

Intel[®] Edison

Audio Setup Guide

May 2015

Revision 001



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Revision History

Revision	Description	Date
001	Initial release.	May 1, 2015

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1 Introduction

This document provides information about the audio subsystem on the Intel[®] Edison platform. It also provides hardware/software overviews, shows different ways to capture/playback audio through the hardware with an Intel[®] Edison board, shows how to play and record audio using Intel[®] Edison, and describes software stacks to develop audio applications or play with audio on the Intel[®] Edison platform.

1.1 References

Reference	Name	Number/location
331188	Intel® Edison Board Support Package User Guide	http://www.intel.com/support/edison/sb/CS-035278.htm
331189	Intel® Edison Compute Module Hardware Guide	http://www.intel.com/support/edison/sb/CS-035274.htm
331190	Intel® Edison Breakout Board Hardware Guide	http://www.intel.com/support/edison/sb/CS-035252.htm
331191	Intel® Edison Kit for Arduino* Hardware Guide	http://www.intel.com/support/edison/sb/CS-035275.htm
331192	Intel [®] Edison Native Application Guide	http://www.intel.com/support/edison/sb/CS-035382.htm
329686	Intel® Galileo and Intel® Edison Release Notes	https://communities.intel.com/docs/DOC-23388
332032	Intel® Edison Software Release Notes	
[GSG]	Intel® Edison Getting Started Guide	W: https://communities.intel.com/docs/DOC-23147 M: https://communities.intel.com/docs/DOC-23148 L: https://communities.intel.com/docs/DOC-23149
331438	Intel® Edison Wi-Fi Guide	http://www.intel.com/support/edison/sb/CS-035380.htm
331704	Intel® Edison Bluetooth* Guide	http://www.intel.com/support/edison/sb/CS-035381.htm
332434	Intel® Edison Audio Setup Guide	(This document)
[YPQSG]	Yocto Project Quick Start Guide	http://www.yoctoproject.org/docs/current/yocto-project-qs /yocto-project-qs.html
[YDM]	Yocto Developer Manual	http://www.yoctoproject.org/docs/current/dev-manual /dev-manual.html
[YKDM]	Yocto Kernel Developer Manual	http://www.yoctoproject.org/docs/latest/kernel-dev /kernel-dev.html

1.2 Terminology

Term	Definition
A2DP	Advanced Audio Distribution Profile
BT	Bluetooth
DSP	Digital signal processing.
HFP	Hands-Free Profile
HSP	Headset Profile
РСМ	Pulse-code modulation
TDM	Time division multiplexing



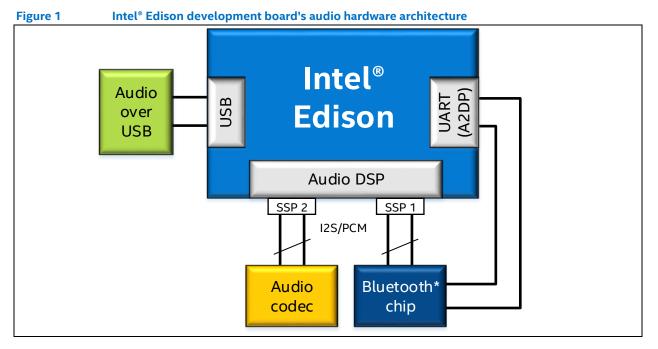


2 Hardware

The Intel[®] Edison development board has hardware capabilities—Bluetooth^{*}, USB, and I2S I/O—that allow you to enable the following audio features:

- Audio over a USB device.
- Audio record from A2DP (sink) remote player.
- Audio playback to an A2DP remote device.
- Local audio playback/record by adding a hardware audio codec + microphone, speakers, or jack headset.

Figure 1 shows the audio architecture of the Intel® Edison development board.



2.1 Hardware specifications

The Intel® Edison controller has two 4-wire ports dedicated to I2S/PCM audio transmission: SSP1 and SSP2.

The SSP1 port is used by the Bluetooth* processor for the hands-free and headset audio profiles (HFP and HSP). It can also be used optionally by the UART for audio transmission. The SSP2 port is available for audio codec hardware.

Both I2S and PCM mode can be used with the audio DSP to record and play audio on an Intel[®] Edison board. The audio DSP has to be the **master** of the bus.

Hardware



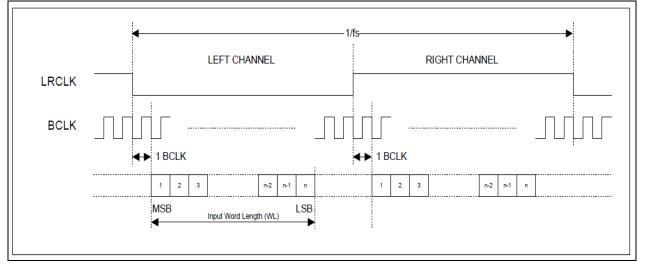
2.1.1 I2S Mode

I2S is a 4 wire digital audio interface (DAI) used in high-fidelity sound equipment, set-top boxes, and various portable devices. For more information, see https://www.kernel.org/doc/Documentation/sound/alsa/soc/DAI.txt.

I2S operating modes:

- I2S MSB
- Left-justified
- Right-justified

Figure 2 Example for I2S left-justified



2.1.2 PCM mode

PCM is also a 4-wire interface, similar to I2S, that can support a flexible protocol. For more information, see *https://www.kernel.org/doc/Documentation/sound/alsa/soc/DAI.txt*.

PCM operating modes:

- Mode A MSB (most significant bit) is transmitted on falling edge of first BCLK after FRAME/SYNC.
- Mode B MSB (most significant bit) is transmitted on rising edge of FRAME/SYNC.

