AN14531

How to Implement USB Audio Mixer on LPC55S69

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Document information

Information	Content
Keywords	AN14531, LPC55S69, USB audio, audio mixer, audio speaker
Abstract	This application note describes how to implement a USB audio mixer on the NXP LPC55S69. It introduces the steps to implement a USB audio mixer based on the NXP SDK examples, the management of the audio ringbuffer, and the measurement of USB audio latency.

1 Introduction

In the application of wired gaming headsets or wireless gaming headsets with USB dongle, a USB audio mixer is required. For example, when playing online games, we want to hear the sounds in the game and the sounds of teammates. It requires that a USB interface can be enumerated as two USB speakers, to hear the sounds of multiple USB audio speakers synchronously, and also to support the function of USB microphone. Taking the wired gaming headset as an example, it consists of two USB audio speaker: **Chat speaker** and **Game speaker**. The USB audio mixer implemented in this application note simply merges the USB audio data received by two USB audio speakers, and then transmits the merged audio data to the audio codec through the **I²S** interface. Then, the customer can hear both sounds at the same time. This application note describes how to implement the USB audio mixer function on the LPC55S69. This USB audio mixer can support USB audio class 1.0 (UAC1.0) and USB audio class 2.0 (UAC2.0).

The solution based on KL27 and NXH3670 wireless gaming headset of NXP has implemented the function of USB audio mixer, but this solution only supports UAC1.0 and the KL27 SDK version used is SDK 2.7. This application note introduces how to implement the USB audio mixer function on the LPC55S69. This USB audio mixer can support the following functions:

- Supporting UAC1.0 and UAC2.0
- Supporting USB audio synchronous mode
- Supporting high-speed USB and full-speed USB interfaces of LPC55S69
- Supporting LPC55S69 SDK v2.15
- Supporting 48 K/16-bit stereo audio format

The USB audio mixer in this application note is implemented based on the LPC55S6-EVK board and `usb_composite_audio_unified_bm` example in LPC55S69 SDK 2.15. The relevant code of the USB dongle mixer in NXH3670 SDK G9.2 is ported to the USB audio example of LPC55S69 SDK 2.15, and the USB descriptor of UAC2.0 is added. The following chapters introduce how to implement the USB audio mixer on LPC55S69.

2 Implementation of USB audio mixer

2.1 LPC55S69 SDK example

The `usb_composite_audio_unified_bm` example in the LPC55S69 SDK 2.15 has implemented a USB composite device. It can support a USB audio speaker, a USB recorder, and a USB HID device. We can use the original USB audio speaker in the SDK as the Chat speaker of the USB audio mixer. Therefore, to realize the function of the USB audio mixer, we only need to add a Game speaker based on LPC55S69 SDK example.

2.2 Add a USB audio speaker

This section describes how to modify the USB description, related variables, and functions to add a USB audio speaker.

2.2.1 Add corresponding USB descriptor

To support a new USB audio speaker (Game speaker) interface, add USB descriptors as shown in Table 1.

Table 1. USB descriptor for USB audio game speaker

USB descriptor type	USB descriptor name	Variable in USB descriptor structure	Descriptor length in UAC1.0	Descriptor length in UAC2.0	Comments
Interface Association Descriptor	Standard Interface Association Descriptor	`iadAudio`	8	8	
Audio Control (AC) Interface Descriptor	Standard AC Interface Descriptor	`control`	9	9	**UAC1.0** byte 7: `bInterface Protocol`, not used, must be set to 0. **UAC2.0** byte 7: `IP_VERSION_02_00`
Audio Class-Specific AC Interface Descriptor	Class-Specific AC Interface Header Descriptor	`controlSub`	9	9	**UAC1.0** byte 7: `bInCollection` byte 8: `baInterfaceNr` UAC2.0: byte 5: `bCategory` byte 8: `bmControls`
	Clock Source Descriptor	`controlSpkr.clock Source`	0	8	UAC1.0 has no CLOCK Source Descriptor.
	Input Terminal Descriptor	`controlSpkr.input Terminal`	12	17	**UAC2.0** byte 7: `bCSourceID` byte 14-15: `bmControls`
	Feature Unit Descriptor		0	18	UAC1.0 has no Feature Unit.
	Output Terminal Descriptor	`controlSpkr.output Terminal`	9	12	**UAC2.0** byte 8 `bCSourceID` byte 9 `bmControls`
Endpoint Descriptor	Endpoint Descriptor	`controlInterrupt Endpoint`	9	0	UAC2.0 does not require an interrupt In endpoint descriptor.
Audio Streaming Interface Descriptor	Standard AS Interface Descriptor (alt 0)	`streamSpkr.altSet0`	9	9	**UAC1.0** byte 7: `bInterface Protocol`, not used, must be set to 0. **UAC2.0** byte 7: `IP_VERSION_02_00`
	Standard AS Interface Descriptor (alt 1)	`streamSpkr.altSet1`	9	9	**UAC1.0** byte 7: `bInterface Protocol`, not used, must be set to 0. **UAC2.0** byte 7: `IP_VERSION_02_00`

