



AN0700

AmebaPro2 Application Note



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USING THIS DOCUMENT

Though every effort has been made to ensure that this document is current and accurate, more information may have become available subsequent to the production of this guide.

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Convention

AmebaPro2 is a high-integrated IC. Its features include 802.11 Wi-Fi, H.264/H.265 video codec, Audio Codec.

This manual introduce users how to develop AmebaPro2, including SDK compiling and downloading image to AmebaPro2.

1 Building Environment

1.1 Setting up GCC Building Environment

1.1.1 Building the project in GCC Building Environment (Windows)

1.1.1.1 Installing mingw with ASDK and setting up the CMake

- (1) Download and extract msys64_v10_3.7z from tools folder
- (2) Check the windows Environment Variable HOME by Command Prompt

```
echo %HOME%
```

NOTE

If %HOME% is not exist, the command will simply print %HOME%

- (3) If your windows already have Environment Variable named HOME, open the file "msys64/etc/post-install/05-home-dir.post" Add "HOME=<PATH_TO_YOUR_MSYS64>/home/<USER_FOLDER>".

```
# If the home directory doesn't exist, create it.
HOME=C:/msys64_v10_3/msys64/home/${USER}
if [ ! -d "${HOME}" ]; then
    if mkdir -p "${HOME}"; then
        echo "Copying skeleton files."
```

NOTE

By default <USER_FOLDER> is \${USER}. To prevent some errors, do not include space characters in <USER_FOLDER>

- (4) Double click "msys2_shell.cmd" from msys64 folder
- (5) After setting up mingw, you need to install cmake. Download cmake in https://github.com/Kitware/CMake/releases/download/v3.20.0-rc1/cmake-3.20.0-rc1-windows-x86_64.msi and install it
- (6) Add location of cmake.exe to PATH of msys2_shell by using vim ~/.bashrc and appending path of cmake.exe to environment variable PATH or using editor to directly append the path to file "msys64/home/<USER_FOLDER>/.bashrc"

```
export PATH=/c/Program\ Files/CMake/bin:$PATH
```

CAUTION

If your PATH contains space characters, remember to use "\\" to escape

NOTE

For the first time adding the CMake PATH, after adding the PATH, you need to re-open the msys2_shell and check by:

```
$ cmake --version
cmake version 3.20.0-rc1
```

```
CMake suite maintained and supported by Kitware (kitware.com/cmake).
```

1.1.1.2 Adding toolchain to msys2

- (1) Like adding PATH for cmake, user can add or change the toolchain in "msys64/home/<USER_FOLDER>/.bashrc".
- (2) Add toolchain PATH by "export PATH=<path to toolchain>:\$PATH".

```
if [ -d "../../asdk-10.3.0" ]; then
    echo "asdk-10.3.0 exist"
    export PATH=/asdk-10.3.0/mingw32/newlib/bin:$PATH
```

NOTE

The recommended toolchain version is 10.3.0

1.1.1.3 Building the project

- (1) Open mingw by double clicking "msys2_shell.cmd".
- (2) Enter the project location: project/realtek_amebapro2_v0_example/GCC-RELEASE.
- (3) Create folder "build" and enter "build" folder.
- (4) Run "cmake .. -G"Unix Makefiles" -DCMAKE_TOOLCHAIN_FILE=../toolchain.cmake" to create the makefile.
- (5) Run "cmake --build . --target flash" to build and generate flash binary.

NOTE

If building successfully, you can see flash_ntz.bin in the build folder

1.1.2 Building the project in GCC Building Environment (LINUX)

1.1.2.1 Add toolchain to the linux PATH

- (1) Extract the toolchain file (the toolchain file may provide in tools folder):

```
tar -jxvf <PATH_TO_YOUR_TOOLCHAIN.tar.bz2> -C <DIR_TO_EXTRACT>
```

- (2) Add toolchain to PATH:

```
export PATH=<PATH_TO_YOUR_TOOLCHAIN>/asdk-10.3.0/linux/newlib/bin:$PATH
```

NOTE

You can add PATH to `~/.bash_profile`

1.1.2.2 Installing cmake for linux

- (1) Install cmake using terminal (like “`sudo apt-get -y install cmake`”), if the installation is successful, you can get the version by “`cmake --version`”.

1.1.2.3 Building the project

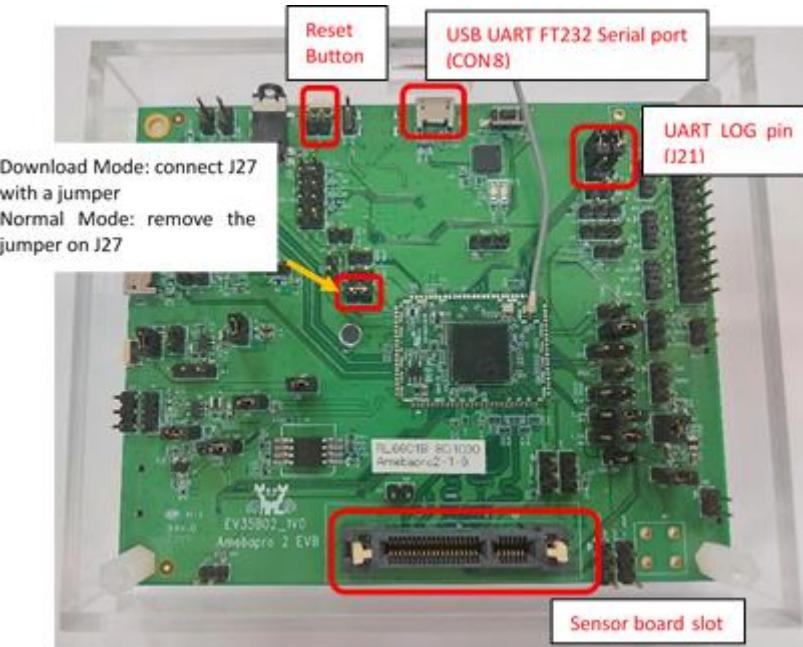
- (1) Open linux terminal and enter the project location: `project/realtek_amebapro2_v0_example/GCC-RELEASE/`.
(2) Create folder “build” and enter “build” folder.
(3) Run “`cmake .. -G"Unix Makefiles" -DCMAKE_TOOLCHAIN_FILE=..../toolchain.cmake`” to create the makefile.
(4) Run “`cmake --build . --target flash`” to build and generate flash binary.

NOTE

- If building successfully, you can see `flash_ntz.bin` in the build folder
- If the ‘build’ folder has been used by others, you can remove ‘build’ folder first to have clean build
- If there’s some permission issues, you can do “`chmod -R 777 <PATH_TO_YOUR_SDK>`”

1.2 Log UART Settings

- (1) To use AmebaPro2 log UART, the user needs to connect jumpers to **J21** for **FT232 (CON8)**.
(2) After using CON8 to connect to PC, you can use console tools (like tera term, MoBaxterm) to get log from EVB by setting baud rate as **115200**.



2 SDK Architecture

In AmebaPro2 sdk, it mainly contains four folders. The folder “component” store the main component source and the folder “project” contains the project makefile, compile flag and some examples. The folder “doc” and “tool” provide the document and tools for assisting you to set up the project.

2.1 Component

Folder	Sub-folder	Description
component	at-cmd	AT-command
	audio	ASP algorithm api audio codec
	bluetooth	bluetooth driver
	example	amazon related examples audio related examples fatfs example mmf examples socket related examples ...
	file_system	Fatfs and Littlefs
	lwip	Lwip API source code
	mbed	mbed API source code
	media	multi-media framework modules muxer and demuxer rtp codec for media
	network	cJSON coap dhcp httpc and httpd iperf mDNS mqtt ping rtsp sntp tftp websocket
	os	freertos: freertos source code os_dep: Realtek encapsulating interface for FreeRTOS, ram usage...
	soc	app: monitor and shell cmsis: cmsis style header file and startup file fwlib: hal drivers and nn api mbed-drivers: mbed API source code misc: driver and utilities
	ssl	ssl stub function and ram map source code
	stdlib	stdlib header files
	usb	usb and uvc header files
	video	ISP and video related api
	wifi	wifi api and wifi config related source code and header files

2.2 Project

Folder	Sub-folder	Description
Project/*	example_sources	examples for peripherals
	GCC-RELEASE	GCC cmake projects
	GCC-RELEASE/application	libraries for (non-trust zone) GCC project
	GCC-RELEASE/bootloader	bootloader project
	GCC-RELEASE/build	pre-build image files (boot.bin) and json files place for building cmake projects and generate flash image file (flash_ntz.bin)
	GCC-RELEASE/mp	for mp
	GCC-RELEASE/ROM	ROM code libraries
	inc	the header files for setting the project compile flag
	src	the main file source code for the project

2.3 Doc and tools

Folder	Sub-folder	Description
tools		PGTool: for downloading image files to AmebaPro2 msys64: for building the environment of AmebaPro2 project
doc		document for AmebaPro2

3 GCC Makefile

3.1 Adding files in CMake project

3.1.1 Adding sources and headers

In the section, we will introduce how to add files to AmebaPro2 project, including adding source, header files, creating and linking the library files.

(1) Adding Source files

Open the application.cmake at “project/realtek_amebapro2_v0_example/GCC-RELEASE/application/”.

Add the source code by append to app_sources:

```
list(
    APPEND app_sources
    ...
    ${PATH_TO_YOUR_SOURCE_FILES}
    ...
)
```

(2) Adding header files

Open the includepath.cmake at “project/realtek_amebapro2_v0_example/GCC-RELEASE/”.

Add the header files by append to inc_path_re:

```
list(
    APPEND inc_path_re
    ...
    ${PATH_TO_YOUR_HEADER_FILES}
    ...
)
```

Also add directory to be included by include_directories (<path to header folder>).

```
include_directories (${PATH_TO_YOUR_INCLUDE_DIR})
```

3.1.2 Adding library files

3.1.2.1 Method 1

You can place the library under “project/realtek_amebapro2_v0_example/GCC-RELEASE/application/output”.

Assume your library file is libABC.a, you can modify application.cmake like:

```
target_link_libraries(
    ${app}
    ...
    ABC
    ...
)
```

3.1.2.2 Method 2

(1) Declare your library

```
add_library (<LIBRARY_NAME> STATIC IMPORTED)
```

(2) Setup location of your library by:

```
set_property (TARGET <LIBRARY_NAME> PROPERTY IMPORTED_LOCATION <PATH_TO_YOUR_LIBRARY>)
```

or

```
set_target_properties (<LIBRARY_NAME> PROPERTY IMPORTED_LOCATION <PATH_TO_YOUR_LIBRARY>)
```

(3) Link to your library

```
target_link_libraries(
    ${app}
    ...
    <LIBRARY_NAME>
    ...
)
```

3.1.3 Building a library

3.1.3.1 Create a cmake file for the library

(1) Set up minimum required cmake version and the project name. Here the output library file name will be libtest.a or libtest.so.

- ```
cmake_minimum_required(VERSION 3.6)
project(test)
set(test test)

(2) Append source files to project source list
list(APPEND test_sources
 ${PROJ_ROOT}/example/test01.c
 ${PROJ_ROOT}/example/test02.c
)

(3) Assign the library type, STATIC means that the library will be built as static-link library (*.a), while SHARED means that the library will be built as dynamic-link library (*.so).
add_library(
 ${test} STATIC ${test_sources}
)

(4) Add the compile flag for the library
list(APPEND test_flags
 CONFIG_BUILD_ALL=1
 CONFIG_BUILD_LIB=1
 ${YOUR_COMPILE_FLAGS}
)

(5) Add the header files need to be included in the library
include(../includepath.cmake)
target_include_directories(${test} PUBLIC
 ${YOUR_INCLUDE_DIRS}
)
```

### 3.1.3.2 Add the cmake and link the library to the project

You can include and link to your library in application.cmake files by:

```
include(./libtest.cmake)
...
target_link_libraries(
 ${app}
 ...
 test
 ...
)
...
```

### 3.1.3.3 Turn off the dependency of the library

If users do not want to rebuild their own library each time when modification is not related to their library, users can open the DependInfo.cmake under GCC-RELEASE/build/application/CMakeFiles/<YOUR\_LIBRARY.dir> and turn on the CMAKE\_DEPENDS\_IN\_PROJECT\_ONLY.

```
Consider dependencies only in project.
set(CMAKE_DEPENDS_IN_PROJECT_ONLY ON)
```

## 3.2 Creating a new application example

The application example folder of AmebaPro2 needs to have app\_example.c and <EXAMPLE\_FOLDER.cmake>. The app\_example.c is the entry of the example and the cmake file is for project build. Here are the steps for building up a new application example.

- (1) Create a folder under “sdk/component/example”, move the source code to the folder and add app\_example.c and <EXAMPLE\_FOLDER.cmake> in the folder.
- (2) Open app\_example.c and call to the entries of example under the function app\_example

```
void app_example(void)
{
 example_audio_helix_aac();
}
```

- (3) Append the source code, header file, compile flag and library needed under the lists in <EXAMPLE\_FOLDER.cmake>.
- (4) After done the previous steps, users can build up the new example project by:

```
cmake .. -G"Unix Makefiles" -DCMAKE_TOOLCHAIN_FILE=../toolchain.cmake -
DEXAMPLE=<EXAMPLE_FOLDER>
cmake --build . --target flash
```

## 3.3 How to use example source code

In this section, we will describe how to use the example source code for AmebaPro2

### 3.3.1 Application example source

AmebaPro2 application's example source codes can be found under folder sdk/component/example.

Each example subfolder contains only one example, the entry function is app\_example(void).

The example entry function is defined as app\_example and only one example is exist in the same project.

Here are steps to build up the example:

- (1) Create example build folder in "project/realtek\_amebapro2\_v0\_example/GCC-RELEASE" and enter it

```
cd project/realtek_amebapro2_v0_example/GCC-RELEASE
mkdir build_example && cd build_example
```

- (2) Use cmake to create makefile for example

```
cmake .. -G"Unix Makefiles" -DCMAKE_TOOLCHAIN_FILE=../toolchain.cmake -
DEXAMPLE=<EXAMPLE_FOLDER_NAME>
```

 NOTE

If the example folder not exist, the "<EXAMPLE\_FOLDER\_NAME> Not Found" message will show, please check the example folder name

- (3) If example configured successfully, run build command to generate flash image

```
cmake --build . --target flash
```

 NOTE

In AmebaPro2 project, when -DEXAMPLE=<EXAMPLE\_FOLDER\_NAME> is used, -DVIDEO\_EXAMPLE=on and -DDOORBELL\_CHIME=on will be set to off after the needed source imported

### 3.3.2 MMF example source

In sdk/component/example/media\_framework, it provides audio-only MMF examples. The examples are based on the Multimedia Framework Architecture and the detail can refer to **Multimedia Framework Architecture**.

### 3.3.3 Peripheral example source

The peripheral example sources are located at the folder sdk/project/realtek\_amebapro2\_v0\_example/example\_sources and basically provide main.c and readme file. The main.c file contains the usage of peripheral function and user should replace it with the original main.c (in SDK/project/realtek\_amebapro2\_v0\_example/src). On the other hand, like application example source, the method to compile example and adjust the important parameters is described in the readme file. After the setting, user can rebuild the project with peripheral example.

### 3.3.4 WiFi example source

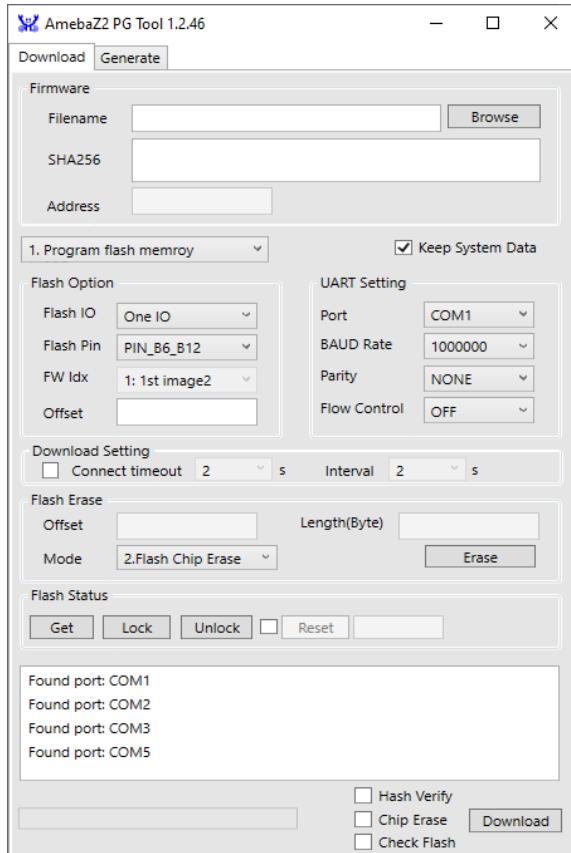
For user to test and development, we provide AT command in AmebaPro2. Users can key in AT command to connect WLAN by the console in PC. AT command can refer to "AN0025 Realtek at command.pdf".

## 4 Image Tool

### 4.1 Introduction

This chapter introduces how to use Image Tool to generate and download images. The image tool - PGTool can be found in tools folder and it has two functions:

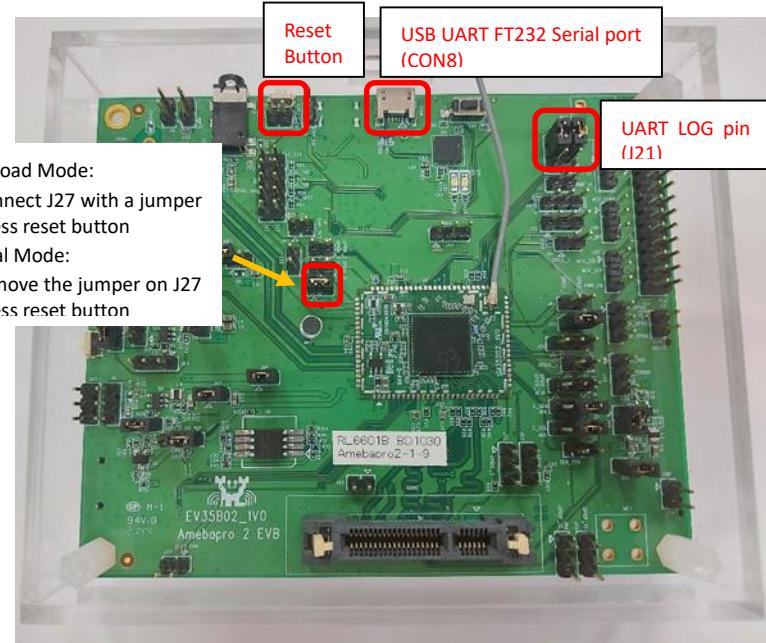
- (1) Download image to an AmebaPro2 device through UART.
- (2) Generate composited image from multiple image files.



### 4.2 Download Environment Setup

#### 4.2.1 Hardware Setup

To download image, the device must to be booted as download mode. Users need to first set up the UART with PC by connecting J21 with jumpers and CON8 with PC. Then, connect J27 with a jumper and press reset button to enter download mode.

**Download Mode:**

- (1) connect J27 with a jumper
  - (2) Press reset button
- Normal Mode:**
- (1) remove the jumper on J27
  - (2) Press reset button

#### 4.2.2 Software Setup

- PC environment requirements: Windows 7 above with FT232 driver.
- PGTool

### 4.3 Image Download

User can download the image to demo board by following steps:

- (1) Boot AmebaPro2 into download mode

**NOTE**

You can check whether your board is in download mode by UART message:

```
== Rt18735b IoT Platform ==
Chip VID: 0, Ver: 0
ROM Version: v3.0
Test Mode: boot_cfg1=0x0

[test mode PG]
test_mode_img_download
Download Image over UART1[tx=4,rx=3] baud=115200
```

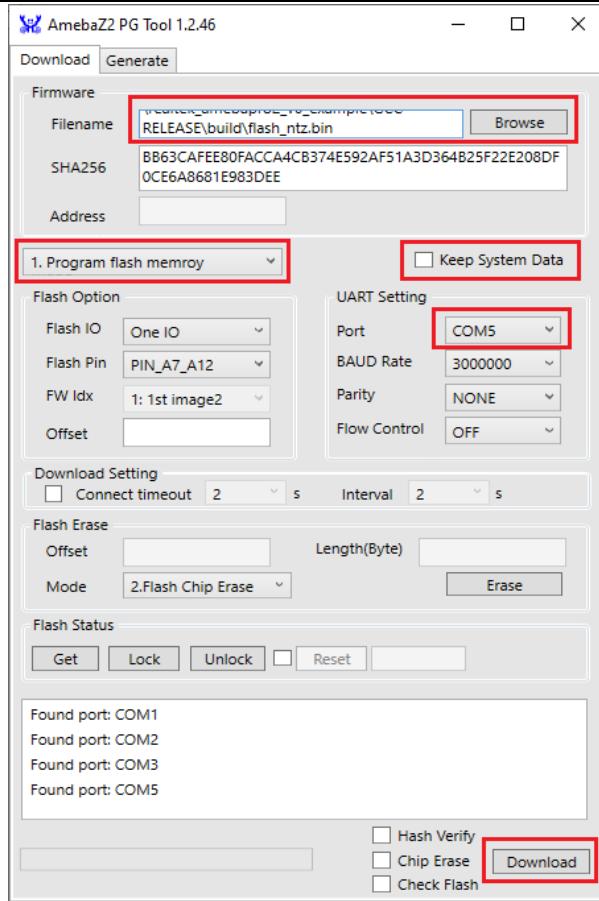
**CAUTION**

*Remember to disconnect log UART console before downloading image*

- (2) Run image tool
- (3) Browse your image: flash\_ntz.bin
- (4) Choose “1. Program flash memory” and select the correct COM port
- (5) Disable “Keep System Data” to prevent some errors
- (6) Press the Download button and the image will start to be downloaded to AmebaPro2 device

**NOTE**

*Flash Pin will be set automatically, we do not need to set*



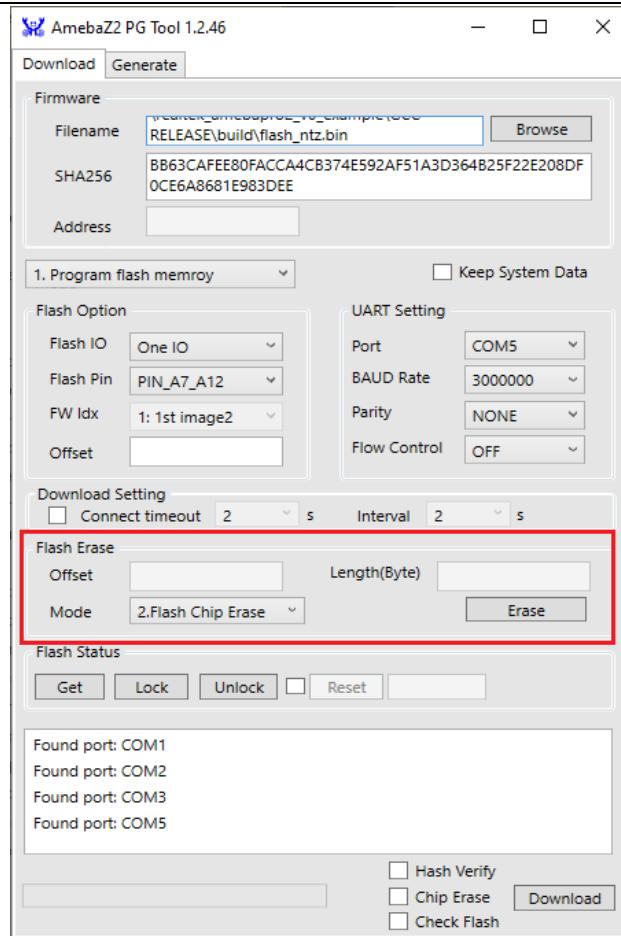
## 4.4 Erase Flash of device

Image tool also provides user to erase the flash of the device, it has two mode:

- (1) Flash Sector Erase: erase the flash from the Offset with a Length of Byte that the user set.
- (2) Flash Chip Erase: erase the whole flash

**NOTE**

*After you press the erase button, it needs time for erasing the flash. Please wait until the PG tool shows the "erase successfully" message*



## 4.5 Command line image tool for NAND flash

Command line image tool supports Windows, Linux and MacOS platforms:

- Windows: uartfwburn.exe
- Linux: uartfwburn.linux
- MacOS: uartfwburn.darwin

**NOTE**

*flash\_control\_info.bin needs to be placed in the same folder as uartfwburn tool*

### 4.5.1 Normal PG mode

Write whole image to NAND flash, run command:

```
uartfwburn -p <COM_PORT> -f flash_ntz.bin -b 3000000 -n pro2
```

**NOTE**

If success, command will print: nand download success

### 4.5.2 Partial image PG mode

Write partial image to NAND flash, run command:

```
uartfwburn -p <COM_PORT> -f flash_ntz.bin -b 3000000 -n pro2 -t 0x81cf
```

-t [type\_id] : pro2 nand flash partial image download, refer to the table below.

| Short name  | Size (Bytes) | Type ID |
|-------------|--------------|---------|
| PT_KEY_CER1 | 2            | 0xe9c2  |
| PT_BL_BRI   | 2            | 0xd1c5  |
| PT_FW1      | 2            | 0xc1c7  |

|           |   |        |
|-----------|---|--------|
| PT_FW2    | 2 | 0xb9c8 |
| PT_ISP_IQ | 2 | 0x89c1 |
| PT_NN_MDL | 2 | 0x81cf |

**i NOTE**

If success, command will print: nand download success

#### 4.5.3 FTL user data PG mode

Write FTL user data to NAND flash, run command:

```
uartfwburn -p <COM_PORT> -f user.bin -b 3000000 -n pro2 -w 800 950
-w [block_s] [block_bs] : pro2 nand FTL write, block_s is block start, block_bs is block backup start.
```

**i NOTE**

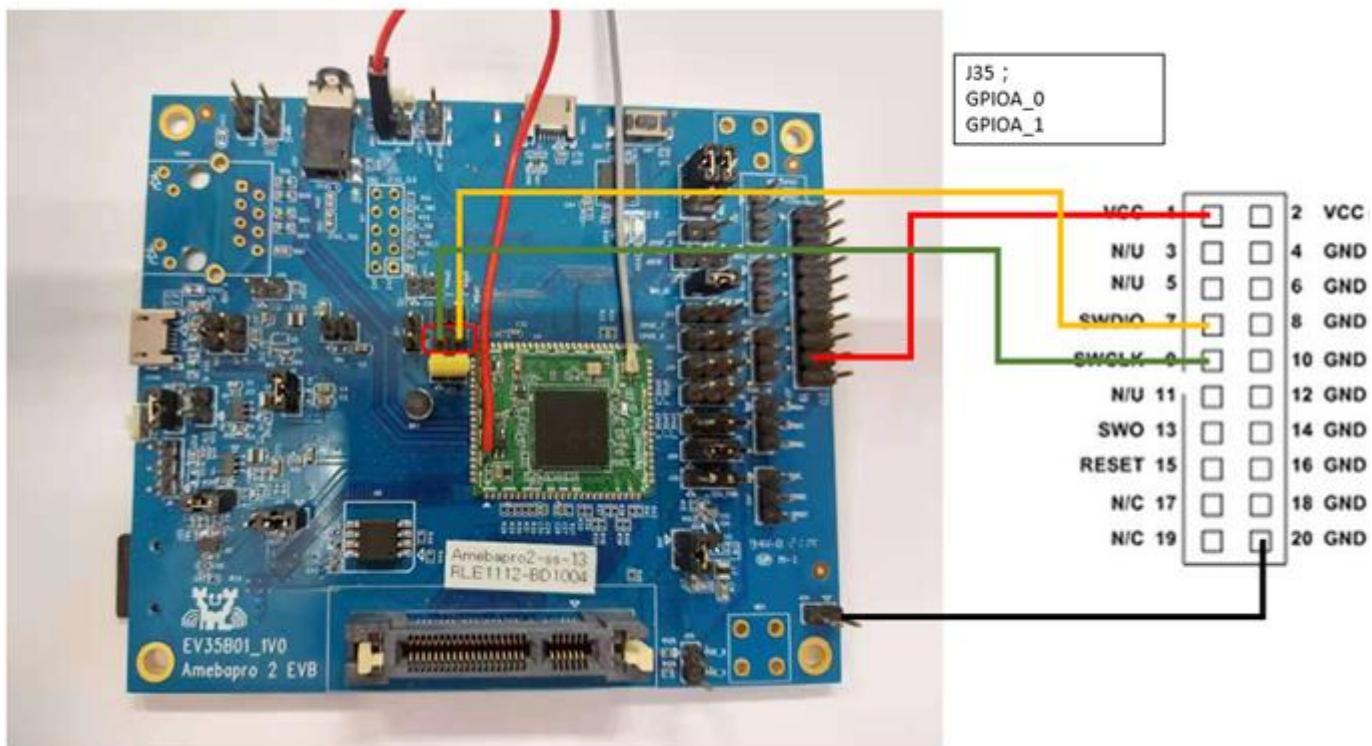
If success, command will print: nand download success

## 5 Using JTAG/SWD to debug

JTAG/SWD is a universal standard for chip internal test. The external JTAG interface has four mandatory pins, TCK, TMS, TDI and TDO, and an optional reset, nTRST. JTAG-DP and SW-DP also require a separate power-on reset: nPOTRST. The external SWD interface requires two pins: bidirectional SWDIO signal and a clock, SWCLK, which can be input or output from the device.

### 5.1 SWD connection

AmebaPro2 supports J-Link debugger. We need to connect the SWD connector to J-Link debugger. The SWD connection is shown as below. After finished these configurations, please connect it to PC side. Note that if you are using Virtual Machine as your platform, please make sure the USB connection setting between VM host and client is correct so that the VM client can detect the device.



#### NOTE

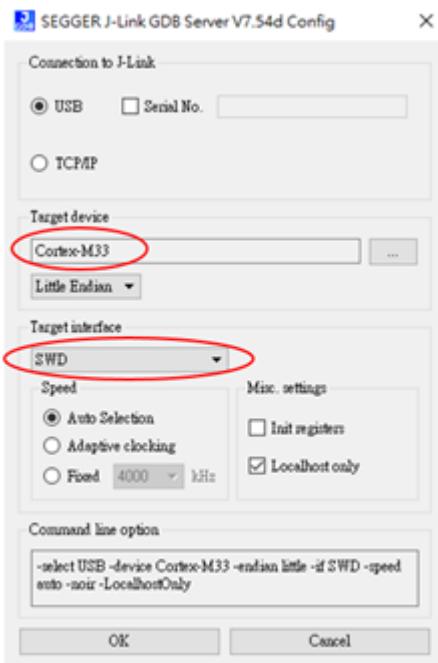
- EVB 1V0 module needs to HW rework, ask FAE for details
- To be able to debugger AmebaPro2 which is powered by Cortex-M33, user needs a J-Link debugger with the latest hardware version (Check [https://wiki.segger.com/Software\\_and\\_Hardware\\_Features\\_Overview](https://wiki.segger.com/Software_and_Hardware_Features_Overview) for details). J-Link EDU with hardware version V10 is used to prepare this document

### 5.2 Software installation

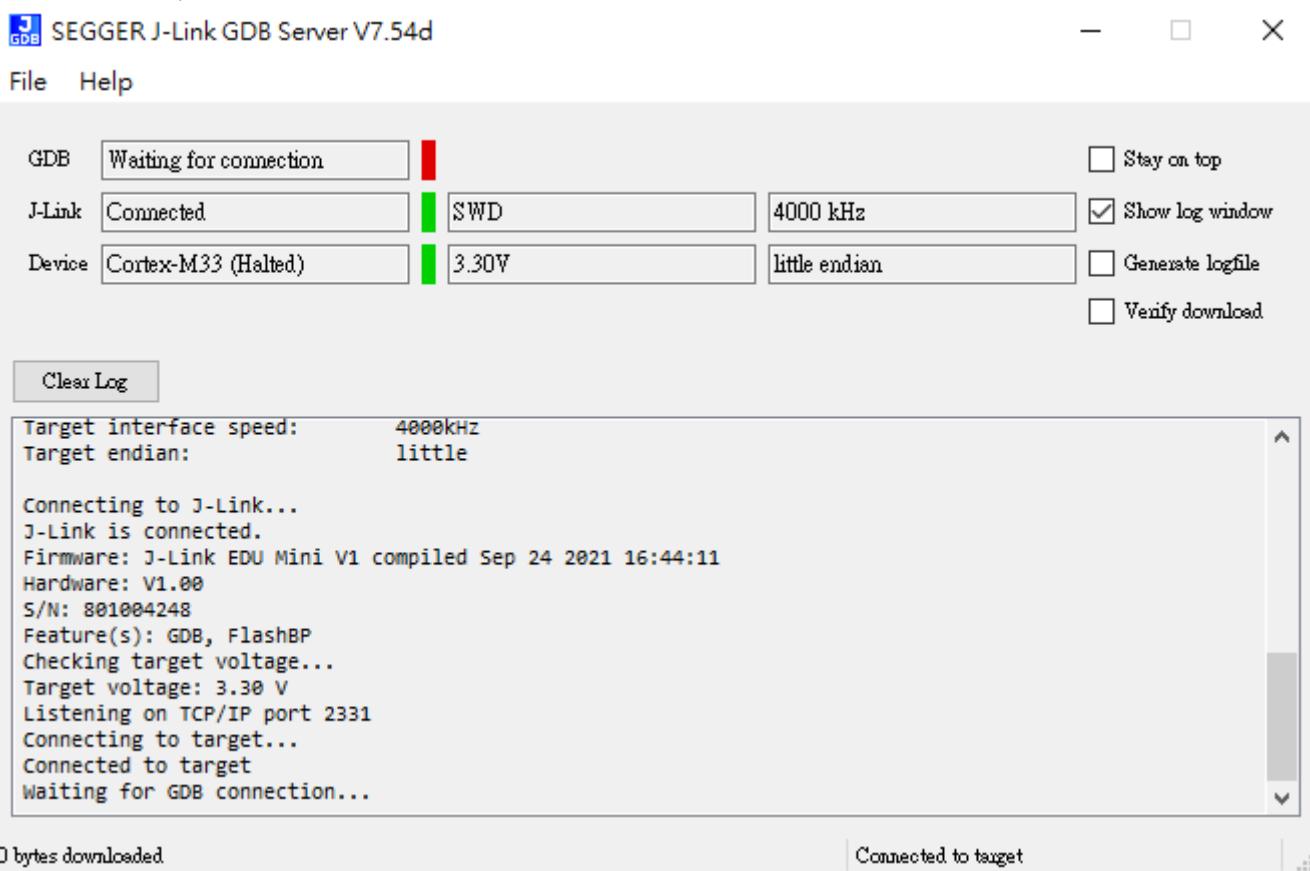
To be able to use J-Link debugger, user needs install J-Link GDB server first. For Windows, please check <http://www.segger.com> and download "J-Link Software and Documentation Pack" (<https://www.segger.com/downloads/jlink>).

### 5.3 Setup environment

To check whether the connection works fine, user can go to the location of SEGGER J-Link tool and run "JLinkGDBServer.exe". Choose target device Cortex-M33 (for AmebaPro2), and target interface SWD. Click "OK"



If connection succeeds, J-Link GDB server must show as below:



If connection fails, J-Link GDB will show:

## SEGGER J-Link GDB Server V7.54d

File Help

GDB Not connected

 Stay on top

J-Link Connected

SWD

 Show log window

Device Not selected

0.00V

little endian

 Generate logfile Verify download

```
J-Link script: none
J-Link settings file: none
-----Target related settings-----
Target device: Cortex-M33
Target interface: SWD
Target interface speed: 4000kHz
Target endian: little

Connecting to J-Link...
J-Link is connected.
Firmware: J-Link EDU Mini V1 compiled Sep 24 2021 16:44:11
Hardware: V1.00
S/N: 801004248
Feature(s): GDB, FlashBP
Checking target voltage...
```

0 bytes downloaded.

Connected to target

## 6 Multimedia Framework Architecture

The Multimedia Framework Architecture version 2(MMFv2) is responsible for handling the connection and management of different media resources on AmebaPro2.

### 6.1 Architecture

The structure of MMFv2 is as shown in the following chart and there are two important entities in the MMFv2, **MM\_MODULE** and **LINKER\_MODULE**:

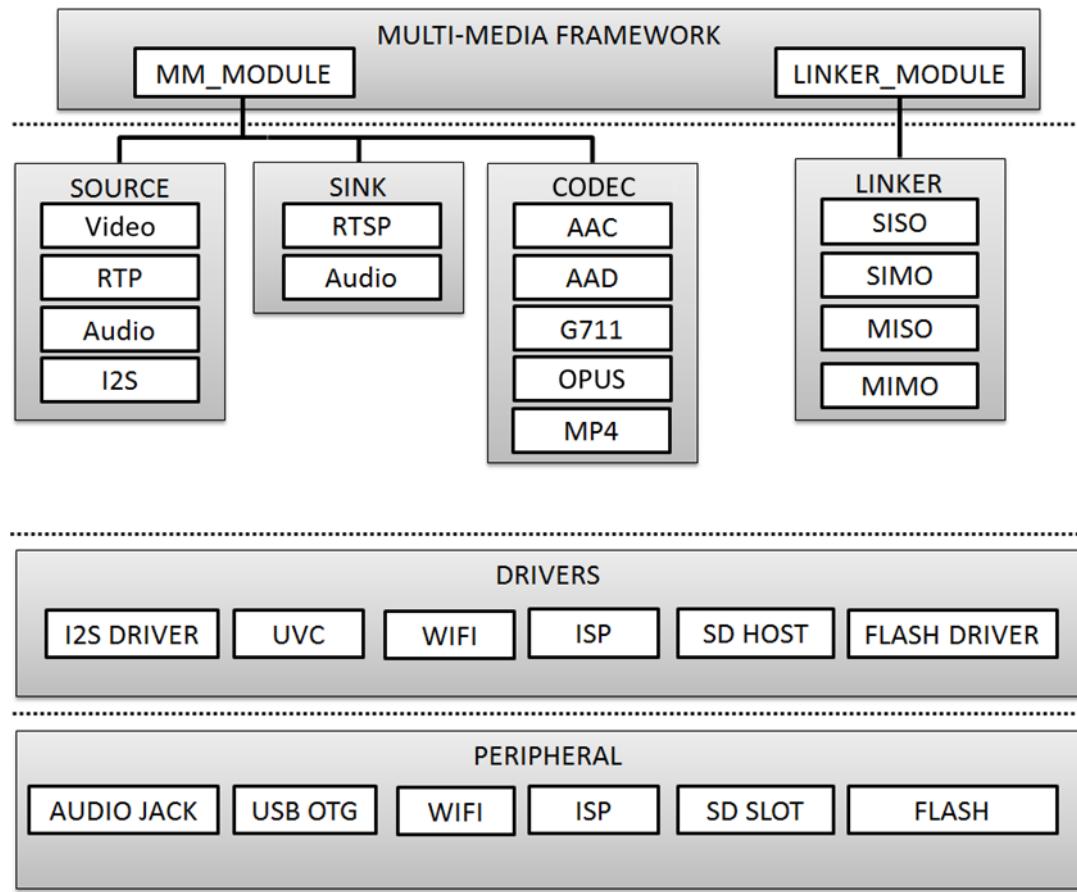
**MM\_MODULE** includes the media source, sink and the codec modules.

- Source module: produce resource, it can be the file input, microphone, camera, or storage.
- Codec module: mainly provide the audio codec, AAC, G711 or opus for customers to do audio encode or decode before sending streaming to sink module. In the mp4 module, it will automatically send the result into storage, SD card or ram disk.
- Sink module: consume resource from the source modules or after encoded/decoded by codec modules, like RTSP or other steaming.

**NOTE**

*The video modules uses VOE to contain the process of sensor catching, ISP and video encoding algorithms (jpeg, H264, HEVC (H265)...).*

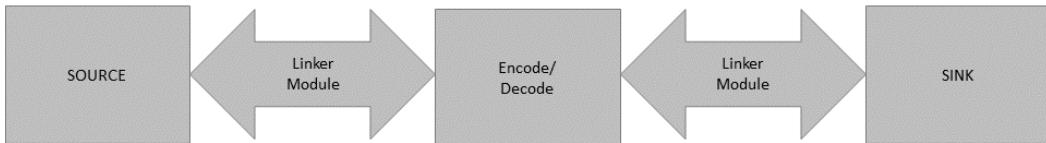
**LINKER\_MODULE** connect different type of module and deal with inter module communication, included siso, simo, miso and mimo.



In order to use the MMFv2, here are some aspects must to be followed.

- Define valid source
- Define valid sink
- Define valid codec (encode/decode) if needed.
- Define valid linker modules to link the above media modules.

The following picture shows the main usage flow to initialize different **MM\_MODULE**, and connect different **MM\_MODULE** through **LINKER\_MODULE**.



### 6.1.1 MM\_Module Prototype

MMFv2 allows users to define customized source, sink and encoder/decoder modules depending on the application. Although implementation details may be different, basic rules of the MMF structure are similar.