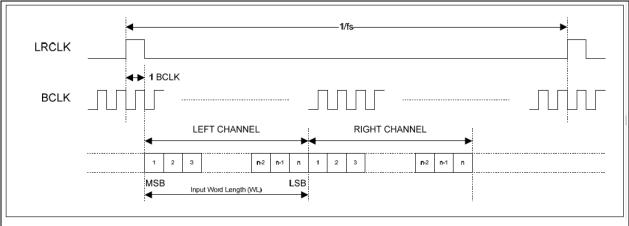


Figure 3 Example PCM mode A

Hardware



2.1.3 Hardware/software capabilities and limitations

The following restrictions apply:

- The audio DSP has to be the master of the bus.
- Not all configurations have been tested.
- The SSPx port supports I2S/PCM/PCM_TDM configurations (1 to 8 slots).

Hardware-supported audio configurations:

- Mode: I2S, PCM, TDM
- Sampling rate: 8, 16, 44.1, and 48 kHz
- Number of bits by channel: 16, 24, or 32 bits
- Number of channels: mono, dual mono, stereo; In TDM mode: 1 to 8 slots

Software-supported configuration on SSP2:

- Mode: TDM
- Sampling rate: 48 kHz
- Bits by channel: 24 bits
- Number of channels: 4 slots

SSP1 port tested configuration:

- I2S/PCM
- 8/16 kHz
- 16 bits
- Mono/dual mono/stereo

SSP2 port tested configuration:

- PCM TDM 4 slots
- 48 kHz
- 24 bits
- Mono/dual mono/stereo

§

Software

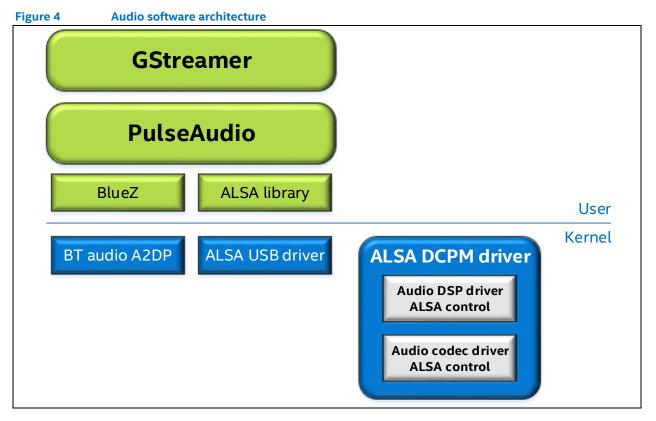


3 Software

The Intel[®] Edison prebuilt image supports the following libraries and tools that allow an Intel[®] Edison board to play back or record audio out of the box:

- **GStreamer**: A player that allows Intel[®] Edison to play WAV files.
- **PulseAudio:** A server manages the audio source/sink device.
- BlueZ: A Bluetooth* library on Linux*.
- Alsa-lib: The Linux* audio library.

Figure 4 shows the Intel® Edison audio software architecture.



3.1 GStreamer

GStreamer is the default player supported on the Intel[®] Edison board. It acts like a framework to develop/create streaming media applications. Designed to make it easy for applications to handle audio and video, GStreamer's framework makes it possible to develop any type of streaming multimedia applications.

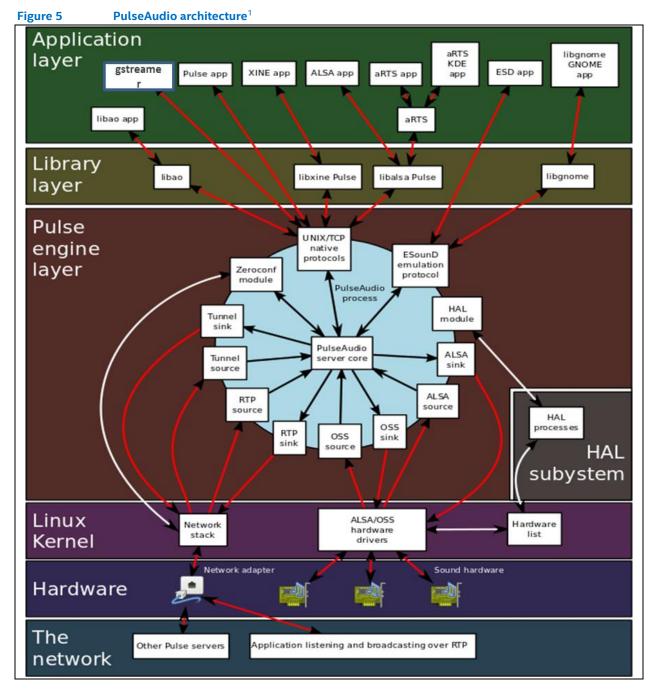
GStreamer already includes components for building a media player that can support a very wide variety of formats, including MP3, Ogg/Vorbis, MPEG-1/2, AVI, and QuickTime*. GStreamer supports a wide variety of codecs via plugins. The default Intel[®] Edison image currently ships with WAV, Vorbis, and FLAC audio formats.

Note: See *http://gstreamer.freedesktop.org* for more information.



3.2 PulseAudio server

PulseAudio is a POSIX-based sound server, designed for Linux* but also ported to other OSes. It features a background process that accepts sound input from one or more **sources** (input/capture devices) and redirects it to one or more **sinks** (output/playback devices).



¹ Diagram by Manuel Amador. Reprinted by permission.

Software



3.2.1 Procedure to use PulseAudio

The PulseAudio server launches when the Intel[®] Edison board boots up. You can stop/restart/start PulseAudio like any *systemd* service (Figure 6).

```
Figure 6
```

```
Pulseaudio start, stop, and restart operations
```

```
    ram@ram-desktop: ~/Nagesh
    root@edison:~# systemctl stop pulseaudio
    root@edison:~# systemctl start pulseaudio
    root@edison:~# systemctl restart pulseaudio
    root@edison:~#
```

We can also launch it by command line and add debug information with -v parameters (Figure 7).

