NNS_SndHeapSaveState. In other words, all data that was loaded after the call to NNS_SndHeapSaveState is deleted.

Figure 4-1 Restoring the Previous State



When the data is deleted, already loaded sounds will not be stopped.

Also, NNS_SndHeapSaveState can be called repeatedly, and the value of the hierarchy level will increase each time the function is called.

4.3.3 Multiple Sound Heaps

Usually, a single sound heap is used to restore the heap, but multiple sound heaps can be also used to restore different heap states. If there are several sound heaps, each sound heap can save and restore an individual state.

To use multiple sound heaps, you only need to create multiple heaps using NNS_SndHeapCreate. When allocating memory for a sound heap, the heap handle must always be specified. To allocate the memory, specify the sound heap by passing the heap handle return value as the argument of the function.

4.4 Player Heap Operations

Normally, the programmer does not have to manage the player heap. The player heap is created using the following function.

BOOL NNS_SndArcPlayerSetup(NNSSndHeapHandle heap);

The sound designer sets the size of the heap needed by the player. The amount of memory is allocated from the sound heap and is passed as an argument in the function to create the player heap.

4.4.1 Deleting the Player Heap

The player heap allocates memory from the sound heap. When the memory region that is allocated to the player heap is released, the player heap is automatically deleted.

5 Stream Playback

This chapter explains stream playback, which was not included in the simple demo.

5.1 Initializing the Stream Library

To play a stream, the stream library must be initialized.

```
NNS_SndArcStrmInit( STREAM_THREAD_PRIO, heap );
```

The first argument is the stream thread priority. The second argument is the sound heap handle, which is used to allocate the stream buffer.

5.1.1 Stream Thread

The stream thread is a thread that loads the data from ROM when necessary.

Data is loaded while stream playback is played; the sound stops if the data is not loaded in time.

Therefore, the data must be loaded quickly. When a stream thread needs to load data, a stream thread can interrupt other processes running on the main loop of the game and load the data.

If you want the stream thread to interrupt the main processing loop and load data, the stream thread must be assigned a higher thread priority. The main loop (main thread) has a default value of 16 Therefore, specify a value that is less than 16.

5.1.2 Stream Buffer

To play a stream, a buffer is required to load data. To play a single stream, the buffer needs approximately 2 to 4 KB of memory. This buffer uses part of the sound heap and is passed as the second argument in the function.

Note that once the stream buffer is released, the stream cannot be played.

5.2 Stream Operations

5.2.1 The Stream Handle

Just as a sound handle is necessary to play a sequence, a stream handle is required to play a stream. NNSSndStrmHandle strmHandle;

The stream handle should be allocated statically for now. Before using the stream handle, initialize a stream function with the following function.

```
void NNS_SndStrmHandleInit( NNSSndStrmHandle* handle );
```

In other respects, a stream handle is identical to a sound handle.

5.2.2 Stream Playback

To play back a stream, call the following function.

```
BOOL NNS_SndArcStrmStart( NNSSndStrmHandle* handle, int strmNo, u32 offset );
```

strmNo is the stream number that specifies which stream to play. offset specifies in milliseconds when to start playing back the stream data. Playback generally starts at the beginning of the stream, so set this parameter to zero.

Similar to a when using a sequence, the stream handle is bound to the stream if playback is successful.

5.2.3 Stopping a Stream

To stop a stream, call the following function.

```
void NNS_SndArcStrmStop( NNSSndStrmHandle* handle, int fadeFrames );
```

fadeFrames specifies the number of frames over which the volume should be gradually lowered before the stream stops. If fadeFrames is set to a value of zero, the stream stops immediately.

5.2.4 Pausing a Stream

There is no function to pause a stream. However, a process similar to pausing a stream is possible. The following procedure allows you to create the effect of pausing.

1. When you want to pause the stream, use the following function to get the current position of playback.

```
u32 NNS_SndArcStrmGetCurrentPlayingPos( NNSSndStrmHandle* handle );
```

- 2. Use NNS_SndArcStrmStop to stop the stream.
- 3. Restart the stream playback by passing the playback location as the offset argument of NNS_SndArcStrmStart and start the stream playback. The playback will start from where the stream stopped.

The stream will not restart at the precise position in the stream.

5.3 Avoid Interrupting Streams

With stream playback, data is loaded in real time. Therefore, if the sound is not loaded in time, the sound will be interrupted. Here are some tips to avoid interrupting streams.

5.3.1 Stream Thread

Stream data is loaded with the stream thread. The basic rule is to maintain stream thread processes without delay.

Here are some circumstances that cause delays in the stream thread processes.

5.3.1.1 Disabling Interrupts

Stream threads cannot run while interrupts are disabled. Interrupts should be disabled only for short periods of time.

5.3.1.2 DMA

Stream threads cannot run while DMA is running. By dividing a large DMA into chunks, the delay in stream thread processes can be reduced.

5.3.1.3 Interrupt Handler Process

Stream threads cannot run while an interrupt handler is being processed. Interrupt handler processes should last only for short periods of time.