The MMFv2 requires users to predefine both source and sink modules through implementing create, destroy, control, handle, new\_item, del\_item and rsz\_item function callbacks. The structure `mmf_module_t` provides the interface for communication between mmf modules. In order to maintain the flexibility and convenience between modules, modules only retain the interface of each type to provide module to access. Function's constant of each module is defined by module itself.

```

typedef struct mm_module_s {
 void* (*create)(void *);
 void* (*destroy)(void *);
 int (*control)(void *, int, int);
 int (*handle)(void *, void *, void *);
 void* (*new_item)(void *);
 void* (*del_item)(void *, void *);
 void* (*rsz_item)(void *, void *, int);
 void* (*vrelease_item)(void *, void *, int);

 uint32_t output_type;
 uint32_t module_type;
 char * name;
} mm_module_t;

```

#### 6.1.1.1 Function description

- **create**

Pointer to the function that loads and initializes the module that you wish to add. For example, for Audio source, it points to the function in which the Audio driver is initialized and the corresponding context is returned.

- **destroy**

Pointer to the function that de-initializes module instance and releases resource. For example, for Audio source, it points to function in which Audio driver is initialized and the corresponding context is released.

- **control**

Pointer to function that sends the control command to the MMF module layer (see `mm_module_ctrl`) or a specific module. For example, for Audio source, it points to function that controls Audio parameters ("sample rate", "word length", "mic gain", etc.) and MMFv2 service task on or off.

- **handle**

Pointer to the function that manipulates media data (how to produce data in source or how to consume data in sink). Data is transferred from source to sink and vice versa by means of OS message queue. Please note that MMF service task reacts differently based on message exchange buffer status.

- **new\_item**

Pointer to the function that creates queue item that will be send to input and output queue, will only be used when setting `MM_CMD_INIT_QUEUE_ITEMS` to `MMQI_FLAG_STATIC`.

- **del\_item**

Pointer to the function that destroys queue item, will only be used when setting `MM_CMD_INIT_QUEUE_ITEMS` to `MMQI_FLAG_STATIC`.

- **rsz\_item**

Pointer to the function decreases memory pool size, will only be used when video (H264, HEVC (H265)...) and AAC module is created.

- **output\_type and module\_type**

`Output_type` indicates output mode. There are `MM_TYPE_NONE`, `MM_TYPE_VSRC`, `MM_TYPE_ASRC`, `MM_TYPE_VDSP`, `MM_TYPE_ADSP`, `MM_TYPE_VSINK`, `MM_TYPE_ASINK`, and `MM_TYPE_AVSSINK` can be used, corresponding to different module usage scenarios, let application know which mode the output is. `module_type` represents the identity of the module, and there are three options can be used `MM_MASK_SRC`, `MM_MASK_DSP` and `MM_MASK_SINK`.

- **name**

Pointer to the module name.

### 6.1.1.2 mm\_module\_ctrl

Here lists some commands defined in MMF module layer. Call by mm\_module\_ctrl (mm\_context\_t \*ctx, int cmd, int arg) to use them.

- MM\_CMD\_INIT\_QUEUE\_ITEMS: initialize static queue item.
- MM\_CMD\_SET\_QUEUE\_LEN: Set one queue's length.
- MM\_CMD\_SET\_QUEUE\_NUM: Set number of queue, not more than 3.
- MM\_CMD\_SELECT\_QUEUE: select queue from multi queues.
- MM\_CMD\_CLEAR\_QUEUE\_ITEMS: clear queue item.

## 6.1.2 Context

MMFv2 context supplies message transfer between different modules. It contains mm\_context\_t, and queue that used to pass data. There are 6 types of status that mm\_context support (MM\_STAT\_INIT, MM\_STAT\_READY, MM\_STAT\_ERROR, MM\_STAT\_ERR\_MALLOC, MM\_STAT\_ERR\_QUEUE, MM\_STAT\_ERR\_NEWITEM), these status are responsible for maintaining the module state to ensure the program runs smoothly.

```
typedef struct mm_context_s {
 union {
 struct {
 xQueueHandle output_ready;
 xQueueHandle output_recycle;
 int32_t item_num;
 };
 mm_conveyor_t port[4];
 };

 mm_module_t* module;

 void* priv; // private data structure for created instance

 // module state
 uint32_t state;
 int32_t queue_num; // number of queue
 int32_t curr_queue;
} mm_context_t;
```

The mm\_context is responsible for maintaining each module entity. MMFv2 support these modules (video, AAC\_encoder, AAC\_decoder, audio, g711, opus, mp4, rtp, rtsp) by default. Each module is independent and corresponding to the individual input/ output queue, state and in the mm\_context of the module to update parameters and delivery entities.

## 6.1.3 Module Inter Connection

This section introduces mm\_siso\_t, mm\_simo\_t, mm\_miso\_t, mm\_mimo\_t and its corresponding create, delete, ctrl, start, stop, pause, resume function, which is responsible for connection and control between modules in mmfv2.

### 6.1.3.1 SISO module (Single Input Single Output)

The SISO module is a unidirectional interface between modules. Input and output are independent. The status of the SISO module is responsible for determining the correct process. The stack\_size is used to determine the size of the handler, while xTaskHandle task, task\_priority and taskname are reserved to control the use of the task, task priority and task name.

```
typedef struct mm_siso_s {
 mm_context_t *input;
 mm_context_t *output;
 int input_port_idx;
 // default is 0, can be set to 1 or 2 or 3 if source module support 2 or more output queue

 uint32_t status;
 uint32_t stack_size;
 uint32_t task_priority;
 char taskname[16];
 xTaskHandle task;
} mm_siso_t;
```

There are some functions in the SISO module responsible for the module inter-connection. By these functions, it will be simple to update the status of the task and are handed over to the task handler for the main processing:

- siso\_create

Pointer to the function that siso\_create declares the space of mm\_siso\_t and returns mm\_siso\_t entity after initialization.

- siso\_delete

Pointer to the function that stops SISO execution and free space of mm\_siso\_t entity.

- siso\_ctrl

Pointer to the function that sends the control command to siso module.  
MMIC\_CMD\_ADD\_INPUT link the input module to the input of the siso module.  
MMIC\_CMD\_ADD\_OUTPUT link the output module to the output of the siso module.  
MMIC\_CMD\_SET\_TASKPRIORITY set the task priority for the linker task. If setting as 0, it will be configured to tskIDLE\_PRIORITY + 1 automatically.  
MMIC\_CMD\_SET\_TASKNAME set the task names for the linker task.  
MMIC\_CMD\_SET\_STACKSIZE add size to the stack\_size of siso.

 **NOTE**

*For consistency, the setting task size will be divided by 4. Make sure setting an enough and valid stack\_size for the task.*

- siso\_start

Pointer to the function that checks whether there is anything in the input and output module before siso start. If the answer is yes, siso task will create a task handler to send data from input module to the output module.

- siso\_stop

Pointer to the function that updates status to MMIC\_STAT\_SET\_EXIT and wait for task handler to switch status to MMIC\_STAT\_EXIT.

- siso\_pause

Pointer to the function that updates status to MMIC\_STAT\_SET\_PAUSE and wait for task handler to switch status to MMIC\_STAT\_PAUSE.

- siso\_resume

Pointer to the function that updates status to MMIC\_STAT\_SET\_RUN and wait for the task handler to switch status to MMIC\_STAT\_RUN.

#### 6.1.3.2 SIMO module (Single Input Multiple Output)

The SIMO module is a unidirectional interface between modules. Input and output are independent, and output\_cnt represents the number of simultaneous output modules. The array – status[4] maintains the state of the SIMO module to check the process is correct in the middle of the transfer, stack\_size is used to determine the size of the handler task for intermediate transfers. Similarly, it also provides xTaskHandle task, task\_priority, taskname for xTaskCreate. Note that each output will be served by one unique task and pause mask will control which output will be blocked.

```
typedef struct mm_simo_s {
 mm_context_t *input;
 int output_cnt;
 mm_context_t *output[4];
 // internal queue to handle reference count and usage log
 mm_simo_queue_t queue;

 uint32_t pause_mask;
 uint32_t status[4];
 uint32_t stack_size;
 uint32_t task_priority;
 char taskname[4][16];
 xTaskHandle task[4];
} mm_simo_t;
```

There are some functions in the SIMO module responsible for the module inter-connection. By these functions, it will be simple to update the status of the task and are handed over to the task handler for the main processing:

- simo\_create

Pointer to the function that simo\_create declares the space of mm\_simo\_t entity and returns mm\_siso\_t after initialization, and simo\_create create a queue head and a queue lock to protect the results of multiple outputs.

- simo\_delete

Pointer to the function that calls simo\_stop() to stop SIMO execution and free space.

- simo\_ctrl

Pointer to the function that sends the control command to simo module.

MMIC\_CMD\_ADD\_INPUT link the input module to the input of the simo module.

MMIC\_CMD\_ADD\_OUTPUT0, MMIC\_CMD\_ADD\_OUTPUT1, MMIC\_CMD\_ADD\_OUTPUT2, MMIC\_CMD\_ADD\_OUTPUT3 link output module to the corresponding output and increase the output\_cnt to record number of output modules.

MMIC\_CMD\_SET\_TASKPRIORITY set the task priority for the linker task. If setting as 0, it will be configured to tskIDLE\_PRIORITY + 1 automatically.

MMIC\_CMD\_SET\_TASKNAME set the task names for the linker task corresponding to MMIC\_CMD\_ADD\_OUTPUTx (x = 0~3).

MMIC\_CMD\_SET\_STACKSIZE add size to simo stack\_size.

 **NOTE**

*For consistency, the setting task size will be divided by 4 and it means each task will only have task\_size/4 for task stack size. Make sure setting an enough and valid stack\_size for the task.*

- simo\_start

Pointer to the function that simo\_start will create corresponding number of task handlers based on simo -> output\_cnt, and each task handler

will be used to send the received data.

- simo\_stop

Pointer to the function that simo\_stop sets each simo status to MMIC\_STAT\_SET\_EXIT, and waits for the task handler to switch each status to MMIC\_STAT\_EXIT.

- simo\_pause

Pointer to the function that simo\_pause will set each simo -> status to MMIC\_STAT\_SET\_PAUSE according to pause\_mask, and wait for the task handler to switch each status to MMIC\_STAT\_PAUSE.

- simo\_resume

Pointer to the function that simo\_resume will set each simo -> status to MMIC\_STAT\_SET\_RUN, and wait for the task handler to switch each status to MMIC\_STAT\_RUN.

#### 6.1.3.3 MISO module (Multiple Input Single Output)

The MISO module is a unidirectional interface between modules. Input and output are independent, and input\_cnt represents the number of simultaneous input modules. The status maintains the state of the MISO module to check the process is correct in the middle of the transfer, stack\_size is used to determine the size of the handler task for intermediate transfers, and finally the xTaskHandle task, task\_priority and taskname are reserved for xTaskCreate to control the use of the task. The pause\_mask can be controlled to block the inputs or the single output.

```
typedef struct mm_miso_s {
 int input_cnt;
 mm_context_t *input[4]; // max 4 input
 int input_port_idx[4];

 mm_context_t *output;

 uint32_t pause_mask;
 uint32_t status;
 uint32_t stack_size;
 uint32_t task_priority;
 char taskname[16];
 xTaskHandle task;
} mm_miso_t;
```

There are some functions in the MISO module responsible for the module inter-connection. By these functions, it will be simple to update the status of the task and are handed over to the task handler for the main processing:

- miso\_create

Pointer to the function that space of mm\_miso\_t is declared in miso\_create and initialized to return mm\_miso\_t entity.

- miso\_delete

Pointer to the function that calls miso\_stop() to stop MISO and free space.

- miso\_ctrl

Pointer to the function that sends the control command to miso module.

MMIC\_CMD\_ADD\_INPUT0, MMIC\_CMD\_ADD\_INPUT1, MMIC\_CMD\_ADD\_INPUT2, MMIC\_CMD\_ADD\_INPUT3 couple input modules to the corresponding miso input and increase the value of input\_cnt for number of input module.

MMIC\_CMD\_ADD\_OUTPUT links the output module to the output of the miso module.

MMIC\_CMD\_SET\_TASKPRIORITY set the task priority for the linker task. If setting as 0, it will be configured to tskIDLE\_PRIORITY + 1 automatically.

MMIC\_CMD\_SET\_TASKNAME set the task names for the linker task.

MMIC\_CMD\_SET\_STACKSIZE add size to miso stack\_size.

##### NOTE

*For consistency, the setting task size will be divided by 4. Make sure setting an enough and valid stack\_size for the task.*

- miso\_start

Pointer to the function that checks whether there is anything in the input and output module before starting. If the answer is yes, a task handler will be created, and the data of the input module will be sent to the output module.

- miso\_stop

Pointer to the function that sets the miso status to MMIC\_STAT\_SET\_EXIT and wait for the task handler to switch the status to MMIC\_STAT\_EXIT.

- miso\_pause

Pointer to the function that miso\_pause will set miso -> status to MMIC\_STAT\_SET\_PAUSE according to pause\_mask, waiting for the task handler to switch status to MMIC\_STAT\_PAUSE.

- miso\_resume

Pointer to the function that miso\_resume will set miso -> status to MMIC\_STAT\_SET\_RUN, waiting for the task handler to switch each status to MMIC\_STAT\_RUN.

#### 6.1.3.4 MIMO module (Multiple Input Multiple Output)

The MIMO module is a unidirectional interface between modules, Input[4] and output[4] represent input and output modules respectively, and input\_cnt represents the number of simultaneous input modules. Input and output support up to 4 outputs at the same time, MIMO module also needs mm\_mimo\_queue\_t queue[4] to maintain the synchronization problem of each input queue. Each mm\_mimo\_queue\_t has a lock and head to record the beginning of each queue and whether a program is already in use. The array, status[4], maintains the state of the MIMO module to determine the correct process in the middle of the transfer, stack\_size is used to determine the size of the handler task for the intermediate transfer, and the xTaskHandle task of xTaskCreate is reserved to control the use of the task. The array, pause\_mask[4], is used to control the input or output streaming for each task.

```
typedef struct mm_mimo_s {
 int input_cnt;
 // depend on input count
 mm_context_t* input[4];
 mm_mimo_queue_t queue[4];
 int output_cnt;
 // depend on output count
 uint32_t pause_mask[4];
 mm_context_t* output[4]; // output module context
 uint32_t output_dep[4]; // output depend on which input, bit mask
 uint32_t input_mask[4]; // convert from output_dep, input referenced by
 which output, bit mask
 uint32_t status[4];
 uint32_t stack_size;
 uint32_t task_priority;
 char taskname[4][16];
 xTaskHandle task[4];
} mm_mimo_t;
```

There are some functions in the MIMO module responsible for the module inter-connection. By these functions, it will be simple to update the status of the task and are handed over to the task handler for the main processing:

- mimo\_create

Pointer to the function mimo\_create declares the space of mm\_mimo\_t entity and returns mm\_mimo\_t after initialization.

- mimo\_delete

Pointer to the function that calls mimo\_stop() to stop the mimo module and free space.

- mimo\_ctrl

Pointer to the function that sends the control command to miso module.

MMIC\_CMD\_ADD\_INPUT0, MMIC\_CMD\_ADD\_INPUT1, MMIC\_CMD\_ADD\_INPUT2, and MMIC\_CMD\_ADD\_INPUT3 link input module to the input corresponding to the mimo module and increase the value of input\_cnt to record the number of input modules.

MMIC\_CMD\_ADD\_OUTPUT0, MMIC\_CMD\_ADD\_OUTPUT1, MMIC\_CMD\_ADD\_OUTPUT2, and MMIC\_CMD\_ADD\_OUTPUT3 couple the output module to the output of the mimo module and increase the value of output\_cnt to record the number of output modules. The inputs corresponding to outputs modules can be set by arg2 of mimo\_ctrl using the union of MMIC\_CMD\_ADD\_INPUTx.

MMIC\_CMD\_SET\_TASKPRIORITY set the task priority for the linker task. If setting as 0, it will be configured to tskIDLE\_PRIORITY + 1 automatically. MMIC\_CMD\_SET\_TASKNAME set the task names for the linker task.

##### NOTE

*For consistency, the setting task size will be divided by 4 and it means each task will only have task\_size/4 for task stack size. Make sure setting an enough and valid stack\_size for the task.*

- mimo\_start

Pointer to the function that mimo\_start will generate corresponding task handler according to output\_cnt to transfer the received data.

- mimo\_stop

Pointer to the function that mimo\_stop will set the mimo status to MMIC\_STAT\_SET\_EXIT according to output\_cnt, and waiting for the task handler switch the status to MMIC\_STAT\_EXIT.

- mimo\_pause

Pointer to the function that mimo\_pause will set each mimo -> status to MMIC\_STAT\_SET\_PAUSE according to pause\_mask, and waiting for the task handler to switch status to MMIC\_STAT\_PAUSE.

- mimo\_resume

Pointer to the function that mimo\_resume will set mimo -> status in the task of MMIC\_STAT\_PAUSE for each status to MMIC\_STAT\_SET\_RUN, and waiting for the task handler to switch each status to MMIC\_STAT\_RUN.

## 6.2 MM\_Module Type and Module Parameter

### 6.2.1 Video

The video module processes the data from sensor and outputs the video streaming data for user.

Here shows the context of the video module.

```
typedef struct video_ctx_s {
 void *parent;

 hal_video_adapter_t *v_adp;
 void *mem_pool;

 video_params_t params;
 int (*snapshot_cb)(uint32_t, uint32_t);
 void (*change_parm_cb)(void *);
 video_state_t state;
} video_ctx_t;
```

- **v\_adp:** Point to the video adapter which will use in the video process.
- **params:** Basic parameters for the video module.
- **snapshot\_cb:** Set the callback function for snapshot, which will be called while doing snapshot. It could be set by using **CMD\_VIDEO\_SNAPSHOT\_CB**.

#### 6.2.1.1 Basic video module parameters setting

Presetting the voe\_heap\_size:

Use **CMD\_VIDEO\_SET\_VOE\_HEAP** to set up the heap size that will be used in the voe process, including the output buffer for ISP (, snapshot) and Encoder, before setting the video parameters.

Here are some video module parameters provided to set.

```
typedef struct video_param_s {
 uint32_t stream_id;
 uint32_t type;
 uint32_t resolution;
 uint32_t width;
 uint32_t height;
 uint32_t bps;
 uint32_t fps;
 uint32_t gop;
 uint32_t rc_mode;
 uint32_t jpeg_qlevel;
 uint32_t rotation;
 uint32_t out_buf_size;
 uint32_t out_rsvd_size;
 uint32_t direct_output;
 uint32_t use_static_addr;
 uint32_t fcs;
 uint32_t use_roi;
 struct video_roi_s {
 uint32_t xmin;
 uint32_t ymin;
 uint32_t xmax;
 uint32_t ymax;
 } roi;
}
```

} video\_params\_t; Use **CMD\_VIDEO\_SET\_PARAMS** to set up the VIDEO parameters.

- **stream\_id:** Select the ISP channel, it can be set from 0~4.
- **type:** Select the video encode type. Currently support HEVC (VIDEO\_HEVC), H264 (VIDEO\_H264), JPEG (VIDEO\_JPEG), NV12 (VIDEO\_NV12), RGB (VIDEO\_RGB), NV16 (VIDEO\_NV16), HEVC+JPEG (VIDEO\_HEVC\_JPEG) and H264+JPEG (VIDEO\_H264\_JPEG).
- **resolution:** Set the video frame resolution. Currently support VIDEO\_QCIF (144\*176), VIDEO\_CIF (288\*352), VIDEO\_WVGA (360\*640), VIDEO\_VGA (480\*640), VIDEO\_D1 (480\*720), VIDEO\_HD (720\*1280), VIDEO\_FHD (1080\*1920), VIDEO\_3M (1536\*2048), VIDEO\_5M (1944\*2592).
- **width:** Set the video frame resolution's width.
- **height:** Set the video frame resolution's height.
- **bps:** Configure the video encoder's bit rate (bits per second).

- fps: Configure the video module output frame rate (frames per second).
- gop: Set the group of the picture which can be seen as the cycle that 1 frame will update.
- rc\_mode: Determine use CBR (1) or VBR (2).
- direct\_output: If set 1, the video module output will not be sent to the video module output ready queue.
- use\_static\_addr: If setting use\_static\_addr to 1, the output\_item data address will directly point to the isp\_addr; while setting to 0, it will allocate a new space for the output item address.

**NOTE**

The video resolution must be less than the sensor's maximum width or height.

#### 6.2.1.2 Video module rate control (RC) adjustment

### 6.2.2 RTSP

```
typedef struct rtsp2_params_s {
 uint32_t type;
 union {
 struct rtsp_video_param_s {
 uint32_t codec_id;
 uint32_t fps;
 uint32_t bps;
 uint32_t ts_flag;
 char* sps;
 char* pps;
 char* lv;
 } v;
 struct rtsp_audio_param_s {
 uint32_t codec_id;
 uint32_t channel;
 uint32_t samplerate;
 } a;
 struct rtsp_audio_opus_param_s {
 uint32_t codec_id;
 uint32_t channel;
 uint32_t samplerate;
 uint32_t max_average_bitrate;
 uint32_t frame_size;
 } a_opus;
 } u;
} rtsp2_params_t;
```

Use **CMD\_RTSP2\_SELECT\_STREAM** to select the RTSP stream index, currently support 0 and 1.

Use **CMD\_RTSP2\_SET\_PARAMS** to set up the RTSP parameters.

- type: Media type, available Video (AVMEDIA\_TYPE\_VIDEO), Audio (AVMEDIA\_TYPE\_AUDIO).
- codec\_id: RTSP supported codec ID, available AV\_CODEC\_ID\_MJPEG, AV\_CODEC\_ID\_H264, AV\_CODEC\_ID\_PCMU, AV\_CODEC\_ID\_PCMA, AV\_CODEC\_ID\_MP4A\_LATM, AV\_CODEC\_ID\_MP4V\_ES, AV\_CODEC\_ID\_H265, AV\_CODEC\_ID\_OPUS, AV\_CODEC\_ID\_RGB888.
- fps: Video frame rate.
- bps: Bit per second
- ts\_flag: H264 rtsp time sync enable switch.
- sps,pps,lv: Set sps, pps and profile level of H264.
- channel: Audio channel.
- samplerate: Audio samplerate.
- max\_average\_bitrate: Set the max\_average\_bitrate for OPUS rtsp.
- frame\_size: Set the using OPUS encode frame size (the unit is msec) which will be related to the timestamp increase of opus rtp packet.

**Current codec table:**

```
static const struct codec_info av_codec_tables[] = {
 {AV_CODEC_ID_MJPEG, "MJPEG", RTP_PT_JPEG, 90000, 0, 0},
 {AV_CODEC_ID_H264, "H264", RTP_PT_DYN_BASE, 90000, 0, 0},
 {AV_CODEC_ID_PCMU, "PCMU", RTP_PT_PCMU, 8000, 1, 0},
```

```
{AV_CODEC_ID_PCMA, "PCMA", RTP_PT_PCMA, 8000, 1, 0},
{AV_CODEC_ID_MP4A_LATM, "MP4A", RTP_PT_DYN_BASE, 8000, 2, 0},
{AV_CODEC_ID_MP4V_ES, "MP4V", RTP_PT_DYN_BASE, 90000, 0, 0},
{AV_CODEC_ID_H265, "H265", RTP_PT_DYN_BASE, 90000, 0, 0},
{AV_CODEC_ID_OPUS, "opus", RTP_PT_DYN_BASE, 48000, 2, 0}
};
```

### 6.2.3 AAC Encoder (AAC)

```
typedef struct aac_params_s {
 uint32_t sample_rate; // 8000
 uint32_t channel; // 1
 uint32_t bit_length; // FAAC_INPUT_16BIT
 uint32_t output_format; // 0: Raw 1: ADTS
 uint32_t mpeg_version; // 0: MPEG4 1: MPEG2

 uint32_t mem_total_size;
 uint32_t mem_block_size;
 uint32_t mem_frame_size;

 int samples_input;
 int max_bytes_output;
} aac_params_t;
```

Use **CMD\_AAC\_SET\_PARAMS** to set up the AAC parameters.

- **sample\_rate:** Sample rate for AAC encoder must be the same as the Audio codec setting. For instance, if using ASR\_8KHZ as the Audio codec sample rate, the sample rate of AAC must be configured to 8000 or the codec result will be unexpected.
- **channel:** Set the audio channel number. The mono is set as 1, while the stereo is set as 2. This setting is related to the Audio codec.
- **bit\_length:** The bit length use in AAC encoder. The bit length configuration must be identical to the Audio codec, like if audio codec word length is equal to WL\_16BIT, which must be set to FAAC\_INPUT\_16BIT.
- **output\_format:** The AAC output format, the default setting is 1 (ADTS).
- **mpeg\_version:** Setting MPEG version, the default setting is 0 (MPEG4).
- **mem\_total\_size:** Memory pool size of AAC encoder output.
- **mem\_block\_size:** Block size used by Memory pool.
- **mem\_frame\_size:** Set maximum FRAME SIZE capacity.
- **samples\_input:** It will be automatically configured when AAC initialization, no need to do setting.
- **max\_bytes\_output:** It will be automatically configured when AAC initialization, no need to do setting.

### 6.2.4 AAC Decoder (AAD)

```
typedef struct aad_params_s {
 uint32_t type; // TYPE_RTP_RAW or TYPE_ADTS
 uint32_t sample_rate; // 8000
 uint32_t channel; // 1
} aad_params_t;
```

Use **CMD\_AAD\_SET\_PARAMS** to set up the AAD parameters.

- **type:** TYPE\_ADTS is used when the source is AAC encoder, TYPE\_RTP\_RAW is used when source is RTP, and TYPE\_TS is not currently supported.
- **sample\_rate:** Need to match source sample rate to decode correctly.
- **channel:** Need to match source channel to decode correctly.

### 6.2.5 Audio Codec

The ASP algorithms, AGC (Automatic gain control), ANS (Adaptive noise suppression), AEC (Acoustic echo cancellation) and VAD (Voice Activity Detection), are included in this module.

```
typedef struct audio_params_s {
 audio_sr sample_rate; // ASR_8KHZ
 audio_wl word_length; // WL_16BIT
 audio_mic_gain mic_gain; // MIC_40DB
 audio_dmic_gain dmic_l_gain; // DMIC_BOOST_24DB
 audio_dmic_gain dmic_r_gain; // DMIC_BOOST_24DB
};
```

```

 int channel; // 1
 int enable_aec; // 0: off 1: on
 int enable_ns; // 0: off, 1: out 2: in 3: in/out
 int enable_agc; // 0: off, 1: output agc
 int enable_vad; // 0: off 1: input vad
 int mix_mode; // 0
 uint8_t use_mic_type; // 0: AMIC 1: LEFT_DMIC 2: RIGHT_DMIC 3:
 STEREO_DMIC
 } audio_params_t;
}

```

Use **CMD\_AUDIO\_SET\_PARAMS** to set up the audio parameters.

- sample\_rate: Currently support 8K (ASR\_8KHZ), 16K, 32K, 44.1K (ASR\_44p1KHZ), 48K, 88.2K, 96K HZ.
- word\_length: Currently support 16 bits (WL\_16BIT), 24 bits (WL\_24BIT).
- mic\_gain: Analog microphone gain value. Support 0, 20, 30, 40 DB.
- dmic\_l\_gain: Left digital gain value. Support 0, 12, 24, 36 DB.
- dmic\_r\_gain: Right digital gain value. Support 0, 12, 24, 36 DB.
- channel: The number of channel is supported. Currently, support mono so set it to 1.
- enable\_aec, enable\_vad: switch of the AEC and VAD while audio RX process. Set 1 to enable them while 0 to disable the function.
- enable\_ns, enable\_agc: switch of ANS and AGC of audio TX and RX. 0 to turn off them, 1 only turn on the process in TX, while 2 to enable them on RX and finally if you want to enable algorithm of TX and RX, set it to 3.
- mix\_mode: enable mix mode of all the logic input, the default setting is 0.
- use\_mic\_type: set the mic type, 0 is the analog microphone, 1 is the left digital mic, 2 is the right digital mic and 3 is the stereo digital mic.

#### **NOTE**

*See the more ASP setting detail in 15.2 Open ASP algorithm*

### 6.2.6 RTP Input

```

typedef struct rtp_param_s {
 uint32_t valid_pt;
 uint32_t port;
 uint32_t frame_size;
 uint32_t cache_depth;
} rtp_params_t;

```

Use **CMD\_AUDIO\_SET\_PARAMS** to set up the audio parameters.

- valid\_pt: Processable RTP payload types. Set 0xFFFFFFFF to handle RTP\_PT\_PCMU (0), RTP\_PT\_PCMA (8) and RTP\_PT\_DYN\_BASE (dynamic, default setting 96).
- port: The port to receive the RTP packet.
- frame\_size: Maximum RTP packet size.
- cache\_depth: The number of caches for RTP packets. The cache handler will send the RTP packet in the cache to the output of the module when the number of packets in the cache >= 50% cache depth.

### 6.2.7 G711 Codec

G711 Encode and G711 Decode use the same parameter structure.

```

typedef struct g711_param_s {
 uint32_t codec_id; // AV_CODEC_ID_PCMA or AV_CODEC_ID_PCMU
 uint32_t buf_len; // output buffer length
 uint32_t mode; // decode or encode
} g711_params_t;

```

Use **CMD\_G711\_SET\_PARAMS** to set up the G711 parameters.

- codec\_id: Set the codec type for G711 encoder/decoder. G711 currently supports PCMU (AV\_CODEC\_ID\_PCMA) and PCMA (AV\_CODEC\_ID\_PCMU) codec modes.
- buf\_len: Determine the length (byte) of the encode buffer.
- mode: Determine whether the G711 codec module is an encoder (G711\_ENCODE) or decoder (G711\_decode).

### 6.2.8 OPUS Encoder (OPUSC)

```

typedef struct opusc_param_s {
 uint32_t sample_rate; // 8000

```

```

 uint32_t channel; // 1
 uint32_t bit_length; // 16
 uint32_t complexity;
 uint32_t use_framesize;

 //VBR CBR setting
 uint32_t bitrate; //default 25000
 uint32_t enable_vbr;
 uint32_t vbr_constraint;
 uint32_t packetLossPercentage;

 uint32_t opus_application;

 int samples_input;
 int max_bytes_output;

} opusc_params_t;

```

Use **CMD\_OPUSC\_SET\_PARAMS** to set up the OPUSC parameters.

- **sample\_rate**: Sample rate for OPUS encoder must be the same as the Audio codec setting. For instance, if using ASR\_8KHZ as the Audio codec sample rate, the sample rate of OPUS must be configured to 8000 or the codec result will be unexpected.
- **channel**: Set the audio channel number. The mono is set as 1, while the stereo is set as 2. This setting is related to the Audio codec.
- **bit\_length**: The bit length use in OPUS encoder. The bit length configuration must be identical to the Audio codec, like if audio codec word length is equal to WL\_16BIT, which must be set to 16.
- **complexity**: Set the opus encoder's complexity, and the value is from 0 (low complexity) to 10 (high complexity). The higher complexity is configured the better quality encoding at a given bitrate but it also means more CPU consumption.
- **use\_framesize**: The frame size contains in one OPUS packet. Since it will be related to the opus rtsp timestamp, if using RTSP, this must be the same as frame\_size in rtsp module. Recommend to be the same or larger than AUDIO\_DMA\_PAGE\_SIZE/(sample\_rate / 1000)/2 but less than 60.
- **bitrate**: Set the bit rate for the opus encoder, the default value is 25000.
- **enable\_vbr**: Enable VBR (variable bit rate) of the opus encoder.
- **vbr\_constraint**: Makes constrained VBR if setting as 1.
- **packetLossPercentage**: Set the percentage of packet loss, the default value is 0.
- **opus\_application**: Set the opus application type, broadcast/high-fidelity application (OPUS\_APPLICATION\_AUDIO), VoIP/videoconference applications (OPUS\_APPLICATION\_VOIP) and lowest-achievable latency (OPUS\_APPLICATION\_RESTRICTED\_LOWDELAY). The default setting is OPUS\_APPLICATION\_AUDIO.
- **samples\_input**: Not need to be set, it will be automatically set in the process of opus encoder.
- **max\_bytes\_output**: Not need to be set, it will be automatically set in the process of opus encoder.

## 6.2.9 OPUS Decoder (OPUSD)

```

typedef struct opusd_param_s {
 uint32_t sample_rate; // 8000
 uint32_t channel; // 1
 uint32_t bit_length; // 16
 uint32_t frame_size_in_msec;
 uint32_t opus_application;
 uint8_t with_opus_enc;

 int samples_input;
 int max_bytes_output;

} opusd_params_t;

```

Use **CMD\_OPUSD\_SET\_PARAMS** to set up the OPUSD parameters.

- **sample\_rate**: The sample of the opus packet will be decoded, must be the same as the audio codec.
- **channel**: **Need to match source channel to decode correctly.**
- **bit\_length**: The audio bit length will be decoded, suggest to set as 16.
- **frame\_size\_in\_msec**: No need to be set, it will be automatically set when using it.
- **opus\_application**: Set the opus application type, broadcast/high-fidelity application (OPUS\_APPLICATION\_AUDIO), VoIP/videoconference applications (OPUS\_APPLICATION\_VOIP) and lowest-achievable latency (OPUS\_APPLICATION\_RESTRICTED\_LOWDELAY). The default setting is OPUS\_APPLICATION\_AUDIO.

- `with_opus_enc`: Set to 1, if the application with opus encoder.
- `samples_input`: Not need to be set, it will be automatically set in the process of opus decoder.
- `max_bytes_output`: Not need to be set, it will be automatically set in the process of opus decoder.

## 6.2.10 MP4

```
typedef struct mp4_param_s {
 uint32_t width;
 uint32_t height;
 uint32_t fps;
 uint32_t gop;

 uint32_t sample_rate;
 uint32_t channel;

 uint32_t record_length;
 uint32_t record_type;
 uint32_t record_file_num;
 char record_file_name[32];
 uint32_t fatfs_buf_size;
 uint32_t mp4_user_callback;
} mp4_params_t
```

Use **CMD\_MP4\_SET\_PARAMS** to set up the MP4 parameters.

- `width`: Set the max video frame width.
- `height`: Set the max video frame height.
- `fps`: Set the frame number per second.
- `gop`: Set the group of the picture which can be seemed as the cycle that I frame will update.
- `sample_rate`: The audio sample rate.
- `channel`: The audio channel number.
- `record_length`: Set the record file length in second.
- `record_type`: Set the record media type, `STORAGE_ALL` (with bot audio and video), `STORAGE_VIDEO` (video only), `STORAGE_AUDIO` (audio only).
- `record_file_num`: Set the number of file that will be recorded.
- `record_file_name`: Set the record file name.
- `fatfs_buf_size`: FATFS cache buffer size.
- `mp4_user_callback`: Configure the user callback function. If enable this, be sure that callback function for open (`CMD_MP4_SET_OPEN_CB`), write (`CMD_MP4_SET_WRITE_CB`), seek (`CMD_MP4_SET_SEEK_CB`) and close (`CMD_MP4_SET_CLOSE_CB`) have been set.

## 6.2.11 I2S

```
typedef struct i2s_param_s {
 int sample_rate; // SR_32KHZ
 int out_sample_rate; // SR_8KHZ
 int word_length; // WL_24b
 int out_word_length; // WL_16b
 audio_mic_gain mic_gain; // MIC_-40DB
 int channel; // 1
 int out_channel;
 int enable_aec; // 0
 int mix_mode; // 0
} i2s_params_t;
```

Use **CMD\_I2S\_SET\_PARAMS** to set up the I2S parameters.

- `sample_rate`: Currently support 8K, 16K, 32K, 44.1K, 48K, 88.2K, 96K, 12K, 24K, 64K 192K, 384K, 7.35K, 11.025K, 14.7K, 22.05K, 58.8K, 176.4K HZ
- `out_sample_rate`: Currently supported sampling rate is the same as the sample rate, but less than or equal to `sample_rate`.
- `word_length`: 16 (WL\_16b), 24 (WL\_24b), 32 (WL\_32b) bits.
- `out_word_length`: Currently supported bit depth is the same as the `word_length`, but less than or equal to `word_length`.
- `mic_gain`: Microphone gain value. Support 0, 20, 30, 40 DB.
- `channel`: Currently supports stereo or mono, please set to 2 or 1, and also supports 5.1 channels (but only support tx).
- `out_channel`: Currently supported channel is the same as the `channel`, but less than or equal to `channel`.

- enable\_aec: The switch of enabling AEC.
- mix\_mode: The switch of enabling mix mode.

## 6.2.12 Httpfs

The httpfs module to construct a HTTP File Server and send the media file on it.

```
typedef struct httpfs_param_s {
 char fileext[4];
 char filedir[32];
 char request_string[128];
 uint32_t fatfs_buf_size;
} httpfs_params_t;
```

Use **CMD\_HTTPFS\_SET\_PARAMS** to set up the HTTPFS parameters.

- fileext: Set the file extension, for example "mp4".
- filedir: Directory where the file is located, for example "VIDEO".
- request\_string: The string of http page, for example "/video\_get.mp4".
- fatfs\_buf\_size: Buffer size of read file.

## 6.2.13 Array

The array module is use to play the small size and predefinition media streaming (like doorbell ring). It can be seemed as a source module.

```
typedef struct array_param_s {
 uint32_t type;
 uint32_t codec_id;
 uint8_t mode;
 union {
 struct array_video_param_s {
 uint32_t fps;
 uint8_t h264_nal_size;
 } v;
 struct array_audio_param_s {
 uint32_t channel;
 uint32_t samplerate;
 uint32_t sample_bit_length;
 uint32_t frame_size;
 } a;
 } u;
} array_params_t;

typedef struct array_s {
 uint32_t data_addr;
 uint32_t data_len;
 uint32_t data_offset;
} array_t;
```

Use the command **CMD\_ARRAY\_SET\_PARAMS** to set up the parameters for the array module.

- type: Media type, available Video (AVMEDIA\_TYPE\_VIDEO), Audio (AVMEDIA\_TYPE\_AUDIO).
- codec\_id: Set the codec ID of the array, like AV\_CODEC\_ID\_MJPEG, AV\_CODEC\_ID\_H264, AV\_CODEC\_ID\_PCMU, AV\_CODEC\_ID\_PCMA, AV\_CODEC\_ID\_MP4A\_LATM, AV\_CODEC\_ID\_MP4V\_ES, AV\_CODEC\_ID\_H265, AV\_CODEC\_ID\_OPUS, AV\_CODEC\_ID\_RGB888.
- mode: set the array play mode, once (ARRAY\_MODE\_ONCE) or repeat (ARRAY\_MODE\_LOOP).
- h264\_nal\_size: Set the NALU length of h264 or h265 media array.
- channel: Set the audio channel.
- samplerate: Set the audio sample rate.
- sample\_bit\_length: bit length for one audio sample.
- frame\_size: Set the using audio frame size (the unit is samples).

Use the command **CMD\_ARRAY\_SET\_ARRAY** to set up the array input.

- data\_addr: Set the media array store address.
- data\_len: Set the media array total size.
- data\_offset: Set the offset that will be started to play and it will also be used to keep the play location while the array module process.

## 6.3 Using the MMF example

Describe how to use the sample program to construct the applicational data stream .

In this section, there will be an introduction to correctly select the mmfv2 sample program and adjust the parameters.

### 6.3.1 Selecting and setting up sample program

For audio only samples, they are in function example\_mmf2\_audio\_only while video joined samples are listed in example\_mmf2\_video\_surport. Pick the example want to open before using it, remove the comment, and recompile. Opening more than two examples at the same time will result in unpredictable program execution results.

#### 6.3.1.1 Requisites and Setup

**Pre-requisites:**

- AmebaPro2 board
- Camera sensor board
- Micro USB cable
- WIFI (for transferring rtsp stream)
- MicroSD card (for saving the mp4 data)

**Hardware setup:**

- Connect the camera sensor board to the AmebaPro2's camera sensor board slot (CON1).
- Connect the PC with the AmebaPro2 CON8 port by the Micro USB cable.
- Insert the MicroSD card to the AmebaPro2's SD card slot.

**Software setup:**

- In project\realtek\_amebapro2\_v0\_example\inc\platform\_opts.h select the usage sensor.
- For audio only example, use "cmake .. -G"Unix Makefiles" -DCMAKE\_TOOLCHAIN\_FILE=../toolchain.cmake -DEXAMPLE=media\_framework" to build up the project.
- For video joined example, use "cmake .. -G"Unix Makefiles" -DCMAKE\_TOOLCHAIN\_FILE=../toolchain.cmake -DVIDEO\_EXAMPLE=on" to build up the project.
- Uncomment the example you want to execute.

The sample program is located at:

Audio only: \component\example\media\_framework\ example\_media\_framework.c

Video joined: \project\realtek\_amebapro2\_v0\_example\src\mmfv2\_video\_example\ video\_example\_media\_framework.c

For example: open mmf2\_video\_example\_joint\_test\_rtsp\_mp4\_init

```
// Joint test RTSP MP4
// H264 -> RTSP (V1)
// H264 -> MP4 (V2)
// AUDIO -> AAC -> RTSP and mp4
// RTP -> AAD -> AUDIO
//mmf2_video_example_joint_test_rtsp_mp4_init();
```

Uncomment the example want to execute

```
// Joint test RTSP MP4
// H264 -> RTSP (V1)
// H264 -> MP4 (V2)
// AUDIO -> AAC -> RTSP and mp4
// RTP -> AAD -> AUDIO
mmf2_video_example_joint_test_rtsp_mp4_init();
```

**NOTE**

*Uncomment two media examples in the same time may cause unexpected result.*

- Compile and execute firmware. The compilation and execution can refer to the previous chapter.