Figure 7 PulseAudio start debug log enabled

👂 亘 💿 ram@ram-desktop: ~/Nagesh root@edison:~# pulseaudio --system -vvv W: [pulseaudio] main.c: Running in system mode, but --disallow-exit not set! W: [pulseaudio] main.c: Running in system mode, but --disallow-module-loading no t set! N: [pulseaudio] main.c: Running in system mode, forcibly disabling SHM mode! N: [pulseaudio] main.c: Running in system mode, forcibly disabling exit idle tim e! D: [pulseaudio] core-rtclock.c: Timer slack is set to 50 us. D: [pulseaudio] core-util.c: setpriority() worked. I: [pulseaudio] core-util.c: Successfully gained nice level -11. I: [pulseaudio] main.c: Found user 'pulse' (UID 995) and group 'pulse' (GID 1000). I: [pulseaudio] main.c: Successfully changed user to "pulse". I: [pulseaudio] main.c: This is PulseAudio 5.0 D: [pulseaudio] main.c: Compilation host: i586-poky-linux-gnu D: [pulseaudio] main.c: Compilation CFLAGS: -O2 -pipe -g -feliminate-unused-deb ug-types -Wall -W -Wextra -Wno-long-long -Wno-overlength-strings -Wunsafe-loop-d ptimizations -Wundef -Wformat=2 -Wlogical-op -Wsign-compare -Wformat-security -N missing-include-dirs -Wformat-nonliteral -Wpointer-arith -Winit-self -Wdeclarati on-after-statement -Wfloat-equal -Wmissing-prototypes -Wredundant-decls -Wmissin g-declarations -Wmissing-noreturn -Wshadow -Wendif-labels -Wcast-align -Wstrictaliasing -Wwrite-strings -Wno-unused-parameter -ffast-math -fno-common -fdiagnos tics-show-option D: [pulseaudio] main.c: Running on host: Linux i686 3.10.17-poky-edison+ #1 SMP PREEMPT Fri Jan 30 14:16:35 CET 2015 D: [pulseaudio] main.c: Found 2 CPUs. I: [pulseaudio] main.c: Page size is 4096 bytes D: [pulseaudio] main.c: Compiled with Valgrind support: no

As a unique "root" user of the Intel[®] Edison image, you must launch PulseAudio with the --system parameter. You can use the -h option to get more parameter capabilities, like --dump-resample-methods, to optimize further desired configurations.

Software



When PulseAudio starts, you may choose to display all your source and sink devices with the **pactl** binary (**P**ulseAudio **Control**) to view their capabilities (Figure 8). Do the following:

1. Enter pactl list.

Figure 8 pactl list command ram@ram-desktop: ~/Nagesh oot@edison:~# pactl list Module #0 Name: module-udev-detect Argument: Usage counter: n/a Properties: module.author = "Lennart Poettering" module.description = "Detect available audio hardware and load m atching drivers" module.version = "5.0" Module #1 Name: module-alsa-card Argument: device_id="0" name="0" card_name="alsa_card.0" namereg_fail=fa lse tsched=yes fixed_latency_range=no ignore_dB=no deferred_volume=yes use_ucm=y es card_properties="module-udev-detect.discovered=1" Usage counter: 0 Properties: module.author = "Lennart Poettering" module.description = "ALSA Card' module.version = "5.0" Module #2 Name: module-native-protocol-unix Argument: Usage counter: n/a **Properties:** module.author = "Lennart Poettering" module.description = "Native protocol (UNIX sockets)" module.version = "5.0"

2. Enter pactl list sinks.

```
Figure 9
                 pactl list sinks command
            root@edison:~# pactl list sinks
            Sink #0
                     State: SUSPENDED
                     Name: alsa_output.0.analog-stereo
                     Description: Loopback Analog Stereo
                     Driver: module-alsa-card.c
                     Sample Specification: s16le 2ch 44100Hz
                     Channel Map: front-left, front-right
                    Owner Module: 1
                     Mute: no
                     Volume: front-left: 65536 / 100% / 0.00 dB,
                                                                         front-right: 65536 / 100%
              0.00 dB
                              balance 0.00
                    Base Volume: 65536 / 100% / 0.00 dB
                    Monitor Source: alsa_output.0.analog-stereo.monitor
                    Latency: 0 usec, configured 0 usec
Flags: HARDWARE DECIBEL_VOLUME LATENCY
                     Properties:
                              alsa.resolution_bits = "16"
                              device.api = "alsa"
device.class = "sound"
                              alsa.class = "generic"
                              alsa.subclass = "generic-mix"
                              alsa.name = "Loopback PCM"
alsa.id = "Loopback PCM"
                              alsa.subdevice = "0'
                              alsa.subdevice_name = "subdevice #0"
                              alsa.device = "0"
```



3. Enter pactl list sources.

Figure 10

pactl list sources command root@edison:~# pactl list sources | grep bluez_source Name: bluez_source.98_0D_2E_C8_BD_2C root@edison:~#

Note: You can use *pactl* to set the default device to use, mute a device, or load a specific module.

PulseAudio server starts with a set of default configurations about audio stream capabilities and the modules to be loaded at start. Audio stream capabilities like sampling rates, resample methods, are defined in the file /etc/pulse/daemon.conf and the modules loaded are defined in /etc/pulse/system.pa.

PulseAudio comes with a set of modules with their own roles. Not all modules available in the PulseAudio build are embedded in the default Intel® Edison image, and the ones present in the image are not necessarily loaded by default when PulseAudio starts. The list of modules ready to be used with the default Intel® Edison image can be retrieved with the following command:

pulseaudio --dump-modules

We can load any of the listed modules with the pactl tool:

pactl load-module loopback-module source=xxx sink=xxx

To embed any module from the PulseAudio library and add it to your custom image, look at this sample file:

\${ROOT EDISON}/device-software/meta-edison-distro/recipesmultimedia/pulseaudio/pulseaudio_5.0.bbappend

This file embeds the A2DP Bluetooth* module.