Table 1. USB descriptor for USB audio game speaker*...continued*

USB descriptor type	USB descriptor name	Variable in USB descriptor structure	Descriptor length in UAC1.0	Descriptor length in UAC2.0	Comments
	Class-Specific AS Interface Descriptor	`streamSpkr.as Interface`	7	16	**UAC1.0** byte 3: `bTerminalLink` byte 4: `bDelay` byte 5: `wFormatTag` **UAC2.0** byte 4: `bmControls` byte 5: `bFormatType` byte 6-9: `bmFormats` byte 10: `bNrChannels` byte 11-14: `bmChannel Config` byte 15: `iChannelNames`
	Class-Specific AS Format Type Descriptor	`streamSpkr.audio Format`	11	6	**UAC1.0** byte 4: `bNrChannels` byte 7 `bSamFreqType` byte 8-11 `tSamFreq`
Audio Streaming Endpoint Descriptors	Standard AS Isochronous Audio Data Endpoint Descriptor	`streamSpkr.iso Endpoint`	9	7	**UAC1.0** byte 7: `bRefresh` byte 8: `bSynchAddress`
	Class-Specific AS Isochronous Audio Data Endpoint Descriptor	`streamSpkr. specificIso Endpoint`	7	8	**UAC1.0** byte 4: `bLockDelayUnits` byte 5-6: `wLockDelay` **UAC2.0** byte 3: `bEndpoint Address` byte 5: `wMaxPacketSize`

In this application note, the `usb_class_audio_headphones_device_descriptor_t` structure is used to represent the USB descriptor of the Game speaker. Figure 1 shows the member variables contained in this structure.

```
typedef struct _usb_class_audio_headphones_device_descriptor {
    usb_descriptor_interface_association_t iadAudio;
    usb_descriptor_interface_t control;
    usb_descriptor_class_specific_ac_interface_headphones_t controlSub;

    usb_class_audio_control_io_descriptor_group_t controlSpkr;
#if USB_DEVICE_CONFIG_AUDIO_CLASS_2_0
#else
    usb_descriptor_ac_interrupt_endpoint_t controlInterruptEndpoint;
#endif

    usb_class_audio_stream_descriptor_group_t streamSpkr;
#if USB_DEVICE_AUDIO_USE_SYNC_MODE
#else
    usb_descriptor_as_iso_sync_endpoint_t feedbackEndpointSpkr;
#endif
} usb_class_audio_headphones_device_descriptor_t;
```

Figure 1. `usb_class_audio_headphones_device_descriptor_t` structure

As shown in Table 1, the descriptors of USB audio game speaker include the interface association descriptor, audio control interface descriptor, audio stream interface descriptor, and audio stream endpoint descriptor. In addition, when comparing UAC1.0 and UAC2.0, the content of the same descriptor may be different. The

Comments column in Table 1 also briefly lists the differences between the descriptors of UAC1.0 and UAC2.0. For a more specific comparison, see the USB spec (USB Audio Class 1.0 and USB Audio Class 2.0) and AN14531SW.

2.2.2 Modify other configurations

In addition to adding the USB descriptors in Table 1, it is also necessary to modify the number of interfaces and endpoints, as well as modify other variables and related callback functions.