#### 6.3.1.2 Currently supported example

- Audio only examples:

| Example             | Description             | Result                                                                                                                                         |
|---------------------|-------------------------|------------------------------------------------------------------------------------------------------------------------------------------------|
| mmf2_example_a_init | audio -> AAC -> RTSP(A) | AmebaPro2's AAC sound stream over the network. The sound received by AmebaPro2 is encoded by AAC and then streamed through the network (rtsp). |

|                                   |                                                 |                                                                                                                                                                   |
|-----------------------------------|-------------------------------------------------|-------------------------------------------------------------------------------------------------------------------------------------------------------------------|
| mmf2_example_audioloop_init       | PCM audio -> PCM audio , audio loopback         | The sound received by AmebaPro2 can be broadcast from the 3.5 audio channel of AmebaPro2, and the PCM transmission is directly used in the procedure.             |
| mmf2_example_g711loop_init        | audio -> G711E -> G711D -> audio                | The sound received by AmebaPro2 can be broadcast from the 3.5 audio channel of AmebaPro2. PCM is encoded by G711 and transmit, then decoded by G711 and playback. |
| mmf2_example_aacloop_init         | audio -> AAC -> AAD -> audio                    | The sound received by AmebaPro2 can be broadcast from the 3.5 audio channel of AmebaPro2. PCM is encoded by AAC and transmit, then decoded by AAD and playback.   |
| mmf2_example_rtp_aad_init         | RTP -> AAD -> audio                             | Stream AAC sound over the network to AmebaPro2 for playback. Streaming audio is decoded by AAD and played through 3.5 audio jack.                                 |
| mmf2_example_2way_audio_init      | audio -> AAC -> RTSP<br>RTP -> AAD -> audio     | Stream AAC sound to AmebaPro2's audio jack via the network and transmit the sound received by AmebaPro2 over the network simultaneously.                          |
| mmf2_example_pcmu_array_rtsp_init | ARRAY (PCMU) -> RTSP (A)                        | Transmitting PCMU sound arrays within AmebaPro2 over the network.                                                                                                 |
| mmf2_example_aac_array_rtsp_init  | ARRAY (AAC) -> RTSP (A)                         | Transfer AAC sound arrays in AmebaPro2 over the network.                                                                                                          |
| mmf2_example_opusloop_init        | audio -> OPUSC -> OPUSD -> audio                | The sound received by AmebaPro2 can be broadcast from the 3.5 audio channel of AmebaPro2. PCM is encoded by OPUS and transmit, then decoded by OPUS and playback. |
| mmf2_example_a_opus_init          | Audio -> OPUSC -> RTSP(A)                       | AmebaPro2's OPUS sound stream over the network. The sound received by AmebaPro2 is encoded by OPUSC and then streamed through the network (rtsp).                 |
| mmf2_example_rtp_opusd_init       | RTP -> OPUSD -> audio                           | Stream OPUSC sound over the network to AmebaPro2 for playback. Streaming audio is decoded by OPUSD and played through 3.5 audio jack.                             |
| mmf2_example_2way_audio_opus_init | audio -> OPUSC -> RTSP<br>RTP -> OPUSD -> audio | Stream OPUS sound to AmebaPro2's audio jack via the network and transmit the sound received by AmebaPro2 over the network simultaneously.                         |
| mmf2_example_pcm_array_audio_init | Array (pcm) -> audio                            | Play the array pcm data through AmebaPro2                                                                                                                         |

- Video only examples:

| Example                             | Description                              | Result                                                                                                                                                         |
|-------------------------------------|------------------------------------------|----------------------------------------------------------------------------------------------------------------------------------------------------------------|
| mmf2_video_example_v1_init          | CH1 Video -> H264/H265 -> RTSP           | Transfer AmebaPro2's H264/HEVC video stream over the network. Video default format: 720P 30FPS.                                                                |
| mmf2_video_example_v2_init          | CH2 Video -> H264/H265-> RTSP            | Transfer AmebaPro2's H264/HEVC video stream over the network. Video default format: 1080P 30FPS.                                                               |
| mmf2_video_example_v3_init          | CH3 Video -> JPEG -> RTSP                | Transfer AmebaPro2's JPEG video stream over the network. Video default format: 1080P 30FPS.                                                                    |
| mmf2_video_example_v1_snapshot_init | CH1 Video -> H264/H265-> RTSP + SNAPSHOT | Transfer AmebaPro2's H264/HEVC video stream over the network and snapshot (JPEG) while streaming.                                                              |
| mmf2_video_example_simo_init        | 1 Video (H264/H265) -> 2 RTSP (V1, V2)   | Transmitting two H264/HEVC video streams from AmebaPro2 over the network, the source of the video is the same video stream. Video default format: 1080P 30FPS. |

|                                         |                                                         |                                                                                                                                                                                                      |
|-----------------------------------------|---------------------------------------------------------|------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------|
| mmf2_video_example_array_rtsp_init      | ARRAY (H264/H265) -> RTSP (V)                           | Transfer H264/HEVC stream array in AmebaPro2 over the network. Video default format: 25FPS.                                                                                                          |
| mmf2_video_example_v1_param_change_init | CH1 Video -> H264/H265-> RTSP (parameter change)        | Transfer AmebaPro2's H264/HEVC video over the network and support dynamic adjustment of video parameters. The parameters of dynamic adjustment are Resolution, Rate Control Mode, Bit Rate in order. |
| mmf2_video_example_h264_array_mp4_init  | ARRAY (H264/H265) -> MP4 (SD card)                      | AmebaPro2 will record H264/HEVC stream array to the SD card for 30 second. Video default format: 25FPS.                                                                                              |
| mmf2_video_example_md_rtsp_init         | CH1 Video -> H264/H265-> RTSP<br>CH4 Video -> RGB -> MD | RTSP video stream over the network.<br>MD detect motion and draw the motion region to RTSP channel.                                                                                                  |

- Video + Audio examples:

| Example                                     | Description                                                                                                                | Result                                                                                                                                                                                                                                                                                                                                                                                                                                                           |
|---------------------------------------------|----------------------------------------------------------------------------------------------------------------------------|------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------|
| mmf2_video_example_av_init                  | 1 Video (H264/H265) and 1 Audio -> AAC -> RTSP                                                                             | Transfer AmebaPro2's H264/HEVC video and AAC sound stream over the network. Video default format: 1080P 30FPS.                                                                                                                                                                                                                                                                                                                                                   |
| mmf2_video_example_av2_init                 | 2 Video (H264/H265) and 1 Audio -> AAC -> 2 RTSP (V1+A, V2+A)                                                              | Transmitting two H264/HEVC videos and AAC audio streams from AmebaPro2 over the network. The source of the videos is different ISP channel. The videos formats are set to 1080P 30FPS (V1) and 720P 30FPS (V2) respectively.                                                                                                                                                                                                                                     |
| mmf2_video_example_av21_init                | 1 Video (H264/H265) and 1 Audio -> 2 RTSP (V+A)                                                                            | Transfer two copies of AmebaPro2's H264/HEVC video (1080P 30FPS) and AAC sound stream through the network, the video source is the same ISP channel.                                                                                                                                                                                                                                                                                                             |
| mmf2_video_example_av_mp4_init              | 1 Video (H264/H265) and 1 Audio -> MP4 (SD card)                                                                           | AmebaPro2 will record three videos (1080P 30FPS) to the SD card for 30 seconds each The default storage name is :<br>AmebaPro2_recording_0.mp4<br>AmebaPro2_recording_1.mp4<br>AmebaPro2_recording_2.mp4                                                                                                                                                                                                                                                         |
| mmf2_video_example_av_rtsp_mp4_init         | Video (H264/H265) -> RTSP and mp4 AUDIO -> AAC -> RTSP and MP4                                                             | (1) Transfer AmebaPro2's H264/HEVC video and AAC sound stream over the network. Video default format: 1080P 30FPS.<br>(2) AmebaPro2 will record three videos (1080P 30FPS+AAC) to the SD card for 30 seconds each. The default storage name is :<br>AmebaPro2_recording_0.mp4<br>AmebaPro2_recording_1.mp4<br>AmebaPro2_recording_2.mp4<br>(3) Streaming AAC sounds to AmebaPro2 via the network.<br>Note: (1) video source of (2) is from the same ISP channel. |
| mmf2_video_example_joint_test_init          | Video (H264/H265) -> RTSP (V1+A)<br>Video (H264/H265) -> RTSP (V2+A)<br>AUDIO -> AAC -> RTSP<br>RTP -> AAD -> AUDIO        | (1) Transmitting two H264/HEVC video streams from AmebaPro2 over the network, the source of the video is the different video stream. Video default format: 1080P 30FPS (V1) and 720P 30FPS (V2).<br>(2) Streaming two copies of AAC sounds to AmebaPro2 via the network.                                                                                                                                                                                         |
| mmf2_video_example_joint_test_rtsp_mp4_init | Video (H264/H265) -> MP4 (V1+A)<br>Video (H264/H265) -> RTSP (V2+A)<br>AUDIO -> AAC -> RTSP and MP4<br>RTP -> AAD -> AUDIO | (1) Transfer AmebaPro2's H264/HEVC video and AAC sound stream over the network. Video default format: 1080P 30FPS.<br>(2) AmebaPro2 will record three videos (720P 30FPS+AAC) to the SD card for 30 seconds each. The default storage name is :                                                                                                                                                                                                                  |

|                                                   |                                                                                                                                |                                                                                                                                                                                                                                                                                                                                               |
|---------------------------------------------------|--------------------------------------------------------------------------------------------------------------------------------|-----------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------|
|                                                   |                                                                                                                                | AmebaPro2_recording_0.mp4<br>AmebaPro2_recording_1.mp4<br>AmebaPro2_recording_2.mp4<br>(3) Streaming AAC sounds to AmebaPro2 via the network.<br>(4) RTP send the audio stream from network to AmebaPro2 and the stream is decoded by AAD and played through 3.5 audio jack.<br>Note: (1) video source of (2) is from different ISP channels. |
| mmf2_video_example_2way_audio_pcm_u_doorbell_init | Video (H264/H265) -> RTSP (V1)<br>AUDIO -> G711E -> RTSP<br>RTP -> G711D -> AUDIO<br>ARRAY (PCMU) -> G711D -> AUDIO (doorbell) | (1) Transmitting AmebaPro2's H264/HEVC stream and PCMU sound stream over the network. Video default format: 1080P 30FPS.<br>(2) PCMU sound can be streamed to AmebaPro2 via the Internet and playback.<br>(3) Play PCMU sound array in AmebaPro2 (default is the doorbell).                                                                   |
| mmf2_video_example_2way_audio_pcm_u_init          | Video (H264/H265) -> RTSP (V1)<br>AUDIO -> G711E -> RTSP<br>RTP -> G711D -> AUDIO                                              | (1) Transmitting AmebaPro2's H264/HEVC stream and PCMU sound stream over the network. Video default format: 1080P 30FPS.<br>(2) PCMU sound can be streamed to AmebaPro2 via the Internet and playback.                                                                                                                                        |
| mmf2_video_example_av_mp4_httpfs_init             | 1 Video (H264) 1 Audio -> MP4 (SD card) Http File Server                                                                       | AmebaPro2 will record a video every 30 seconds and save it to the SD card (1080P 30FPS+AAC). The default is to record 60 files, and repeat the recording after the end.<br>The default storage name is:<br>mp4_record_0.mp4~mp4_record_29.mp4 Also open Http File Server for client to do playback.                                           |
| mmf2_video_example_h264_pcmu_array_mp4_init       | 1 Video array and 1 Audio array(pcmu) -> MP4 (SD card)                                                                         | Save 1 video stream and 1 pcmu audio stream to mp4 file (the record file may not play on some player)                                                                                                                                                                                                                                         |
| mmf2_video_example_demuxer_rtsp_init              | Demux a mp4 file in SD card (based on record file name) to 1 Video and 1 Audio -> RTSP                                         | Demux a mp4 file (suggest to use a file created by AmebaPro2) and send the video and audio data through rtsp                                                                                                                                                                                                                                  |

- Video + NN examples:

| Example                            | Description                                                   | Result                                                                                                                                                                             |
|------------------------------------|---------------------------------------------------------------|------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------|
| mmf2_video_example_vipnn_rtsp_init | CH1 Video -> H264/H265-> RTSP<br>CH4 Video -> RGB -> NN       | RTSP video stream over the network.<br>NN do object detection and draw the bounding box to RTSP channel.<br>Please see NN chapter for more details                                 |
| mmf2_video_example_md_nn_rtsp_init | CH1 Video -> H264/H265-> RTSP<br>CH4 Video -> RGB -> MD -> NN | RTSP video stream over the network.<br>MD module detect motion. If there is motion detected, it will trigger NN module to detect object and draw the bounding box to RTSP channel. |

- Audio + NN examples:

| Example                               | Description | Result                                                                                                                                |
|---------------------------------------|-------------|---------------------------------------------------------------------------------------------------------------------------------------|
| mmf2_video_example_audio_vipnn_init.c | AUDIO -> NN | The sound received by AmebaPro2 can be transmitted to NN engine to do sound classification.<br>Please see NN chapter for more details |

### 6.3.1.3 Execution and testing

Before executing example, it is necessary to set up console tool first (Tera Term, MobaXterm or PuTTY.....) and configure serial port baud to 115200. Once the setting is completed, AmebaPro2 is also connected with the PC and booted to get the Log message output of AmebaPro2.

- For examples with rtsp stream, we must first set up AmebaPro2 to connect with the network. Use AT command below to do the connect with an AP device:

```
ATW0=<Name of WIFI SSID> => Set the WiFi AP SSID to be connected
ATW1=<Password> => Set the WiFi AP password, if needed
ATWC => Initiate the connection
```

- 6.3.1.4 When the “RTSP stream enabled” message shown on console, it indicates that the RTSP server is already running. You can use VLC player to check the rtsp stream. For rtsp usage can refer to 6.3.3.

## 6.3.2 MMF AT command

MMF video examples provide commands for user to refer the audio reset and de-initialize the MMF and linker modules

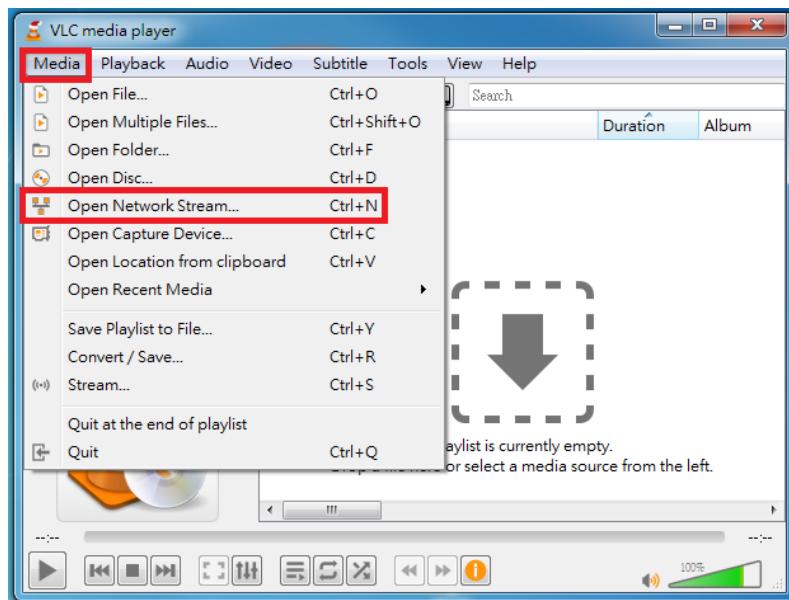
- UC=TD: use for de-initialize the whole flow of corresponding examples
- UC=TSR: reset the whole system

## 6.3.3 VLC media player settings

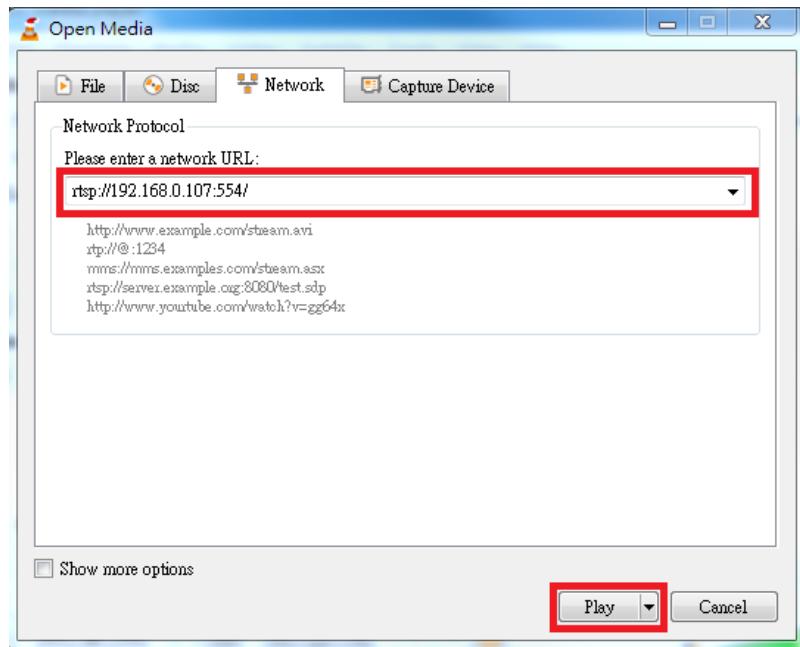
For RTSP examples, you can use VLC media player to receive or transmit the stream. Download VLC media player from website <https://www.videolan.org/>.

### 6.3.3.1 Stream audio/video from AmebaPro2 to VLC player

- Click “Media” -> “Open Network Stream”.

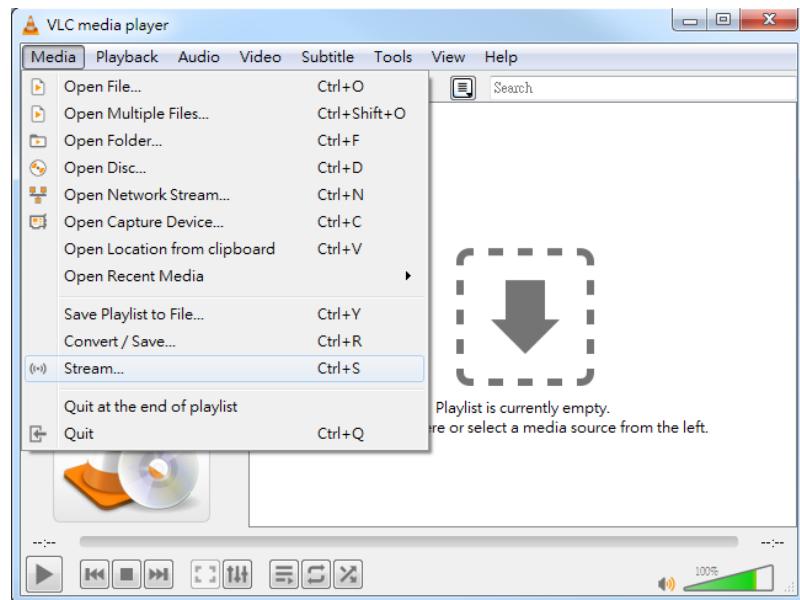


- Enter "rtsp://xxx.xxx.xxx.xxx:yyy/", where xxx.xxx.xxx.xxx is the Ameba IP address and yyy is the RTSP server port (default is 554), and click "Play".

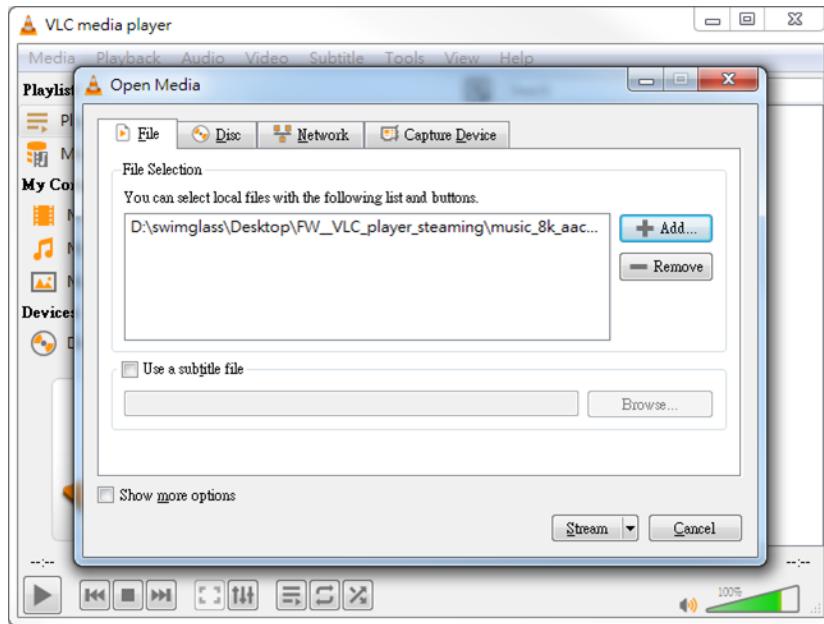


#### 6.3.3.2 Stream audio from VLC player to AmebaPro2

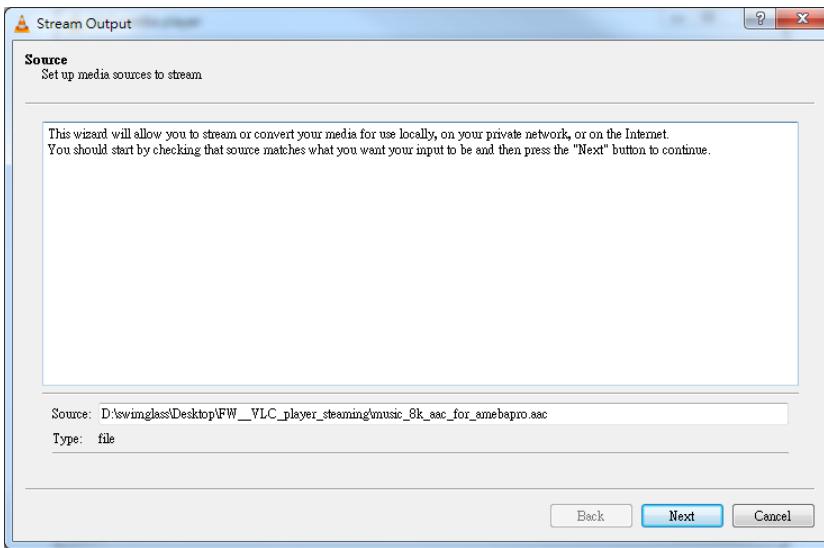
- Click "Media" -> "Stream".



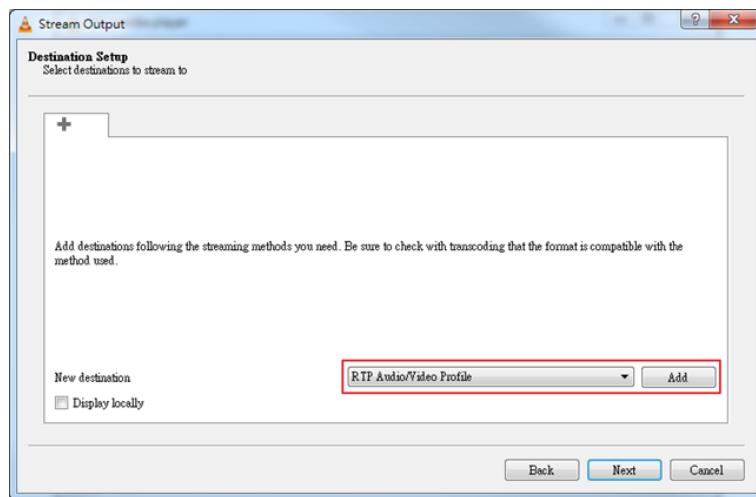
- Select "File", choose the file by "Add" and finally click the "Stream". (If the startup example is RTP -> AAD -> AUDIO please select the audio file with the file name .aac (The file format must be the same as the AAC decoder setting, the default is mono, sampling rate = 8k Hz). If the startup example is RTP -> G711D -> AUDIO, please select the audio file with the file extension .wav). If the startup example is RTP -> OPUSD -> AUDIO, please select the audio file with the file name .opus)



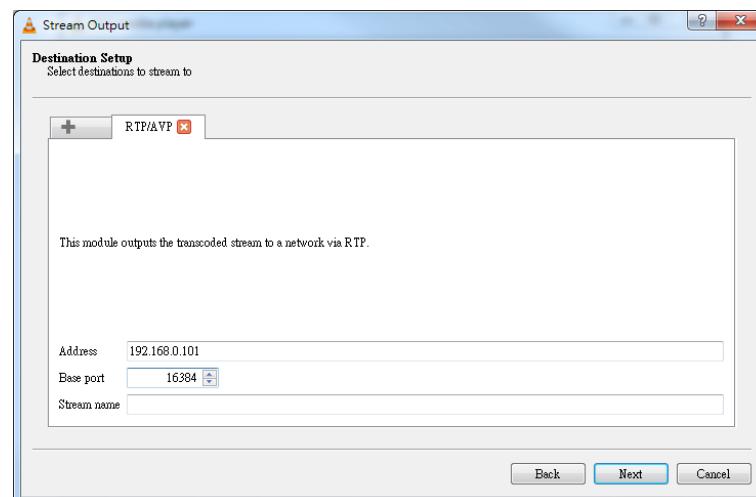
- You will see your select file after push "Stream". Check it and click "Next".



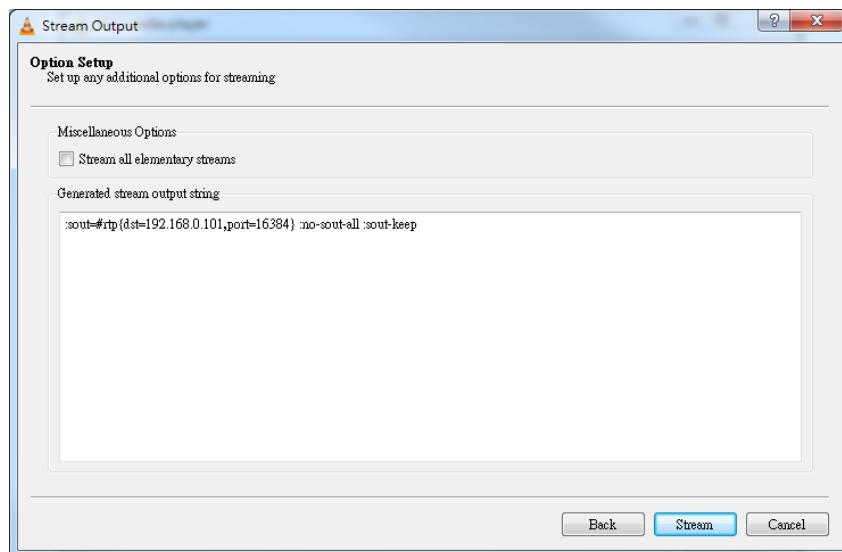
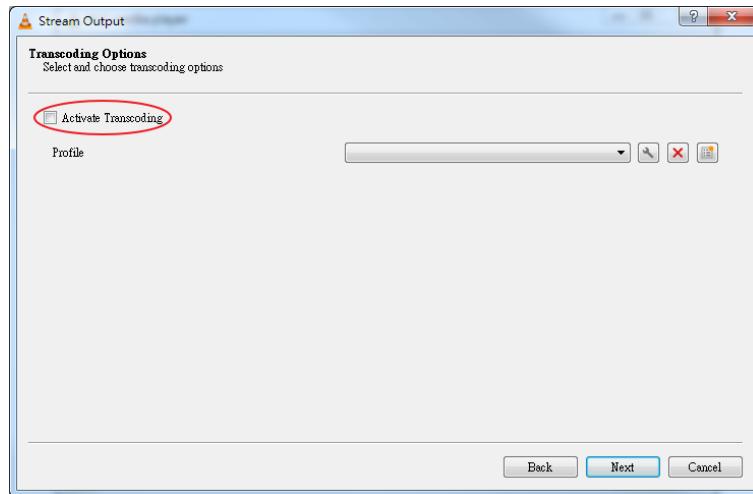
- Select "RTP Audio/Video Profile", and click "Add".



- Enter AmebaPro's IP Address in "Address" field, with "Base port" set to 16384, and click "Next".

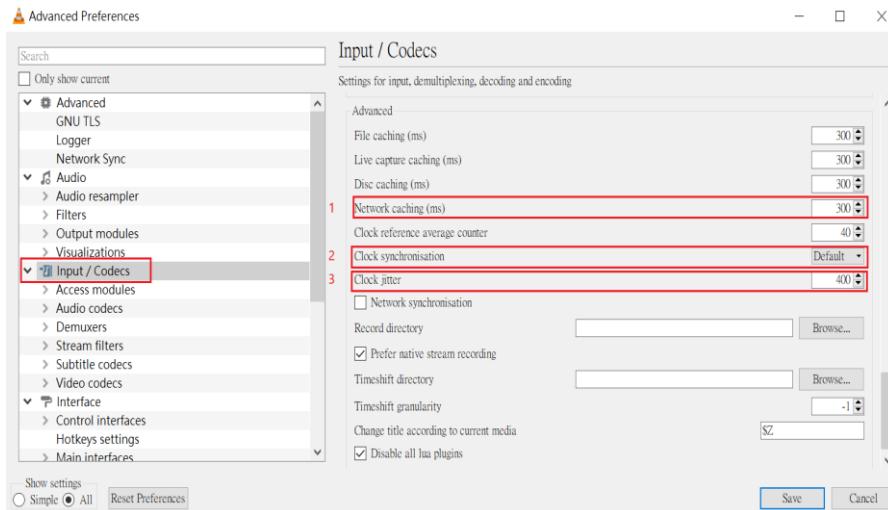


- Confirm “Activate Transcoding” is unchecked, and click “Next” -> “Stream”. Then the sound can be heard on AmebaPro2 3.5mm audio jack.

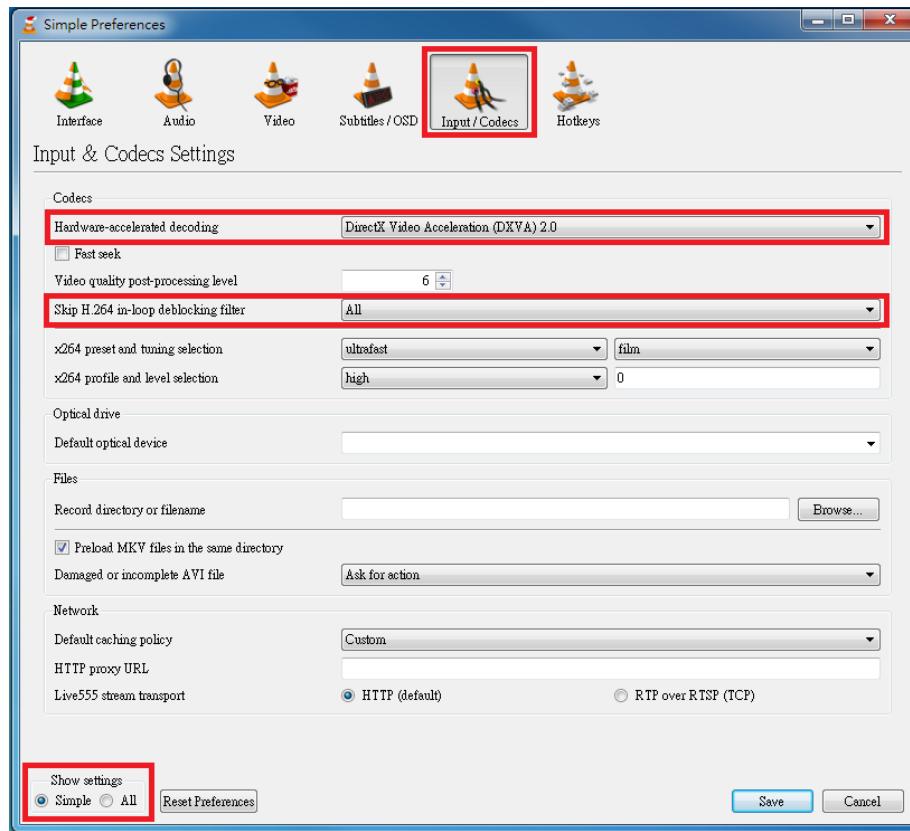


#### 6.3.3.2.1 Adjust latency (buffer) related settings

- Click “Tools” -> “Preferences” -> “Show settings: All” (lower left corner) -> “Input/ Codecs”, (1) set “Network caching” to 300ms (recommended), (2)set “Clock synchronisation” to Default, (3) set “Clock jitter” to 400ms (recommended).



- Click “Tools” -> “Preferences” -> “Show settings: Simple” (lower left corner) -> “Input/ Codecs”. Enable “Hardware-accelerated decoding” if available, and set “Skip H.264 in-loop deblocking filter” to “All”.



- VLC have a pts\_delay buffer by "network buffer" and "clock jitter". The maximum value of this buffer is equal to "network buffer" plus "clock jitter". The video display on the VLC side will delay due to the increase of pts\_delay buffer. By reducing the "network cache" and "clock jitter" can achieve the effect of shortening the delay.

### 6.3.4 Echo Cancellation

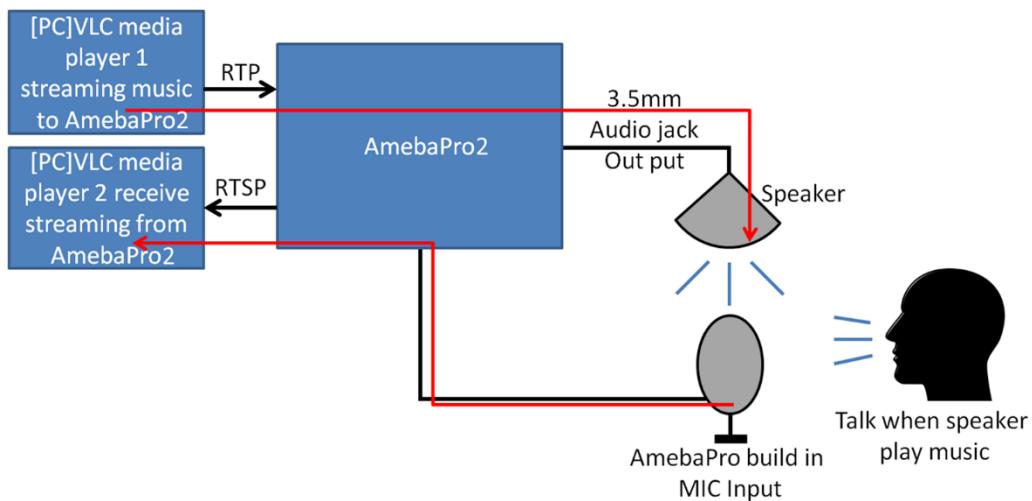
Echo cancellation is default provided in the audio part of MMFv2. To test whether the echo cancellation function is correct, use VLC media player to verify it on the computer.

## Usage Note (Refer to Audio optimization chapter):

- Sample rate must be 8K Hz/16K Hz
- Frame size must be the multiplies of 10ms ( suggest to be 10ms or 20ms about 160 samples and 320 samples)
- Two input signals must keep unchanged during AEC\_process.
- Time for executing AEC\_process must be under 10ms or 20ms (up on the frame size).
- Please check microphone and speaker signal and make sure there is no clipping signal.

The verification method is as follows:

- Use VLC media player on the PC to stream voice signal to AmebaPro2.
- Put AmebaPro2 speaker next to AmebaPro2 built-in Mic and speak at the same time.
- Then pass the received sound to the VLC media player on the PC via AmebaPro2 to see if the sound in step 1 is small enough or even disappear.



## 7 Memory Layout

### 7.1 Programming Space

| Start Address | Size   | Cache Support | IP Function |
|---------------|--------|---------------|-------------|
| 0x0000_0000   | 32 KB  | -             | ITCM ROM    |
| 0x0001_0000   | 128 KB | -             | ITCM SRAM   |

### 7.2 Data Space

| Start Address | Size  | Cache Support | IP Function |
|---------------|-------|---------------|-------------|
| 0x2000_0000   | 16 KB | -             | DTCM ROM    |
| 0x2001_0000   | 80 KB | -             | ITCM SRAM   |

### 7.3 Extension Memory Space

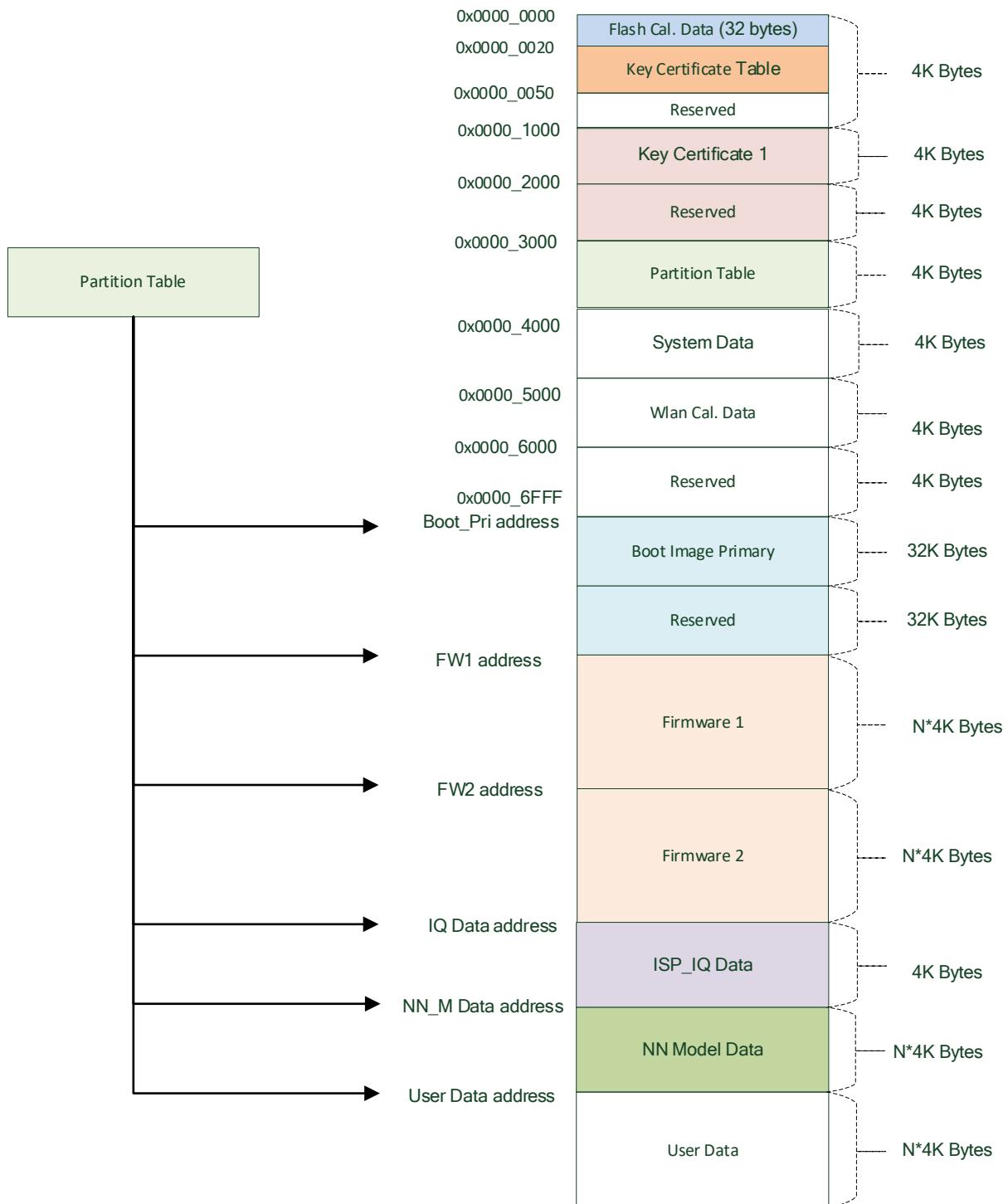
| Name  | Physical Address        | Size   | IP Function           |
|-------|-------------------------|--------|-----------------------|
| Flash | 0x0800_0000~0x0FFF_FFFF | 128 MB | External flash memory |
| DRAM  | 0x0700_0000~0x7FFF_FFFF | 128 MB | Extended DRAM memory  |

 NOTE

- The external flash memory address base is from 0x0800\_0000 to 0x0FFF\_FFFF
- The extended DRAM address base is from 0x0700\_0000 to 0x7FFF\_FFFF
- Both of flash and DRAM access are cacheable design

## 8 Flash Layout

### 8.1 NOR Flash Layout overview



#### 8.1.1 Blocks used in SDK

- File System

Place in the NOR flash:0xE00000

Content: The littlefs or the fatfs file system.

- System Data for BT and Wi-Fi SSID

Content: The following picture shows the flash layout of system data, signature and version are used to record the status of system data. The BT parameter size is 4bytes. The actual wifi-ssid(wifi fast connection data) is placed in 0xf00020-0x1000000.