5.3.1.4 Higher Priority Threads

Stream threads cannot run while higher priority threads are being processed. Higher priority thread processes need to last only for short periods of time or the priority of stream threads needs to be raised.

5.3.2 Accessing the DS Card/Backup Media

Stream data cannot be loaded while the DS Card or backup media is accessed. Therefore, divide the bandwidth between the DS Card and stream data.

5.3.3 Stream Buffer

If the buffer size for stream playback is large, the sound will not be interrupted even if the stream thread process speed is reduced. However, when the buffer size is larger, the process for one stream thread requires more bandwidth, and a negative effect on the lower priority threads may occur.

5.3.4 Simultaneous Playback

When multiple streams are played simultaneously, the stream thread processes increase and require more bandwidth. The sound may be interrupted even with a short delay in the processes. Exercise caution when playing back multiple streams simultaneously.

6 Cautions

6.1 Sound Processes in Sleep Mode

When entering Sleep Mode, the library automatically performs the processes to stop the sound features. In other words, the programmer simply calls the PM_GosleepMode function to enter Sleep Mode and the necessary processes are carried out by the library.

Similarly, the library carries out all the processes needed to restart sound features when returning from Sleep Mode. However, pay attention to the following.

6.1.1 Sequence Playback

The sequence process resumes from the place where it was before entering Sleep Mode, but the channel that was playing sounds in the sequence restarts from the very beginning of its waveform data.

If this is a problem, the programmer must pause the playback before Sleep Mode and cancel the pause after Sleep Mode. However, the channel that was playing sounds in the sequence will not play upon returning from Sleep Mode.

6.1.2 Stream Playback

There is a process that stops steam playback before entering Sleep Mode and resumes play when returning from Sleep Mode. At this time, any data left in the Stream buffer will be destroyed and sounds will partially skip.

There is no effective workaround to this problem.

6.1.3 Sound Capture

A process is performed that stops capture before entering Sleep Mode and resumes capture upon returning. At that time, any data left in the capture buffer will be destroyed, so there will be no capture data when returning from Sleep Mode.

There is no effective workaround to this problem.

6.2 Cautions When Using TWL-SDK SND Functions

NITRO-Composer is implemented by using the TWL-SDK SND functions. For this reason, the following cautions must be observed when using the TWL-SDK SND functions directly.

6.2.1 Using Player

NITRO-Composer uses sequence command functions such as $\mathtt{SND_StartSeq}$ to play back sequences.

While using NITRO-Composer, calling sequence command functions is prohibited.

6.2.2 Using Channels

When the NITRO-Composer NNS_SndLockChannel function is called, NITRO-Composer can no longer use the specified channel. You can use SND functions to manipulate this channel only.

6.2.3 Using Sound Capture

When the NITRO-Composer NNS_SndLockCapture function is called, NITRO-Composer can no longer use the specified sound capture. You can use SND functions to manipulate this sound capture only.

However, the NNS_SndLockChannel function must be called simultaneously because sound capture uses Channel 1 and Channel 3.

6.2.4 Using Sound Alarm

Because NITRO-Composer uses the sound alarm internally, call the NITRO-Composer NNS_SndAllocAlarm function and get the number of an alarm that can be used for a sound alarm when using NITRO-Composer.

When you are finished using an alarm, release it with NNS_SndFreeAlarm.

6.2.5 Using the SNDEX Functions

The SNDEX functions can be used independently of NITRO-Composer.

However, note that there are functions that impact NITRO-Composer sound output, such as the SNDEX_SetVolume function.

6.3 Cautions When Using DSP

DSP can be used independently of NITRO-Composer.

However, note that when DSP output volume is increased with the <code>SNDEX_SetDSPMixRate</code>, the output volume from NITRO-Composer decreases.

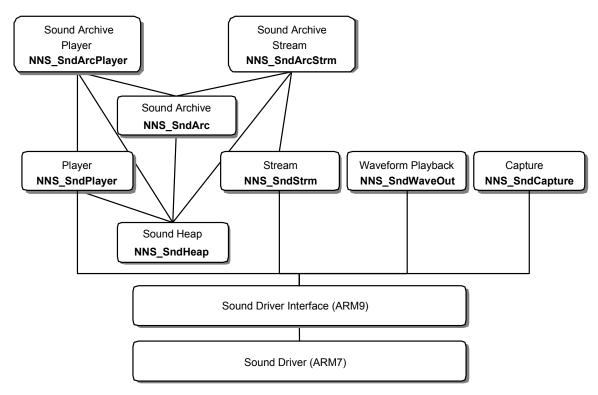
7 Library Organization

This chapter describes the organization of the NITRO-Composer library.

7.1 Library Organization

The NITRO-Composer library is composed of several libraries as shown below.

Figure 7-1 Library Organization Diagram



The libraries with NNS_Snd prefixes are the sound libraries included in TWL-System. Usually, the programmer uses these library functions. The libraries are described in the following sections.

7.2 Player Library

The Player library is the most basic library for playing back sequences. The function prefix set is NNS_SndPlayer.