δ



4 **Producing Audio with an Intel® Edison Device**

This section explains various ways to get sound with an Intel[®] Edison board using loopback methods, which are interactive and allow an Intel[®] Edison board to behave as a connected device.

There are three ways to play audio through an Intel® Edison device:

- Loopback A2DP stream to an A2DP remote device.
- Loopback audio from and over a USB headset.
- Loopback A2DP stream over a USB sound device.

4.1 Loopback audio from A2DP source to an A2DP Bluetooth headset

Figure 11 shows what this process looks like graphically.



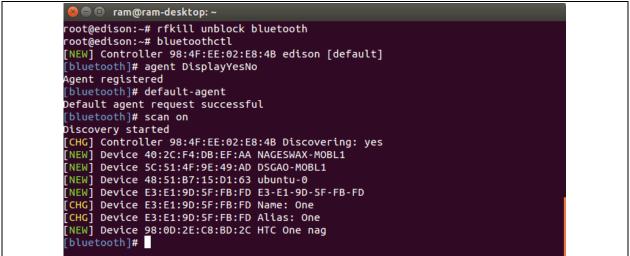


Note: By default, an Intel[®] Edison device's Bluetooth* name is **BlueZ 5.24**. This is the name that may appear on your player devices.

Step-by-step guide

1. Start the *bluetoothctl* controller (Figure 11).

Figure 11 Enabling Bluetooth* with the unblock command



2. Pair the Intel® Edison device with the audio source device (Android device). In this case, HTC One nag.



Figure 12 Using the bluetoothctl command to scan and pair with the audio devices as source

😣 🖨 💷 ram@ram-desktop: ~
[bluetooth]# trust 98:0D:2E:C8:BD:2C
[CHG] Device 98:0D:2E:C8:BD:2C Trusted: yes
Changing 98:0D:2E:C8:BD:2C trust succeeded
[CHG] Device 40:2C:F4:DB:EF:AA RSSI: -68
[bluetooth]# pair 98:0D:2E:C8:BD:2C
Attempting to pair with 98:0D:2E:C8:BD:2C
[CHG] Device 98:0D:2E:C8:BD:2C Connected: yes
Request confirmation
[agent] Confirm passkey 602891 (yes/no): yes
[CHG] Device 98:0D:2E:C8:BD:2C Modalias: bluetooth:v000Fp0000d0000
[CHG] Device 98:0D:2E:C8:BD:2C UUIDs:
00001105-0000-1000-8000-00805f9b34fb
00001106-0000-1000-8000-00805f9b34fb
0000110a-0000-1000-8000-00805f9b34fb
0000110c-0000-1000-8000-00805f9b34fb
00001112-0000-1000-8000-00805f9b34fb
00001116-0000-1000-8000-00805f9b34fb
0000111f-0000-1000-8000-00805f9b34fb 0000112f-0000-1000-8000-00805f9b34fb
00001121-0000-1000-8000-0080519D341D 00001132-0000-1000-8000-00805f9b34fb
00001200-0000-1000-8000-00805f9b34fb
00001200-0000-1000-8000-00805f9b34fb
00001801-0000-1000-8000-00805f9b34fb
00006675-7475-7265-6469-616c62756d70
[CHG] Device 98:0D:2E:C8:BD:2C Paired: yes
Pairing successful
[CHG] Device 98:0D:2E:C8:BD:2C Connected: no

3. Connecting Intel[®] Edison device with the audio source device (Figure 12).

```
Figure 13
                Connect the Bluetooth audio source device
            [bluetooth]# connect 98:0D:2E:C8:BD:2C
            Attempting to connect to 98:0D:2E:C8:BD:2C
            [CHG] Device 98:0D:2E:C8:BD:2C Connected: yes
            Connection successful
            [NEW] Device 40:2C:F4:F4:32:53 40-2C-F4-F4-32-53
             bluetooth]# scan off
             CHG] Device 40:2C:F4:F4:32:53 RSSI is nil
             [CHG] Device C3:03:1B:1F:FC:9D RSSI is nil
             [CHG] Device 98:0D:2E:C8:BD:2C RSSI is nil
             CHG] Device E3:E1:9D:5F:FB:FD RSSI is nil
CHG] Device 48:51:B7:15:D1:63 RSSI is nil
             CHG] Device 5C:51:4F:9E:49:AD RSSI is nil
            [CHG] Device 40:2C:F4:DB:EF:AA RSSI is nil
            Discovery stopped
            [CHG] Controller 98:4F:EE:02:E8:4B Discovering: no
            [bluetooth]# exit
```

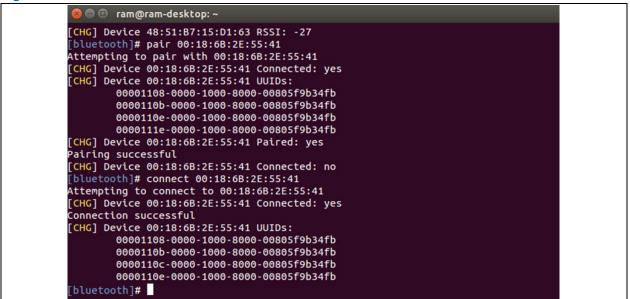
4. Scan the A2DP receiver (sink) device (the Bluetooth* headset) (Figure 14).