Place in the NOR flash:

|                            | <b>0x00</b>               | <b>0x04</b> | <b>0x08</b> | <b>0x0C</b> |
|----------------------------|---------------------------|-------------|-------------|-------------|
| 0xf00000                   | Signature(SYSd)           | Version     | RSVD        | RSVD        |
| 0xf00010                   | BT Para Data              | RSVD        | RSVD        | RSVD        |
| 0xf00020<br>~<br>0x1000000 | Wifi fast connection data |             |             |             |

- Secure Storage

Content: The secure\_storage example stores/loads an AES GCM encrypted user data in flash. The plaintext of user data is like the following structure. The structure can be defined based on the requirement of user data. For the detail of secure\_storage example, please refer to Section “2.2 Secure Storage” of “UM0700 Realtek AmebaPro2 trustzone” document.

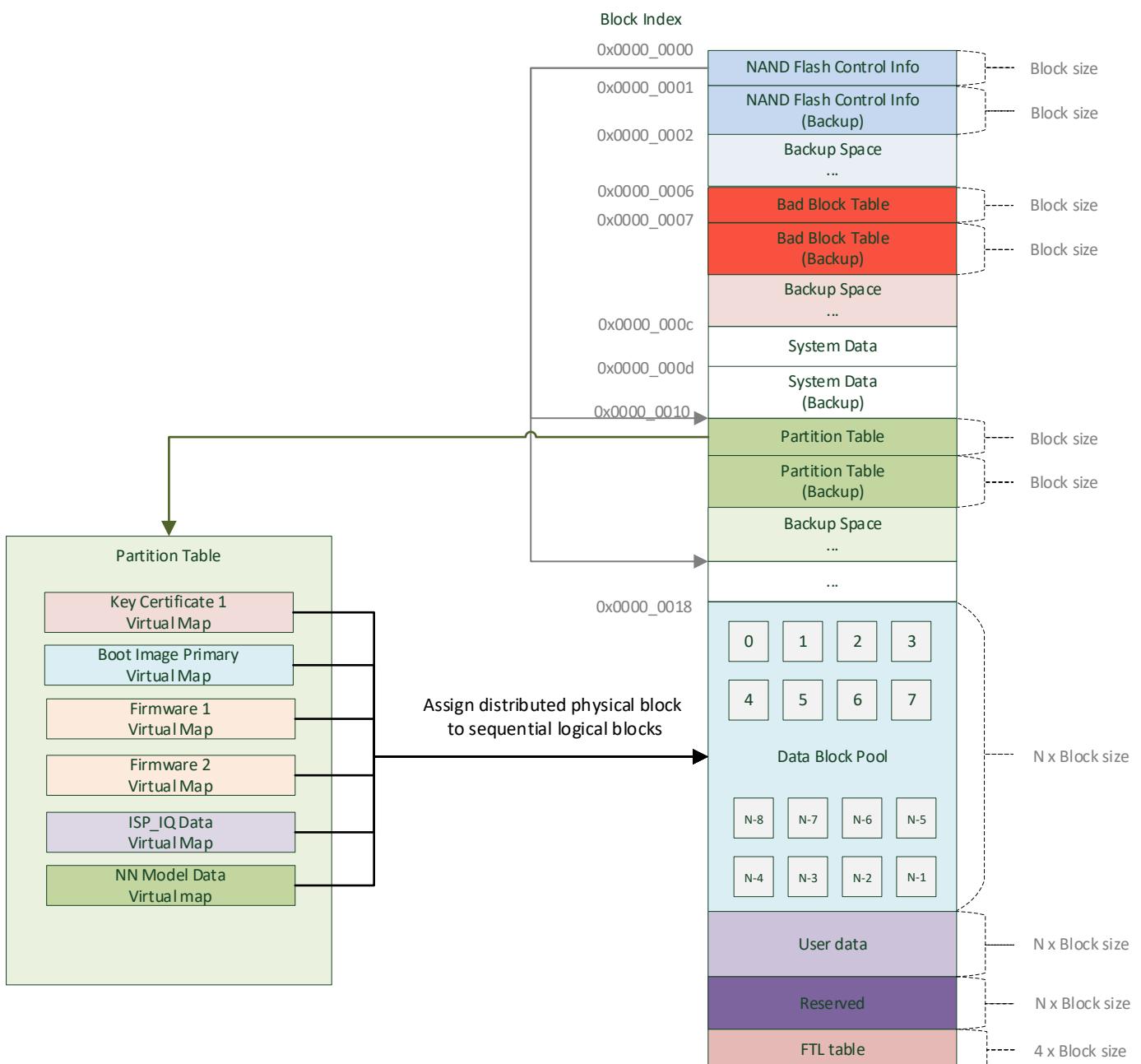
```
typedef struct user_data_s {
 char client_key[512];
} user_data_t;
```

Place in the NOR flash:

The flash address used by secure storage is defined in platform\_opts.h and can be modified based on requirement.

```
#define SECURE_STORAGE_BASS (0xF00000 + 0x4000) // 4KB
```

## 8.2 NAND Flash Layout overview



### 8.2.1 Blocks used in SDK

- File System

Content: The littlefs file system.

Place in the NAND flash: 0x4000000

- System Data for BT and Wi-Fi SSID

Content: The following picture shows the flash layout of system data, signature and version are used to record the status of system data. The BT parameter size is 4bytes. The actual wifi-ssid(wifi fast connection data) is placed in 0xf00020 – 0x1000000.

Place in the NAND flash:

|          | 0x00            | 0x04    | 0x08 | 0x0C |
|----------|-----------------|---------|------|------|
| 0xf00000 | Signature(SYSD) | Version | RSVD | RSVD |

|                             |                           |      |      |      |
|-----------------------------|---------------------------|------|------|------|
| 0xf00010                    | BT Para Data              | RSVD | RSVD | RSVD |
| 0xf00020<br>~<br>0x10000000 | Wifi fast connection data |      |      |      |

- Secure Storage

Content: The secure\_storage example stores/loads an AES GCM encrypted user data in flash. The plaintext of user data is like the following structure. The structure can be defined based on the requirement of user data. For the detail of secure\_storage example, please refer to Section "2.2 Secure Storage" of "UM0700 Realtek AmebaPro2 trustzone" document.

```
typedef struct user_data_s {
 char client_key[512];
} user_data_t;
```

Place in the NAND flash:

The flash address used by secure storage is defined in platform\_opts.h and can be modified based on requirement.

```
#define SECURE_STORAGE_BASS (0x7A00000 + 0x4000) // 4KB
```

## 8.3 NAND Flash

The NAND flash 1G-bit memory array is organized into 65,536 programmable pages of 2,048-bytes each. The entire page can be programmed at one time using the data from the 2,048-Byte internal buffer. Pages can be erased in groups of 64 (128KB block erase). The NAND flash has 1,024 erasable blocks.

1Gb SLC NAND Flash: 1G-bit / 128M-byte

On chip ECC for memory array

|                                                               |                 |
|---------------------------------------------------------------|-----------------|
| Page Data Buffer (2048 Byte)                                  | Spare(64 Byte)  |
| Block (64 Pages, 64*2048 Byte)                                | 64*64 Byte      |
| Total flash 1024 Blocks (65536 Pages, 1024 Blocks * 64 Pages) | 1024*64*64 Byte |

| Page Structure (2112 Byte) | Page Data Buffer (2048 Byte) ECC Protected |         |         |         | Spare(64 Byte) |        |        |        |
|----------------------------|--------------------------------------------|---------|---------|---------|----------------|--------|--------|--------|
|                            | Sector0                                    | Sector1 | Sector2 | Sector3 | Spare0         | Spare1 | Spare2 | Spare3 |

The first byte of spare0 is bad block marker.

### 8.3.1 NAND Flash mbed API

NAND Flash mbed API is used to access Flash physical location.

Please refer to snand\_api.h & snand\_api.c

```
/**
 * @brief Init Flash
 * @param obj: address of the flash object
 * @retval none
 */
void snand_init(snand_t *obj);

/**
 * @brief Erase flash block, usually 1 block = 64K bytes
 * Please refer to flash data sheet to confirm the actual block size.
 * The actual address which being erased always aligned with block size.
 * @param address: Specifies the starting address to be erased.
 */
```

```
* @retval SUCCESS, FAIL
*/
int snand_erase_block(snand_t *obj, uint32_t address);

/**
 * @brief Read a stream of data from specified address via user mode
 * @param obj: Specifies the parameter of flash object.
 * @param address: Specifies the address to be read.
 * @param len: Specifies the length of the data to read.
 * @param data: Specified the address to save the readback data.
 * @retval SUCCESS, FAIL
*/
int snand_page_read(snand_t *obj, uint32_t address, uint32_t Length, uint8_t *data);

/**
 * @brief Write a stream of data to specified address
 * @param obj: Specifies the parameter of flash object.
 * @param address: Specifies the address to be programmed.
 * @param Length: Specifies the length of the data to write.
 * @param data: Specified the pointer of the data to be written.
 * If the address is in the flash, full address is required, i.e. SPI_SNAND_BASE +
Offset
 * @retval SUCCESS, FAIL
*/
int snand_page_write(snand_t *obj, uint32_t address, uint32_t Length, uint8_t *data);
```

### 8.3.1.1 NAND flash mbed example

This example demonstrates how use mbed API to scan bad block and read/write a NAND flash.

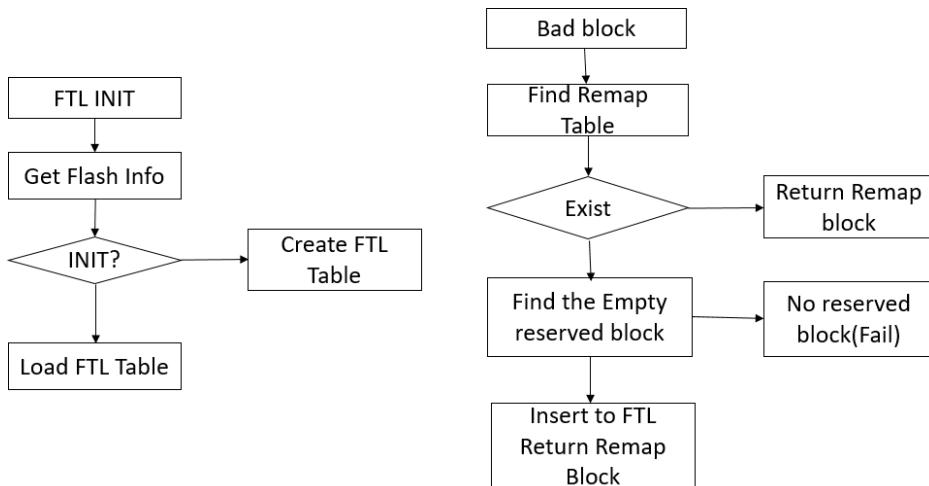
The example is located in:

“\project\realtek\_amebapro2\_v0\_example\example\_sources\nand\_flash\”

Copy main.c to src folder, compile project, and the download the binary.

### 8.3.2 NAND FTL

The flash translation layer (FTL) performs logical-to-physical address. It is block-mapping method; you do not need to deal with the bad block operation. It reserved 5% reserved blocks to replace the bad block. The detail procedure is as below.

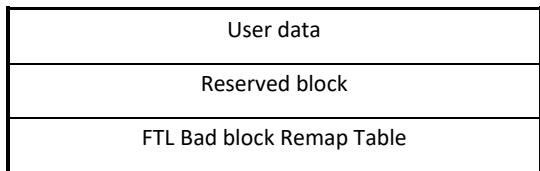


FTL table

| Start Tag |      | Bad block Number |      | Remap block Number |      | End Tag |      |
|-----------|------|------------------|------|--------------------|------|---------|------|
| BB        |      | Number           |      | Number             |      | bb      |      |
| 0xFF      | 0xFF | 0xFF             | 0xFF | 0xFF               | 0xFF | 0xFF    | 0xFF |
| 0xFF      | 0xFF | 0xFF             | 0xFF | 0xFF               | 0xFF | 0xFF    | 0xFF |

Bad block is 800 and the remap block is 950 for the example

| Byte0 | Byte1 | Byte2 | Byte3 | Byte4 | Byte5 | Byte6 | Byte7 |
|-------|-------|-------|-------|-------|-------|-------|-------|
| BB    |       | 800   |       | 950   |       | bb    |       |
| 0xFF  |
| 0xFF  |



### 8.3.2.1 FTL API

- `ftl_common_read` – Read the data from flash.
- `ftl_common_write` – Write the data into flash
- `ftl_common_erase` – Erase the sector or block from flash.
- `ftl_erase_sector` – Erase the 4k sector from flash.
- `ftl_common_info` – Get the flash type, page size, block size and block count from flash.

### 8.3.2.2 FTL Example

We can use the ATCMD and Littlefs as the example.

About the atcmd, please modify the platform\_opts.h to enable the ATCMD example. We provide the below command to r/w the flash, it is located at the atcmd\_ftl.c.

```
#define CONFIG_FTL 0 //support FTL AT command
```

- AFWD – Write the data into flash.
- AFRD – Read the data from flash.
- AFTR – Select the block and page to read the Nand flash data.
- AFTR – Select the sector to read the Nor flash data.
- AFTW – Select the block and page to write the data into nand flash.
- AFTW – Select the sector to write the data into nor flash.
- AFTE – Select the block to erase the Nand flash.
- AFTE – Select the sector to erase the nor flash.

About the file system, please select the littlefs to execute the example.

```
$ cmake .. -G"Unix Makefiles" -DCMAKE_TOOLCHAIN_FILE=../toolchain.cmake -DEXAMPLE=std_file
```

## 9 OTA

Over-the-air programming (OTA) provides a methodology to update device firmware remotely via TCP/IP network connection.

### 9.1 OTA Operation Flow for NOR Flash

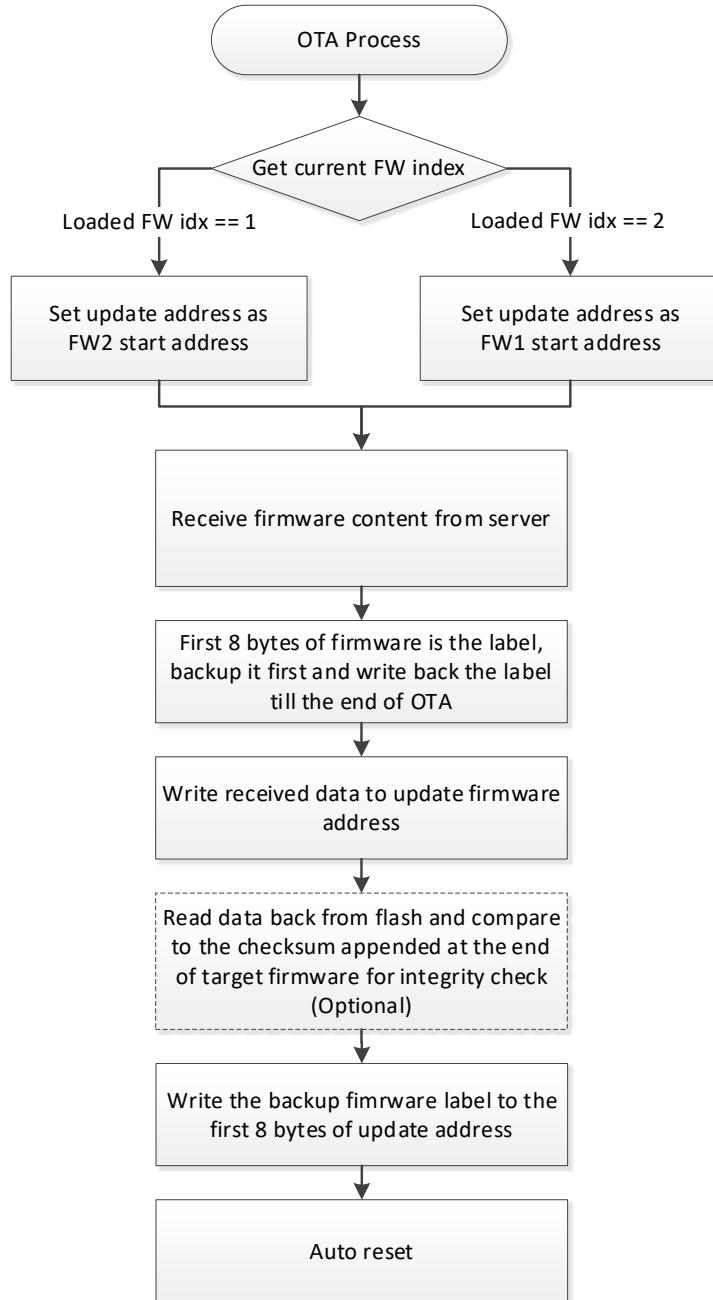


Figure 9-1 OTA Process Flow for NOR Flash

The first 8 bytes of firmware image would be a label. During the step of “Write received data to update firmware address”, the 8 bytes label need set to 0xffffffffffff. That means the label is invalid. The backup label needs to be written back at the end of OTA process to prevent device booting from incomplete firmware.

## 9.2 OTA Operation Flow for NAND Flash

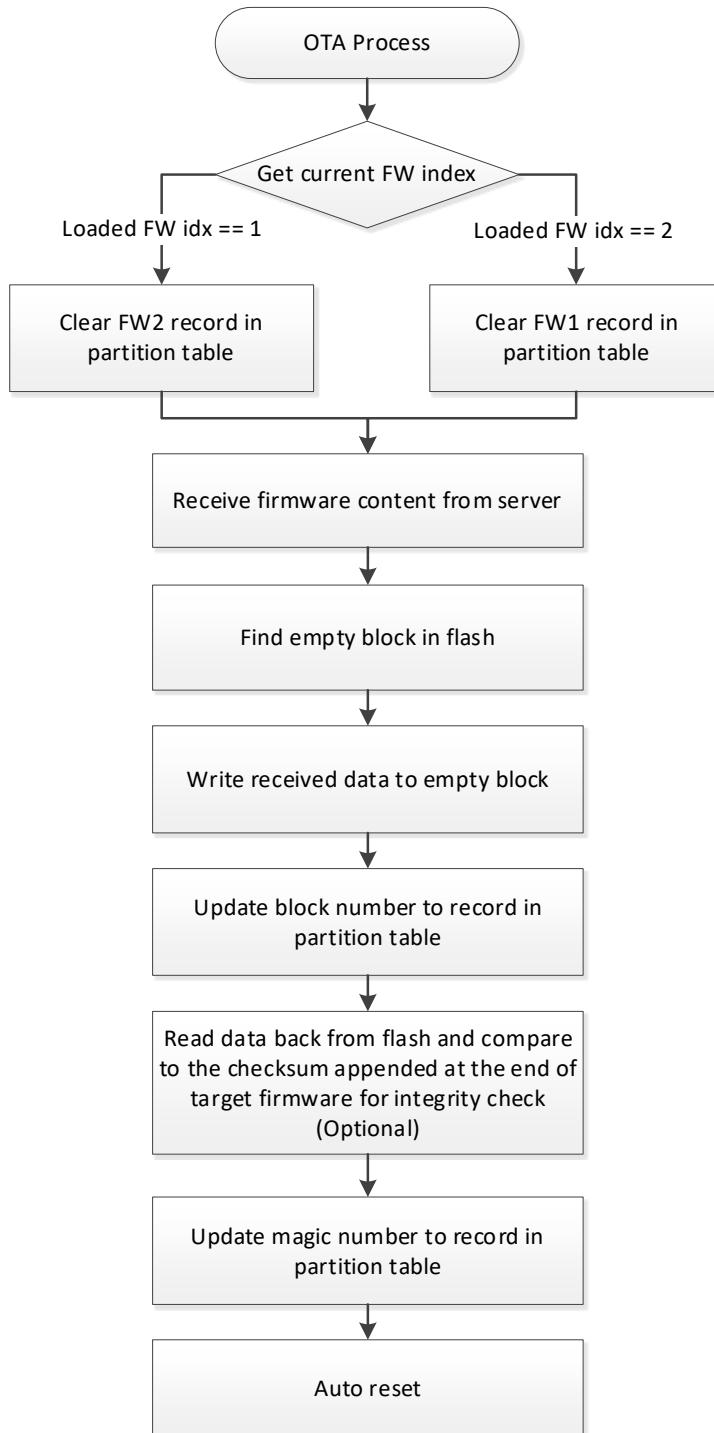


Figure 9-2 OTA Process Flow for NAND Flash

During the step of “Write received data to empty block”, the magic number of this new record in partition table is invalid. The magic number is updated at the end of OTA process to make this record valid.

## 9.3 OTA Checksum Mechanism

The first 8 bytes firmware label for NOR flash OTA process and the magic number of record in partition table for NAND flash OTA process are used to notice the bootloader the overall OTA process is done without any network disconnection or re-boot during the OTA. However, firmware label or magic number cannot guarantee the content of firmware image is correct.

User can design a mechanism to calculate the hash of target OTA firmware for integrity check during the OTA update process. For the default OTA example in SDK, there is USE\_CHECKSUM option for this integrity check purpose. During image build, SDK would append 4 bytes checksum at the end of firmware image to become OTA image (ota.bin). When performing OTA routine, right after the firmware is

downloaded and programmed into flash, it would read back all the programmed data from flash and compare with the checksum value from target firmware if USE\_CHECKSUM enabled. In such way, it can ensure the downloaded firmware is transferred completely and correct. For the detail implementation, please refer to OTA example ota\_8735b.c in SDK:

```
#define USE_CHECKSUM 1
```

## 9.4 Boot Process Flow

Boot loader will select latest firmware based on firmware version and timestamp, and load it.

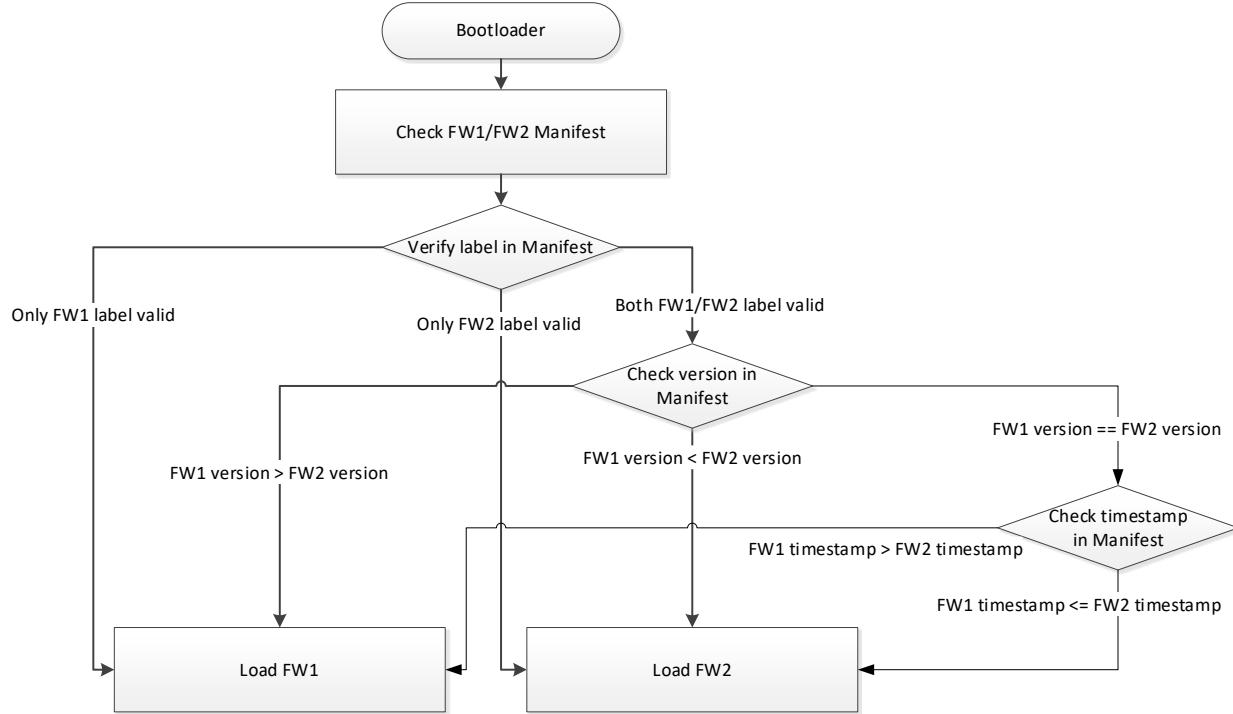


Figure 9-3 Boot Process Flow

## 9.5 Upgraded Partition

In AmebaPro2 OTA update procedure, Firmware 1 and Firmware 2 are swapped to each other.

The Firmware 1/Firmware 2 partition addresses and length are stored in partition records, defined in ‘amebapro2\_partitiontable.json’ under ‘project\realtek\_amebapro2\_v0\_example\GCC-RELEASE\mp’. Please adjust it according to your firmware size.

```
"fw1": {
 "start_addr" : "0x100000",
 "length" : "0x300000",
 "type": "PT_FW1",
 "valid": true
},
"fw2": {
 "start_addr" : "0x400000",
 "length" : "0x300000",
 "type": "PT_FW2",
 "valid": true
},
```

For NOR flash, OTA firmware is written to the partition start address in flash, and OTA firmware size is checked with the partition length. For NAND flash, OTA firmware is written to empty blocks distributed in flash, and OTA firmware size is checked with the partition length.

## 9.6 Firmware Image Output

After building project source files in SDK, it would generate firmware as ‘firmware.bin’, and OTA firmware as ‘ota.bin’ which is firmware.bin with 4 bytes checksum appended at the end.

### 9.6.1 OTA Firmware Swap Behavior

When device executes OTA procedure, it would update another firmware partition, rather than the current running firmware partition. The OTA firmware swap behavior should be looked like as below figure if the updated firmware keeps using newer firmware version and timestamp.

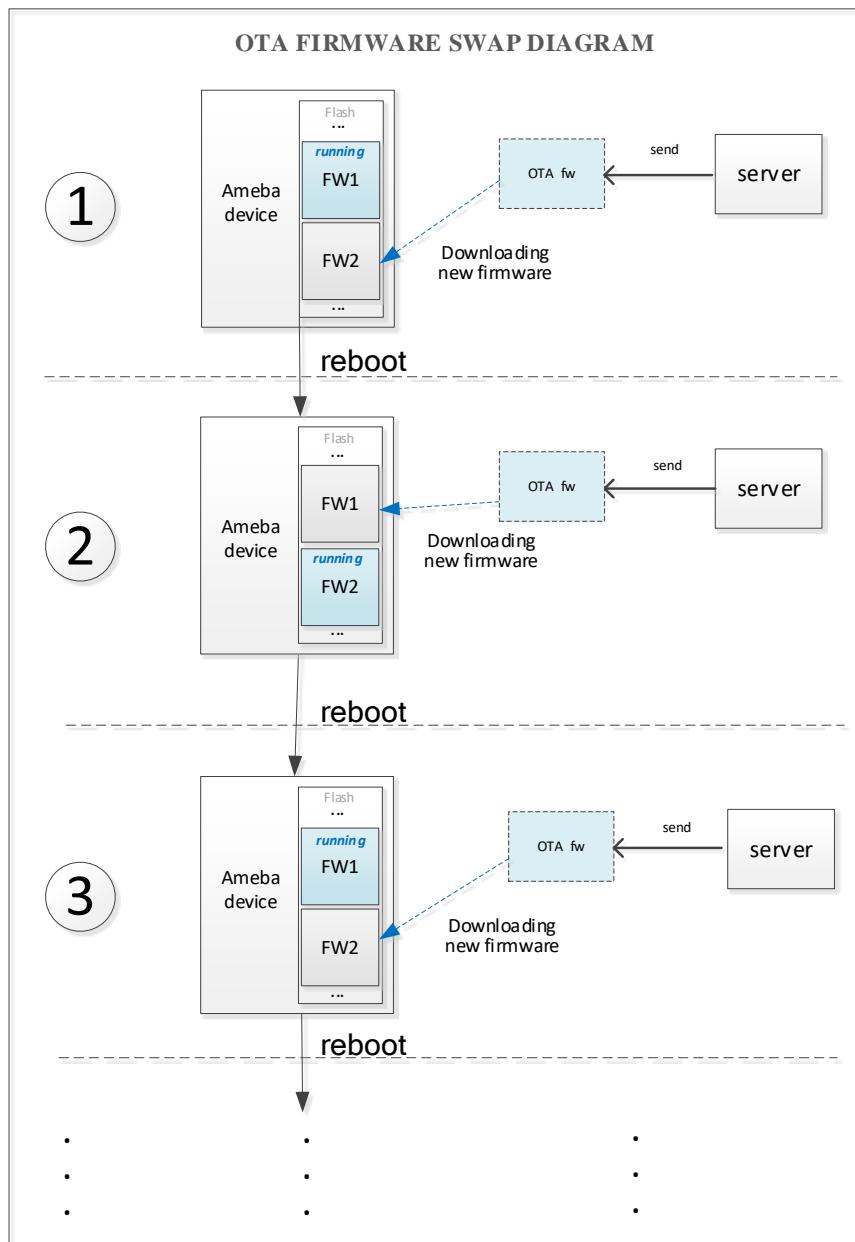


Figure 9-4 OTA Firmware SWAP Procedure

### 9.6.2 Version and Timestamp

AmebaPro2 bootloader boots to Firmware 1 or Firmware 2 based on firmware version and timestamp. Please check the version and timestamp of generated OTA firmware are expected.

In firmware image, the version is a 32bytes value in little endian order. The version can be configured in 'amebapro2\_firmware\_ntz.json' under 'project\realtek\_amebapro2\_v0\_example\GCC-RELEASE\mp'.

```

"MANIFEST": {
 "label": "RTL8735B",
 "vrf_alg": "NA_VRF_CHECK",
 "tlv": [
 ...
 {"type": "TYPE_ID", "length": 2, "value": "IMG_FWHS_S"},

 {"type": "VERSION",

 "length": 32,

```

```
"value": "FFFFFFFFFFFFFFF...FFFFFFFFFFF" },
 {"type": "TIMST", "length": 8, "value": "auto"},
```

The version which higher bit is zero presents higher version. For example, the version of 'FFFFFFFFFFFFFFF...FFFFFFFFFFF' is zero in bit 0, version of 'EFFFFFF...FFFFFFFFFFF' is zero in bit 4, and version of 'FFF...FFFFFFFFFFF' is zero in bit 8. Then, it will be version with bit 8 zero > version with bit 4 zero > version with bit 0 zero.

The timestamp is an 8bytes value in little endian order. The timestamp which presents image build time will be automatically generated when image build.

## 9.7 Implement OTA over Wi-Fi

### 9.7.1 OTA Using Local Download Server Base on Socket

The example shows how device updates image from a local download server. The local download server sends image to device based on network socket.

**NOTE**

Make sure both device and PC are connecting to the same local network.

#### 9.7.1.1 Build OTA Application Image

Enable CONFIG\_OTA\_UPDATE flag in 'project\realtek\_amebapro2\_v0\_example\inc\platform\_opts.h' to support ATWO AT command for OTA with local download server.

```
#define CONFIG_OTA_UPDATE 1
```

Download the firmware to AmebaPro2 board to execute OTA.

#### 9.7.1.2 Setup Local Download Server

Step 1: Build new ota.bin and place it to 'tools\DownloadServer' folder.

Step 2: Edit 'tools\DownloadServer\start.bat' file for server port and OTA file name

```
@echo off
DownloadServer 8082 ota.bin
set /p DUMMY=Press Enter to Continue ...
```

Step 3: Execute 'tools\DownloadServer\start.bat'.



Figure 9-5 Download Server

#### 9.7.1.3 Execute OTA Procedure

After device connects to AP, enter command: ATWO=IP[PORT]. Please note that the device and your PC need under the same AP. The IP in ATWO command is the IP of your PC.

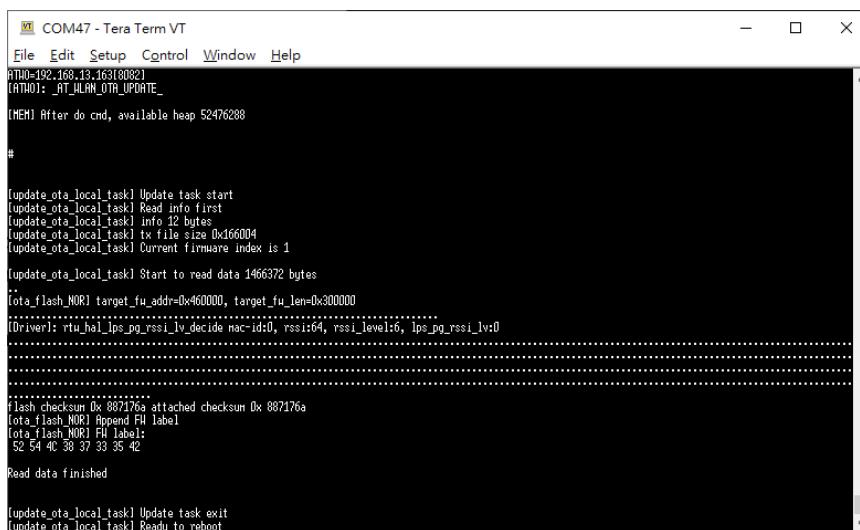


Figure 9-6 ATWO command

After finishing OTA download, device will reboot automatically, and the bootloader will boot to new firmware according to firmware version and timestamp.

## 9.7.2 OTA Using Local Download Server Based on HTTP

This example shows how device updates image from a local http download server. The local http download server will send the http response which data part is 'ota.bin' after receiving the http request.

### NOTE

Make sure both device and PC are connecting to the same local network.

#### 9.7.2.1 Build OTA Application Image

Set server IP, port, and resource in ota\_http example (component\example\ota\_http\example\_ota\_http.c).

```
#define PORT 8082
#define HOST "192.168.1.100"
#define RESOURCE "ota.bin"
```

Build firmware with ota\_http example.

```
cmake .. -G"Unix Makefiles" -DCMAKE_TOOLCHAIN_FILE=../toolchain.cmake
-DEXAMPLE=ota_http
```

Download the firmware to AmebaPro2 board to execute OTA.

#### 9.7.2.2 Setup Local HTTP Download Server

Step 1: Build new ota.bin and place it to 'tools\DownloadServer(HTTP)' folder.

Step 2: Edit 'tools\DownloadServer\start.bat' file for server port and OTA file name

```
@echo off
DownloadServer 8082 ota.bin
set /p DUMMY=Press Enter to Continue ...
```

Step 3: Execute 'tools\DownloadServer(HTTP)\start.bat'.

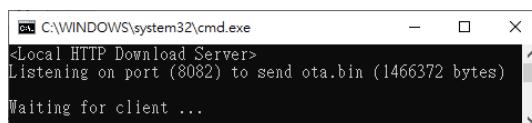


Figure 9-7 HTTP Download Server

#### 9.7.2.3 Execute OTA Procedure

Reboot the device and connect to AP, it should execute ota\_http example automatically to start the OTA update through HTTP protocol.

After finishing OTA download, device will reboot automatically, and the bootloader will boot to new firmware according to firmware version and timestamp.

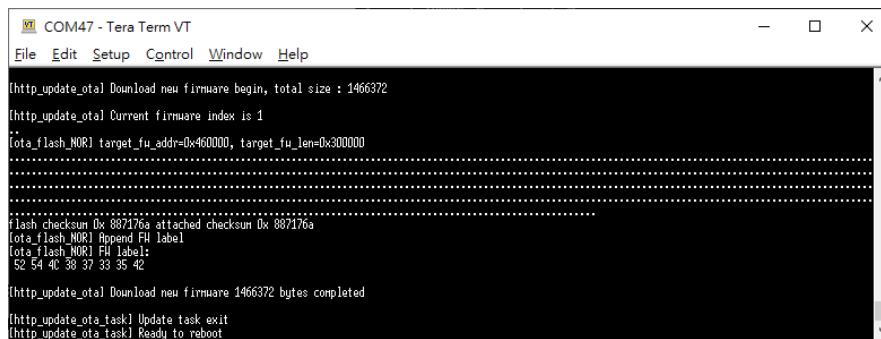


Figure 9-8 OTA HTTP Example

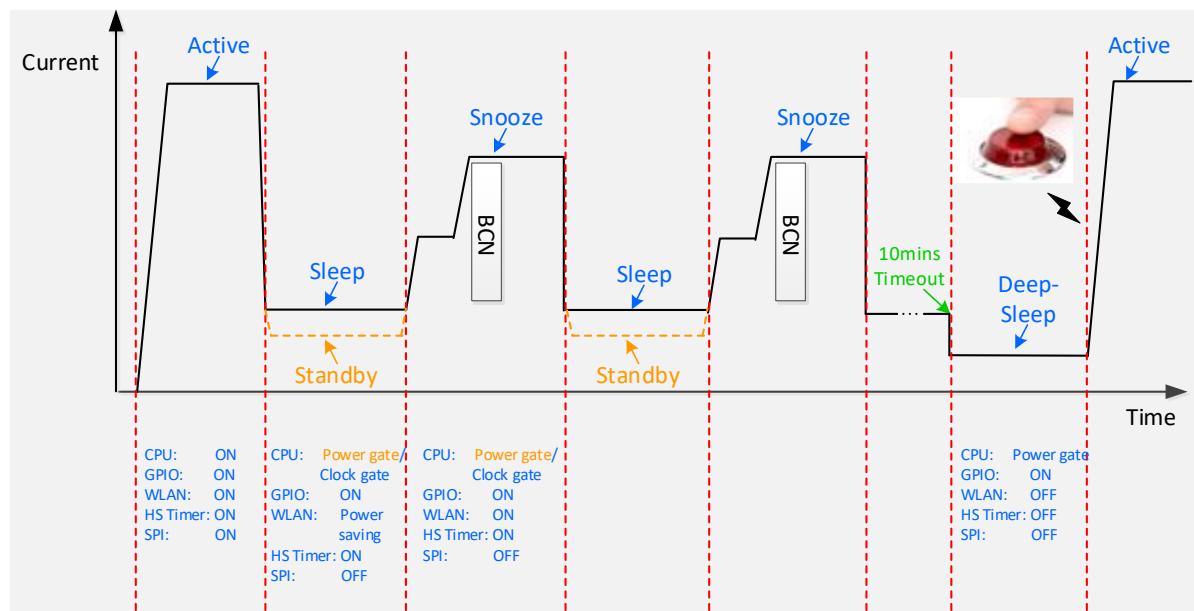
# 10 Power Save

## 10.1 Overview

### 10.1.1 Application Scenario

AmebaPro2 achieves low power consumption with a combination of several proprietary technologies. The power-saving architecture features six reduced power modes of operation: active, sleep, standby, snooze, deepsleep, shutdown mode. With the elaborate architecture, the battery life of whole IOT system could be extended.

For reading pen application, it can divide into three-scenario. First, press the power button to power on reading pen to active mode and then connect to the cloud to download data. Second, once the reading pen without any activity for 2 minutes, the system will go to sleep mode (for system fast resume and keep WIFI connect) or standby mode (for lower power consumption and keep WIFI connect) and regularly wake up to receive WLAN beacon while into snooze mode. At last, without using the reading pen exceeds 10 minutes, the system will into deep sleep mode and waiting for any button signals to wake up system into active mode. The application scenario flow was shown in

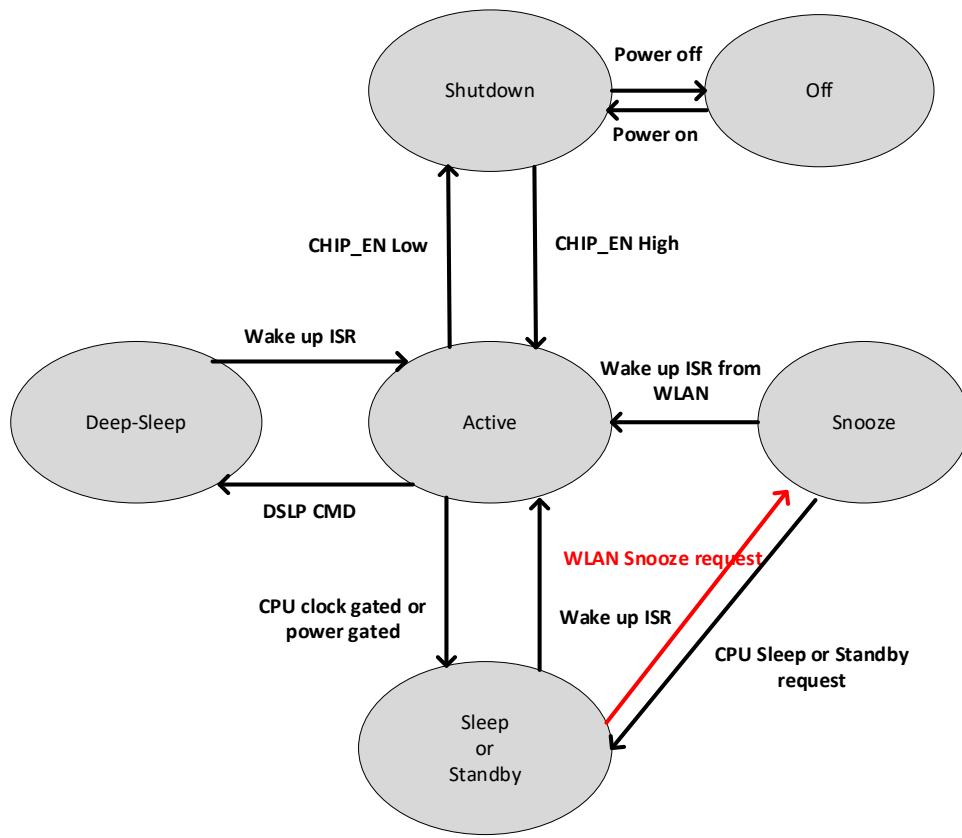


### 10.1.2 Features

- Active Mode:** The CPU is in active mode and all peripherals are available.
- Sleep Mode:** The CPU is in clock-gated and can be woken up by most of peripherals. The system resume time could be much faster than the Standby mode and the WLAN could be ON or power saving mode in this state.
- Standby Mode:** The CPU is in power-gated and can be woken up by most of peripherals. The power consumption could be lower than the Sleep mode and the WLAN could be ON or power saving mode in this state.
- Snooze Mode:** In this mode system can regularly wake up to receive WLAN beacon without software intervention. The significant difference between Snooze and Sleep/Standby mode is WLAN capability and could be receive and transmit beacon in this state.
- DeepSleep Mode:** The lowest power consumption than the other power mode except for shutdown mode, it can be only woken up by LP Timer or GPIO.
- Shutdown Mode:** The CPU will be shutdown while CHIP\_EN was Low.