2.2.2.1 Modify the number of interfaces and endpoints

To implement a USB audio Game speaker, add two USB interfaces: the audio control interface and the audio stream interface. The new interface configuration is as below.

- #define USB_AUDIO_CHAT_CONTROL_INTERFACE_INDEX (0)
- #define USB_AUDIO_RECORDER_STREAM_INTERFACE_INDEX (1)
- #define USB_AUDIO_CHAT_SPEAKER_STREAM_INTERFACE_INDEX (2)
- **#define USB_AUDIO_GAME_CONTROL_INTERFACE_INDEX (3)**
- **#define USB_AUDIO_GAME_SPEAKER_STREAM_INTERFACE_INDEX (4)**
- #define USB_HID_CONSUMER_CONTROL_INTERFACE_INDEX (5)

It is also necessary to add two USB endpoints for the USB audio Game speaker, one is the audio control endpoint and the other is the audio stream endpoint. The new endpoint configuration is as below.

- #define USB_AUDIO_CHAT_CONTROL_ENDPOINT (6)
- #define USB_AUDIO_CHAT_SPEAKER_STREAM_ENDPOINT (1)
- #define USB_AUDIO_RECORDER_STREAM_ENDPOINT (3)
- **#define USB_AUDIO_GAME_CONTROL_ENDPOINT (7)**
- **#define USB_AUDIO_GAME_SPEAKER_STREAM_ENDPOINT (2)**
- #define USB_HID_CONSUMER_CONTROL_ENDPOINT (4)

2.2.2.2 Modify related variables and functions

In addition to modifying the number of interfaces and endpoints, add some variables related to the USB audio Game speaker used in the USB enumeration process, as shown in Table 2.

Table 2. Added variables for USB audio Game speaker

USB Game speaker related variables
`g_UsbDeviceAudioGameSpeakerEntity`
`g_UsbDeviceAudioGameSpeakerEntities`
`g_UsbDeviceAudioGameSpeakerControInterface`
`g_UsbDeviceAudioGameSpeakerInterfaces`
`g_UsbDeviceAudioInterfaceListGameSpeaker`
`g_UsbDeviceAudioClassGameSpeaker`
`g_CompositeClassConfig`

For more information about the modification of related variables, see AN14531SW.

To handle the requests related to the USB audio game interface, modify functions shown in Table 3.

Table 3. Functions that need to be modified

Function name	Description
`USB_DeviceCallback()`	Add the processing of Set Interface request of game interface
`APPInit()`	Add `audioGameSpeakerHandle` related configuration
`USB_DeviceAudioRequest()`	Update audio specific request handling
`USB_DeviceAudioSpeakerSetInterface()`	Add processing for set Game interface request

Note: *Table 2 and Table 3 only show some of the variables and functions to be modified or added. For more code modifications, see AN14531SW.*

2.3 ringbuffer management

This application note uses the `ringbuffer` management mechanism from the NXH3670 SDK (USB dongle mixer project). Each audio speaker uses a `ringbuffer` to manage audio data. Figure 2 shows the flow of audio data.

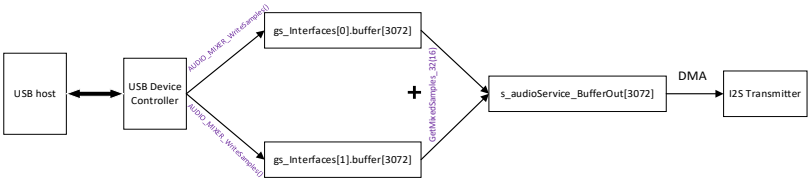

Figure 2. Audio data flow

The following source files are used for the management of audio `ringbuffer`.