Figure 14	Using the bluetoothctl command to scan for audio receive device (sink)	
	😣 🖻 🗊 ram@ram-desktop: ~	
	root@edison:~# bluetoothctl	
	[NEW] Controller 98:4F:EE:02:E8:4B edison [default]	
	[NEW] Device 40:2C:F4:F4:32:53 40-2C-F4-F4-32-53	
	[NEW] Device 98:0D:2E:C8:BD:2C HTC One nag	
	[NEW] Device E3:E1:9D:5F:FB:FD One	
	[NEW] Device 48:51:B7:15:D1:63 ubuntu-0	
	[NEW] Device 5C:51:4F:9E:49:AD DSGAO-MOBL1	
	[NEW] Device 40:2C:F4:DB:EF:AA NAGESWAX-MOBL1	
	[bluetooth]# agent DisplayYesNo	
	Agent registered	
	[bluetooth]# default-agent	
	Default agent request successful	
	[bluetooth]# scan on	
	Discovery started	
	[CHG] Controller 98:4F:EE:02:E8:4B Discovering: yes	
	[CHG] Device 40:2C:F4:DB:EF:AA RSSI: -50	
	[CHG] Device 5C:51:4F:9E:49:AD RSSI: -79	
	[CHG] Device 48:51:B7:15:D1:63 RSSI: -28	
	[NEW] Device 49:85:B8:3D:B5:04 49-85-B8-3D-B5-04	
	[NEW] Device 00:18:6B:2E:55:41 LG HBS700	
1	[CHG] Device 49:85:B8:3D:B5:04 RSSI: -76	
	[bluetooth]#	

5. Pair and connect the A2DP receiving device (Figure 15).

Figure 15 Pair and connect the audio sink device with the Intel[®] Edison device



6. Check if both your A2DP devices are recognized in PulseAudio, then get the source and sink name of both your devices (Figure 16).



Figure 16	Using the pactl list command to display the BlueZ (source and sink Bluetooth*) devices	
root@e	dison:~# pactl list grep bluez	
	Name: module-bluez5-discover	
	Name: module-bluez5-device	
	Argument: path=/org/bluez/hci0/dev_98_0D_2E_C8_BD_2C	
	Argument: source="bluez_source.98_0D_2E_C8_BD_2C" source_dont_move="true	
" sink	_input_properties="media.role=music"	
	Name: module-bluez5-device	
	Argument: path=/org/bluez/hci0/dev_00_18_6B_2E_55_41	
	Name: bluez_sink.00_18_6B_2E_55_41	
	Driver: module-bluez5-device.c	
	Monitor Source: bluez_sink.00_18_6B_2E_55_41.monitor	
	device.api = "bluez"	
	bluez.path = "/org/bluez/hci0/dev_00_18_6B_2E_55_41"	
	bluez.class = "0x240404"	
	bluez.alias = "LG HBS700"	
	Name: bluez_source.98_0D_2E_C8_BD_2C	
	Driver: module-bluez5-device.c	
	device.api = "bluez"	
	bluez.path = "/org/bluez/hci0/dev_98_0D_2E_C8_BD_2C"	
	bluez.class = "0x5a020c"	
	bluez.alias = "HTC One nag"	
	Name: bluez_sink.00_18_6B_2E_55_41.monitor	
	Driver: module-bluez5-device.c	
	Monitor of Sink: bluez_sink.00_18_6B_2E_55_41	
	device.api = "bluez"	
	bluez.path = "/org/bluez/hci0/dev_00_18_6B_2E_55_41"	
	bluez.class = "0x240404"	
	bluez.alias = "LG HBS700"	

7. Set command to loopback A2DP source device and A2DP sink device (Bluetooth* headset).

Figure 17 Load loopback module with source and sink device as parameters

root@edison:~# pactl load-module module-loopback source=bluez_source.98_0D_2E_C8_BD_
2C sink=bluez_sink.00_18_6B_2E_55_41
15
root@edison:~# ls

8. Play the audio on the source device (Android device). You should be able to hear the audio in the Bluetooth* headset.



4.2 Loopback audio from A2DP smartphone over a USB headset

Figure 18 shows what this process looks like graphically.

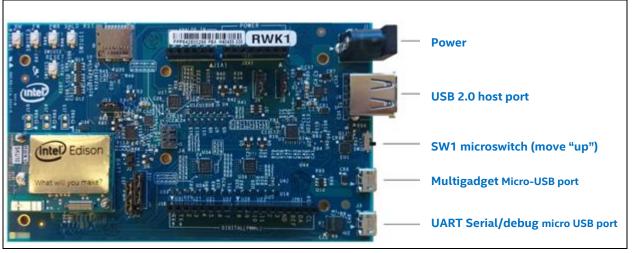


Step-by-step guide

Figure 19

1. Activate USB host mode by moving SW1 up toward the power socket. (You need an external power source in this mode.)

SW1 on an Intel[®] Edison board on an Arduino* expansion board



- 2. Plug in your USB headset.
- 3. Verify that the device is correctly mounted (Figure 20).

Figure 20 Using the cat command to check whether USB headset is detected

root@edison:~# cat /proc/asound/pcm 00-00: Loopback PCM : Loopback PCM : playback 8 : capture 8 00-01: Loopback PCM : Loopback PCM : playback 8 : capture 8 01-00: USB Audio : USB Audio : playback 1 : capture 1

4. Since a USB headset is plugged in, you can see the number of playback and capture streams handled by this device.

Figure 21	Using the pactl command to get the USB source and sink device names
root@edison	:~# pactl list grep Name grep usb
	a_output.usb-Plantronics_C320-M_0DC6315F9635B04FAAFAEBC7E98DFE2D-00-C320M.analog-stereo
	a_output.usb-Plantronics_C320-M_0DC6315F9635B04FAAFAEBC7E98DFE2D-00-C320M.analog-stereo.monitor
	a_input.usb-Plantronics_C320-M_0DC6315F9635B04FAAFAEBC7E98DFE2D-00-C320M.analog-stereo
Name: als	a_card.usb-Plantronics_C320-M_0DC6315F9635B04FAAFAEBC7E98DFE2D-00-C320M

5. Connect your A2DP player device (Android smartphone) and check if it is listed in PulseAudio as a source device (Figure 22).