### 10.1.3 Power Mode and Power Consumption

The mode transition diagram is given in.



In Figure, the power mode can be divided into 6 states except for “off” state and the each power In Figure, the power mode can be divided into 6 states except for “off” state and the each power consumption was shown in Figure. The introduction of each power mode, clock-gated and power-gated state will be in the following sections. Clock/power gated state could be regarded as a status of any hardware.

## 10.2 Deep Sleep Mode

- CHIP\_EN keeps high. User can invoke Deep Sleep API to force into deep sleep mode. By using specified interrupts to wake up system.
- The following wake flow: Wake up ISR is high -> PMC -> enable CPU -> Reboot flow.

### 10.2.1 Wakeup Source

Aon GPIO, RTC, comparator, Aon Timer  
Aon GPIO: GPIOA0~GPIOA3  
Comparator: GPIOA0~GPIOA3

## 10.3 Standby Mode

- CHIP\_EN keeps high. User can invoke Standby API to force into deep sleep mode. By using specified interrupts to wake up system.
- The following wake flow: Wake up ISR is high -> PMC -> enable CPU -> Fast reboot flow.

### 10.3.1 Wakeup Source

Aon GPIO, RTC, comparator, Aon Timer, Gtimer0, PWM, Pon GPIO, Uart0, Wlan  
Aon GPIO: GPIOA0~GPIOA3  
Pon GPIO: GPIOF0~GPIOF17  
Comparator: GPIOA0~GPIOA3

## 10.4 Sleep Mode

- CHIP\_EN keeps high. User can invoke Sleep API to force system into deep sleep mode. By using specified interrupts to wake up system.
- The following wake flow: Wake up ISR is high -> PMC -> enable CPU -> Execution of instructions continues.

### 10.4.1 Wakeup Source

Aon GPIO, RTC, comparator, Aon Timer, Gtimer0, PWM, Pon GPIO, Uart0, Wlan

Aon GPIO: GPIOA0~GPIOA3

Pon GPIO: GPIOF0~GPIOF17

Comparator: GPIOA0~GPIOA3

## 10.5 Snooze Mode

- CHIP\_EN keeps high. By using specified interrupts to wake up system.
- The following wake flow: WLAN power on request-> Receive particular beacon-> Wake up ISR is high -> PMC -> enable CPU -> Execution of instructions continues or fast reboot flow.

### 10.5.1 Wakeup Source

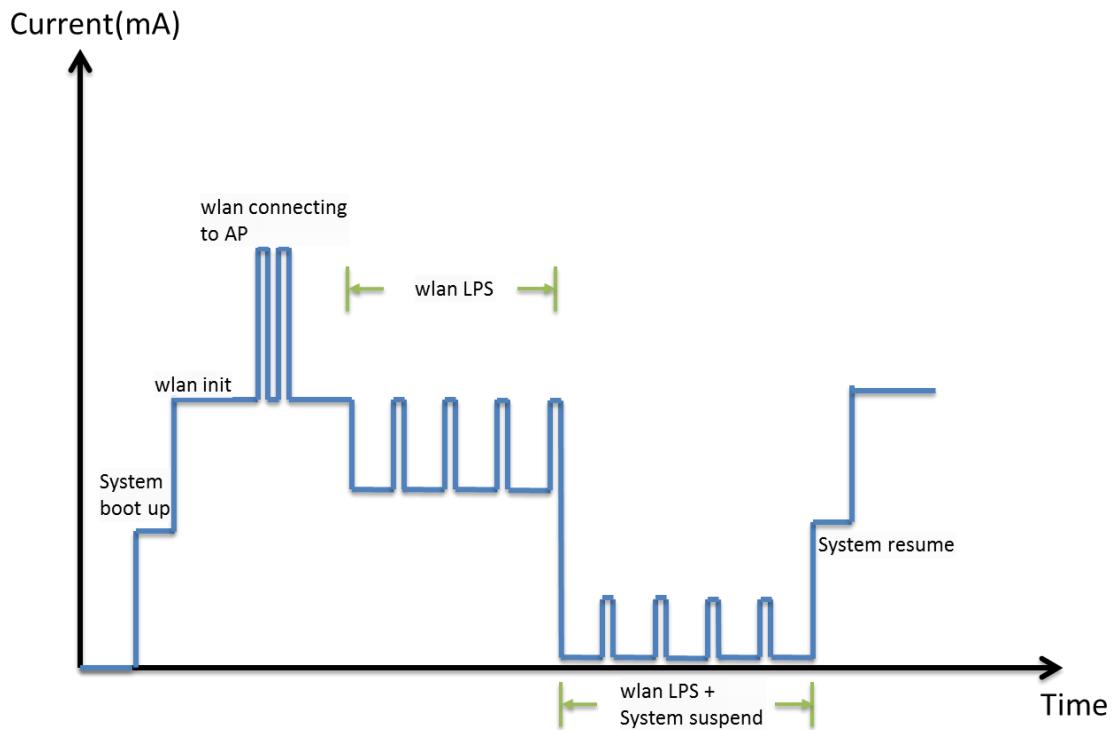
In snooze mode, the only wake up source is WLAN. The wakeup condition could be configured by WLAN driver according to system application. Once the event triggered, WLAN hardware would raise interrupt to PMC that could change hardware state.

### 10.5.2 Wakeup from WLAN

If user choose wakeup from WLAN in Standby mode, he needs to configure wlan before invoke Standby.

After user configure wlan and invoke Standby, system would enter suspend state. In this state, MCU would periodically check wlan state and decide if it needs to wakeup. After wlan receive the wakeup packet that matches the wakeup pattern, then wakeup.

The whole progress is like below diagram:



- At first system boot up, initialize wlan, connecting to AP.
- After connected to AP, wlan would enter LPS state if there is no heavy data traffic. In LPS state, wlan would listen beacon for every 100ms (if DTIM is 1). So you could see power consumption rise for every 100ms. Power consumption would drop after wlan receive the beacon and the TIM field has no packet for this device, then wlan would turn off RF and try to keep in low power state.
- If user try to make system save more power with wlan associate idle, he could invoke Standby. You could see the power consumption drops more in wlan LSP with system suspend.

Please note that if you want to measure the power consumption when system suspend with wlan LPS, you have to make sure the voltage regulator

of power supply and current meter could handle the voltage drop and rise between hundreds of micro amp and dozens of milliamp.

### 10.5.3 Keep-alive mechanism

Wowlan mode can perform TCP keep alive and MQTT ssl ping request keep alive by setting a fixed pattern.

#### 10.5.3.1 TCP keep alive

In first, set 2 parameters: IP & port of TCP server. The setting would become effective after system enter wake on wlan mode. By default it sends a TCP packet every 30s, and resend after 10s if STA does not receive TCP ACK. You can modify the interval in the setting.

In the offload setting, it also adds a wake on wlan pattern that matches TCP packet with same source port in tcp keep alive, and match TCP flag with PSH+ACK. These setting would allow TCP server wakeup STA by sending a packet to STA.

#### 10.5.3.2 MQTT SSL keep alive

Just like tcp keep alive, we only need to set the IP & port of MQTT server and SSL Key offload after TLS connection, and fill in {0xc0, 0x00} in the tcp payload to send ping request packets at a fixed time for MQTT keep alive .

### 10.5.4 Wakeup from pattern

When the user needs to use the remote wake-up function, the system can leave Standby mode when wlan receives a matching data packet. The SDK has been configured with the ICMP pattern to wake up, and supports user-set custom patterns.

The part of the Wakeup pattern comparison includes the data of Destination, BSSID, Source in the MAC Header, and Destination IP Address from the Protocol Type of the LLC Header to the Destination IP Address of the IP Header. To set a custom pattern, users need to set (1) Pattern content and (2) Mask: Pattern content to compare Byte, the above two items must be set in the wowlan\_pattern\_t structure:

```
typedef struct wowlan_pattern {
 unsigned char eth_da[6];
 unsigned char eth_sa[6];
 unsigned char eth_proto_type[2];
 unsigned char header_len[1];
 unsigned char ip_proto[1];
 unsigned char ip_sa[4];
 unsigned char ip_da[4];
 unsigned char src_port[2];
 unsigned char dest_port[2];
 unsigned char flag2[1];
 unsigned char mask[6];

 unsigned char window[2]; //Reserved
 unsigned char checksum[2]; //Reserved
 unsigned char urgent_pointer[2]; //Reserved
 unsigned char payload[64];
 unsigned char payload_mask[9];
} wowlan_pattern_t;
```

After setting the pattern and mask, you need to set to wlan using the wifi\_wowlan\_set\_pattern API.

Take the TCP data packet as an example, assuming the Ameba MAC address is 00: E0: 4C: 87: 00: 00, the following description sets the TCP Unicast data packet whose receiver is Ameba as the wake-up packet. First, the MAC Destination of the comparison packet needs to be Ameba, so set the eth\_da field of wowlan\_pattern\_t to 00: E0: 4C: 87: 00: 00. Next, set the Protocol type of the LLC Header to IP Protocol: {0x08, 0x00}, Version + Length: {0x45}, the Protocol Type of the IP Header to TCP: {0x06}, and set the Destination IP to Ameba's IP.

After the above fields are set, they will be converted into HW comparison format, as follows:

| eth_da<br>(6) | eth_sa<br>(6) | eth_proto_type<br>(2) | header_len<br>(1) | Rsvd<br>(8) | ip_proto<br>(1) | Rsvd<br>(2) | ip_sa<br>(4) | ip_da<br>(4) | src_port<br>(2) | Dest_port<br>(2) | Flag2<br>(1) |
|---------------|---------------|-----------------------|-------------------|-------------|-----------------|-------------|--------------|--------------|-----------------|------------------|--------------|
|---------------|---------------|-----------------------|-------------------|-------------|-----------------|-------------|--------------|--------------|-----------------|------------------|--------------|

Therefore, the HW Pattern after the TCP Pattern conversion in this example is:

|                      |                      |             |                            |          |                |                |             |       |    |    |   |
|----------------------|----------------------|-------------|----------------------------|----------|----------------|----------------|-------------|-------|----|----|---|
| 00 e0 4c 87<br>00 00 | 00 00 00<br>00 00 00 | 08 00<br>45 | 00 00 00 00 00<br>00 00 00 | 06<br>00 | 00 00<br>00 00 | c0 a8 00<br>c4 | 00 64<br>18 | e1 6c | 11 | 11 | 1 |
|----------------------|----------------------|-------------|----------------------------|----------|----------------|----------------|-------------|-------|----|----|---|

Next, the user needs to set the Mask. 1 bit in the Mask corresponds to 1 byte of the HW pattern, and the byte corresponding to the 1 bit in the Mask will be added for comparison. The following explains how Mask is composed:

First use the bit sequence mask to identify the bytes to be compared:

|        |        |    |   |          |   |    |      |      |    |    |   |
|--------|--------|----|---|----------|---|----|------|------|----|----|---|
| 111111 | 000000 | 11 | 1 | 00000000 | 1 | 00 | 0000 | 1111 | 11 | 11 | 1 |
|--------|--------|----|---|----------|---|----|------|------|----|----|---|

This bit sequence is composed of 1 byte every 8 bits:

|          |          |          |          |        |   |
|----------|----------|----------|----------|--------|---|
| 11111100 | 00001110 | 00000001 | 00000011 | 111111 | 1 |
|----------|----------|----------|----------|--------|---|

HW is compared from bit0 of each byte, so the bit order of each byte must be reversed.

|          |          |          |          |          |          |
|----------|----------|----------|----------|----------|----------|
| 00111111 | 01110000 | 10000000 | 11000000 | 00111111 | 10000000 |
|----------|----------|----------|----------|----------|----------|

Convert from step 3 to Hex: {0x3f, 0x70, 0x80, 0xc0, 0x3f, 0x80} to get the final Mask.

#### 10.5.4.1 Wakeup pattern payload

Set the payload\_mask, 1 bit in the Mask corresponds to 1 byte of the HW pattern, and the byte corresponding to the bit of Mask 1 will be added to the comparison. Assuming that a payload of 10 bytes is set, the following explains how the Mask is composed:

Every 8 bits of the bit sequence form 1 byte, and the first 6 bits are reserved:

|          |          |          |          |          |          |          |          |          |
|----------|----------|----------|----------|----------|----------|----------|----------|----------|
| 00000011 | 11111111 | 00000000 | 00000000 | 00000000 | 00000000 | 00000000 | 00000000 | 00000000 |
|----------|----------|----------|----------|----------|----------|----------|----------|----------|

HW is compared from the bit0 of each byte, so the bit order of each byte must be reversed

|          |          |          |          |          |          |          |          |          |
|----------|----------|----------|----------|----------|----------|----------|----------|----------|
| 11000000 | 11111111 | 00000000 | 00000000 | 00000000 | 00000000 | 00000000 | 00000000 | 00000000 |
|----------|----------|----------|----------|----------|----------|----------|----------|----------|

Convert to Hex from step 2: {0xc0, 0xff, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00}, get the final Mask.

### 10.5.5 Wakeup from SSL pattern

Wake up from ssl mode support tls cipher suite aes256, sha384.

After establishing a TLS connection, you can wake up by setting ssl key offload and ssl wakeup mode.

When the user needs to use the remote wake-up function, when wlan receives a matching data packet, the system can exit the Standby mode.

#### 10.5.5.1 SSL key offload

The part of the SSL Key offload comparison includes the ssl\_ctr, ssl\_iv, ssl\_enc\_key, ssl\_dec\_key and ssl hmac key. To set a SSL Key offload, users need to set wifi\_set\_ssl\_offload() after TLS connection, the above items must be set in the SSL Key offload structure:

|                         |
|-------------------------|
| SSL_CTR (8 bytes)       |
| SSL_IV (16 bytes)       |
| SSL_ENC_KEY (32 bytes)  |
| SSL_DEC_KEY (32 bytes)  |
| SSL_HMAC_KEY (48 bytes) |
| SSL_IS_ETM (1 bytes)    |

#### 10.5.5.2 SSL wakeup pattern

SSL Wakeup pattern support 8 groups, each pattern supports up to 64 Bytes. The user can fill in the patterns used for waking up in sequence through wifi\_wowlan\_set\_ssl\_pattern(). After waking up, the wakeup reason can know which pattern was awakened.

#### 10.5.5.3 SSL wakeup pattern

SSL Wakeup pattern support 8 groups, each pattern supports up to 64 Bytes. The user can fill in the patterns used for waking up in sequence through wifi\_wowlan\_set\_ssl\_pattern(). After waking up, the wakeup reason can know which pattern was awakened.

# 11 System API

## 11.1 System reset

Reset the system

```
/***
 * @brief system software reset.
 * @retval none
 */
void sys_reset(void)
```

## 11.2 Get boot select

Identify the system is boot from NAND or NOR flash

```
/***
 * @brief Get boot select function.
 * @retval boot select device
 * @note
 * BootFromNORFlash = 0,
 * BootFromNANDFlash = 1,
 * BootFromUART = 2
 */
uint8_t sys_get_boot_sel(void)
```

## 11.3 JTAG/SWD disable

JTAG/SWD is enabled by default. Turn off JTAG/SWD to release more pins to the application

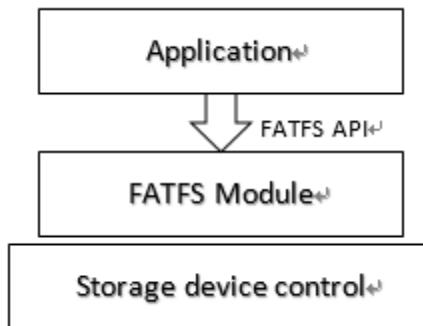
```
/***
 * @brief Turn off the JTAG/SWD function.
 * @retval none
 */
void sys_jtag_off(void)
```

# 12 File System

## 12.1 FATFS File System

Storage is a key feature of embedded system. AmebaPro2 provides flexible method of storage management. In this chapter, three kinds of application scenarios (SD/RAM/flash) will be mentioned.

### 12.1.1 FATFS Module



AmebaPro2 utilizes FAT File system Module to provide access to low level storage devices. Applications can manage and operate the file system through FATFS API.

### 12.1.2 FATFS API

AmebaPro2 SDK uses open source FATFS module. The application interface provides various functions for applications to manipulate the file system.

#### (1) File Access

- f\_open - Open/Create a file
- f\_close - Close an open file
- f\_read - Read data from the file
- f\_write - Write data to the file
- f\_lseek - Move read/write pointer, Expand size
- f\_truncate - Truncate file size
- f\_sync - Flush cached data
- f\_forward - Forward data to the stream
- f\_expand - Allocate a contiguous block to the file
- f\_gets - Read a string
- f\_putc - Write a character
- f\_puts - Write a string
- f\_printf - Write a formatted string
- f\_tell - Get current read/write pointer
- f\_eof - Test for end-of-file
- f\_size - Get size
- f\_error - Test for an error

#### (2) Directory Access

- f\_opendir - Open a directory
- f\_closedir - Close an open directory
- f\_readdir - Read an directory item
- f\_findfirst - Open a directory and read the first item matched
- f\_findnext - Read a next item matched

#### (3) File and Directory Management

- f\_stat - Check existance of a file or sub-directory
- f\_unlink - Remove a file or sub-directory
- f\_rename - Rename/Move a file or sub-directory

- f\_chmod - Change attribute of a file or sub-directory
  - f\_ftime - Change timestamp of a file or sub-directory
  - f\_mkdir - Create a sub-directory
  - f\_chdir - Change current directory
  - f\_chdrive - Change current drive
  - f\_getcwd - Retrieve the current directory and drive
- (4) Volume Management and System Configuration
- f\_mount - Register/Unregister the work area of the volume
  - f\_mkfs - Create an FAT volume on the logical drive
  - f\_fdisk - Create logical drives on the physical drive
  - f\_getfree - Get total size and free size on the volume
  - f\_getlabel - Get volume label
  - f\_setlabel - Set volume label
  - f\_setcp - Set active code page

More details about the usage of FATFS API, please visit [http://elm-chan.org/fsw/ff/00index\\_e.html](http://elm-chan.org/fsw/ff/00index_e.html)

### 12.1.3 FATFS Example

The example for NOR flash and RAM file system need to be assigned the start address to run:

In FLASH\_FATFS.c

```
define FLASH_APP_BASE 0x180000
#define FLASH_BLOCK_SIZE 512
#define FLASH_SECTOR_COUNT 256
#define SECTOR_SIZE_FLASH 4096
```

In FATFS\_RAMDISK\_API.c

```
#define RAM_DISK_SZIE 1024*1024*10
#define SECTOR_SIZE_RAM 512
#define SECTOR_COUNT_RAM (RAM_DISK_SZIE/512)
```

Please execute the example\_fatfs.c to run the example.

### 12.1.4 FATFS Behavior Description

In this example, we demonstrate how to use FATFS on AmebaPro2 flash memory and manage files and directories in the file system.

First, we use FATFS API to register flash disk driver and get a drive number for the flash drive. We use this drive number as its path and mount to a FATFS object.

Next, the example list files currently exist in the flash memory, clear all files and directories, and list files again to check if the drive is all clean and empty.

Next, the example uses f\_mkdir API to create a directory named "ameba\_dir" in the root of the filesystem and use f\_open to create a file named "ameba\_dir\_file" in ameba\_dir. Then list files to show the created directory and file.

Next, we create a file named "ameba\_root\_file" at the root of the drive, and use f\_write API to try to write some content to the file. Then use f\_read API to read from the file to check if the content written to the file can be read back correctly.

Finally we list all files and directories in the drive.

### 12.1.5 Dual Fat File system (File system on both SD Card and Flash)

Please modify the example\_fatfs.h to enable the SD and FLASH function.

```
#define CONFIG_FATFS_IF_SD 1
#define CONFIG_FATFS_IF_FLASH 1
```

In this example, we demonstrate how to use FATFS on both AmebaPro2 flash memory and SD card, and manage files and directories in the two filesystems.

First, we use FATFS API to register flash disk driver and SD disk driver, and each drive gets a drive number. We use the drive number as drive path and mount flash drive and SD drive, each with a FATFS object.

Next, the example clears files currently exist in both drives, and list files again to check if the drives are all clean and empty.

Next, the example tests operations on the SD drive. We create a new file("sd\_file") and perform read/write to the file, then create a new directory("sd\_dir") and open a new file in the directory("sd\_file2").

Next, the example tests similar operations on the flash drive. Create a new file("flash\_file") and perform read/write to the file. Then we create a new directory("flash\_dir") and open a new file in the directory("flash\_file2").

Finally we list all files and directories in each drive.

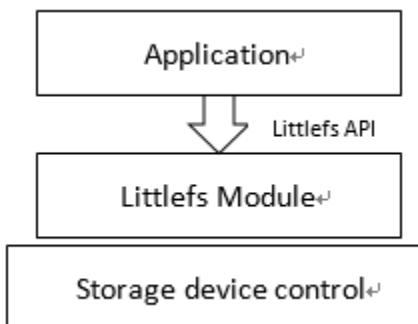
## 12.2 LITTLEFS File System

The little file system (LittleFS) is a fail-safe file system designed for embedded systems, specifically for microcontrollers that use external flash storage.

There are three challenges for embedded storage on microcontrollers and flash storage: power loss, wear and limited RAM and ROM. This file system provides the solution to these challenges.

- Bounded RAM/ROM - This file system works with a limited amount of memory. It avoids recursion and limits dynamic memory to configurable buffers that can be provided statically.
- Power-loss resilient - We have designed this for operating systems that may have random power failures. It has strong copy-on-write guarantees and keeps storage on disk in a valid state.
- Wear leveling - Because the most common form of embedded storage is erodible flash memories, the file system provides a form of dynamic wear leveling for systems that cannot fit a full flash translation layer.

### 12.2.1 LITTLEFS Module



AmebaPro2 utilizes Littlefs File system Module to provide access to low level storage devices. Applications can manage and operate the file system through Littlefs API.

### 12.2.2 LITTLEFS API

#### (1) File Access

- `lfs_file_open` - Open a file.
- `lfs_file_opencfg` - Open a file with extra configuration.
- `lfs_file_close` - Close a file.
- `lfs_file_sync` - Synchronize a file on storage.
- `lfs_file_read` - Read data from file.
- `lfs_file_write` - Write data to file
- `lfs_file_seek` - Change the position of the file.
- `lfs_file_truncate` - Truncates the size of the file to the specified size
- `lfs_file_tell` - Return the position of the file.
- `lfs_file_rewind` - Change the position of the file to the beginning of the file.
- `lfs_file_size` - Return the size of the file.

#### (1) Directory Access

- `lfs_mkdir` - Create a directory.
- `lfs_dir_open` - Open a directory.

- `lfs_dir_close` - Close a directory
- `lfs_dir_read` - Read an entry in the directory.
- `lfs_dir_seek` - Change the position of the directory.
- `lfs_dir_tell` - Return the position of the directory.
- `lfs_dir_rewind` - Change the position of the directory to the beginning of the directory.

(2) File and Directory Management

- `lfs_remove` - Removes a file or directory.
- `lfs_rename` - Rename or move a file or directory.
- `lfs_stat` - Find info about a file or directory.
- `lfs_getattr` - Get a custom attribute.
- `lfs_setattr` - Set custom attributes.
- `lfs_removeattr` - Removes a custom attribute
- `lfs_fs_size` - Finds the current size of the file system.
- `lfs_fs_traverse` - Traverse through all blocks in use by the file system.

(3) Volume Management and System Configuration

- `lfs_format` - Format a block device with the littlefs.
- `lfs_mount` - Mounts a littlefs.
- `lfs_unmount` - Unmounts a littlefs.
- `lfs_migrate` - Attempts to migrate a previous version of littlefs.

### 12.2.3 LITTLEFS Example

It can support the NOR and NAND flash. It depend on different file operation.

Please run the example\_littlefs to run the example. The behavior is the same as fatfs example.

Both of them need to assign the start address and block size. Please reference the `lfs_nor_api.c` and `lfs_nand_api.c` to do the setup.

# 13 Audio optimization

The following chapters describe the software and hardware optimization solutions of AmebaPro2 audio.

## 13.1 Audio setting

### 13.1.1 Gain setting

#### 13.1.1.1 Analog microphone gain setting

The audio analog input gain can be namely divided into analog gain and digital gain.

- Analog mic gain

it supports 0, 20, 30, 40 dB for the gain optimization.

User can use `audio_mic_analog_gain` or set the parameter `mic_gain` for audio module to set it.

- ADC gain can be used to set the input (analog to) digital gain – ADC Volume.

the range is -17.625dB (0x00) ~ 30dB (0x7F)

User can use the function `audio_adc_digital_vol` or use `CMD_AUDIO_SET_ADC_GAIN` to control audio module to use the function.

A digital gain configuration is offered to control the audio output gain. Customers can set a reasonable gain value via DAC Volume to obtain the appropriate audio output volume according to their needs. Basically setting the gain to 0dB (0xAF), the output amplitude will meet the board audio output volume requirements. Note that a sound breakage will happen when the output gain is setting too large.

If the analog gain is too large, analog gain will affect the sound effect and noise will be obvious.

Recommend that customers can first configure the digital gain. If the audio signal gain need to increase but the digital gain achieves the maximum range, then configure the analog gain.

#### 13.1.1.2 Digital microphone gain setting

- Left mic gain

Left dmic gain and it supports 0, 12, 24, 36 dB for the gain optimization.

User can use `audio_l_dmic_gain` or set the parameter `dmc_l_gain` for audio module to set it.

- Right mic gain

Right dmic gain and it supports 0, 12, 24, 36 dB for the gain optimization.

User can use `audio_r_dmic_gain` or set the parameter `dmc_r_gain` for audio module to set it.

#### 13.1.1.3 Speaker gain setting

- DAC gain can be used to set the output digital (to analog) gain – DAC Volume.

the range is -65.625dB (0x00) ~ 0dB (0xAF)

User can use the function `audio_dac_digital_vol` or use `CMD_AUDIO_SET_DAC_GAIN` to control audio module to use the function.

## 13.1.2 HPF setting

In AmebaPro2, it provides a high pass filter for user to filter the low frequency noise.

Here is the function:

```
void audio_adc_l_hpf(audio_t *obj, BOOL en, audio_hpf_fc hpf_fc);
```

the parameters mean:

- obj: Audio object define in application software.
- en: enable the high pass frequency or not.
- hpf\_fc: set the cutoff frequency, the value is from 0~7; fc ~ 5e-3 / (hpf\_fc + 1) \* fs.

## 13.1.3 EQ setting

In AmebaPro2, it also provides five sets of biquad filters in three sides for left digital mic (analog mic), right digital mic and audio output. One biquad filter can switch to high-pass, low-pass, band-pass, notch, peak, low shelf, and high shelf filter by register settings.

Here are some tips for user to use the EQ:

- Select the biquad filter

User can use the following websites to configure the preferred filter type, sample rate, cutoff frequency, Q value and Gain first:

<https://www.earlevel.com/main/2021/09/02/biquad-calculator-v3/>

- Get the registers' value of selecting filter

User can use AmebaPro2\_EQ\_tool.exe to generate register settings. For example, if we choose a high pass filter with cutoff frequency 200Hz and Q value 0.707, user can type the setting and get the registers' value (0x1e45618, 0x1c000000, 0x2000000, 0x3c72d61, 0x1e35d500) for this setting.

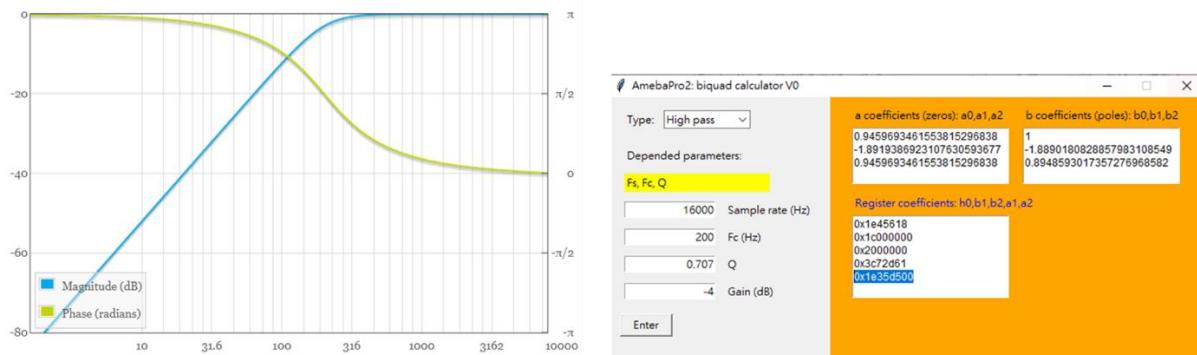


Figure 13-1 EQ setting

- Set the register value got from AmebaPro2\_EQ\_tool.exe

After getting the registers' value, user can use the following function to apply the filter setting on AmebaPro2 on left digital mic (analog mic), right digital mic and audio output. Note that there are 5 filters could be set in each side.

```
void audio_input_l_eq(audio_t *obj, audio_eq eq, BOOL en, u32 h0, u32 b0, u32 b1, u32 a0, u32 a1);
void audio_input_r_eq(audio_t *obj, audio_eq eq, BOOL en, u32 h0, u32 b0, u32 b1, u32 a0, u32 a1);
void audio_output_l_eq(audio_t *obj, audio_eq eq, BOOL en, u32 h0, u32 b0, u32 b1, u32 a0, u32 a1);
```

Here are the parameters:

- obj: Audio object define in application software.
- eq: Select the EQ number, can be 0~4.
- en: enable the eq filter or not
- h0, b0, b1, a0, a1: the registers' value gotten from AmebaPro2\_EQ\_tool.exe.

### 13.1.4 Other setting

Here are some commands about the module audio setting:

- CMD\_AUDIO\_SET\_RESET

will be re-initialize the audio setting and also the ASP algorithms. If you do some changes need to reset the audio configuration, like change the sample rate, reset the audio to switch the configuration.

- CMD\_AUDIO\_SET\_SAMPLERATE

can set the sample rate. After using this command, a reset is needed to apply the sample rate configuration on audio and ASP algorithms.

#### NOTE

If using audio codec, be sure the sample rate is fitting the sample rate used in audio codec.

- CMD\_AUDIO\_SET\_TRX

provide a way to stop and re-start the audio without re-initialize the audio system and ASP algorithms. Set 0 to stop the tx and rx progresses or 1 to start them.

## 13.2 Audio ASP algorithm

The following table shows some common audio problem with their causes and also the adjustment using ASP algorithm.

| Situation        | Algorithm | Influence End    | Cases                                                                                                                                                                        |
|------------------|-----------|------------------|------------------------------------------------------------------------------------------------------------------------------------------------------------------------------|
| Distortion       | AGC       | transmitting end | <ul style="list-style-type: none"> <li>The ambient sound is too high</li> <li>Headphone preGain</li> <li>Compression_gain_db of AGC is too large</li> </ul>                  |
| Low audio volume | AGC       | transmitting end | <ul style="list-style-type: none"> <li>The original input volume is too low</li> <li>Compression_gain_db of AGC is too small</li> <li>AGC is not working properly</li> </ul> |
| Echo or howling  | AEC       | transmitting end | <ul style="list-style-type: none"> <li>Too close between transmitting and receiving end device</li> <li>Volume too large or mic too sensitive</li> </ul>                     |

|                    |                |                  |                                                                                                                                                                           |
|--------------------|----------------|------------------|---------------------------------------------------------------------------------------------------------------------------------------------------------------------------|
|                    |                |                  | <ul style="list-style-type: none"> <li>● AEC is not turn on</li> <li>● AEC parameters is not setting correctly (frame_size, sample rate, set_sndcard_delay_ms)</li> </ul> |
| Intermittent voice | AEC、NS         | transmitting end | <ul style="list-style-type: none"> <li>● NS or AEC suppression</li> </ul>                                                                                                 |
| Noise floor        | NS             | transmitting end | <ul style="list-style-type: none"> <li>● NS mode setting too low</li> <li>● Caused by the environment, NS can't do well</li> </ul>                                        |
| Mechanical sound   | Network、Device | Receiving end    | <ul style="list-style-type: none"> <li>● Poor network environment</li> <li>● Device sampling is unstable or device hardware problem</li> </ul>                            |

### 13.2.1 Open ASP algorithm

The codes and functions related to the ASP algorithm are shows in the table.

Enable ENABLE\_ASP in module\_audio.h and use the 3A (AGC: Automatic gain control; ANS: Adaptive noise suppression; AEC: Acoustic echo cancellation) and VAD (Voice Activity Detection) algorithms to obtain better audio effects.

- The parameters, sample\_rate and mic\_gain, and the initialization of NS (enable\_ns), AEC (enable\_aec), AGC (enable\_agc), VAD (enable\_vad) and other algorithms will be setting at CMD\_AUDIO\_APPLY and CMD\_AUDIO\_SET\_RESET.

To enable ASP function user can use the following parameters in audio\_params\_t:

```
===== Open ASP algorithm (module_audio.h) =====
#define ENABLE_ASP 1

typedef struct audio_params_s {
 audio_sr sample_rate; // ASR_8KHZ
 audio_wl word_length; // WL_16BIT
 audio_mic_gain mic_gain; // MIC_40DB

 int channel; // 1
 int enable_aec; // 0: off 1: on
 int enable_ns; // 0: off, 1: out 2: in 3: in/out
 int enable_agc; // 0: off, 1: output agc
 int enable_vad; // 0: off 1: input vad
 int mix_mode; // 0
 //...
} audio_params_t;
```

In "AEC.h" it defined some function for the ASP setting. The following table shows the functions for setting the ASP algorithm:

| Function      | Related module | Parameters                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                       | Note                                                                                                                                                                                                         |
|---------------|----------------|------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------|--------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------|
| AEC_init      | NS, AEC, AGC   | <ul style="list-style-type: none"> <li>● frame_size: setting the frame size for the AEC module, the unit is "sample"</li> <li>● sample_freq: audio sample rate (support 8k and 16k)</li> <li>● aec_core: select the core for AEC module, there are WEBRTC_AEC, WEBRTC_AECM and SPEEX_AEC. Suggest to use WEBRTC_AECM</li> <li>● speex_filter_length: for WEBRTC_AEC and WEBRTC_AECM, it equal to sndcard_delay_ms * (sample_rate/1000); while for SPEEX_AEC, it equal to filter_length</li> <li>● agc_mode: set the AGC mode for AGC module, the value is 0~3</li> <li>● compression_gain_db: set the compression gain in db for AGC module</li> <li>● limiter_enable: enable the limiter for AGC module</li> <li>● ns_mode: set the aggressive level (the larger the more aggressive) for NS module, the value is from 0~3</li> <li>● snd_amplification: set the amplification for the output result</li> </ul> | <ul style="list-style-type: none"> <li>● For mic side ASP</li> <li>● The AEC process include NS, AEC and AGC.</li> <li>● If use AEC process, additional NS and AGC process is no need in mic side</li> </ul> |
| AEC_set_level | AEC            | <ul style="list-style-type: none"> <li>● level: the aggressive level (the larger the more aggressive) for AEC module, the level is from 0~4</li> </ul>                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                           | <ul style="list-style-type: none"> <li>● For mic side ASP</li> </ul>                                                                                                                                         |

|                                 |              |                                                                                                                                                                                                                                                                                                                           |                                                                                                                                                                                |
|---------------------------------|--------------|---------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------|--------------------------------------------------------------------------------------------------------------------------------------------------------------------------------|
| AEC_process                     | NS, AEC, AGC | <ul style="list-style-type: none"> <li>• farend: the array input for the far-end data</li> <li>• nearend: the array input for the far-end data</li> <li>• out: the space to save the ASP processed data</li> </ul>                                                                                                        | <ul style="list-style-type: none"> <li>• For mic side ASP</li> </ul>                                                                                                           |
| AEC_destory                     | NS, AEC, AGC |                                                                                                                                                                                                                                                                                                                           | <ul style="list-style-type: none"> <li>• For mic side ASP</li> <li>• For destroy the modules initialed in AEC_init</li> </ul>                                                  |
| set_sndcard_delay_ms_for_webRTC | AEC          | <ul style="list-style-type: none"> <li>• ms: set sndcard delay delay for AEC module</li> </ul>                                                                                                                                                                                                                            | <ul style="list-style-type: none"> <li>• For mic side ASP</li> <li>• Only will effect on AEC core is WEBRTC_AEC or WEBRTC_AECM</li> <li>• Can be set after AEC_init</li> </ul> |
| AGC2_init                       | AGC          | <ul style="list-style-type: none"> <li>• sample_freq: audio sample rate (support 8k and 16k)</li> <li>• agc_mode: set the AGC mode for AGC module, the value is 0~3</li> <li>• compression_gain_db: set the compression gain in db for AGC module</li> <li>• limiter_enable: enable the limiter for AGC module</li> </ul> | <ul style="list-style-type: none"> <li>• For mic side ASP</li> </ul>                                                                                                           |
| AGC2_process                    | AGC          | <ul style="list-style-type: none"> <li>• frame_size: setting the frame size for the AGC module, the unit is "sample"</li> <li>• out: the data will be used to do AGC process, the data will directly be modified</li> </ul>                                                                                               | <ul style="list-style-type: none"> <li>• For mic side ASP</li> </ul>                                                                                                           |
| AGC2_destory                    | AGC          |                                                                                                                                                                                                                                                                                                                           | <ul style="list-style-type: none"> <li>• For mic side ASP</li> <li>• For destroy the modules creates in AGC2_init</li> </ul>                                                   |
| AGC_init                        | AGC          | <ul style="list-style-type: none"> <li>• sample_freq: audio sample rate (support 8k and 16k)</li> <li>• agc_mode: set the AGC mode for AGC module, the value is 0~3</li> <li>• compression_gain_db: set the compression gain in db for AGC module</li> <li>• limiter_enable: enable the limiter for AGC module</li> </ul> | <ul style="list-style-type: none"> <li>• For output (speaker) side ASP</li> </ul>                                                                                              |
| AGC_process                     | AGC          | <ul style="list-style-type: none"> <li>• frame_size: setting the frame size for the AGC module, the unit is "sample"</li> <li>• out: the data will be used to do AGC process, the data will directly be modified</li> </ul>                                                                                               | <ul style="list-style-type: none"> <li>• For output (speaker) side ASP</li> </ul>                                                                                              |
| AGC_destory                     | AGC          |                                                                                                                                                                                                                                                                                                                           | <ul style="list-style-type: none"> <li>• For output (speaker) side ASP</li> <li>• For destroy the modules creates in AGC_init</li> </ul>                                       |
| NS2_init                        | NS           | <ul style="list-style-type: none"> <li>• sample_freq: audio sample rate (support 8k and 16k)</li> <li>• ns_mode: set the aggressive level (the larger the more aggressive) for NS module, the value is from 0~3</li> </ul>                                                                                                | <ul style="list-style-type: none"> <li>• For mic side ASP</li> </ul>                                                                                                           |
| NS2_process                     | NS           | <ul style="list-style-type: none"> <li>• frame_size: setting the frame size for the AGC module, the unit is "sample"</li> <li>• out: the data will be used to do AGC process, the data will directly be modified</li> </ul>                                                                                               | <ul style="list-style-type: none"> <li>• For mic side ASP</li> </ul>                                                                                                           |
| NS2_destory                     | NS           |                                                                                                                                                                                                                                                                                                                           | <ul style="list-style-type: none"> <li>• For mic side ASP</li> <li>• For destroy the modules creates in NS2_init</li> </ul>                                                    |
| NS_init                         | NS           | <ul style="list-style-type: none"> <li>• sample_freq: audio sample rate (support 8k and 16k)</li> <li>• ns_mode: set the aggressive level (the larger the more aggressive) for NS module, the value is from 0~3</li> </ul>                                                                                                | <ul style="list-style-type: none"> <li>• For output (speaker) side ASP</li> </ul>                                                                                              |

|             |     |                                                                                                                                                                                                                          |                                                                                                                                     |
|-------------|-----|--------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------|-------------------------------------------------------------------------------------------------------------------------------------|
| NS_process  | NS  | <ul style="list-style-type: none"> <li>frame_size: setting the frame size for the AGC module, the unit is "sample"</li> <li>out: the data will be used to do AGC process, the data will directly be modified</li> </ul>  | <ul style="list-style-type: none"> <li>For output (speaker) side ASP</li> </ul>                                                     |
| NS_destory  | NS  |                                                                                                                                                                                                                          | <ul style="list-style-type: none"> <li>For output (speaker) side ASP</li> <li>For destroy the modules creates in NS_init</li> </ul> |
| VAD_init    | VAD | <ul style="list-style-type: none"> <li>sample_freq: audio sample rate (support 8k and 16k)</li> <li>vad_mode: set the aggressive level (the larger the more aggressive) for VAD module, the value is from 0~3</li> </ul> | <ul style="list-style-type: none"> <li>For mic side ASP</li> </ul>                                                                  |
| VAD_process | VAD | <ul style="list-style-type: none"> <li>frame_size: setting the frame size for the AGC module, the unit is "sample"</li> <li>out: the data will be used to do AGC process, the data will directly be modified</li> </ul>  | <ul style="list-style-type: none"> <li>For mic side ASP</li> </ul>                                                                  |
| VAD_destory | VAD |                                                                                                                                                                                                                          | <ul style="list-style-type: none"> <li>For mic side ASP</li> <li>For destroy the modules creates in NS_init</li> </ul>              |

### 13.2.1.1 ASP algorithm usage

Here are the configurations for ASP algorithm:

- 8K and 16K audio sample rate are supported in the ASP algorithms.
- When enable\_ns is set, 0 is to disable NS, 1 is to enable NS\_init() for Speaker, 2 means enable NS2\_init() for MIC, 3 is to enable both directions.
- When enable\_agc is set, 0 is to disable AGC, 1 is to enable AGC\_init() for Speaker, 2 is to enable AGC2\_init() for MIC, and 3 is to enable both directions.
- Set enable\_aec 0/1 to disable/enable the AEC\_init().
- In AEC\_process, it will also run NS and AGC algorithms, so it will be unnecessary to apply additional NS and AGC for MIC (NS2\_init() and AGC2\_init()) if open enable\_aec.
- Set enable\_vad to 0/1 to disable/enable VAD\_init().