- *audio_ringbuffer.h*
- *audio_ringbuffer.c*
- *audio_mixer.h*
- *audio_mixer.c*
- *audio_tx.h*
- *audio_tx.c*

The *audio_mixer.c* file defines the `gs_Interfaces[2]` structure array, which includes two structures corresponding to the two audio speakers: Chat and Game. Each structure has a buffer variable. Both buffers are `ringbuffer` with a length of 3072 bytes, which are used to store the USB audio data received by the Chat and Game speakers. In the USB interrupt service routine, the `AUDIO_MIXER_WriteSamples()` function is called to copy the received USB audio packet to `gs_Interfaces[x].buffer[3072]`. When the filling value of the ringbuffer reaches the set threshold, DMA transmission starts. Before each new DMA transfer, the `audio_GetAndTransmitSamples` function is called (the `audio_GetAndTransmitSamples` function calls the `GetMixedSamples_32(16)` function) to merge the audio data in `gs_Interfaces[0].buffer` and `gs_Interfaces[1].buffer`, and copy the merged data to the `s_audioService_BufferOut[3072]` array. Then, DMA transfer is started to move the mixed audio data to the I2S TX FIFO for playback.

2.4 USB audio synchronization

The USB audio device in this application note works in synchronous mode, which means that the USB device must follow the Start Of Frame (SOF) signal of the USB host to match the playback speed of the audio data on the USB device side with the USB SOF signal. In this application note, the Ctimer timer is used to capture the USB SOF signal, and the fractional division coefficient of the audio clock (Audio PLL) is adjusted in the

Ctimer capture interrupt service routine to match the audio clock with the USB SOF signal. For the specific implementation method, see the `CTIMER_SOF_TOGGLE_HANDLER_PLL()` function in AN14531SW.

2.5 Audio latency measurement

The audio latency in this application note is related to the filling level of the ringbuffer when the DMA transfer starts. The length of each DMA transfer is the length of a USB audio packet. For the UAC1.0 device with two channels 48K/16bit audio format, the length of each USB audio packet is 192 bytes, so the length of each DMA transfer is 192 bytes. As shown in Figure 3, when the filling level of the ringbuffer is greater than or equal to four DMA transfer lengths. DMA transfer starts, that is, at least four USB audio packets must be received before DMA transfers can be started. For UAC1.0 devices, the USB host sends a USB audio packet every 1ms, that is, the audio latency from the USB host starting to send audio data to the USB device starting to transmit audio data to the Codec is about 3-4 ms. For high-speed UAC2.0 devices, the minimum interval of USB audio packets is 125 µs, that is, a USB audio packet is sent in each microframe, and the size of this audio packet is 24 bytes. In the LPC55S69 SDK, the USB audio packet interval of the high-speed UAC2.0 device is set to four microframes by default. That is, an audio packet is sent every 0.5 ms, and each audio packet is 96 bytes, so the audio latency should be 1.5 - 2 ms. For high-speed UAC2.0 devices, if the audio packet interval is adjusted to 125 µs and the initial fill level is adjusted to 2-3 USB audio packet lengths, the audio latency can theoretically be less than 1 ms.

```
if (!inInterface->isActive && (inInterface->fillLevel >= (AUDIO_GetTxDMATransferSize() * 4) )) {
    inInterface->isActive = true;
    gs_ActiveCount++;

    /* if this was the first interface to become active, then the mixer is ready to start outputting samples */
    if ((gs_ActiveCount == 1) && (gs_ReadyStateChangedCb != NULL)) {
        gs_ReadyStateChangedCb(true);
    }
}
}
```

Start I2S TX DMA transfer

Figure 3. Timing to start DMA transfer

For full-speed USB, you can use a logic analyzer to measure the USB_DP/DM signal and the I^2S signal to calculate the USB to I^2S audio latency, as shown in Figure 4.