Figure 22	Using the bluetoothctl scan command	
	😣 🖨 🗉 ram@ram-desktop: ~	
	root@edison:~# rfkill unblock bluetooth	
	root@edison:~# bluetoothctl	
	[NEW] Controller 98:4F:EE:02:E8:4B edison [default]	
	[bluetooth]# agent DisplayYesNo	
	Agent registered	
	[bluetooth]# default-agent	
	Default agent request successful	
	[bluetooth]# scan on	
	Discovery started	
	[CHG] Controller 98:4F:EE:02:E8:4B Discovering: yes	
	[NEW] Device 40:2C:F4:DB:EF:AA NAGESWAX-MOBL1	
	[NEW] Device 5C:51:4F:9E:49:AD DSGAO-MOBL1	
	[NEW] Device 48:51:B7:15:D1:63 ubuntu-0	
	[NEW] Device E3:E1:9D:5F:FB:FD E3-E1-9D-5F-FB-FD	
	[CHG] Device E3:E1:9D:5F:FB:FD Name: One	
	[CHG] Device E3:E1:9D:5F:FB:FD Alias: One	
	[NEW] Device 98:0D:2E:C8:BD:2C HTC One nag	
	[bluetooth]#	

Figure 23

Audio source device pairing and connecting

😕 🖨 🗉 ram@ram-desktop: ~	
<pre>[bluetooth]# trust 98:0D:2E:C8:BD:2C</pre>	
<pre>[CHG] Device 98:0D:2E:C8:BD:2C Trusted: yes</pre>	
Changing 98:0D:2E:C8:BD:2C trust succeeded	
<pre>[CHG] Device 40:2C:F4:DB:EF:AA RSSI: -68</pre>	
[bluetooth]# pair 98:0D:2E:C8:BD:2C	
Attempting to pair with 98:0D:2E:C8:BD:2C	
<pre>[CHG] Device 98:0D:2E:C8:BD:2C Connected: yes</pre>	
Request confirmation	
[agent] Confirm passkey 602891 (yes/no): yes	
[CHG] Device 98:0D:2E:C8:BD:2C Modalias: bluetooth:v000Fp0000d0000)
[CHG] Device 98:0D:2E:C8:BD:2C UUIDs:	
00001105-0000-1000-8000-00805f9b34fb	
00001106-0000-1000-8000-00805f9b34fb	
0000110a-0000-1000-8000-00805f9b34fb	
0000110c-0000-1000-8000-00805f9b34fb	
00001112-0000-1000-8000-00805f9b34fb	
00001116-0000-1000-8000-00805f9b34fb 0000111f-0000-1000-8000-00805f9b34fb	
00001111-0000-1000-8000-0080519D341D 0000112f-0000-1000-8000-00805f9b34fb	
00001132-0000-1000-8000-0080519D341D	
00001200-0000-1000-8000-00805f9b34fb	
00001200-0000-1000-8000-0080519D341D	
00001801-0000-1000-8000-00805f9b34fb	
00006675-7475-7265-6469-616c62756d70	
[CHG] Device 98:0D:2E:C8:BD:2C Paired: yes	
Pairing successful	
[CHG] Device 98:0D:2E:C8:BD:2C Connected: no	
[NEW] Device C3:03:1B:1F:FC:9D LOIS	
[CHG] Device 40:2C:F4:DB:EF:AA RSSI: -60	
[bluetooth]# connect 98:0D:2E:C8:BD:2C	

6. Check if your A2DP device is recognized in PulseAudio and get the source name of your device. (It should start with *bluez_source*.)



Figure 24 Using the pactl list command to get the audio source

root@edison:~# pactl list sources grep bluez_source	
Name: bluez_source.98_0D_2E_C8_BD_2C	
root@edison:~#	

7. Loopback the A2DP source device (Android device) to the USB playback device so that the audio playing on the Android device is routed to the USB headset connected to Intel[®] Edison device (Figure 25).

Figure 25 Using the pactl list command to get the audio source root@edison:~# pactl load-module module-loopback source=bluez_source.98_0D_2E_C8 _BD_2C sink=alsa_output.usb-Plantronics_Plantronics_C320-M_0DC6315F9635B04FAAFAE BC7E98DFE2D-00-C320M.analog-stereo 16 root@edison:~#

8. Play an audio file on your A2DP audio source device and check for audio on the headset.

4.3 Loopback audio from and over a USB headset

Step-by-step guide

- 1. Activate the USB host mode by moving the SW1 to the right position. (You need an external power source in this mode.)
- 2. Plug in your USB headset.
- 3. Check if the device is correctly mounted (Figure 26).

```
Figure 26 Using the pactl list command to get the audio source
```

```
root@edison:~# cat /proc/asound/pcm
00-00: Loopback PCM : Loopback PCM : playback 8 : capture 8
00-01: Loopback PCM : Loopback PCM : playback 8 : capture 8
01-00: USB Audio : USB Audio : playback 1 : capture 1
```

4. Since a USB headset is plugged in, you can see the number of playback and capture streams handled by this device.

Figure 27 A2DP audio loopback

oot@edison:~# pactl list | grep Name | grep usb Name: <mark>alsa_output.usb-Plantronics_C320-M_0DC6315F9635B04FAAFAEBC7E98DFE2D-00-C320M.analog-stereo</mark> Name: alsa_output.usb-Plantronics_C320-M_0DC6315F9635B04FAAFAEBC7E98DFE2D-00-C320M.analog-stereo.monitor Name: alsa_input.usb-Plantronics_C320-M_0DC6315F9635B04FAAFAEBC7E98DFE2D-00-C320M.analog-stereo Name: alsa_input.usb-Plantronics_C320-M_0DC6315F9635B04FAAFAEBC7E98DFE2D-00-C320M.analog-stereo

5. Loopback the USB source device (USB Mic) to the USB playback device so that the audio received from the MIC is routed to the USB headset connected to Intel[®] Edison device (Figure 28).