This library is used to change parameters for sequences and stopping sequences. However, the sound archive player functions from the upper library are used for only playback. The sequence playback functions in the player library are used to simply run sequences. For actual sequence playback, complex processing (for example, data loading) is required. Therefore, sequence playback is processed with the sound archive player.

The player library also includes some functions that operate the sound handles. The function set prefix is NNS_SndHandle. The sound handle and player are closely related. Refer to section 3.5.1 Sound Handle for an overview of the sound handle.

The main functions of the Player library are shown in Table 7-1.

Table 7-1 Player Library Functions

Function Name	Description
NNS_SndPlayerStopSeq	Stops the sequence.
NNS_SndPlayerPause	Pauses or restarts the sequence.
NNS_SndPlayerSetTempoRatio	Changes the tempo of the sequence.
NNS_SndPlayerSetVolume	Changes the volume of the sequence.
NNS_SndPlayerSetTrackVolume	Changes the volume of the sequence track.
NNS_SndPlayerSetTrackPitch	Changes the pitch of the sequence track.
NNS_SndPlayerSetTrackPan	Changes the pan (location) of the sequence track.

7.3 Sound Archive Player Library

The Sound Archive Player library plays back the sequence using the sound archive. The function set prefix is NNS_SndArcPlayer.

The Sound Archive Player library is located in the upper-level libraries of the player library and the sound archive library. Sequences can be easily played back using these functionalities.

The main functions of the Sound Archive Player library are shown in Table 7-2.

Table 7-2 Sound Archive Player Library Functions

Function Name	Description
NNS_SndArcPlayerSetup	Sets up the player using the settings in the sound archive.
NNS_SndArcPlayerStartSeq	Plays back the sequence.
NNS_SndArcPlayerStartSeqArc	Plays back the sequence archive.

7.4 Sound Archive Stream Library

Use the Sound Archive Stream library to play stream data in a sound archive. The function names begin with NNS_SndArcStrm.

This library is an upper-level library of the stream library and the sound archive library. The Sound Archive Stream library uses the functionality of the stream library and the sound archive library to make playing streams simple.

The main functions of the Sound Archive Stream library are shown in Table 7-3.

Table 7-3 Sound Archive Stream Library Functions

Function Name	Description
NNS_SndArcStrmInit	Initializes the sound archive stream library.
NNS_SndArcStrmStart	Plays a stream.
NNS_SndArcStrmStop	Stops a stream.

7.5 Stream Library

The Stream Library is a low-level library for the playback of streams. The function names begin with NNS_SndStrm.

This library is used to play data that is received through communications in real time, and for stream playback of waveform data in unique stream data format.

The main functions of the Stream library are shown in Table 7-4.

Table 7-4 Stream Library Functions

Function Name	Description
NNS_SndStrmInit	Initializes a stream.
NNS_SndStrmAllocChannel	Allocates a channel for stream playback.
NNS_SndStrmFreeChannel	Frees a stream playback channel.
NNS_SndStrmSetup	Prepares for stream playback.
NNS_SndStrmStart	Plays a stream.
NNS_SndStrmStop	Stops a stream.

7.6 Sound Archive Library

The Sound Archive library loads sound data in the sound archives and retrieves parameters. The function set prefix is NNS_SndArc.

The main functions of the Sound Archive library are shown in Table 7-5.

Table 7-5 Sound Archive Library Functions

Function Name	Description
NNS_SndArcInit	Initializes a sound archive.
NNS_SndArcLoadGroup	Loads sound data in units of the group.

7.7 Sound Heap Library

The Sound Heap library manages the sound heap. The function set prefix is NNS_SndHeap.

The main functions of the Sound Heap library are shown in Table 7-6.

Table 7-6 Sound Heap Library Functions

Function Name	Description
NNS_SndHeapCreate	Creates the sound heap.
NNS_SndHeapClear	Clears all heap memory.
NNS_SndHeapSaveState	Saves the heap state.
NNS_SndHeapLoadState	Restores the heap state.

7.8 Capture Library

Both TWL and Nintendo DS are equipped with a sound capture feature. The Capture library is used for generating effects using the capture feature (for example, reverb). The function set prefix is NNS_SndCapture.

The main functions of the Capture library are shown in Table 7-7.

Table 7-7 Capture Library Functions

Function Names	Description
NNS_SndCaptureStartReverb	Starts the reverb.
NNS_SndCaptureStopReverb	Stops the reverb.
NNS_SndCaptureStartEffect	Starts the effect.
NNS_SndCaptureStopEffect	Stops the effect.

7.9 Waveform Playback Library

The Waveform Playback library provides functionality to play waveform data without using sequence playback. The function set prefix is NNS_SndWaveOut.

The Waveform Playback library is used for playing sampled data captured with a microphone. Use this stream library to play back waveform data that is generated in real time.

The main functions of the Waveform Playback library are shown in Table 7-8.

Table 7-8 Waveform Playback Library Functions

Function Names	Description
NNS_SndWaveOutAllocChannel	Allocates a channel for playing waveform.
NNS_SndWaveOutStart	Starts waveform playback.
NNS_SndWaveOutStop	Stops waveform playback.

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