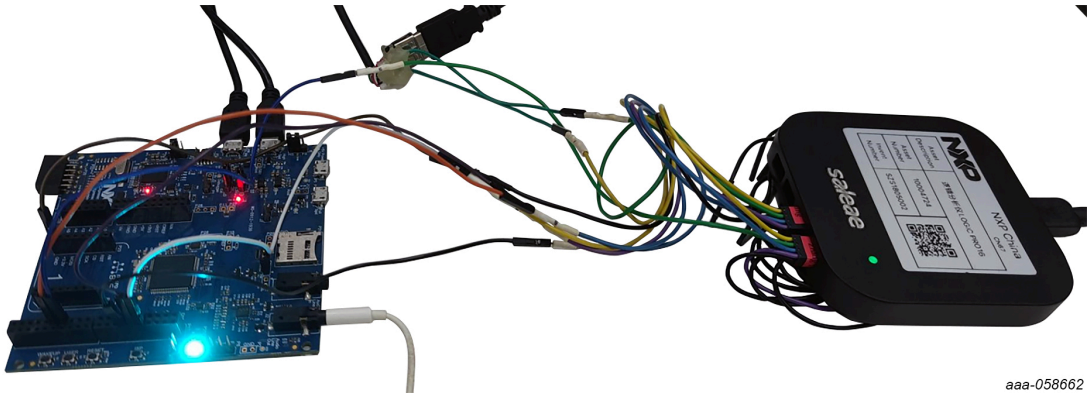

Figure 4. Measure USB to I^2S audio latency using a logic analyzer

When the threshold of the initial fill value of the ringbuffer is *AUDIO_GetTxDMATransferSize()* * 4, that is, DMA transfer starts after receiving four USB audio packets. Figure 5 shows the audio delay measurement results of UAC1.0. The audio latency from USB to I^2S is 3.4 ms.

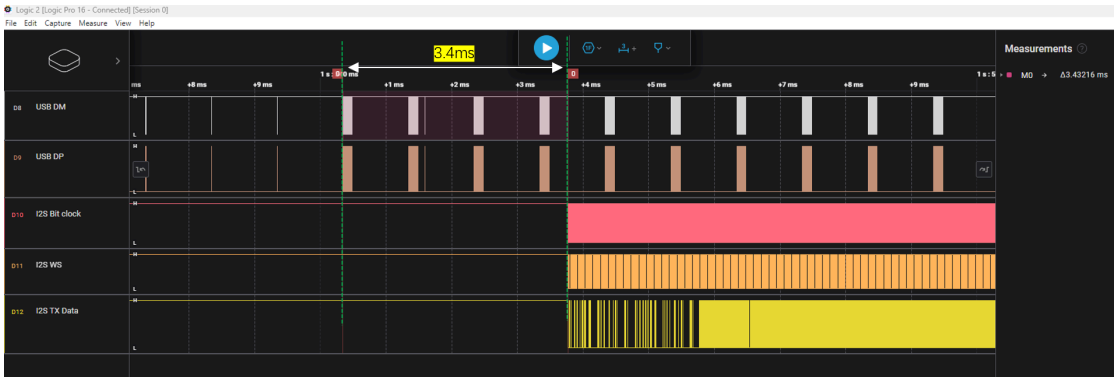

Figure 5. Measure audio latency of UAC1.0 device

When using the high-speed USB interface and enabling UAC2.0 mode, the high-speed USB clock is 480 M. Since the maximum sample rate of the logic analyzer used is 500 M, it cannot accurately sample USB DP/DM signals. Currently, we can use the GPIO toggle method to measure the USB to I^2S latency. Toggle a GPIO (P1_7) every time a USB audio packet is received.

```
case kUSB_DeviceAudioEventStreamRecvResponse:
    /* endpoint callback length is USB_CANCELLED_TRANSFER_LENGTH (0xFFFFFFFFU) when transfer is canceled */
    if ((g_deviceAudioComposite->audioUnified.attach) &&
        (ep_cb_param->length != (USB_CANCELLED_TRANSFER_LENGTH)))
    {
        USB_LatencyDebug_ReceivingAudioFromHost(true);   toggle P1_7
```

Figure 6. Toggle GPIO P1_7 after receiving USB audio packet

Then toggle another GPIO (P1_6) when starting a DMA transfer and when the DMA transfer is completed.

```
s_audio_GetSamplesCb = getMixedSamplesCb;
/* get and write samples into the output ring buffer */
audio_GetAndTransmitSamples(nbSamples);


HAL_AudioTransferSendNonBlocking((hal_audio_handle_t)&audioTxHandle[0], &s_TxTransfer);   Start TX DMA transfer
AUDIO_LatencyDebug_SendingAudioToI2s(true);
AUDIO_LatencyDebug_SendingAudioToI2s(false);   toggle P1_6
```

Figure 7. Toggle GPIO P1_6 when starting a DMA transfer and when a DMA transfer is completed

The time from receiving the first USB audio packet to starting DMA transfer is the audio latency from USB to I^2S. As shown in Figure 8, the interval between USB audio packets is 500 µs, that is, a USB audio packet is sent every four microframes. The length of each USB audio packet is 96 bytes. After receiving four USB audio packets, DMA transfer starts. The audio latency from USB to I^2S is 1.65 ms.