### 13.2.1.2 AEC setting

The AEC algorithm includes three parts: delay adjustment strategy, linear echo estimation, and nonlinear echo suppression.

- Use CMD\_AUDIO\_RUN\_AEC to dynamically switch the use of AEC\_process().
- Use CMD\_AUDIO\_SET\_AEC\_ENABLE to determine whether AEC\_init() is enabled during audio reset.
- CMD\_AUDIO\_SET\_AEC\_LEVEL can set the strength of cancellation.
- Note that if using WEBRTC\_AECM (default) as AEC\_CORE to do initialization, the AEC strength level is 0 ~ 4 and the minimum strength is 0; while using WEBRTC\_AEC as AEC\_CORE to do initialization, the AEC strength level is 0 ~ 2 and the minimum strength is 0.
- Default value of WEBRTC\_AECM is 3, and default value of WEBRTC\_AEC is 1
- Please make sure **that the level setting should be after the audio is initialized**.
- The parameters agc\_mode, compression\_gain\_db and limiter\_enable of AEC\_init are for setting the agc processing mode, compression gain (default 18dB) and limiter (default 0 => disable) of AGC algorithm in AEC\_process.
- The parameters ns\_mode of AEC\_init are for setting the NS strength of NS algorithm in AEC\_process.

### 13.2.1.3 AEC effect

Here is the estimation result of AEC algorithm on the device.

- For the audio setting, the MIC gain, ADC gain and DAC gain are set as 40dB, 12dB (0x4F) and 0dB (0xAF).
- Only the NS process with NS mode level 3 is applied before AEC algorithm to decrease the noise of the rx input.
- The ACOM is obtained by the average 1 minutes speech result.

For single talk test, the ACOM will be calculated after 1 second for waiting the AEC algorithm frozen.

**NOTE**

The echo loss is not larger than 10 dB.

The table of single talk ACOM of male and female speech:

| Input signal level | -10 dBm0 | -15 dBm0 | -20 dBm0 | -25 dBm0 | -30 dBm0 |
|--------------------|----------|----------|----------|----------|----------|
| Male speech        | 71.27 dB | 68.60 dB | 70.60 dB | 63.19 dB | 57.89 dB |
| Female speech      | 66.23 dB | 66.23 dB | 61.40 dB | 61.36 dB | 51.92 dB |

**NOTE**

According to the ITU G.167, for mobile radio systems, ACOM (single talk) shall be at least [45 dB] when no acoustic noise is added at the Sin interface.

For double talk test, the ACOM will be calculated right after the far end talk end (about 30 seconds) for waiting the AEC algorithm frozen.

The table of double talk ACOM of male and female speech:

| Input signal level | -10 dBm0 | -15 dBm0 | -20 dBm0 | -25 dBm0 | -30 dBm0 |
|--------------------|----------|----------|----------|----------|----------|
| Male speech        | 75.30 dB | 68.89 dB | 67.52 dB | 61.71 dB | 50.96 dB |
| Female speech      | 60.74 dB | 59.38 dB | 62.36 dB | 56.32 dB | 50.65 dB |

**NOTE**

According to the ITU G.167, for mobile radio systems, ACOM (double talk) shall be at least [30 dB] when no acoustic noise is added at the Sin interface.

#### 13.2.1.4 NS setting

The NS algorithm is aimed at decrease the noise or environment sound, so it is recommended to use before other ASP algorithms.

- Use CMD\_AUDIO\_SET\_NS\_ENABLE to determine whether NSx\_init() is enabled during audio reset.
- Setting parameter NS\_MODE in NSx\_init() to adjust ns aggressive level. The value is from 0 to 3 and the default value is 3.
- Use CMD\_AUDIO\_RUN\_NS to dynamically switch the use of NSx\_process().

#### 13.2.1.5 AGC setting

The AGC algorithm is used to balance the audio volume of signal streaming.

- Use CMD\_AUDIO\_SET\_AGC\_ENABLE to determine whether AGCx\_init() is enabled during audio reset.
- The parameter of agc\_mode can choose the AGC mode to initial, compression\_gain\_db (default 18/24 for in/not in the AEC) can setting max gain of AGC and limiter\_enable (default 0) to restrict the signal level.
- Use CMD\_AUDIO\_RUN\_AGC can dynamically switch the use of AGCx\_process().

#### 13.2.1.6 VAD setting

The VAD algorithm is applied to do voice enhancement.

- Use CMD\_AUDIO\_SET\_VAD\_ENABLE to determine whether VAD\_init() is enabled during audio reset.
- The parameter VAD MODE can set the aggressive level of VAD. The value can be configured from 0 to 3 and default is 1.
- Use CMD\_AUDIO\_RUN\_VAD to dynamically switch the use of VAD\_process().

### 13.3 Audio test tool

AmebaPro2 provide an example for audio testing.

User can use the following steps to build up the audio test tool image

- Step1: cd project\realtek\_ameapro2\_v0\_example\GCC-RELEASE
- Step2: mkdir build
- Step3: cd build
- Step4: cmake .. -G"Unix Makefiles" -DCMAKE\_TOOLCHAIN\_FILE=../toolchain.cmake -DAUDIO\_TEST\_TOOL=on
- Step5: cmake --build . --target flash

The following shows the command of the test tool

| command | parameters | description |
|---------|------------|-------------|
|---------|------------|-------------|

|          |                                                                                                                                               |                                                                                                                                                                                                                                                                                                                                                          |
|----------|-----------------------------------------------------------------------------------------------------------------------------------------------|----------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------|
| AUMMODE  | <ul style="list-style-type: none"> <li>[mic_mode]=amic/l_dmic/r_dmic/stereo_dmic</li> </ul>                                                   | Set up the microphone type                                                                                                                                                                                                                                                                                                                               |
| AUMG     | <ul style="list-style-type: none"> <li>[mic_gain]=0~3</li> </ul>                                                                              | Set up the analog mic gain (0: 0dB, 1: 20dB, 2: 30dB, 3:40dB)                                                                                                                                                                                                                                                                                            |
| AUMB     | <ul style="list-style-type: none"> <li>[mic_bias]=0~2</li> </ul>                                                                              | Set up the mic bias (0: 0.9, 1: 0.86, 2: 0.75)                                                                                                                                                                                                                                                                                                           |
| AUMLG    | <ul style="list-style-type: none"> <li>[left_dmic_gain]=0~3</li> </ul>                                                                        | Set up the left digital mic (0: 0dB, 1: 12dB, 2: 24dB, 3:36dB)                                                                                                                                                                                                                                                                                           |
| AUMRG    | <ul style="list-style-type: none"> <li>[right_dmic_gain]=0~3</li> </ul>                                                                       | Set up the right digital mic (0: 0dB, 1: 12dB, 2: 24dB, 3:36dB)                                                                                                                                                                                                                                                                                          |
| AUADC    | <ul style="list-style-type: none"> <li>[ADC_gain]=0x00~0x7F</li> </ul>                                                                        | Set up audio input digital gain level, the gain level is up 0.375dB/step. The max and min gains are 30dB and -17.625dB.                                                                                                                                                                                                                                  |
| AUHPF    | <ul style="list-style-type: none"> <li>[cutoff num]=0~8</li> </ul>                                                                            | Set up the mic first HPF cutoff frequency. The cutoff frequency is equal to $5e-3 / (\text{cutoff num} + 1) * \text{fs}$ (sample rate frequency).                                                                                                                                                                                                        |
| AUMLEQ   | <ul style="list-style-type: none"> <li>[eq num]=0~4</li> <li>[register h0],[register a1],[register a2],[register b1],[register b2]</li> </ul> | Set up the EQ for analog or left digital microphone (PDM rising trigger). There are 5 EQs ([eq num]) can be used (the EQ0 is used for a HPF default). The register setting can generate by AmebaPro2_EQ_tool.exe.                                                                                                                                        |
| AUMREQ   | <ul style="list-style-type: none"> <li>[eq num]=0~4</li> <li>[register h0],[register a1],[register a2],[register b1],[register b2]</li> </ul> | Set up the EQ for analog or right digital microphone (PDM falling trigger). There are 5 EQs ([eq num]) can be used (the EQ0 is used for a HPF default). The register setting can generate by AmebaPro2_EQ_tool.exe.                                                                                                                                      |
| AUAEC    | <ul style="list-style-type: none"> <li>[enable]=0 or 1</li> </ul>                                                                             | Open the AEC or not.                                                                                                                                                                                                                                                                                                                                     |
| AUNS     | <ul style="list-style-type: none"> <li>[NS_level]=-1~3</li> </ul>                                                                             | Set up the noise suppression level for audio input. The larger level is, the more aggressive NS process. Close the NS module by setting [NS_level] to -1.                                                                                                                                                                                                |
| AUSPNS   | <ul style="list-style-type: none"> <li>[NS_level_SPK]=-1~3</li> </ul>                                                                         | Set up the noise suppression level for audio output. The larger level is, the more aggressive NS process. Close the NS module by setting [NS_level_SPK] to -1.                                                                                                                                                                                           |
| AUSPEQ   | <ul style="list-style-type: none"> <li>[eq num]=0~4</li> <li>[register h0],[register a1],[register a2],[register b1],[register b2]</li> </ul> | Set up the EQ for audio output. There are 5 EQs ([eq num]) can be used (the EQ0 is used for a HPF default). The register setting can generate by AmebaPro2_EQ_tool.exe.                                                                                                                                                                                  |
| AUDAC    | <ul style="list-style-type: none"> <li>[DAC_gain]=0x00~0xAF</li> </ul>                                                                        | Set up the audio output digital gain, the gain level is up 0.375dB/step. The max and min gains are 0dB and -65.625dB.                                                                                                                                                                                                                                    |
| AUSPM    | <ul style="list-style-type: none"> <li>[enable_mute]=0 or 1</li> </ul>                                                                        | Mute the audio output ([enable_mute]=1) or unmute the audio output ([enable_mute]=0).                                                                                                                                                                                                                                                                    |
| AUTXMODE | <ul style="list-style-type: none"> <li>[tx_mode]=noplay/playback/playtone/playmusic</li> <li>[audio_tone(Hz)]</li> </ul>                      | Set up the audio output mode, there are four modes supported now. The noplay mode will stop sending data to audio output .The playmusic mode will start to play the music (support 8k or 16k). The playback mode will audio input will directly send to audio output. The audio tone mode will start playing the audio tone setting by [audio_tone(Hz)]. |
| AUAMPIN  | <ul style="list-style-type: none"> <li>[pin_name]=pin name</li> <li>[on/off]=1/0</li> </ul>                                                   | Set up or down the audio amplifier pin.                                                                                                                                                                                                                                                                                                                  |
| AUSR     | <ul style="list-style-type: none"> <li>[sample_rate]=8000, 16000, 32000, 44100, 48000, 88200, 96000</li> </ul>                                | Set up the audio input and output sample rate.                                                                                                                                                                                                                                                                                                           |
| AUTRX    | <ul style="list-style-type: none"> <li>[enable]=0 or 1</li> </ul>                                                                             | Set down or up the audio input and output.                                                                                                                                                                                                                                                                                                               |
| AURES    | <ul style="list-style-type: none"> <li>[reset_enable]</li> </ul>                                                                              | Reset the audio module to enable the previous audio setting.                                                                                                                                                                                                                                                                                             |
| AUFFTS   | <ul style="list-style-type: none"> <li>[FFT_EN]=0 or 1</li> </ul>                                                                             | Enable print audio input FFT result, but the play back mode is not supported.                                                                                                                                                                                                                                                                            |
| AUFILE   | <ul style="list-style-type: none"> <li>[filename]</li> </ul>                                                                                  | Set up the audio save file name. The length                                                                                                                                                                                                                                                                                                              |
| AUSDL    | <ul style="list-style-type: none"> <li>[enable_sd_download]=0 or 1</li> </ul>                                                                 | Enable the save file download to SD card or not. If enabling the SD download, device will copy the save data to SD card after each record.                                                                                                                                                                                                               |

|       |                                                                                                                                        |                                                                                                                                                  |
|-------|----------------------------------------------------------------------------------------------------------------------------------------|--------------------------------------------------------------------------------------------------------------------------------------------------|
| AUREC | <ul style="list-style-type: none"><li>● [record_time]</li><li>● [RECORD_TYPE1], [RECORD_TYPE2],<br/>[RECORD_TYPE3]=RX,TX,ASP</li></ul> | Start record the audio data for record time. RECORD_TYPE can select to save RX (audio input), TX (audio output) and ASP (audio input after ASP). |
|-------|----------------------------------------------------------------------------------------------------------------------------------------|--------------------------------------------------------------------------------------------------------------------------------------------------|

# 14 NN

AmebaPro2 has an NN H/W engine to accelerate the neural network inference process. NN models obtained from different AI framework can be deployed on AmebaPro2 easily.

## 14.1 NN module

The NN mmf module – vipnn is provided to process the input RGB frame from video module, and do NN inference. Then, NN inference result will be stored in network output tensor. Since the output tensor format of each model are different, vipnn module will also do the post-process work for the NN output tensor to extract the information and convert to understandable message.

### 14.1.1 VIPNN module

The context of the video module shows as following:

```
typedef struct vipnn_ctx_s {
 void *parent;
 vip_network network;
 vip_buffer_create_params_t vip_param_in[MAX_IO_NUM];
 vip_buffer_create_params_t vip_param_out[MAX_IO_NUM];
 vip_buffer input_buffers[MAX_IO_NUM];
 vip_buffer output_buffers[MAX_IO_NUM];
 vipnn_params_t params;
 vipnn_status_t status;
 char network_name[64];
 int input_count;
 int output_count;
 vipnn_preproc_t pre_process;
 vipnn_postproc_t post_process;
 disp_postprcess_t disp_postproc;
 vipnn_cascaded_mode_t cas_mode;
 bool module_out_en;
 vipnn_measure_t measure;
} vipnn_ctx_t;
```

Description of parameter in vipnn\_ctx\_t:

- network: an opaque handle to the new network object if the request is executed successfully.
- vip\_param\_in: parameter of network input tensor.
- vip\_param\_out: parameter of network output tensor.
- input\_buffers: buffer for model input tensor.
- output\_buffers: buffer for model output tensor.
- params: basic parameters for the vipnn module.
- status: record status of vipnn.
- network\_name: nn network name.
- input\_count: the number of input tensor in the NN network.
- output\_count: the number of output tensor in the NN network.
- pre\_process: pre-process function for processing the data before passing to NN inference.
- post\_process: post-process function for decoding the data from NN inference.
- disp\_postproc: Set the callback function for display the NN result on video frame. It could be set by using CMD\_VIPNN\_SET\_DISPPOST.
- module\_out\_en: enable module output.
- measure: time measurement

#### 14.1.1.1 Basic vipnn module parameters setting

Here are some vipnn module parameters provided to set.

```
typedef struct vipnn_param_s {
 int model_type;
 char model_file[64];
 uint8_t *model_mem;
 uint32_t model_size;
 int fps;
 int in_width, in_height;
 rect_t roi;
 int m_width, m_height; // should read from model, not user setting
}
```

```

 nn_data_param_t *in_param;
 nnmodel_t *model;
} vipnn_params_t;
...
nn_data_param_t in_param = {
 .img = {
 .width = NN_WIDTH,
 .height = NN_HEIGHT,
 .rgb = 0,
 .roi = {
 .xmin = 0,
 .ymin = 0,
 .xmax = NN_WIDTH,
 .ymax = NN_HEIGHT,
 }
 }
};

```

Use CMD\_VIPNN\_SET\_IN\_PARAMS to set up the NN input parameters.

- img.width: input frame width.
- img.height: input frame height.
- img.rgb: enable reverse channel (1: RGB ↔ BGR, 0: disable).
- img.roi: ROI of input frame (xmin, ymin, xmax, ymax).

#### 14.1.1.2 Set NN model to vipnn module

Use CMD\_VIPNN\_SET\_MODEL to set up the NN model:

```

vipnn_ctx = mm_module_open(&vipnn_module);
if (vipnn_ctx) {
 ...
 mm_module_ctrl(vipnn_ctx, CMD_VIPNN_SET_MODEL, (int)&yolov3_tiny_fwfs);
 ...
}

```

**NOTE**

If using Yolov4-tiny, just set model as yolov3\_tiny here.

#### 14.1.1.3 Set NN result display callback function

User can register a call back function to so display the NN result or do their own customized additional post-processing. Use CMD\_VIPNN\_SET\_DISPPOST to set up callback function for display the NN result:

```

static void nn_result_display (void *p, void *img_param)
{
 objdetect_res_t *res = (objdetect_res_t *)p;
 nn_data_param_t *im = (nn_data_param_t *)img_param;

 /* Process or display the result here */
}

...
...

vipnn_ctx = mm_module_open(&vipnn_module);
if (vipnn_ctx) {
 ...
 mm_module_ctrl(vipnn_ctx, CMD_VIPNN_SET_DISPPOST, (int)nn_result_display);
 ...
}

```

#### 14.1.1.4 Set NN object detection threshold

There are two threshold values related to NN post-processing result – confidence & NMS threshold.

Confidence is the score of the bounding box. Use CMD\_VIPNN\_SET\_SCORE\_THRES to set up confidence score threshold:

```

static float nn_confidence_thresh = 0.5;
mm_module_ctrl(vipnn_ctx, CMD_VIPNN_SET_CONFIDENCE_THRES, (int)&nn_confidence_thresh);

```

For the same class, if the IOU (Intersection over union) of two bounding box larger then NMS threshold, these two objects will be considered the same object. Use CMD\_VIPNN\_SET\_NMS\_THRES to set up NMS threshold:

```

static float nn_nms_thresh = 0.3;

```

```
mm_module_ctrl(vipnn_ctx, CMD_VIPNN_SET_NMS_THRESH, (int)&nn_nms_thresh);
```

## 14.2 Model Zoo

Currently, the SDK provides several deployed models. They are listed in following table:

Table 14-1 Pro2 model list

| Category             | Model                      | Description                                                                                                                                                       |
|----------------------|----------------------------|-------------------------------------------------------------------------------------------------------------------------------------------------------------------|
| Object detection     | Yolov4-tiny<br>Yolov3-tiny | <a href="https://github.com/AlexeyAB/darknet">https://github.com/AlexeyAB/darknet</a>                                                                             |
| Sound classification | YAMNet                     | <a href="https://github.com/tensorflow/models/tree/master/research/audioset/yamnet">https://github.com/tensorflow/models/tree/master/research/audioset/yamnet</a> |

### 14.2.1 Object detection model

SDK provides object detection model for user to evaluate – Yolov4-tiny.

#### 14.2.1.1 Yolov4-tiny

YOLO (you only look once) is a neural network algorithm for object detection, implemented with darknet architecture. Yolo is well-known for its lightweight, less dependent and efficient in algorithms.

For more information, see Yolo's Github maintain by its authors: <https://github.com/AlexeyAB/darknet>

 NOTE

*SDK also provide older Yolov3-tiny model. However, Yolov4-tiny is recommended for its better performance.*

### 14.2.2 Sound classification model

A pre-trained sound classification model is provided in SDK – YAMNet.

#### 14.2.2.1 YAMNet

YAMNet is a model that can predicts 521 audio event classes based on the AudioSet.

For more information, see TensorFlow official Github:

<https://github.com/tensorflow/models/tree/master/research/audioset/yamnet>

- yamnet\_fp16: official model used to predict 521 sounds
- yamnet\_s: self-trained model to predict 2 alarm sounds – CO, Smoke

## 14.3 NN result format

After NN model inference, the inference result will be store in NN output tensor. These output tensors should be decoded in post-processing. `vipnn_res_t` structure is used to store the post-processing result:

```
typedef struct vipnn_res_s {
 union {
 objdetect_res_t od_res; // for object detection
 };
 int type;
} vipnn_res_t;
```

#### Object detection

For the object detection result, the post-processing will parse the object position and probability from the output tensor, and fill the results to an `objdetect_res_t` structure:

```
#define MAX_DETECT_OBJ_NUM 30
typedef struct objdetect_res_s {
 int obj_num;
 union {
 float result[MAX_DETECT_OBJ_NUM * 6];
 detobj_t res[MAX_DETECT_OBJ_NUM];
 };
}
```

```

 };
} objdetect_res_t;

```

Description of parameter in objdetect\_res\_t:

- obj\_num: indicate the number of object detected in current frame.
- result: record the class\_index, probability and bounding box position for each object as format in Figure 14-1.
  - c: class\_index
  - p: probability
  - tx, ty, bx, by: bounding box(top\_x, top\_y, bottom\_x, bottom\_y)

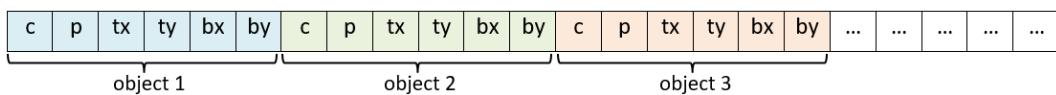


Figure 14-1 object detection format

## 14.4 NN memory evaluation

This section shows how to evaluate NN model size and DDR usage.

### 14.4.1 Evaluate memory usage of model

Please use Verisilicon\_SW\_VIP\_NBInfo tool to evaluate the ddr memory usage of the model on PC. Take yolov4-tiny for example, it requires at least 8MB ddr memory

```

Memory Info

Total Read Only Memory (bytes): 3737536
Total Command buffer (bytes): 167552
Total Load States (bytes): 34176
Total NN and TP instruction (bytes): 132864
Total PPU instruction (bytes): 512

Total Operation Memory (bytes): 3939264
Total Input Memory (bytes): 519168
Total Output Memory (bytes): 215552
Memory Pool (bytes): 2769920
Video memory heap node reserved (bytes): 20480

Total Video Memory (bytes): 7464448
Total System Memory (bytes): 247964

```

Therefore, we have to make sure the NN ddr region in link script is enough for this model.

Check and modify in "project\realtek\_ameapro2\_v0\_example\GCC-RELEASE\application\rtl8735b\_ram.ld

```
/* DDR memory */

VOE (rwx) : ORIGIN = 0x70000000, LENGTH = 0x70100000 /* 1MB */
DDR (rwx) : ORIGIN = 0x70100000, LENGTH = 0x73000000 /* 49MB */
NN (rwx) : ORIGIN = 0x73000000, LENGTH = 0x74000000 /* 16MB */
```

### 14.4.2 Evaluate model size

Please make sure the NN region in partition table is larger than your model size, so that the model can be downloaded to flash correctly. Take yolov4-tiny for example, the model size is 4MB



Figure 14-2 model network binary

The nn region length in "project\realtek\_ameapro2\_v0\_example\GCC-RELEASE\mp\ameapro2\_partitiontable.json" should not less than 4MB

```
"nn": {
 "start_addr" : "0x770000",
 "length" : "0x700000",
 "type": "PT_NN_MDL",
 "valid": true
},
```

## 14.5 Using the NN MMF example with VIPNN module

The NN example is a part of mmf video joined example. Please uncomment the example want to execute.

(project/realtek\_amebapro2\_v0\_example/src/mmfv2\_video\_example/video\_example\_media\_framework.c)

```
mmf2_video_example_vipnn_rtsp_init();
//mmf2_video_example_audio_vipnn_init();
```

### Current supported VIP NN examples

Table 14-2 NN examples

| Example                               | Description                                              | Result                                                                                                   |
|---------------------------------------|----------------------------------------------------------|----------------------------------------------------------------------------------------------------------|
| mmf2_video_example_vipnn_rtsp_init    | CH1 Video -> H264/HEVC -> RTSP<br>CH4 Video -> RGB -> NN | RTSP video stream over the network.<br>NN do object detection and draw the bounding box to RTSP channel. |
| mmf2_video_example_audio_vipnn_init.c | AUDIO -> NN                                              | The sound received by AmebaPro2 can be transmitted to NN engine to do sound classification.              |

### 14.5.1 Set RGB video resolution as model input size

If setting the RGB resolution according to NN model input tensor shape, it can avoid image resizing and save pre-processing time.

```
#define YOLO_MODEL 1
#define USE_NN_MODEL YOLO_MODEL
...
#if (USE_NN_MODEL==YOLO_MODEL)
#define NN_WIDTH 416
#define NN_HEIGHT 416
static float nn_confidence_thresh = 0.4;
static float nn_nms_thresh = 0.3;
#else
#error Please set model correctly. (YOLO_MODEL)
#endif
...
static video_params_t video_v4_params = {
 .stream_id = NN_CHANNEL,
 .type = NN_TYPE,
 .resolution = NN_RESOLUTION,
 .width = NN_WIDTH,
 .height = NN_HEIGHT,
 .bps = NN_BPS,
 .fps = NN_FPS,
 .gop = NN_GOP,
 .direct_output = 0,
 .use_static_addr = 1
};
```

### 14.5.2 Choose NN model

Please check the desired models are selected in amebapro2\_fwf\_nn\_models.json. For example, if we want to use yolov4\_tiny and YAMNet\_s: Go to “project/realtek\_amebapro2\_v0\_example/GCC-RELEASE/mp/amebapro2\_fwf\_nn\_models.json” and set model yolov4\_tiny - “MODEL0” and YAMNet\_s - “MODEL2” be used:

```
{
 "msg_level":3,
```

```

 "PROFILE": ["FWFS"],
 "FWFS": {
 "files": [
 "MODEL0",
 "MODEL2"
]
 },
 "MODEL0": {
 "name" : "yolov4_tiny.nb",
 "source": "binary",
 "file": "yolov4_tiny.nb"
 },
 "MODEL1": {
 "name" : "yamnet_fp16.nb",
 "source": "binary",
 "file": "yamnet_fp16.nb"
 },
 "MODEL2": {
 "name" : "yamnet_s.nb",
 "source": "binary",
 "file": "yamnet_s.nb"
 }
}

```

**NOTE**

After choosing the model, user have to check the ddr memory and flash size usage of models. Please refer 14.4.1 and 14.4.2 to do evaluation.

### 14.5.3 Build NN example

Since it's a part of video mmf example, user should use the following command to generate the makefile.

Generate the makefile for the NN project:

```
cmake .. -G"Unix Makefiles" -DCMAKE_TOOLCHAIN_FILE=../toolchain.cmake -DVVIDEO_EXAMPLE=ON
```

Then, use the following command to generate an image with NN model inside:

```
cmake --build . --target flash_nn
```

After running the command above, you will get the flash\_ntz.nn.bin (including the model) in “project\realtek\_amebapro2\_v0\_example\GCC-RELEASE\build”

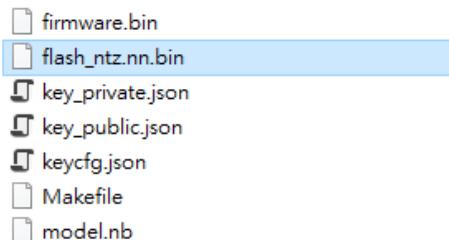


Figure 14-3 image with NN model

Then, use the image tool to download it to AmebaPro2.

### 14.5.4 Update NN model on flash

If user just want to update the NN model instead of updating whole firmware, the following command can be used to update NN section on flash partially:

**Nand flash**

```
$.\uartfburn.exe -p COM? -f .\flash_ntz.nn.bin -b 3000000 -n pro2 -t 0x81cf
```

**Nor flash**

```
$.\uartfburn.exe -p COM? -f .\flash_ntz.nn.bin -b 3000000 -t 0x81cf
```

## 14.5.5 Validate NN example

Refer the following section to validate nn examples.

### 14.5.5.1 Object detection example

While running the example, you may need to configure WiFi connection by using these commands in uart terminal.

```
ATW0=<WiFi_SSID> : Set the WiFi AP to be connected
ATW1=<WiFi_Password> : Set the WiFi AP password
ATWC : Initiate the connection
```

If everything works fine, you should see the following logs

```
...
[VOE]RGB3 640x480 1/5
[VOE]Start Mem Used ISP/ENC: 0 KB/ 0 KB Free= 701
hal_rtl_sys_get_clk 2
GCChipRev data = 8020
GCChipDate data = 20190925
queue 20121bd8 queue mutex 71691380
npu gck vip_drv_init, video memory heap base: 0x71B00000, size: 0x01300000
yuv in 0x714cee00
[VOE][process_rgb_yonly_irq][371]Errrrgb ddr frame count overflow : int_status 0x00000008
buf_status 0x00000010 time 15573511 cnt 0
input 0 dim 416 416 3 1, data format=2, quant_format=2, scale=0.003660, zero_point=0
output 0 dim 13 13 255 1, data format=2, scale=0.092055, zero_point=216
output 1 dim 26 26 255 1, data format=2, scale=0.093103, zero_point=216

input count 1, output count 2
input param 0
 data_format 2
 memory_type 0
 num_of_dims 4
 quant_format 2
 quant_data , scale=0.003660, zero_point=0
 sizes 1a0 1a0 3 1 0 0
output param 0
 data_format 2
 memory_type 0
 num_of_dims 4
 quant_format 2
 quant_data , scale=0.092055, zero_point=216
 sizes d d ff 1 0 0
output param 1
 data_format 2
 memory_type 0
 num_of_dims 4
 quant_format 2
 quant_data , scale=0.093103, zero_point=216
 sizes 1a 1a ff 1 0 0

in 0, size 416 416
VIPNN opened
siso_array_vipnn started
nn tick[0] = 47
object num = 0
nn tick[0] = 46
object num = 0
...
```

Then, open VLC and create a network stream with URL: rtsp://192.168.x.xx:554

If everything works fine, you should see the object detection result on VLC player.



Figure 14-4 VLC validation

#### 14.5.5.2 Audio classification example

If everything works fine, you should see the following logs

```
...
Deploy YAMNET_S
fci part tbl start 10
fci part tbl dup cnt 8
update page size 2048 page per block 64
type_name NN_MDL, file_name yamnet_s.nb
open: part_rec 7043d6a0, part_recs_cnt 1, type_id 81cf
file yamnet_s.nb, len 678336
network 70431540
input 0 dim 1 64 96 1, data format=1, quant_format=0, none-quant
output 0 dim 3 1 0 0, data format=1, none-quant

input count 1, output count 1
input param 0
 data_format 1
 memory_type 0
 num_of_dims 4
 quant_format 0
 quant_data , none-quant
 sizes 1 40 60 1 0 0
output param 0
 data_format 1
 memory_type 0
 num_of_dims 2
 quant_format 0
 quant_data , none-quant
 sizes 3 1 0 0 0 0

in 0, size 1 64
VIPNN opened
siso_audio_vipnn started
YAMNET_S tick[0] = 2
class 1, prob 1.00
YAMNET_S tick[0] = 2
class 1, prob 1.00
YAMNET_S tick[0] = 1
class 1, prob 1.00
YAMNET_S tick[0] = 1
class 1, prob 1.00
YAMNET_S tick[0] = 1
class 1, prob 1.00
...
...
```

User can use audio sample to validate the result. Use CO & smoke audio smaple in "project/realtek\_amebapro2\_v0\_example/src/test\_model/audio\_test\_sample" to verify the result.  
YAMNet\_s can recognize 3 audio classes:

- (1) class 0: CO
- (2) class 1: Others
- (3) class 2: Smoke

# 15 ISP

AmebaPro2 has a video engine integrating following function block: sensor driver, image signal processor, OSD and motion detection. The following chapters describe the software flow to setup AmebaPro2 ISP function.

## 15.1 Sensor Driver

AmebaPro2 supports single sensor input with ISP engine processing and output up to 4 channel with independent configuration. Based on AmebaPro2 Spec, ISP can support up to 5m resolution (but fps need to be calculated by bandwidth limitation). In this chapter, we will introduce sensor driver information such as sensor list and sensor configuration.

### 15.1.1 Sensor list

RTK will update support sensor driver regularly. User can contact RTK-FAE to get newest sensor list. The following table shows support list with sensor information.

Table 15-1 Support sensor list on AmebaPro2

| Vendor      | Sensor   | Max Resolution and FPS | Description |
|-------------|----------|------------------------|-------------|
| Galaxycore  | GC2053   | 1920x1080 * 30         |             |
| Galaxycore  | GC4653   | 2560x1440 * 15         |             |
| PrimeSensor | PS5258   | 1920x1080 * 30         |             |
| PrimeSensor | PS5270   | 1536x1536 * 30         |             |
| SmartSens   | SC2336   | 1920x1080 * 30         |             |
| SmartSens   | SC301IOT | 2048x1536 * 20         |             |
| SOI         | JXF37P   | 1920x1080 * 30         |             |
| SOI         | JXF51    | 1536x1536 * 30         |             |
| Sony        | IMX307   | 1920x1080 * 30         | With HDR    |
| Sony        | IMX327   | 1920x1080 * 30         | With HDR    |
| ImageDesign | MIS2008  | 1920x1080 * 30         |             |

### 15.1.2 Sensor configuration

User can check supported sensor list and sensor settings is in “project\realtek\_amebapro2\_v0\_example\inc\sensor.h”:

```
#define SENSOR_DUMMY 0x00
#define SENSOR_SC2336 0x01
#define SENSOR_GC2053 0x02
...
#define USE_SENSOR SENSOR_GC2053
```

For each sensor, AmebaPro2 need to configuration 3 binary file

- Fast Camera start flow binary file
- Sensor Driver binary file
- Image Quality binary file

User can check support sensor list and set sensor in “project\realtek\_amebapro2\_v0\_example\GCC-RELEASE\mp\ amebapro2\_sensor\_set.json”, and take GC2053 with index 02 for example

```
"ISP_SENSOR_SETS": {
 ...
 "sensor_sets": [
 "SENSOR_SET0",
 "SENSOR_SET1",
 "SENSOR_SET2", // GC2053 index in sensor.h
 "SENSOR_SET3",
 "SENSOR_SET4",
 "SENSOR_SET5"
],
 ...
 "SENSOR_SET2": { // binary for GC2053
 "merge_en": true,
 ...
 }
}
```

```

 "fcs_data": {
 "source": "binary",
 "file": "fcs_data_gc2053.bin"
 },
 "iq_data": {
 "source": "binary",
 "file": "iq_gc2053.bin"
 },
 "sensor_data": {
 "source": "binary",
 "file": "sensor_gc2053.bin"
 }
},

```

### 15.1.3 How to add a new sensor

Customer need to contact with RTK-FAE for new sensor support. After RTK-RD successfully brings up sensor with general IQ, RTK will prepare integrated patch based on customer SDK version, and this may take 3~4 weeks working day.

## 15.2 Image Quality

User can bring up sensor with basic image quality on RTK-EVT or user's DUT. Image Quality will vary based on selected lens and optical structure. For different application, end customer will also have different image quality criteria. Based on each project, user can check subjective and objective image quality criteria or compare with target DUT. For advanced image quality tuning support, user can contact RTK-FAE.

### 15.2.1 Use UVC Example

AmebaPro2 ISP can support compressed (H264 / H265 / JPG) and uncompressed (NV16 / NV12) image through UVC (wired transmission), and user can check video on pc with Potplayer, Amcap or RTK-realcam. For uncompressed format, user need to install RTK decoder to get video on computer. User can use following flow to build UVC example. Generate the makefile for the UVC project:

```
cmake .. -G"Unix Makefiles" -DCMAKE_TOOLCHAIN_FILE=../toolchain.cmake -DVVIDEO_EXAMPLE=ON
```

Then, use the following command to generate an image:

```
cmake --build . --target flash
```

### 15.2.2 Image Quality Criteria

For first draft image quality version, RTK will provide image quality patch following RTK criteria. User can check table for detailed.

Table 15-2 Objective image quality on AmebaPro2

| Category           | Condition     | Criteria                        |
|--------------------|---------------|---------------------------------|
| Lens Shading       | D65 & CWF & A | Relative illumination > 80%     |
|                    | D65 & CWF & A | R/G [0.9~1.1]                   |
|                    | D65 & CWF & A | B/G [0.9~1.1]                   |
|                    | D65 & CWF & A | B/G [0.9~1.1]                   |
| Color Checker      | D65 & CWF & A | Saturation [100%~120%]          |
|                    | D65 & CWF & A | Mean $\Delta C \leq 10$         |
|                    | D65 & CWF & A | Max $\Delta C \leq 30$          |
|                    | D65 & CWF & A | Mean $\Delta E \leq 20$         |
|                    | D65 & CWF & A | Max $\Delta E \leq 30$          |
| Auto White Balance | D65 & CWF & A | #20~#23 Max $\Delta S \leq 0.1$ |
| Resolution (1080P) | D65           | Center Horizontal: $\geq 1000$  |
|                    |               | Center Vertical: $\geq 1000$    |
|                    |               | Corner Horizontal $\geq 600$    |
|                    |               | Corner Vertical $\geq 600$      |
| Dynamic Range      | D65           | Max Y $\geq 200$                |
|                    |               | Step $\geq 14$                  |

|              |              |      |
|--------------|--------------|------|
| Defect Pixel | Dark & White | None |
|--------------|--------------|------|

For advanced image quality such as customized objective image criteria or quality benchmark with target DUT, user can contact with RTK-FAE for tuning support.

## 15.3 OSD2

### 15.3.1 OSD2 introduction

The text image display consists of hardware maps, drivers and provided Libs. Users use the provided API to create instances, set alphanumeric and image properties, and place alphanumerics or images on streaming images. Users can replace different font files, and then provide enough memory for the OSD to convert alphanumerics into color images according to the font size and text length. The input and output image buffers here need to be physically continuous memory.

### 15.3.2 Configuration

- Display image
- Display alphanumeric
- Display date and time
- Alphanumeric rotation, stroke, transparency
- Font library capability;
  - Supports up to 3 sets of different fonts
  - Each font group supports up to 1 single-character glyph file and 1 double-character glyph file
- Each stream can display up to 6 sets of OSD Block

**NOTE**

*The starting address of the image Array must be 16Byte align with the Hardware DMA limit*

### 15.3.3 OSD example

OSD example is included in RTSP (-DVVIDEO\_EXAMPLE=ON) and UVCD (-DEXAMPLE=media\_uvcd) examples, and it is located at the path "component\video\osd2\isp\_osd\_example.c"

Take UVCD for example, before building the firmware, run below command to create the makefile.

```
cmake .. -G"Unix Makefiles" -DCMAKE_TOOLCHAIN_FILE=../toolchain.cmake -DCUTVER=B -
DEXAMPLE=media_uvcd
```

- Execution and testing
  - Open the ISP AT command in platform\_otps.h:

```
#define CONFIG_ISP 1
```

  - Build code and load image.
  - Connect the USB cable to the AmebaPro2 CON port and the other end to the PC.
  - Open potplayer, enter atcmd "ATIO=task,0,0,28,56" will show results.
    - ◆ Command parameters:"ATIO=task,0,0(channel),28(char\_width),56(char\_height)"

### 15.3.4 OSD Enumerations and Data Structures

In this chapter, we will list OSD API with function parameter introduction.