Figure 28 A2DP audio loopback

```
root@edison:~# pactl load-module module-loopback source=alsa_input.usb-Plantroni
cs_Plantronics_C320-M_0DC6315F9635B04FAAFAEBC7E98DFE2D-00-C320M.analog-stereo si
nk=alsa_output.usb-Plantronics_Plantronics_C320-M_0DC6315F9635B04FAAFAEBC7E98DFE
2D-00-C320M.analog-stereo
15
root@edison:~#
```

6. The user can play an audio file or speak through the microphone and verify that the sound is audible on the headset. By default, it should be audible as soon as the setup is complete.



4.4 Audio record and play through USB headset

Step-by-step guide

- 1. Activate the USB host mode by moving the SW1 to right position. (You need an external power source in this mode.)
- 2. Plug your USB headset.
- 3. Check if the device is correctly mounted (Figure 29).

Figure 29 Using the pactl list command to get the audio source

root@edison:~# cat /proc/asound/pcm
00-00: Loopback PCM : Loopback PCM : playback 8 : capture 8
00-01: Loopback PCM : Loopback PCM : playback 8 : capture 8
01-00: USB Audio : USB Audio : playback 1 : capture 1

4. Since a USB headset is plugged in, you can see the number of playback and capture streams handled by this device.

Figure 30 A2DP audio lists

oot@edison:~# pactl list | grep Name | grep usb Name: alsa_output.usb-Plantronics_C320-M_0DC6315F9635B04FAAFAEBC7E98DFE2D-00-C320M.analog-stereo Name: alsa_output.usb-Plantronics_C320-M_0DC6315F9635B04FAAFAEBC7E98DFE2D-00-C320M.analog-stereo.monitor Name: alsa_input.usb-Plantronics_C320-M_0DC6315F9635B04FAAFAEBC7E98DFE2D-00-C320M.analog-stereo Name: alsa_input.usb-Plantronics_C320-M_0DC6315F9635B04FAAFAEBC7E98DFE2D-00-C320M.analog-stereo

5. Set the default output sink device as USB headset (Figure 31).

Figure 31 Set default audio output device

root@edison:~# pactl set-default-sink alsa_output.usb-Plantronics_Plantronics_C3 20-M_0DC6315F9635B04FAAFAEBC7E98DFE2D-00-C320M.analog-stereo

6. The device is ready for recording and playing of audio. Run the *arecord* command to record audio from the MIC and store it into local file on the Intel[®] Edison device (Figure 32).

Figure 32 Using the arecord command to record audio

root@edison:~# arecord -D hw:1,0 -f dat audio_record.wav
Recording WAVE 'audio_record.wav' : Signed 16 bit Little Endian, Rate 48000 Hz, Stereo
CAborted by signal Interrupt
root@edison:~# mplayer audio_record.wav

- **Note:** Check the card ID and device ID of the USB headset MIC by using arecord -1 command and change the card ID and device ID number accordingly in the above command.
 - 7. After recording for some time, press Ctrl+C and run the GStreamer command to play the *audio_record.wav* audio file (Figure 33).

Figure 33 GStreamer to play the recorded audio

root@edison:~# gst-launch-1.0 filesrc location=audio_record.wav ! wavparse ! pulsesink
Setting pipeline to PAUSED
Pipeline is PREROLLING
Pipeline is PREROLLED
Setting pipeline to PLAYING
New clock: GstPulseSinkClock
Got EOS from element "pipeline0".
Execution ended after 0:00:16.418637665
Setting pipeline to PAUSED
Setting pipeline to READY
Setting pipeline to NULL
Freeing pipelin <u>e</u>
root@edison:~#

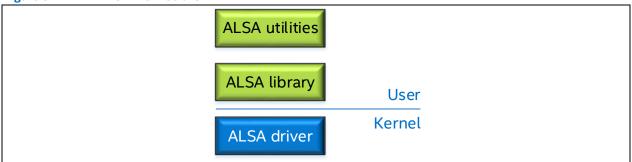
8. You should be able to play the recorded audio from the USB headset.



5 Advanced Linux Sound Architecture (ALSA)

ALSA is a framework and integral part of the Linux kernel that provides APIs for sound card device drivers. Some of the goals of the ALSA project at its inception were automatic configuration of sound card hardware and graceful handling of multiple sound devices in a system. ALSA is released under the GNU General Public License (GPL) and the GNU Lesser General Public License (LGPL).





5.1 ALSA user space

5.1.1 ALSA utilities

The ALSA utilities package contains a set of utility interfaces that control the sound card operations like get information, create a capture, or a playback stream. These are some of the most commonly used utilities:

- amixer Display, get control on Alsa driver controls.
- arecord Create a capture stream from an audio device.
- **aplay** Play a file and play back stream to an audio device.
- **alsactl** Store or restore an audio device configuration.
- **Note:** See http://www.linuxfromscratch.org/blfs/view/6.3/multimedia/alsa-utils.html for more information. These utilities depend on Alsa-IIb.

5.1.2 ALSA lib

The ALSA lib interface APIs are nothing but the interface to the ALSA drivers. Developers working on audio who want to use ALSA need to use these interfaces to achieve native ALSA support for their applications. See http://www.alsa-project.org/alsa-doc/alsa-lib for more information.



5.2 ALSA kernel space

5.2.1 ALSA-driver

The ALSA-driver layer is designed to make it easy to implement and handle audio devices.