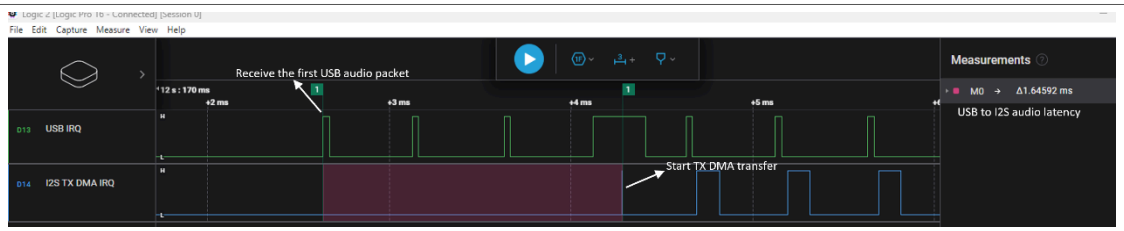

Figure 8. Measure audio latency on high-speed UAC2.0 device

To get a shorter audio latency, set the interval of the USB audio packet to 125 µs, that is, send a USB audio packet in each microframe, and reduce the initial fill threshold of the ringbuffer. However, the minimum threshold is the length of two USB audio packets, because we must prepare at least two DMA transfer data before starting DMA transfer to form a link DMA transfer. After the first DMA transfer is completed, the second DMA transfer is started immediately to ensure continuous playback of audio data.

3 USB audio mixer testing

The USB audio mixer implemented in this application note supports UAC1.0 and UAC2.0 modes, and supports full-speed USB and high-speed USB interfaces. Customers can set the following macro definitions to select different working modes. These macros are defined in the *usb_device_config.h* file.

#define USB_DEVICE_CONFIG_LPCIP3511FS (1U)

#define USB_DEVICE_CONFIG_LPCIP3511HS (0U)

#define USB_DEVICE_CONFIG_AUDIO_CLASS_2_0 (0U)

After setting the desired mode, you can start compiling the project and download the compiled program to the LPC55S69-EVK through the onboard debugger interface (P6) or the external debugger interface (P7). Press the RESET button (S4) on the board to start running the program. Pay attention to connect the correct USB port to the PC or other USB host. The full-speed USB interface is P10, and the high-speed USB interface is P9. The USB host recognizes a composite USB audio device, as shown in Figure 9.

Figure 9. USB audio mixer counted by the USB host

Open two audio players on the USB host and select NXP Dongle mix (Chat) speaker and NXP Dongle mix (Game) speaker to play the audio. Then you can connect a 3.5 mm headphone to the headphone jack (J2) of the LPC55S69-EVK board to hear a mixed audio.

4 Conclusion

This application note introduces how to implement a USB audio mixer on the LPC55S69-EVK. Based on the `usb_composite_audio_unified_bm` example in the LPC55S69 SDK v2.15, port the descriptors of the USB dongle mixer in the NXH3670 SDK to this basic project, and add the UAC2.0 function. In addition, the ringbuffer management mechanism, the implementation of the USB audio synchronization mode, and the measurement of audio latency are introduced. Customers can implement the USB audio mixer function on the LPC55S69 and other NXP MCU platforms based on this application note and AN14531SW.

5 Reference

1. NXH3670 SDK Gaming Package
2. *Getting started with NxH3670 gaming use case* (document AN12360)
3. *LPC55S6x/LPC55S2x/LPC552x User manual* (document UM11126)
4. USB spec, Audio Device Document 1.0
5. USB spec, Audio Device Rev. 2.0

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7 Revision history

Table 4 summarizes the revisions to this document.

Table 4. Revision history

Document ID	Release date	Description
AN14531 v1.0	13 January 2025	Initial public release

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