#### 15.3.4.1 OSD Data Structures

Table 15-3 OSD data structure

| Data Structures    | Introduction                 |
|--------------------|------------------------------|
| <osd_text_info_st> | Text type OSD parameters.    |
| <rt_font_st>       | Fonts parameters             |
| <osd_pict_st>      | Picture type OSD parameters. |
| <rt_osd2_info_st>  | OSD parameters.              |

Table 15-4 OSD data structure: osd\_text\_info\_st

| Parameter | Type       | Introduction                                |
|-----------|------------|---------------------------------------------|
| <chn_id>  | int        | Channel ID: 0~2                             |
| <blk_idx> | int        | Block index: 0~5                            |
| <font>    | rt_font_st | Please refers to table of rt_font_st        |
| <start_x> | uint32_t   | x-coordinate of start point                 |
| <start_y> | uint32_t   | y-coordinate of start point                 |
| <rotate>  | uint32_t   | Please refers to enumeration of rt_rotate_t |
| <str>     | char *     | String content                              |

Table 15-5 OSD data structure: rt\_font\_st

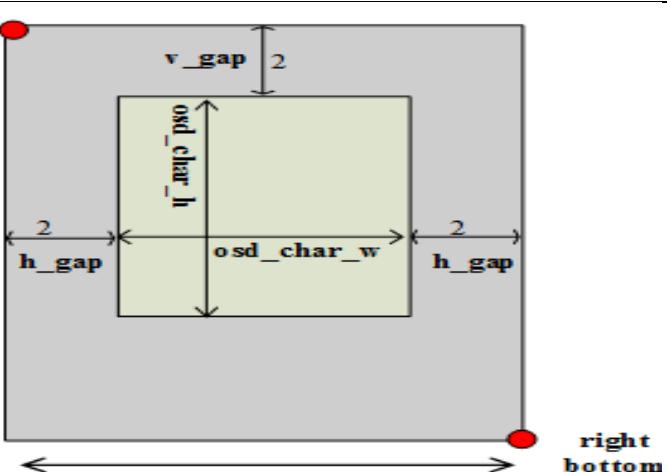
| Parameter     | Type             | Introduction                                                                                                                                                                                                                           |
|---------------|------------------|----------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------|
| <block_alpha> | uint8_t          | Transparent value: 0~15.                                                                                                                                                                                                               |
| <ch_color>    | uint32_t         | Character color in RGB.                                                                                                                                                                                                                |
| <bg_enable>   | uint8_t          | Enable background: 0~1.                                                                                                                                                                                                                |
| <bg_color>    | uint32_t         | Background color in RGB.                                                                                                                                                                                                               |
| <h_gap>       | uint8_t :4       |  <p>The meaning of the field in the osd structure is shown in the figure. The horizontal interval is h_gap, and the vertical interval is v_gap.</p> |
| <v_gap>       | uint8_t :4       |                                                                                                                                                                                                                                        |
| <time_fmt>    | rts_osd_time_fmt | Time format, please refer to introduction of rts_osd_time_fmt                                                                                                                                                                          |
| <date_fmt>    | rts_osd_date_fmt | Date format, please refer to introduction of rts_osd_date_fmt                                                                                                                                                                          |

Table 15-6 OSD data structure: osd\_pict\_st

| Parameter | Type            | Introduction    |
|-----------|-----------------|-----------------|
| <chn_id>  | int             | Channel ID: 0~2 |
| <osd2>    | rt_osd2_info_st | OSD parameters. |

Table 15-7 OSD data structure: rt\_osd2\_info\_st

| Parameter    | Type             | Introduction                                                |
|--------------|------------------|-------------------------------------------------------------|
| <blk_idx>    | int              | Block index: 0~5                                            |
| <blk_fmt>    | rts_osd2_blk_fmt | Block format: Please refers to enumeration rts_osd2_blk_fmt |
| <start_x>    | uint32_t         | x-coordinate of start point                                 |
| <start_y>    | uint32_t         | y-coordinate of start point                                 |
| <end_x>      | uint32_t         | x-coordinate of end point                                   |
| <end_y>      | uint32_t         | y-coordinate of end point                                   |
| <color_1bpp> | uint32_t         | Set the RGB color when format is RTS OSD2 BLK_FMT_1BPP      |
| <buf>        | uint8_t *        | Image buffer                                                |
| <len>        | uint32_t         | Image buffer length                                         |

**15.3.4.2 OSD Enumerations**

Table 15-8 OSD Enumerations

| Enumerations       | Introduction                              |
|--------------------|-------------------------------------------|
| <rt_rotate_t>      | Rotation angle, include 0, 90, 180, 270.. |
| <rts_osd_time_fmt> | Time format                               |
| <rts_osd_date_fmt> | Date format                               |
| <rts_osd2_blk_fmt> | Block format                              |

Table 15-9 OSD data structure: rt\_rotate\_t

| Definition       | Introduction                   |
|------------------|--------------------------------|
| <RT_ROTATE_0>    | None rotation                  |
| <RT_ROTATE_90R>  | Rotate 90 degree to the right  |
| <RT_ROTATE_180R> | Rotate 180 degree to the right |
| <RT_ROTATE_270R> | Rotate 270 degree to the right |
| <RT_ROTATE_90L>  | Rotate 90 degree to the left   |
| <RT_ROTATE_180L> | Rotate 180 degree to the left  |
| <RT_ROTATE_270L> | Rotate 270 degree to the left  |

Table 15-10 OSD data structure: rts\_osd\_time\_fmt

| Definition          | Type             | Introduction     |
|---------------------|------------------|------------------|
| <osd_time_fmt_no>   | Not display time | Not display time |
| <osd_time_fmt_24>   | hh:mm:ss         | 14:32:58         |
| <osd_time_fmt_12>   | hh:mm:ss         | 02:32:58         |
| <osd_time_fmt_12_1> | Phh:mm:ss        | P02:32:58        |
| <osd_time_fmt_12_2> | PMhh:mm:ss       | PM02:32:58       |
| <osd_time_fmt_12_3> | PM~hh:mm:ss      | PM~02:32:58      |
| <osd_time_fmt_12_4> | hh:mm:ssPM       | 02:32:58PM       |
| <osd_time_fmt_12_5> | hh:mm:ss~PM      | 02:32:58~PM      |
| <osd_time_fmt_12_6> | hh:mm:ss~~PM     | 02:32:58~~PM     |
| <osd_time_fmt_12_7> | hh:mm:ss~~~PM    | 02:32:58~~~PM    |

Table 15-11 OSD data structure: rts\_osd\_date\_fmt

| Definition        | Type             | Example          |
|-------------------|------------------|------------------|
| <osd_date_fmt_no> | Not display date | Not display date |
| <osd_date_fmt_0>  | dd/MM/yyyy       | 26/05/2015       |
| <osd_date_fmt_1>  | dd/MM/yy         | 26/05/15         |
| <osd_date_fmt_2>  | d/M/yy           | 26/5/15          |
| <osd_date_fmt_3>  | M/d/yyyy         | 5/26/2015        |
| <osd_date_fmt_4>  | M/d/yy           | 5/26/15          |
| <osd_date_fmt_5>  | MM/dd/yy         | 05/26/15         |
| <osd_date_fmt_6>  | MM/dd/yyyy       | 05/26/2015       |
| <osd_date_fmt_7>  | yyyy/M/d         | 2015/5/26        |
| <osd_date_fmt_8>  | yyyy-M-d         | 2015-5-26        |
| <osd_date_fmt_9>  | yyyy-MM-dd       | 2015-05-26       |
| <osd_date_fmt_10> | yyyy/MM/dd       | 2015/05/26       |
| <osd_date_fmt_11> | yy-MM-dd         | 15-05-26         |
| <osd_date_fmt_12> | yy/M/d           | 15/5/26          |
| <osd_date_fmt_13> | yy-M-d           | 15-5-26          |
| <osd_date_fmt_14> | yy/MM/dd         | 15/05/26         |
| <osd_date_fmt_15> | yyyy.mm.dd       | 2015.05.26       |
| <osd_date_fmt_16> | dd.mm.yyyy       | 26.05.2015       |
| <osd_date_fmt_17> | mm.dd.yyyy       | 05.26.2015       |
| <osd_date_fmt_18> | mm-dd-yyyy       | 05-26-2015       |
| <osd_date_fmt_19> | dd-mm-yyyy       | 26-05-2015       |
| <osd_date_fmt_20> | dd-mm-yyyy www   | 26-05-2015 Tue   |

|                   |                |                |
|-------------------|----------------|----------------|
| <osd_date_fmt_21> | dd/mm/yyyy www | 26/05/2015 Tue |
| <osd_date_fmt_22> | dd.mm.yyyy www | 26.05.2015 Tue |

Table 15-12 OSD data structure: rts\_osd2\_blk\_fmt

| Definition                  | Introduction                             |
|-----------------------------|------------------------------------------|
| <RTS OSD2_BLK_FMT_1BPP>     | Format in 1BPP, pixel size: 1 bit.       |
| <RTS OSD2_BLK_FMT_RGBA1111> | Format in RGBA1111, pixel size: 4 bit.   |
| <RTS OSD2_BLK_FMT_RGBA2222> | Format in RGBA2222, pixel size: 1 byte.  |
| <RTS OSD2_BLK_FMT_RGBA5551> | Format in RGBA5551, pixel size: 2 bytes. |
| <RTS OSD2_BLK_FMT_RGBA4444> | Format in RGBA4444, pixel size: 2 bytes. |
| <RTS OSD2_BLK_FMT_RGBA8888> | Format in RGBA8888, pixel size: 4bytes.  |

If the block type is rts\_osd2\_type\_date, rts\_osd2\_type\_time or rts\_osd2\_type\_text, block format is always RGBA1111. If the block type is rts\_osd2\_type\_pict, below all block format are supported.

## 15.3.5 OSD API

### 15.3.5.1 rts\_osd\_init

Initial function is used to create OSD data, font lib and set the time-zone for the indicated stream.

Table 15-13 OSD API: rts\_osd\_init

| Parameter       | Type | Introduction                |
|-----------------|------|-----------------------------|
| <chn_id>        | int  | Stream channel ID.          |
| <char_resize_w> | int  | Character size in width.    |
|                 | int  | Character size in height.   |
| <timezone_s>    | int  | Time-zone, unit in seconds. |
| <chn_id>        | int  | Stream channel ID.          |

### 15.3.5.2 rts\_osd\_deinit

De-initialize the OSD data of indicated stream.

Table 15-14 OSD API: rts\_osd\_deinit

| Parameter | Type | Introduction       |
|-----------|------|--------------------|
| <chn_id>  | int  | Stream channel ID. |

### 15.3.5.3 rts\_osd\_set\_info

It sets OSD data of indicated stream and block. Each video stream has a separate OSD module. Each OSD module supports up to 6 blocks, a block is an area in the image for displaying characters or images, which represented by the structure “osd\_text\_info\_st” or “osd\_pict\_st”. English and digital width of a word are inconsistent with Chinese in display. English and array use a single, the width and font files are saved in the single font file. The Chinese display takes up double width, and the font file is saved in the double wide font file. For the detail of “osd\_text\_info\_st” and “osd\_pict\_st”, refer to previous instructions

Table 15-15 OSD API: rts\_osd\_set\_info

| Parameter  | Type   | Introduction                                                                                                                                                                                                                                                                       |
|------------|--------|------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------|
| <osd_type> | int    | Types include <ul style="list-style-type: none"> <li>● rts_osd2_type_date,</li> <li>● rts_osd2_type_time,</li> <li>● rts_osd2_type_pict,</li> <li>● rts_osd2_type_text.</li> </ul>                                                                                                 |
| <osd_info> | void * | Block detail description, which includes “osd_text_info_st*” and “osd_pict_st*”. <ul style="list-style-type: none"> <li>● “osd_text_info_st*” includes rts_osd2_type_date, rts_osd2_type_time, rts_osd2_type_text</li> <li>● “osd_pict_st*” includes rts_osd2_type_pict</li> </ul> |

**15.3.5.4 rts\_osd\_get\_timezone**

Get the time-zone.

Parameter: None.

**15.3.5.5 rts\_osd\_set\_timezone**

Set the time-zone.

Table 15-16 OSD API: rts\_osd\_set\_timezone

| Parameter    | Type | Introduction                       |
|--------------|------|------------------------------------|
| <timezone_s> | int  | The value of time-zone in seconds. |

**15.3.5.6 rts\_osd\_isp\_refresh\_datetime**

Refresh date-time. All stream use the same date-time information.

Parameter: None.

**15.3.5.7 rts\_osd\_block\_hide**

The function used to hide the indicated block.

Table 15-17 OSD API: rts\_osd\_block\_hide

| Parameter    | Type | Introduction                       |
|--------------|------|------------------------------------|
| <timezone_s> | int  | The value of time-zone in seconds. |

**15.3.5.8 rts\_osd\_block\_show**

The function used to show the indicated block.

Table 15-18 OSD API: rts\_osd\_block\_show

| Parameter | Type | Introduction     |
|-----------|------|------------------|
| <chn_id>  | int  | Channel ID: 0~2  |
| <idx>     | int  | Block index: 0~5 |

**15.3.5.9 rts\_set\_char\_size**

This function used to change character size dynamically.

Table 15-19 OSD API: rts\_set\_char\_size

| Parameter       | Type | Introduction              |
|-----------------|------|---------------------------|
| <chn_id>        | Int  | Stream channel ID.        |
| <char_resize_w> | Int  | Character size in width.  |
| <char_resize_h> | int  | Character size in height. |

**15.3.5.10 rts\_osd\_task**

OSD task function.

Parameter: None.

**NOTE**

Please use *xTaskCreate* to create the task.

## 15.4 ISP Control API

In this chapter, we list all ISP control API at application layer. User can use these API to do customized image tuning. For all API, we divide them into 5 category: AE, AWB, image tuning, mode, WDR and dehaze. And we will also show an example to evaluate lens through ISP API.

## 15.4.1 ISP Control API (AE)

### 15.4.1.1 isp\_set\_exposure\_mode

Table 15-20 ISP API: isp\_set\_exposure\_mode

| Parameter | Type | Introduction                                                    |
|-----------|------|-----------------------------------------------------------------|
| <val>     | int  | The mode of exposure, value is 0 or 1.<br>(0: manual, 1: Auto). |

### 15.4.1.2 isp\_get\_exposure\_mode

Table 15-21 ISP API: isp\_get\_exposure\_mode

| Parameter | Type  | Introduction                                                            |
|-----------|-------|-------------------------------------------------------------------------|
| <pval>    | Int * | Retrieve the mode of exposure, value is 0 or 1.<br>(0: manual, 1: auto) |

### 15.4.1.3 isp\_set\_power\_line\_freq

Table 15-22 ISP API: isp\_set\_power\_line\_freq

| Parameter | Type | Introduction                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                               |
|-----------|------|--------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------|
| <val>     | int  | <p>Anti-flicker mode.<br/>Range: 0 ~ 3<br/>0: Disable, 1: 50Hz, 2: 60Hz, 3: Auto</p> <p>Remark:</p> <ul style="list-style-type: none"> <li>1. Auto mode:<br/>(A) Auto mode include the algorithm of flicker detection, the detection fail rate might result in flicker problem.</li> <li>2. 50 Hz<br/>(A) The lowest exposure time to stop flicker is 10ms. If lower, flicker might happen.<br/>(B) IQ parameters can hold the flicker off, but side effect is over exposure under high brightness environment.<br/>(C) If IQ parameters cannot stop flicker, some FPS settings, such as 25, 20, or 10 can stop the moving. (Banding still exist.)</li> <li>3. 60 Hz<br/>(A) The lowest exposure time to stop flicker is 8.33ms. If lower, flicker might happen.<br/>(B) IQ parameters can hold the flicker off, but side effect is over exposure under high brightness environment.<br/>(C) If IQ parameters cannot stop flicker, some FPS settings, such as 30, 24, 20, 15, or 12 can stop the moving. (Banding still exist.)</li> </ul> |

### 15.4.1.4 isp\_get\_power\_line\_freq

Table 15-23 ISP API: isp\_get\_power\_line\_freq

| Parameter | Type | Introduction                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                               |
|-----------|------|----------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------|
| <val>     | int  | <p>Anti-flicker mode.<br/>Range: 0 ~ 3<br/>0: Disable, 1: 50Hz, 2: 60Hz, 3: Auto</p> <p>Remark:</p> <ul style="list-style-type: none"> <li>1. Auto mode:<br/>(A) Auto mode include the algorithm of flicker detection, the detection fail rate might result in flicker problem.</li> <li>2. 50 Hz<br/>(A) The lowest exposure time to stop flicker is 10ms. If lower, flicker might happen.<br/>(B) IQ parameters can hold the flicker off, but side effect is over exposure under high brightness environment.</li> </ul> |

|  |  |                                                                                                                                                                                                                                                                                                                                                                                                                                                                                                    |
|--|--|----------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------|
|  |  | (C) If IQ parameters cannot stop flicker, some FPS settings, such as 25, 20, or 10 can stop the moving. (Banding still exist.)<br>3. 60 Hz<br>(A) The lowest exposure time to stop flicker is 8.33ms. If lower, flicker might happen.<br>(B) IQ parameters can hold the flicker off, but side effect is over exposure under high brightness environment.<br>(C) If IQ parameters cannot stop flicker, some FPS settings, such as 30, 24, 20, 15, or 12 can stop the moving. (Banding still exist.) |
|--|--|----------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------|

#### 15.4.1.5 isp\_set\_exposure\_time

Table 15-24 ISP API: isp\_set\_exposure\_time

| Parameter | Type  | Introduction                                       |
|-----------|-------|----------------------------------------------------|
| <pval>    | Int * | Retrieve the mode of Anti-flicker.<br>Range: 0 ~ 3 |

#### 15.4.1.6 isp\_get\_exposure\_time

Table 15-25 ISP API: isp\_get\_exposure\_time

| Parameter | Type | Introduction                                                                          |
|-----------|------|---------------------------------------------------------------------------------------|
| <val>     | int  | The exposure time, unit is us.<br>Range is 1~100,000.<br>Adjustable precision is +-1. |

#### 15.4.1.7 isp\_set\_ae\_gain

Table 15-26 ISP API: isp\_set\_ae\_gain

| Parameter | Type | Introduction                                                 |
|-----------|------|--------------------------------------------------------------|
| <val>     | int  | Gain value.<br>Range: 256~32768<br>Adjustable precision: +-1 |

#### 15.4.1.8 isp\_get\_ae\_gain

Table 15-27 ISP API: isp\_get\_ae\_gain

| Parameter | Type  | Introduction                            |
|-----------|-------|-----------------------------------------|
| <pval>    | Int * | Retrieve gain value<br>Range: 256~32768 |

### 15.4.2 ISP Control API (AWB)

#### 15.4.2.1 isp\_set\_awb\_ctrl

Table 15-28 ISP API: isp\_set\_awb\_ctrl

| Parameter | Type | Introduction                                              |
|-----------|------|-----------------------------------------------------------|
| <val>     | int  | Mode of white balance.<br>0: Manual temperature, 1: Auto. |

**NOTE**

The API of manual temperature is not supported.

#### 15.4.2.2 isp\_get\_awb\_ctrl

Table 15-29 ISP API: isp\_get\_awb\_ctrl

| Parameter | Type  | Introduction                                               |
|-----------|-------|------------------------------------------------------------|
| <pval>    | Int * | Retrieve the mode of white balance.<br>0: Manual, 1: Auto. |

#### 15.4.2.3 isp\_set\_wb\_temperature

Table 15-30 ISP API: isp\_set\_wb\_temperature

| Parameter | Type | Introduction                                                                 |
|-----------|------|------------------------------------------------------------------------------|
| <val>     | int  | white balance temperature<br>Range: 1000~10000.<br>Adjustable precision: +-1 |

#### 15.4.2.4 isp\_get\_wb\_temperature

Table 15-31 ISP API: isp\_get\_wb\_temperature

| Parameter | Type  | Introduction                                   |
|-----------|-------|------------------------------------------------|
| <pval>    | Int * | Retrieve the current white balance temperature |

#### 15.4.2.5 isp\_set\_red\_balance

Table 15-32 ISP API: isp\_set\_red\_balance

| Parameter | Type | Introduction                                                                      |
|-----------|------|-----------------------------------------------------------------------------------|
| <val>     | int  | Red balance value based on 256.<br>Range: 256~2047.<br>Adjustable precision: +-1. |

#### 15.4.2.6 isp\_get\_red\_balance

Table 15-33 ISP API: isp\_get\_red\_balance

| Parameter | Type  | Introduction                    |
|-----------|-------|---------------------------------|
| <pval>    | Int * | Retrieve the red balance value. |

#### 15.4.2.7 isp\_set\_green\_balance

Table 15-34 ISP API: isp\_set\_green\_balance

| Parameter | Type | Introduction                                                                       |
|-----------|------|------------------------------------------------------------------------------------|
| <val>     | int  | Green balance value based on 256.<br>Range: 256~2047.<br>Adjustable precision: +-1 |

NOTE

Usually this value is set 256 as default.

#### 15.4.2.8 isp\_get\_green\_balance

Table 15-35 ISP API: isp\_get\_green\_balance

| Parameter | Type  | Introduction                     |
|-----------|-------|----------------------------------|
| <pval>    | Int * | Retrieve the green balance value |

#### 15.4.2.9 isp\_set\_blue\_balance

Table 15-36 ISP API: isp\_set\_blue\_balance

| Parameter | Type  | Introduction |
|-----------|-------|--------------|
| <pval>    | Int * |              |

|       |     |                                                                                    |
|-------|-----|------------------------------------------------------------------------------------|
| <val> | int | Blue balance value based on 256.<br>Range: 256~2047.<br>Adjustable precision: +-1. |
|-------|-----|------------------------------------------------------------------------------------|

#### 15.4.2.10 isp\_get\_blue\_balance

Table 15-37 ISP API: isp\_get\_blue\_balance

| Parameter | Type  | Introduction                     |
|-----------|-------|----------------------------------|
| <pval>    | Int * | Retrieve the blue balance value. |

### 15.4.3 ISP Control API (Image Tuning)

#### 15.4.3.1 isp\_set\_brightness

Table 15-38 ISP API: isp\_set\_brightness

| Parameter | Type | Introduction                                                                          |
|-----------|------|---------------------------------------------------------------------------------------|
| <val>     | int  | The brightness value of the image.<br>Range: -64 to 64.<br>Adjustable precision: +-1. |

#### 15.4.3.2 isp\_get\_brightness

Table 15-39 ISP API: isp\_get\_brightness

| Parameter | Type  | Introduction                                                 |
|-----------|-------|--------------------------------------------------------------|
| <pval>    | int * | Retrieves the current brightness value.<br>Range: -64 to 64. |

#### 15.4.3.3 isp\_set\_contrast

Table 15-40 ISP API: isp\_set\_contrast

| Parameter | Type | Introduction                                                           |
|-----------|------|------------------------------------------------------------------------|
| <val>     | int  | image contrast value.<br>Range: 0~100.<br>Adjustable precision is +-1. |

#### 15.4.3.4 isp\_get\_contrast

Table 15-41 ISP API: isp\_get\_contrast

| Parameter | Type  | Introduction                                     |
|-----------|-------|--------------------------------------------------|
| <pval>    | Int * | Get the current contrast value.<br>Range: 0~100. |

#### 15.4.3.5 isp\_set\_saturation

Table 15-42 ISP API: isp\_set\_saturation

| Parameter | Type | Introduction                                                    |
|-----------|------|-----------------------------------------------------------------|
| <val>     | int  | ISP saturation.<br>Range: 0 to 100.<br>Adjustable accuracy: +1. |

**15.4.3.6 isp\_get\_saturation**

Table 15-43 ISP API: isp\_get\_saturation

| Parameter | Type  | Introduction                                    |
|-----------|-------|-------------------------------------------------|
| <pval>    | Int * | Get the current saturation.<br>Range: 0 to 100. |

**15.4.3.7 isp\_set\_gamma**

Table 15-44 ISP API: isp\_set\_gamma

| Parameter | Type | Introduction                                                        |
|-----------|------|---------------------------------------------------------------------|
| <val>     | int  | Gamma coefficient.<br>Range: 100~500.<br>Adjustable precision: +-1. |

**15.4.3.8 isp\_get\_gamma**

Table 15-45 ISP API: isp\_get\_gamma

| Parameter | Type  | Introduction                                         |
|-----------|-------|------------------------------------------------------|
| <pval>    | Int * | Retrieve the current Gamma coefficient from 100~500. |

**15.4.3.9 isp\_set\_sharpness**

Table 15-46 ISP API: isp\_set\_sharpness

| Parameter | Type | Introduction                                                   |
|-----------|------|----------------------------------------------------------------|
| <val>     | int  | Sharpness of isp<br>Range: 0~100.<br>Adjustable precision: +-1 |

**15.4.3.10 isp\_get\_sharpness**

Table 15-47 ISP API: isp\_get\_sharpness

| Parameter | Type  | Introduction                                   |
|-----------|-------|------------------------------------------------|
| <pval>    | Int * | Retrieve the current sharp value from 0 to 100 |

**15.4.3.11 isp\_set\_denoise\_level**

Table 15-48 ISP API: isp\_set\_denoise\_level

| Parameter | Type | Introduction                                                             |
|-----------|------|--------------------------------------------------------------------------|
| <val>     | int  | The level of noise reduction.<br>Range: 0~8<br>Adjustable precision: +-1 |

**15.4.3.12 isp\_get\_denoise\_level**

Table 15-49 ISP API: isp\_get\_denoise\_level

| Parameter | Type  | Introduction                                         |
|-----------|-------|------------------------------------------------------|
| <pval>    | Int * | Retrieve the level of noise reduction.<br>Range: 0~8 |

## 15.4.4 ISP Control API (Mode)

### 15.4.4.1 isp\_set\_iq\_table

Table 15-50 ISP API: isp\_set\_iq\_table

| Parameter | Type | Introduction                                                    |
|-----------|------|-----------------------------------------------------------------|
| <val>     | int  | Index of IQ-table.<br>Range: 0 ~ 3<br>Adjustable precision: +-1 |

### 15.4.4.2 isp\_get\_iq\_table

Table 15-51 ISP API: isp\_get\_iq\_table

| Parameter | Type  | Introduction                         |
|-----------|-------|--------------------------------------|
| <pval>    | Int * | Retrieve the index value of IQ-table |

### 15.4.4.3 isp\_set\_day\_night

Table 15-52 ISP API: isp\_set\_day\_night

| Parameter | Type | Introduction                                            |
|-----------|------|---------------------------------------------------------|
| <val>     | int  | The value of day/night mode. 0: day mode, 1: night mode |

### 15.4.4.4 isp\_get\_day\_night

Table 15-53 ISP API: isp\_get\_day\_night

| Parameter | Type  | Introduction                                                        |
|-----------|-------|---------------------------------------------------------------------|
| <pval>    | Int * | Retrieve the value of day/night mode.<br>0: day mode, 1: night mode |

### 15.4.4.5 isp\_set\_gray\_mode

Table 15-54 ISP API: isp\_set\_gray\_mode

| Parameter | Type | Introduction                                                 |
|-----------|------|--------------------------------------------------------------|
| <val>     | int  | The value of gray/color mode.<br>0: color mode, 1: gray mode |

### 15.4.4.6 isp\_get\_gray\_mode

Table 15-55 ISP API: isp\_get\_gray\_mode

| Parameter | Type  | Introduction                                                           |
|-----------|-------|------------------------------------------------------------------------|
| <pval>    | Int * | Retrieve the value of gray/color mode.<br>0: color mode , 1: gray mode |

## 15.4.5 ISP Control API (WDR)

### 15.4.5.1 isp\_set\_wdr\_mode

Table 15-56 ISP API: isp\_set\_wdr\_mode

| Parameter | Type | Introduction                                                |
|-----------|------|-------------------------------------------------------------|
| <val>     | int  | WDR mode.<br>Range: 0 ~ 2<br>0: Disable, 1: Manual, 2: Auto |

**15.4.5.2 isp\_get\_wdr\_mode**

Table 15-57 ISP API: isp\_get\_wdr\_mode

| Parameter | Type  | Introduction                                    |
|-----------|-------|-------------------------------------------------|
| <pval>    | Int * | Retrieve the value of WDR mode.<br>Range: 0 ~ 2 |

**15.4.5.3 isp\_set\_wdr\_level**

Table 15-58 ISP API: isp\_set\_wdr\_level

| Parameter | Type | Introduction                                             |
|-----------|------|----------------------------------------------------------|
| <val>     | int  | WDR level.<br>Range: 0~100.<br>Adjustable precision: +-1 |

**15.4.5.4 isp\_get\_wdr\_level**

Table 15-59 ISP API: isp\_get\_wdr\_level

| Parameter | Type  | Introduction                                      |
|-----------|-------|---------------------------------------------------|
| <pval>    | Int * | Retrieve the value of WDR level.<br>Range: 0~100. |

**15.4.6 ISP Control API (Dehaze)****15.4.6.1 isp\_set\_dehaze**

Table 15-60 ISP API: isp\_set\_dehaze

| Parameter | Type | Introduction                                               |
|-----------|------|------------------------------------------------------------|
| <val>     | int  | The value of enable/disable mode.<br>0: disable, 1: enable |

**15.4.6.2 isp\_get\_dehaze**

Table 15-61 ISP API: isp\_get\_dehaze

| Parameter | Type  | Introduction                                                         |
|-----------|-------|----------------------------------------------------------------------|
| <pval>    | Int * | Retrieve the value of enable/disable mode.<br>0: disable , 1: enable |

**15.4.6.3 isp\_set\_dehaze\_level**

Table 15-62 ISP API: isp\_set\_dehaze\_level

| Parameter | Type | Introduction                                                      |
|-----------|------|-------------------------------------------------------------------|
| <val>     | int  | The level of dehaze.<br>Range: 0~255<br>Adjustable precision: +-1 |

**15.4.6.4 isp\_get\_dehaze\_level**

Table 15-63 ISP API: isp\_get\_dehaze\_level

| Parameter | Type  | Introduction                                  |
|-----------|-------|-----------------------------------------------|
| <pval>    | Int * | Retrieve the level of dehaze.<br>Range: 0~255 |

### 15.4.7 Lens evaluation

For lens performance evaluation, user may need to configuration isp. And we have prepared quick start guide.

Table 15-64 ISP API: Lens evaluation flow

| ISP API               | Description         | Flow                                                               |
|-----------------------|---------------------|--------------------------------------------------------------------|
| isp_set_exposure_mode | 0: Manual, 1:Auto   |                                                                    |
| isp_set_awb_ctrl      | 0: Manual, 1:Auto   |                                                                    |
| isp_get_exposure_time | Exposure (unit: us) |                                                                    |
| isp_get_ae_gain       | Gain (unit: 256=1x) |                                                                    |
| isp_get_red_balance   | Gain (unit: 256=1x) |                                                                    |
| isp_get_blue_balance  | Gain (unit: 256=1x) |                                                                    |
| isp_set_exposure_time | Exposure (unit: us) |                                                                    |
| isp_set_ae_gain       | Gain (unit: 256=1x) | For golden lens, use auto mode to get AE & AWB information.        |
| isp_set_red_balance   | Gain (unit: 256=1x) |                                                                    |
| isp_set_blue_balance  | Gain (unit: 256=1x) |                                                                    |
|                       |                     | For competitor lens, use manual mode and set AE & AWB information. |

## 15.5 Motion Detection

Motion detection is achieved by computing YRGB information.

### 15.5.1 Motion Detection Architecture

Motion detection architecture is shown in Figure 15-5. First, it will confirm whether the value of the automatic exposure (AE) is stable. After stabilization, get 16x16 YRGB value to initialize the background model. The 16x16 YRGB value is calculated by averaging the RGB image generated by from video channel 4. After initialization, calculate the difference between the YRGB values and the background model and the average difference of the entire image. Use the difference information to determine whether to trigger motion detection and update the background model immediately.

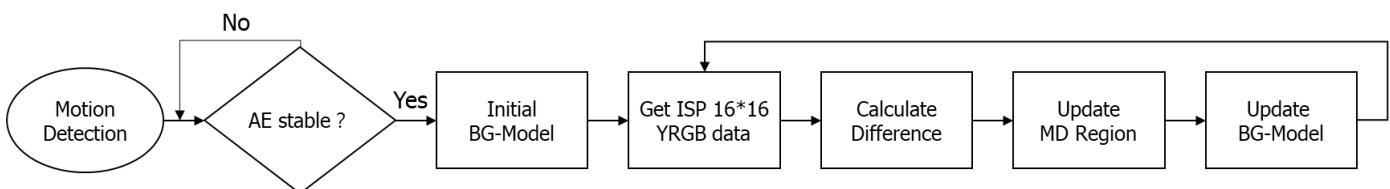


Figure 15-5 Motion detection architecture

#### 15.5.1.1 16x16 YRGB Data

The 16x16 YRGB value is calculated by averaging the RGB image generated by from video channel 4. Each value corresponds to the average YRGB value for dividing the image into a 16x16 frame. As shown in Figure 15-6. Calculating motion with 16x16 YRGB data has some advantages, such as (1) saving computation time, (2) taking into account color information, (3) filtering out noise

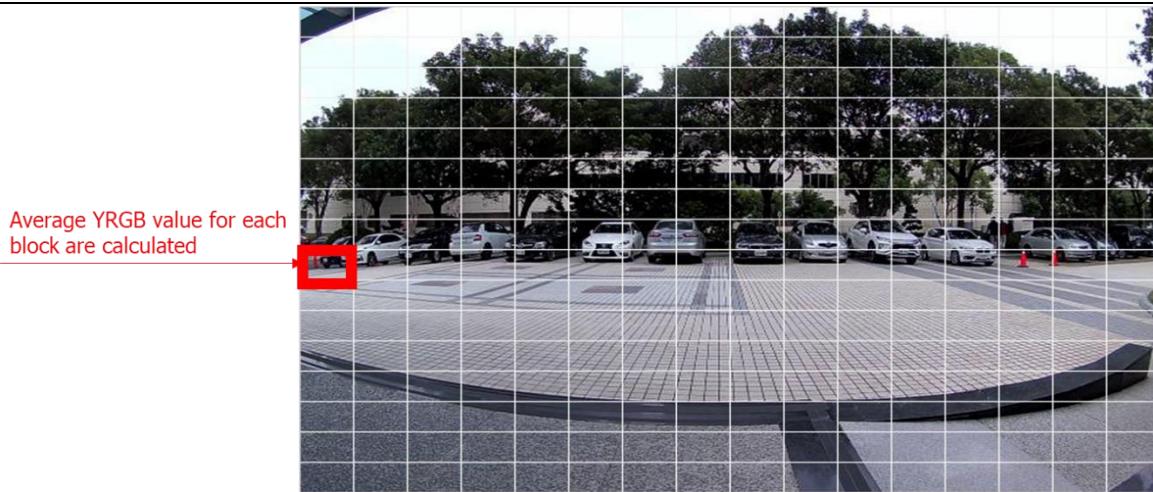


Figure 15-6 16x16 YRGB data

#### 15.5.1.2 Background Model

The calculation method of the background model is to calculate the average value of the current YRGB value and the recorded background model. This method can preserves background features, improve the sensitivity of motion detection, and update the background model in real time to avoid the problem of false alarm caused by the background change.

#### 15.5.1.3 YRGB Difference Calculation

The motion information is obtained by calculating the difference and average difference between the YRGB value of each frame and the background model. When the YRGB value difference is greater than the threshold, motion detection is triggered. The threshold is not a fixed value, but dynamically set with reference to the average difference value of each frame.

Usually the same difference through whole image is caused by noise or light change. By calculating the average difference value of a frame, the area with the difference smaller than the average difference can be filtered out. Take Figure 15-7 as an example. Initially, all the background value are 1. Motion occurred in the black region, and light change simultaneously. By calculating the average difference change of the whole image, we get average difference of 1.02. After filtering the difference value less than 1.02, we detect black areas where actual motion occurs.

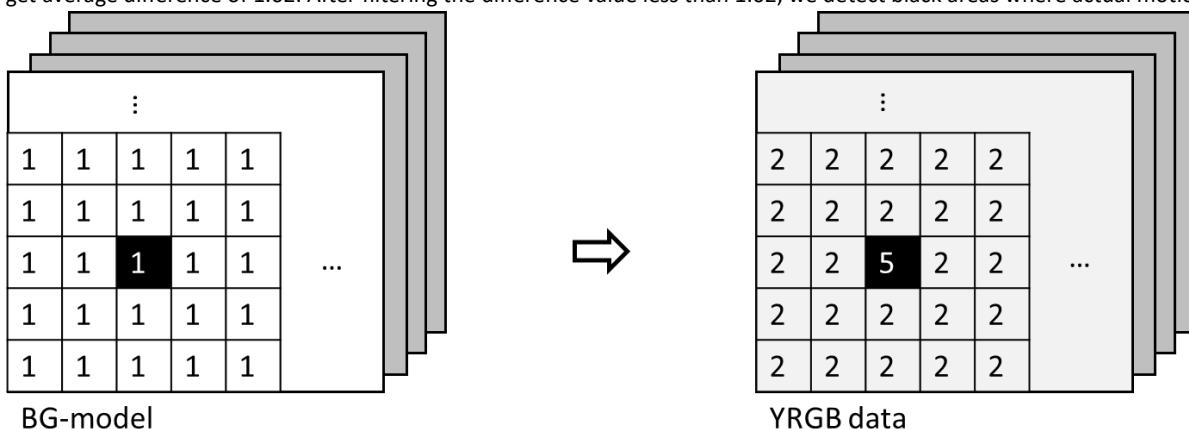


Figure 15-7 YRGB difference calculation

## 15.5.2 MD Module

The context of the md module shows as following:

```
typedef struct md_context_s {
 ...
 int md_result[col * row];
 int md_mask[col * row];
 int md_trigger_block_threshold;
 double Tauto;
 motion_detect_bgmodel_t md_bgmodel;
 motion_detect_YRGB_data_t YRGB_data;
 motion_detect_threshold_t *md_threshold;
```

```

} md_context_t;
...
typedef struct md_ctx_s {
 void *parent;
 md_param_t *params;
 md_context_t *motion_detect_ctx;
 md_disp_postproc disp_postproc;
 bool md_out_en;
} md_ctx_t;

```

Description of parameter in md\_context\_t:

- md\_result: motion detection result array. 0: no motion, 1:motion detected
- md\_mask: motion detection mask array. set 0: disable MD, 1: enable MD
- md\_trigger\_block\_threshold: md triggered when at least n motion block triggered
- md\_bgmodel: current background model
- YRGB\_data: current YRGB value
- md\_threshold: motion detection threshold. Lower value for higher sensitivity.
- disp\_postproc: Set the callback function for display the MD result on video frame.

Some md module parameters provided in module\_md.h.

```

#define col 16
#define row 16

//motion detect every n frames
#define MOTION_DETECT_INTERVAL 2

//enable dynamic threshold or not
#define DYNAMIC_THRESHOLD 0

//start MD after AE stable
#define MD_AFTER_AE_STABLE 1

//dynamic increase sensitivity when too light or too dark
#define BRIGHT_THRESHOLD 180
#define DARK_THRESHOLD 35
#if DARK_THRESHOLD > BRIGHT_THRESHOLD
#error "Motion Detection: DARK_THRESHOLD shouldn't greater than BRIGHT_THRESHOLD"
#endif

//turn off motion detect while too dark
#define TURN_OFF_THRESHOLD 8

```

- col : motion detection column resolution. The value is fixed
- row: motion detection row resolution. The value is fixed
- MOTION\_DETECT\_INTERVAL: motion detect every n frames
- DYNAMIC\_THRESHOLD: set 1 to enable dynamic threshold
- MD\_AFTER\_AE\_STABLE: set 1 to enable motion detection after AE stable
- BRIGHT\_THRESHOLD: Only use when set DYNAMIC\_THRESHOLD to 1. Dynamic increase sensitivity when too bright.
- DARK\_THRESHOLD: Only use when set DYNAMIC\_THRESHOLD to 1. Dynamic increase sensitivity when too dark.
- TURN\_OFF\_THRESHOLD: turn off motion detect while too dark

Here are some md module parameters provided to set.

```

typedef struct md_param_s {
 int width;
 int height;
} md_param_t;

```

Use CMD\_VIPNN\_SET\_IN\_PARAMS to set up the NN input parameters.

- width: input frame width.
- height: input frame height.

### 15.5.2.1 Set MD Threshold

The sensitivity of motion detection can be adjusted by tuning Tbase and Tlum.

- Tbase: motion detect base threshold. It is recommended to set Tbase in the range of 0~20
- Tlum: motion detect light sensitivity threshold. It is recommended to set Tlum in the range of 0~5

Generally, when the light changes little, the sensitivity can be improved by lowering the Tbase value. When the light changes frequently and

drastically, the sensor needs to adjust the image to adapt to the new brightness, resulting in false motion detection. Increasing the value of Tlum can reduce the problem.

Under normal conditions, we recommend setting Tbase = 5, Tlum = 3 to detect motion within 4m. Set Tbase = 2, Tlum = 3 to detect motion within 8m. Set Tbase = 1, Tlum = 3 to detect motion within 11m. However, the performance of motion detection will be affected by the camera placement angle and image distortion, it should be tested on a case-by-case basis.