As the Intel[®] Edison board is based on an SoC, we will focus on the ASOC (ALSA system-on-a-chip) implementation, which was designed to provide better ALSA support for embedded system-on-chip processors. ASOC basically splits an embedded audio system into multiple reusable component drivers.

- **The audio codec** Contains all audio controls (like mixers, gain, etc.), audio interface capabilities (support/configuration of I2S), I/O, and power management definition (DAPM). Must be platform independent.
- The platform driver Support of the DMA engine, Digital Audio Interface drivers (I2S/PCM/AC97), and if necessary, audio DSP controls + DAPM support.
- **The machine driver** This is the glue between the two previous parts and the management of external events (like an event when a headset is plugged in).

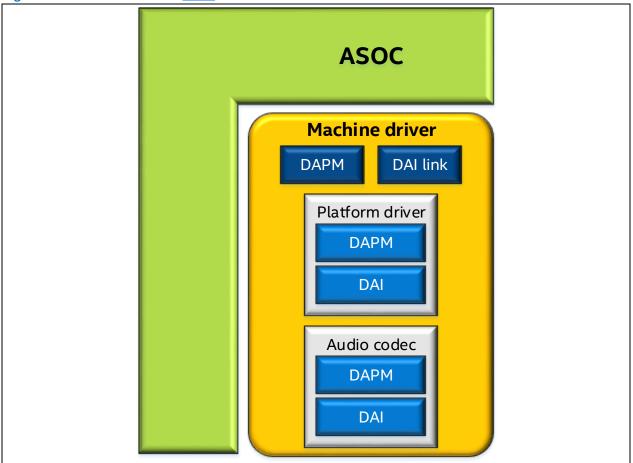


Figure 35 ALSA driver in <u>ASOC</u>

See https://git.kernel.org/cgit/linux/kernel/git/torvalds/linux.git/tree/Documentation/sound/alsa/soc/overview.txt for more information.



6 Audio DSP

The Intel® Atom[™] processor on the Intel® Edison board has an embedded audio DSP. (See section 2.1 for in-depth hardware specification details.) The SSP2 port allows platform makers to plug in an audio codec to add sound enhancement to the Intel® Edison board. The four wires can be easily accessed with the breakout board on **J19** (pin 11/10) and J20 (pin 11/10) connectors. Refer to the Intel® Edison Breakout Board Hardware Guide (331190) for details (http://www.intel.com/support/edison/sb/CS-035252.htm). The SSP1 port is used by the Bluetooth* chip. It can be used for the HSP and HFP profiles in PCM mode (vs. HCI mode, where the audio is sent through the UART).

This section is mainly intended for kernel developers.

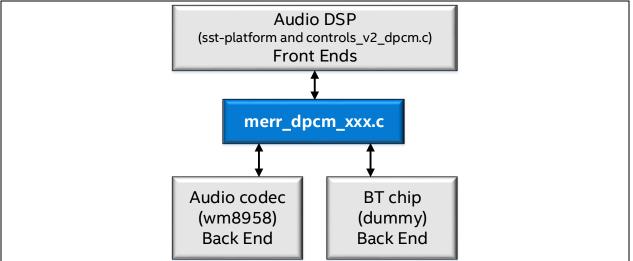
You can view the driver reference code at in the Intel® Edison kernel source tree here:

\${ROOT EDISON}/linux/sound/soc/intel

The Intel driver supports dynamic PCM (DPCM). It allows you to switch during runtime the DAI back end (like audio codec, Bluetooth* chip) connected to a DAI front end (media port, voip port of audio DSP).

See https://git.kernel.org/cgit/linux/kernel/git/torvalds/linux.git/tree/Documentation/sound/alsa/soc/DPCM.txt for more information.





6.1 Implementing a dummy codec with the audio DSP

The audio DSP exposes ALSA control to route, configure audio stream from/to the available Front End (FE) to the connected Back End (BE) and the audio DSP, as seen in section 2.1, is **master** of the SSP ports. So even if there is no audio codec, we can drive the SSP2 port and check the signals with an oscilloscope.

6.1.1 Create a machine driver with a dummy codec

1. Look at the dummy implementation of the driver at the following location:

\${ROOT_EDISON}/linux/sound/soc/intel/board/merr_dpcm_dummy.c

There are two significant parts to this file:

Audio DSP



This structure defines the **FE (front end)** and **BE (back end)** supported by the dummy DPCM machine driver.

```
static const struct snd_soc_dapm_route map[] = {
  { "Dummy Playback", NULL, "codec_out0" },
  { "Dummy Playback", NULL, "codec_out1" },
  { "codec_in0", NULL, "Dummy Capture" },
  { "codec_in1", NULL, "Dummy Capture" },
};
```

This structure defines the interconnections between the FE (audio DSP) I/O widgets codec_in and codec_out:

\${root edison}/linux/sound/soc/intel/platform-libs/controls v2 dpcm.c

...and the BE I/O (dummy codec) widgets linked for dummy playback and dummy capture:

\${root edison}/linux/sound/soc/soc-utils.c

2. To register the new machine driver, refer to :

```
${root_edison}/arch/x86/platform/intel-
mid/device_libs/platform_mrfld_audio.c
void *mrfld sst audio platform data(void *info)
```

This is a basic implementation of a dummy codec with the audio DSP. It allows user to stream an audio playback to the SSP2 port.

You can find the default audio DSP configuration associated in the *asound.state* file located in the */var/lib/alsa* folder. This configuration has to be set for any codec implementation as it configures the audio DSP path to capture or play back an audio stream through the SSP2 port, and before any open devices.

6.1.2 Testing

By default, the dummy machine driver is selected by the kernel command line with the argument "audio_codec=dummy". Set it to "wm8958" if you have a **wm8958 codec** connected to the Intel[®] Edison board.

Use GStreamer to play a Wav file on the ALSA device and check the signals with an oscilloscope.