User can set desired threshold value by using CMD\_MD\_SET\_MD\_THRESHOLD:

```
motion_detect_threshold_t md_thr = {
 .Tbase = 2,
 .Tlum = 3
};
md_ctx = mm_module_open(&md_module);
if (md_ctx) {
 ...
 mm_module_ctrl(md_ctx, CMD_MD_SET_MD_THRESHOLD, (int)&md_thr);
}
```

#### 15.5.2.2 Get MD Threshold

User can get MD threshold value by using CMD\_MD\_GET\_MD\_THRESHOLD:

```
motion_detect_threshold_t md_thr;
md_ctx = mm_module_open(&md_module);
if (md_ctx) {
 ...
 mm_module_ctrl(md_ctx, CMD_MD_GET_MD_THRESHOLD, (int)&md_thr);
}
```

#### 15.5.2.3 MD Dynamic Threshold Mode

The MD dynamic threshold mode can be enabled by setting DYNAMIC\_THRESHOLD = 1. When the brightness of the image is too bright or too dark, it will be more difficult to trigger motion detection because the color change is less obvious. When the image brightness is greater than BRIGHT\_THRESHOLD or less than DARK\_THRESHOLD, the threshold for motion detection is lowered to improve the motion detection ability.

#### 15.5.2.4 Set MD Mask

User can disable motion detect in desired region by setting motion detection mask array. Setting 0 to disable MD, 1 to enable MD.

Take care lane image as an example. If we want to filter out the car motion when far from camera, we can disable MD in the upper part of the image, as shown in Figure 15-8.

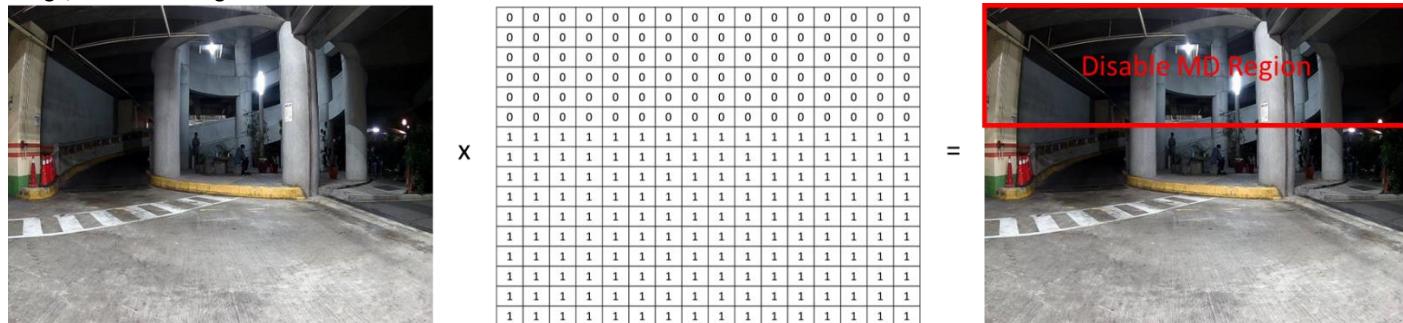


Figure 15-8 MD Mask

User can set MD mask value by using CMD\_MD\_SET\_MD\_MASK.

```
if (md_ctx) {
 ...
 mm_module_ctrl(md_ctx, CMD_MD_SET_MD_MASK, (int)&md_mask);
}
int md_mask [16 * 16] = {
 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1,
 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1,
 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1,
 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1,
 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1,
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 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1,
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 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1,
 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1}
```

```
1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1,
1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1,
1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1,
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1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1,
1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1,
1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1,
1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1,
1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1,
1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1,
1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1, 1,
};

md_ctx = mm_module_open(&md_module);
if (md_ctx) {
 ...
 mm_module_ctrl(md_ctx, CMD_MD_SET_MD_MASK, (int)&md_mask);
}
```

### 15.5.2.5 Get MD Mask

User can get MD mask value by using CMD\_MD\_GET\_MD\_MASK:

```
int md_mask [16 * 16];
md_ctx = mm_module_open(&md_module);
if (md_ctx) {
 ...
 mm_module_ctrl(md_ctx, CMD_MD_GET_MD_MASK, (int)&md_mask);
}
```

### 15.5.2.6 Get MD Result

User can get MD result value by using CMD\_MD\_GET\_MD\_RESULT.

It will return a 16x16 array, which corresponding to whole image. Get value 0 when motion is detected, 1 when no motion is detected.

```
int md_result [16 * 16];
md_ctx = mm_module_open(&md_module);
if (md_ctx) {
 ...
 mm_module_ctrl(md_ctx, CMD_MD_GET_MD_RESULT, (int)&md_result);
}
```

### 15.5.2.7 Set MD Trigger Block Threshold

User can set motion detect trigger block threshold by using CMD\_MD\_SET\_TRIG\_BLK.

This command only used in motion trigger NN example. Trigger NN detection when at least n motion block are triggered

```
md_ctx = mm_module_open(&md_module);
if (md_ctx) {
 ...
 mm_module_ctrl(md_ctx, CMD_MD_SET_TRIG_BLK, 3);
}
```

### 15.5.2.8 Set MD result display callback function

User can register a call back function to so display the MD result. Use CMD\_MD\_SET\_DISPPOST to set up callback function for display the MD result:

```
static void md_process(void *md_result)
{
 int *md_res = (int *) md_result;
 /* Process or display the result here */
}
md_ctx = mm_module_open(&md_module);
if (md_ctx) {
 ...
 mm_module_ctrl(md_ctx, CMD_MD_SET_DISPPOST, (int)md_process);
}
```

## 15.5.3 MD Example

The MD example is a part of mmf video joined example. Please uncomment the example want to execute.  
(project/realtek\_amebapro2\_v0\_example/src/mmfv2\_video\_example/video\_example\_media\_framework.c)

```
mmf2_video_example_md_rtsp_init();
//mmf2_video_example_md_nn_rtsp_init();
```

Table 15-65 MD example

| Example                            | Description                                                    | Result                                                                                                                                                                             |
|------------------------------------|----------------------------------------------------------------|------------------------------------------------------------------------------------------------------------------------------------------------------------------------------------|
| mmf2_video_example_md_rtsp_init    | CH1 Video -> H264/HEVC -> RTSP<br>CH4 Video -> RGB -> MD       | RTSP video stream over the network.<br>MD detect motion and draw the motion region to RTSP channel.                                                                                |
| mmf2_video_example_md_nn_rtsp_init | CH1 Video -> H264/HEVC -> RTSP<br>CH4 Video -> RGB -> MD -> NN | RTSP video stream over the network.<br>MD module detect motion. If there is motion detected, it will trigger NN module to detect object and draw the bounding box to RTSP channel. |

### 15.5.3.1 Build MD Example

Since it's a part of video mmf example, user should use the following command to generate the makefile.

Generate the makefile for the MD project:

```
cmake .. -G"Unix Makefiles" -DCMAKE_TOOLCHAIN_FILE=../toolchain.cmake -DVIDEO_EXAMPLE=ON
```

Then, use the following command to generate an image:

```
cmake --build . --target flash
```

After running the command above, you will get the flash\_ntz.bin in "project\realtek\_amebapro2\_v0\_example\GCC-RELEASE\build". Then, use the image tool to download it to AmebaPro2.

### 15.5.3.2 Build MD & NN Example

Since it's a part of video mmf example, user should use the following command to generate the makefile.

Generate the makefile for the MD project:

```
cmake .. -G"Unix Makefiles" -DCMAKE_TOOLCHAIN_FILE=../toolchain.cmake -DVIDEO_EXAMPLE=ON
```

If running mmf2\_video\_example\_md\_nn\_rtsp\_init example, use the following command to generate an image with NN model inside:

```
cmake --build . --target flash_nn
```

After running the command above, you will get the flash\_ntz.nn.bin in "project\realtek\_amebapro2\_v0\_example\GCC-RELEASE\build". Then, use the image tool to download it to AmebaPro2.

### 15.5.3.3 Validate MD example

While running the example, you may need to configure WiFi connection by using these commands in uart terminal.

```
ATW0=<WiFi_SSID> : Set the WiFi AP to be connected
ATW1=<WiFi_Password> : Set the WiFi AP password
ATWC : Initiate the connection
```

If everything works fine, you should see the following logs. Motion detection result will show in logs.

```
[VOE]RGB3 640x480 1/10
[VOE]zoom default setting
It is SENSOR_GC2053
[VOE]release s4 isp buffer 0
[VOE]release s4 isp buffer 1
Set MD Threshold: Tbase = 2.000000, Tlum = 3.000000
Set MD Mask:
1 1 1 1 1 1 1 1 1 1 1 1 1 1 1 1
1 1 1 1 1 1 1 1 1 1 1 1 1 1 1 1
1 1 1 1 1 1 1 1 1 1 1 1 1 1 1 1
1 1 1 1 1 1 1 1 1 1 1 1 1 1 1 1
1 1 1 1 1 1 1 1 1 1 1 1 1 1 1 1
1 1 1 1 1 1 1 1 1 1 1 1 1 1 1 1
1 1 1 1 1 1 1 1 1 1 1 1 1 1 1 1
1 1 1 1 1 1 1 1 1 1 1 1 1 1 1 1
1 1 1 1 1 1 1 1 1 1 1 1 1 1 1 1
1 1 1 1 1 1 1 1 1 1 1 1 1 1 1 1
1 1 1 1 1 1 1 1 1 1 1 1 1 1 1 1
1 1 1 1 1 1 1 1 1 1 1 1 1 1 1 1
1 1 1 1 1 1 1 1 1 1 1 1 1 1 1 1
```

```
1 1 1 1 1 1 1 1 1 1 1 1 1 1 1 1 1
1 1 1 1 1 1 1 1 1 1 1 1 1 1 1 1 1
1 1 1 1 1 1 1 1 1 1 1 1 1 1 1 1 1
1 1 1 1 1 1 1 1 1 1 1 1 1 1 1 1 1
1 1 1 1 1 1 1 1 1 1 1 1 1 1 1 1 1
```

siso\_rgb\_md started

Calculate YRGB after 47ms.

```
0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0
0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0
0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0
0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0
0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0
0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0
0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0
0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0
0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0
0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0
0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0
0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0
0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0
0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0
0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0
0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0
0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0
0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0
```

Calculate YRGB after 0ms.

Calculate YRGB after 46ms.

```
1 0 0 0 0 0 1 0 0 0 0 0 0 0 0 0 0
0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0
0 0 1 1 0 0 0 0 0 0 0 0 0 0 0 0 0
0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0
0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0
0 0 0 0 0 0 1 0 0 0 0 0 0 0 0 0 0
0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0
0 0 0 0 0 1 1 0 0 0 0 0 0 0 0 0 0
0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0
0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0
0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0
0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0
0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0
0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0
0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0
0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0
0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0 0
```

If desire to see the motion detected region, set MD\_DRAW to 1.

```
#define MD_DRAW 1
```

Then, open VLC (or PotPlayer) and create a network stream with URL: rtsp://192.168.x.xx:554

When motion detected, it will draw the motion detected region.



```
Deploy YOLO
network 71c263c0
input 0 dim 416 416 3 1, data format=2, quant_format=2, scale=0.003660, zero_point=0
input buffer 0 = 71c27490, vid memory 733b9700
output 0 dim 13 13 255 1, data format=2, scale=0.104320, zero_point=200
output buffer 0 = 71c274d8, vid memory 73438300
output 1 dim 26 26 255 1, data format=2, scale=0.102169, zero_point=197
output buffer 1 = 71c27520, vid memory 73442b80

input count 1, output count 2
input param 0
 data_format 2
 memory_type 0
 num_of_dims 4
 quant_format 2
 quant_data , scale=0.003660, zero_point=0
 sizes 1a0 1a0 3 1 0 0
output param 0
 data_format 2
 memory_type 0
 num_of_dims 4
 quant_format 2
 quant_data , scale=0.104320, zero_point=200
 sizes d d ff 1 0 0
output param 1
 data_format 2
 memory_type 0
 num_of_dims 4
 quant_format 2
 quant_data , scale=0.102169, zero_point=197
 sizes 1a 1a ff 1 0 0

in 0, size 416 416
VIPNN opened

Calculate YRGB after 48ms.
siso_rgb_md started
siso_md_nn started
Text/Logo OSD Test
nn_rect_ch:0, nn_rect_txt_w:16, nn_rect_txt_h:32.
font resize new size: 4768.
font resize new size: 3688.
font resize from 32 64 to 16 32.
font resize from 64 64 to 32 32.
font resize:21.

Calculate YRGB after 0ms.

Calculate YRGB after 49ms.
Motion Detected
YOLO tick[0] = 84
object num = 3
2,c0:855 296 1041 585

Calculate YRGB after 0ms.
YOLO tick[0] = 76
object num = 3
2,c0:852 297 1039 586

Calculate YRGB after 47ms.
Motion Detected
YOLO tick[0] = 85
object num = 4
1,c0:853 293 1038 584
2,c0:536 127 1609 955
```

Then, open VLC (or PotPlayer) and create a network stream with URL: rtsp://192.168.x.xx:554

When motion detected, it will trigger object detection, and draw the detection result

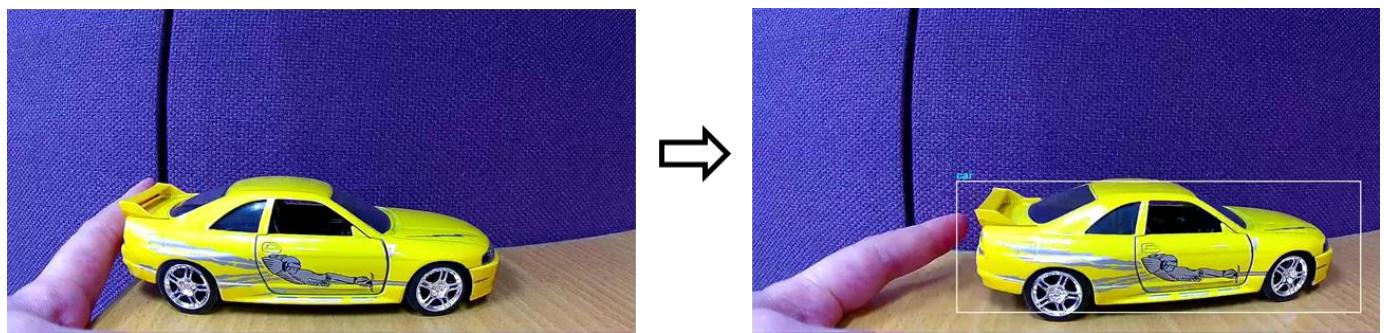


Figure 15-10 MD & NN Example

# 16 Bluetooth

## 16.1 BT Example

AmebaPro2 BT examples provide a set of BT functionality, such as BT Config, BT Peripheral, BT Central and BT Beacon.

BT Config example demonstrates how to transfer SSID profile from Mobile Phone to device.

For more information, please refer to [UM0201 Ameba Common BT Application User Manual EN.pdf](#).

This section illustrates how to build and run BT examples in our SDK, including GCC and IAR environment.

### 16.1.1 GCC Project

- (1) Enter SDK path: project\realtek\_amebapro2\_v0\_example\inc and modify file platform\_opts\_bt.h to enable BT.

```
#define CONFIG_BT 1 //This define must be enabled.
//Enable corresponding define for which example want to be used.
//Here using example bt peripheral for instance.
#define CONFIG_BT_CONFIG 0
#define CONFIG_BT_AIRSYNC_CONFIG 0
#define CONFIG_BT_PERIPHERAL 1
#define CONFIG_BT_CENTRAL 0
#define CONFIG_BT_SCATTERNET 0
#define CONFIG_BT_BEACON 0
```

- (2) Build images

- (3) Use ImageTool to download images to your board.

**NOTE**

You can select one BLE example or all the examples at once. If all the examples are selected at once, the integrated image can support all BLE test commands, and the scatternet configuration will display only when both peripheral and central are selected.

### 16.1.2 Examples List

#### 16.1.2.1 ble\_peripheral

This example shows how to create and run GATT service on GATT server.

##### 16.1.2.1.1 Image Generation

- (1) To run ble\_peripheral example, turn on the following flags defined in

```
\project\realtek_amebapro2_v0_example\inc\platform_opts_bt.h
#define CONFIG_BT 1
#if CONFIG_BT
#define CONFIG_FTL_ENABLED
#define CONFIG_BT_CONFIG 0
#define CONFIG_BT_PERIPHERAL 1
#define CONFIG_BT_CENTRAL 0
#define CONFIG_BT_SCATTERNET 0
#define CONFIG_BT_BEACON 0
#define CONFIG_BT_MESH_PROVISIONER 0
#define CONFIG_BT_MESH_DEVICE 0
```

- (2) Build image and download image to your board.

##### 16.1.2.1.2 Test Procedure

- (1) After download image to your AmebaPro2 board, reset it. The default device name is BLE\_PERIPHERAL.

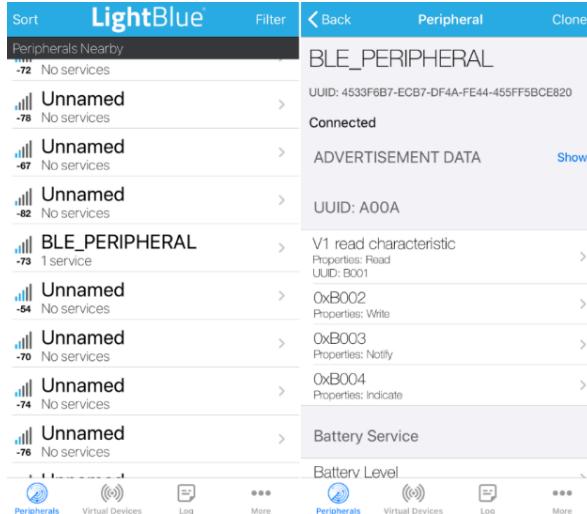
- (2) Download apps such as “LightBlue” or “nRF Connect” and use as GATT Client to connect it.

- (3) ATBp is an AT command for BT Peripheral. Using “ATBp=1” to initialize BT Peripheral stack, which can send advertising package out and scannable by other devices. Below is the BT peripheral example initialization success log.

```
hci_borad_controller_reset:346(info) BT Reset OK!
amebapro2_uart_set_bdrate:72(info)
Set baudrate to 921600 success!
[BLE peripheral] GAP stack ready local bd addr: 0x
[MEM] After do cmd, available heap 46760992
#
89:51:12:36:28:11
```

```
GAP adv start
```

- (4) Search for BLE\_PERIPHERAL device and connect to it.



### 16.1.2.2 ble\_central

This example shows how to discover service on GATT server.

#### 16.1.2.2.1 Image Generation

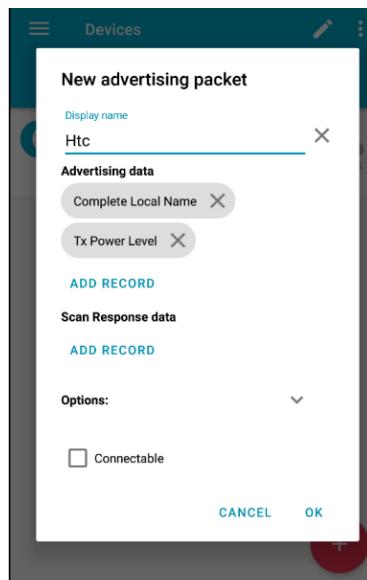
- (1) To run ble\_central example, turn on the following flags defined in \project\realtek\_amebapro2\_v0\_example\inc\platform\_opts\_bt.h

```
#define CONFIG_BT 1
#ifndef CONFIG_BT
#define CONFIG_BT_FTL_ENABLED
#define CONFIG_BT_CONFIG 0
#define CONFIG_BT_PERIPHERAL 0
#define CONFIG_BT_CENTRAL 1
#define CONFIG_BT_SCATTERNET 0
#define CONFIG_BT_BEACON 0
#define CONFIG_BT_MESH_PROVISIONER 0
#define CONFIG_BT_MESH_DEVICE 0
```

- (2) Build image and download image to your board.

#### 16.1.2.2.2 Test Procedure

- (1) After download image to your AmebaPro2 board, reset it.  
(2) Download app “nRF Connect” and use as GATT Server to be connected.  
(3) Add new advertising packet and set its additional data.



- (4) ATBc is an AT command for BT Central. Using “ATBc=1” to turn BT Central stack ON.  
(5) Using “ATBS=1” to scan available BT devices nearby.

- 
- (6) Using “ATBC=P/R, BLE\_BD\_ADDR” to connect to the device.

BT Central scan and connect log:

```
#ATBS=1
Start scan, scan_filter_policy = 0, scan_filter_duplicate = 1 [MEM] After do cmd,
available heap 46756320
#
GAP scan start
ADVType | AddrType |BT_Addr |rss
CON_UNDIRECT random 4f:6e:3e:75:56:2e -80
GAP_ADTYPE_FLAGS: 0x1a
GAP_ADTYPE_MANUFACTURER_SPECIFIC: company_id 0x4c, len 24
ADVType | AddrType |BT_Addr |rss
CON_UNDIRECT random 70:20:ca:98:7a:88 -74
GAP_ADTYPE_FLAGS: 0x1a
GAP_ADTYPE_POWER_LEVEL: 0x18
GAP_ADTYPE_MANUFACTURER_SPECIFIC: company_id 0x4c, len 7
#ATBS=0 Stop scan
[MEM] After do cmd, available heap 46756320
GAP scan stop
ATBC=R, 665544778899 [MEM] After do cmd, available heap 46756320
cmd_con, DestAddr: 0x66:0x55:0x44:0x77:0x88:0x99
```

For more AT commands used for BT Central, please refer to user manual [UM0201 Ameba Common BT Application User Manual EN.pdf](#).

### 16.1.2.3 ble\_scatternet

BLE Scatternet is the coexistence of BLE Central mode and BLE Peripheral mode. Once BLE Scatternet stack initialized, AT command of BLE Central and BLE Peripheral are available. This example shows how to turn BLE Scatternet on.

#### 16.1.2.3.1 Image Generation

- (1) To run ble\_central example, turn on the following flags defined in \project\realtek\_amebapro2\_v0\_example\inc\platform\_opts\_bt.h

```
#define CONFIG_BT 1
#if CONFIG_BT
#define CONFIG_FTL_ENABLED
#define CONFIG_BT_CONFIG 0
#define CONFIG_BT_PERIPHERAL 0
#define CONFIG_BT_CENTRAL 0
#define CONFIG_BT_SCATTERNET 1
#define CONFIG_BT_BEACON 0
#define CONFIG_BT_MESH_PROVISIONER 0
#define CONFIG_BT_MESH_DEVICE 0
```

- (2) Build image and download image to your board.

#### 16.1.2.3.2 Test Procedure

- (1) After download image to your AmebaPro2 board, reset it.  
(2) Using “ATBf=1” to turn BT Scatternet stack ON.  
(3) Once see the following message, you can continue input other AT command of BT Scatternet mode as well as BT Central mode and BT Peripheral mode.

```
hci_borad_controller_reset:346(info) BT Reset OK!
amebapro2_uart_set_bdrate:72(info) Set baudrate to 921600 success!
local bd addr: 0x89:51:12:36:28:11
[MEM] After do cmd, available heap 46754528
#
GAP adv start
```

For other AT commands used for BT Scatternet, please refer to [UM0201 Ameba Common BT Application User Manual EN.pdf](#).

### 16.1.2.4 bt\_beacon

This example shows how to send BLE Beacons. AmebaPro2 provides two types of Beacon: Apple iBeacon and Radius Networks AltBeacons.

#### 16.1.2.4.1 Image Generation

- (1) To run ble\_central example, turn on the following flags defined in \project\realtek\_amebapro2\_v0\_example\inc\platform\_opts\_bt.h

```
#define CONFIG_BT 1
#if CONFIG_BT
#define CONFIG_FTL_ENABLED
#define CONFIG_BT_CONFIG 0
#define CONFIG_BT_PERIPHERAL 0
#define CONFIG_BT_CENTRAL 0
#define CONFIG_BT_SCATTERNET 0
#define CONFIG_BT_BEACON 1
#define CONFIG_BT_MESH_PROVISIONER 0
#define CONFIG_BT_MESH_DEVICE 0
```

- (2) Build image and download image to your board.

#### 16.1.2.4.2 Test Procedure

- (1) Choose beacon type by using “ATBJ=1,1” or “ATBJ=1,2” command.

```
ATBJ
[ATBJ] Start BT I_Beacon: ATBJ=1,1
[ATBJ] Start BT Alt_Beacon: ATBJ=1,2
[ATBJ] Stop BT Beacon: ATBJ=0
```

- (2) You can use apps such as “LightBlue” or “nRF Connect” to observe beacons. “Locate” observe beacon by its adv UUID. Below screenshot is taken using Android “nRF Connect”.

### 16.1.2.5 bt\_config

BT Config provides a simple way for Wi-Fi device to associate to AP easily.

#### 16.1.2.5.1 Image Generation

- (1) To run ble\_central example, turn on the following flags defined in \project\realtek\_amebapro2\_v0\_example\inc\platform\_opts\_bt.h

```
#define CONFIG_BT 1
#if CONFIG_BT
#define CONFIG_FTL_ENABLED
#define CONFIG_BT_CONFIG 1
#define CONFIG_BT_PERIPHERAL 0
#define CONFIG_BT_CENTRAL 0
#define CONFIG_BT_SCATTERNET 0
#define CONFIG_BT_BEACON 0
#define CONFIG_BT_MESH_PROVISIONER 0
#define CONFIG_BT_MESH_DEVICE 0
```

- (2) Build image and download image to your board.

#### 16.1.2.5.2 APP Installation

Search “Easy WiFi Config” in the application store. You can install Android or iOS as your phone OS.



#### 16.1.2.5.3 Test Procedure

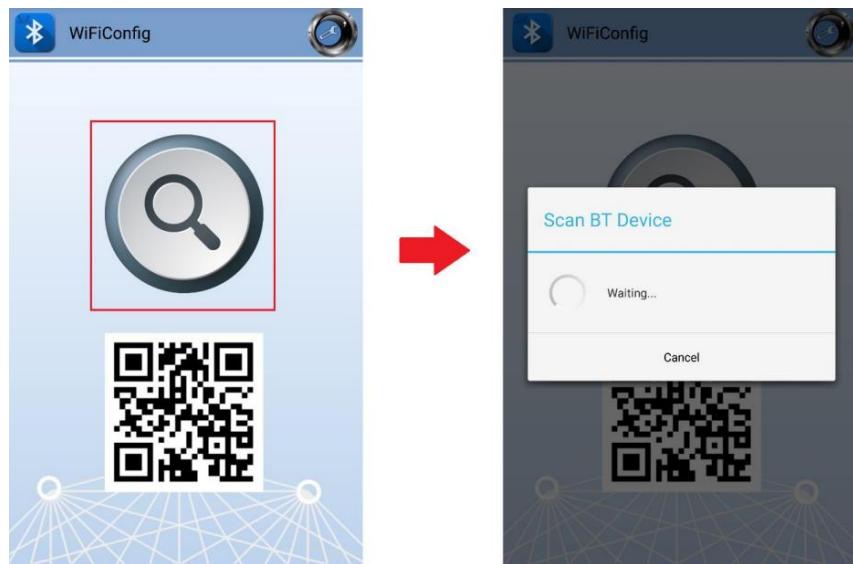
- (1) ATBB is an AT command for BT Config. Using “ATBB=1” to enter BT Config mode, which allows BT Config APP to discover and connect to AmebaPro2. Reset your AmebaPro2 board, and input command “ATBB=1”.  
(2) Once see the following message, you can open BT Config APP to associate AP.

BT Initialize and start adv log:

```
[BT Config Wifi] BT Config Wifi ready
[BT Config Wifi] ADV started
```

- (3) Click the BT config icon to launch it. Scan and connect with AmebaPro2 BT using BT Config app.

Display on BT config app:

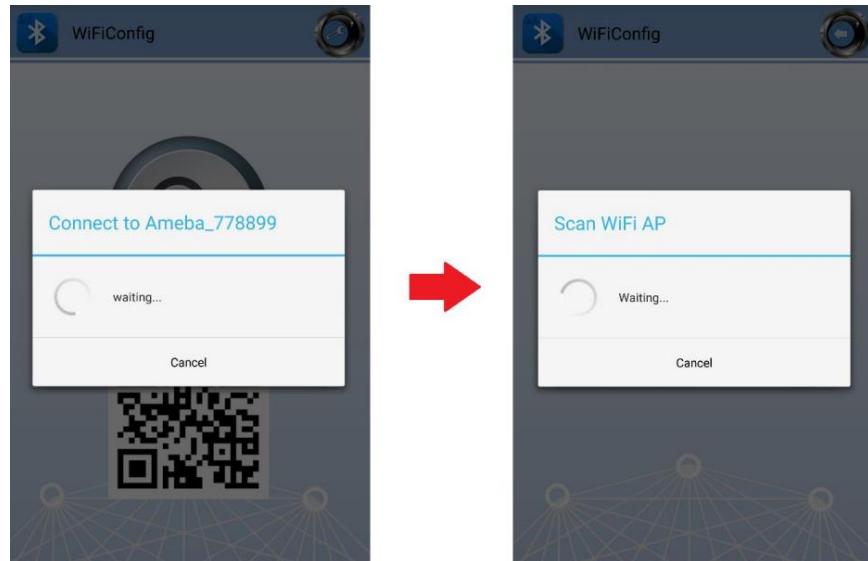


- (4) Once BT Config APP connected to AmebaPro2, below log will be show. When connection is established AmebaPro2 will start searching for AP.

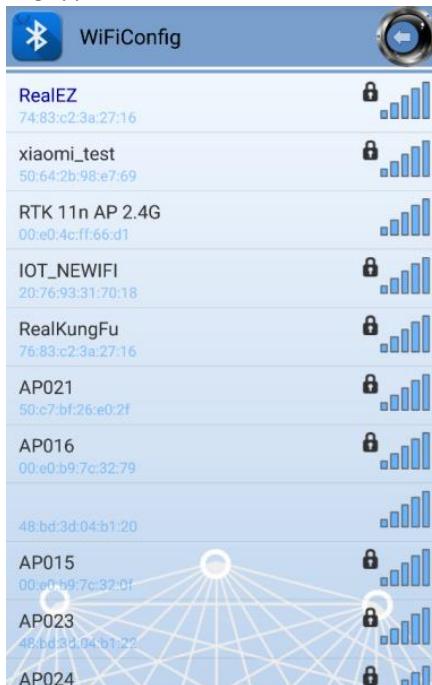
BT Connection log:

```
[BT Config Wifi] Bluetooth Connection Established
[BT Config Wifi] Band Request
[BT Config Wifi] Scan Request
[BT Config Wifi] Scan 2.4G AP
[BT Config Wifi] Scan 5G AP
```

Display on BT config app:



Scanned and reachable APs will be show on BT config app:

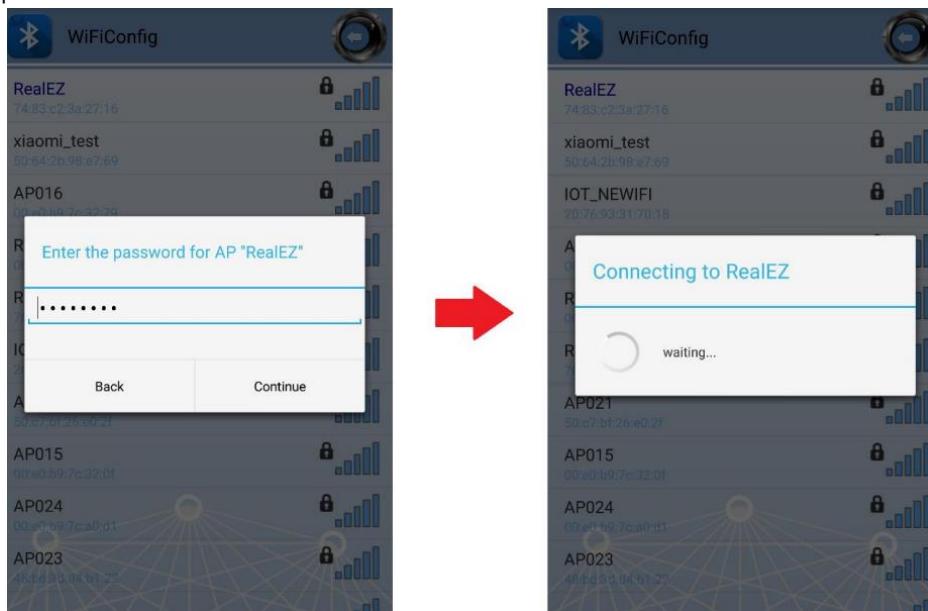


(5) Select an AP to connect to and input password (if any).

AP Connection log:

```
[BT Config Wifi] Connect Request
[Driver]: set BSSID: 90:94:e4:c5:d3:f0
[Driver]: set ssid [Test_ap]
[Driver]: start auth to 90:94:e4:c5:d3:f0
[Driver]: auth success, start assoc
[Driver]: association success(res=7)
[Driver]: set pairwise key to hw: alg:4(WEP40-1 WEP104-5 TKIP-2 AES4)
[Driver]: set group key to hw: alg:2(WEP40-1 WEP104-5 TKIP-2 AES-4) keyid:1
[BT Config Wifi] Connected after 3458ms.
Interface 0 IP address : 192.168.0.102 [BT Config Wifi]
Got IP after 3500ms.
```

Display on BT config app:



(6) When AmebaPro2 is connected to an AP, user can confirm connection or select another AP. Click "Confirm" to confirm AP connection. Click "Try another AP" to go back to Wi-Fi scan list page and choose another AP to connect to. After confirming BT config result, Bluetooth connection is disconnected, AmebaPro2 becomes undiscoverable to BT Config APP.

## BT Disconnect log:

```
[BT Config Wifi] Bluetooth Connection Disconnected
[BT Config Wifi] ADV started
[BT Config Wifi] [BC_status_monitor] wifi connected, delete BC_cmd_task and
BC_status_monitor
[BT Config Wifi] ADV stopped
```

Display on BT config app:



(7) You can use "ATBB=1" to restart BT Config mode again.

| Command | Introduction    |
|---------|-----------------|
| ATBB=1  | Start BT Config |
| ATBB=0  | Stop BT Config  |

**NOTE**

Enter BT Config mode will disconnect existing Wi-Fi connection.

Please refer to BT Config APP User Guide for more details

# 17 System Resource Evaluation

This section will explain how to calculate used system resource in GCC project.

## 17.1 Memory Section

We can find the memory configuration in “rtl8735b\_ram.ld”:

- .ram.code\_rodata : This section is read-only data in SRAM
- .ram.data : This section is read-write data in SRAM
- .ram.bss : This section is data has no initial values in SRAM
- .ddr.rodata : This section is read-only data in DDR Memory
- .ddr.data : This section is read-write data in DDR Memory
- .ddr.bss : This section is data has no initial values in DDR Memory

## 17.2 Memory Size

User can refer “application.ntz.map” to observe them after project build. This file can be found in “project\realtek\_amebapro2\_v0\_example\GCC-RELEASE\build\application” folder.

### 17.2.1 Memory Size in SRAM

There are several sections in SRAM, including “.ram.code\_text”, “.ram.data”, “.ram.code\_rodata”, “.ram.bss” and “.non\_secure.bss”. We can sum up these sections:

“.ram.code\_text” has size 0xdef0:

```
.ram.code_text 0x0000000020100b00 0xdef0
 0x0000000020100b00 . = ALIGN (0x4)
 0x0000000020100b00 __etext2 = .
 0x0000000020100b00 . = ALIGN (0x20)
 0x0000000020100b00 __ram_entry_text_start__ = .
```

“.ram.data” has size 0x9b4:

```
.ram.data 0x000000002010e9f0 0x9b4
 0x000000002010e9f0 __fw_img_start__ = .
 0x000000002010e9f0 __etext = .
 0x000000002010e9f0 __data_start__ = .
```

“.ram.code\_rodata” has size 0x5d8:

```
.ram.code_rodata
 0x000000002010f3a8 0x5d8
 0x000000002010f3a8 . = ALIGN (0x4)
 0x000000002010f3a8 __ram_code_rodata_start__ = .
```

“.ram.bss” has size 0x139fc:

```
.ram.bss 0x000000002010f980 0x139fc
 0x000000002010f980 . = ALIGN (0x4)
 0x000000002010f980 __bss_start__ = .
```

“.non\_secure.bss” has size 0x4:

```
.non_secure.bss
 0x000000002012337c 0x4
 0x0000000020123380 . = ALIGN (0x10)
```

So user totally use  $0xdef0 + 0x9b4 + 0x5d8 + 0x139fc + 0x4 = 0x2287c = 138KB$  memory in SRAM.

And the total SRAM space is defined at project\realtek\_amebapro2\_v0\_example\GCC-RELEASE\application\rtl8735b\_ram.ld:

```
RAM (rwx) : ORIGIN = 0x20100B00, LENGTH = 0x20177B00 - 0x20100B00 /* 476KB */
```

For this case, the SRAM total size is 476KB. So user still have free SRAM space about  $476KB - 138KB = 338KB$ .

### 17.2.2 Memory Size in DDR Memory (ERAM)

There are several sections in DDR Memory, “.ddr.bss”, “.ddr.text”, “.ddr.data” and “.ddr.rodata”. We can sum up these sections:

“.ddr.bss” has size 0x1424d8:

|          |                    |                               |
|----------|--------------------|-------------------------------|
| .ddr.bss | 0x0000000070100000 | 0x1424d8                      |
|          | 0x0000000070100000 | . = ALIGN (0x4)               |
|          | 0x0000000070100000 | <u>__eram_bss_start__</u> = . |

".ddr.text" has size 0x7d278:

|           |                    |                                |
|-----------|--------------------|--------------------------------|
| .ddr.text | 0x00000000702424e0 | 0x7d278                        |
|           | 0x00000000702424e0 | . = ALIGN (0x4)                |
|           | 0x00000000702424e0 | <u>__eram_text_start__</u> = . |

".ddr.data" has size 0x4e08:

|           |                    |                                |
|-----------|--------------------|--------------------------------|
| .ddr.data | 0x00000000702bf79c | 0x4e08                         |
|           | 0x00000000702bf79c | . = ALIGN (0x4)                |
|           | 0x00000000702bf79c | <u>__eram_data_start__</u> = . |

".ddr.rodata" has size 0x512a:

|             |                    |                                  |
|-------------|--------------------|----------------------------------|
| .ddr.rodata | 0x00000000702c45a4 | 0x512af                          |
|             | 0x00000000702c45a4 | . = ALIGN (0x4)                  |
|             | 0x00000000702c45a4 | <u>__eram_rodata_start__</u> = . |

we cannot calculate the heap usage now. However, we can get its value after running an application on AmebaPro2.

DDR memory total size = .ddr.bss + .ddr.text + .ddr.data + .ddr.rodata + heap usage + heap available

We can get the heap available size by entering AT command – “ATW?” in uart console:

```
[MEM] After do cmd, available heap 55318304
```

For this example, the heap available size is 54893248 = 53606 KB = 52 MB

And the total ERAM space is defined at project\realtek\_amebapro2\_v0\_example\GCC-RELEASE\application\rtl8735b\_ram.ld:

```
DDR (rwx) : ORIGIN = 0x70100000, LENGTH = 0x73900000 - 0x70100000 /* 56MB */
```

For this case, the DDR memory total size is 0x73900000 - 0x70100000 = 0x3800000 = 56MB.

So we can calculate the heap usage now:

heap usage = DDR memory total size - .ddr.bss - .ddr.text - .ddr.data - .ddr.rodata - heap available = 0x3800000(56MB) - 0x1424d8 - 0x7d278 - 0x4e08 - 0x512af - 54893248 = 1,641,785 B = 1.6 MB

## 17.3 Code Size

The size of flash\_ntz.bin:

```
$ ls -al GCC-RELEASE/build/flash_ntz.bin
-rw-r--r-- 1 user Domain Users 4599808 May 27 16:11 flash_ntz.bin
```

For this case, the code size is about 4.4MB

## 17.4 CPU Utilization

CPU utilization can be evaluated by AT command - “ATSS” in uart console.

It will show the amount of time each task has spent in the Running state (how much CPU time each task has consumed).

```
[ATSS]: _AT_SYSTEM_CPU_STATS_
log_service 161 <1%
IDLE 4983981 99%
Tmr Svc 0 <1%
TCP_IP 0 <1%
cmd_thread 1246 <1%
rtw_interru 0 <1%
rtw_recv_ta 0 <1%
rtw_xmit_ta 0 <1%
```

## 18 FCS and multi sensor

The fast camera start (FCS) can speed up the sensor bring up time. The sensor starts from rom code and initial the Video engine from bootloader. It can speed up the time to get the first frame. The multi sensor can support auto sensor recognition. The sensor firmware is located at the flash; it will try the sensor list to get the current sensor driver. The first time will take time to scan the sensor, and then the result will be stored at the flash.

### 18.1 FCS



#### 18.1.1 How to use the fcs mode

Modify the project\realtek\_amebapro2\_v0\_example\GCC-RELEASE\mp\amebapro2\_isp\_iq.json

```

"VARIABLE": {
 "*type*": "0x00 (INVALID) is reserved for invalid type, 0x01 (SENSOR_ID) is
 reserved for sensor id. default offset is 2048 from manifest start",
 "tlv": [
 {"type": "SENSOR_ID", "length": 1, "value": "3"} //value : sensor id
]
}

```

```
},
```

It need to change the value to your sensor id.

Modify the file from component\video\driver\RTL8735B\video\_user\_boot.c

```
void user_boot_config_init(void *parm)
{
 //Insert your code into here
 //dbg_printf("user_boot_config_init\r\n");
}
```

If you need to execute the operation at boot loader step, you can add your code at the API.

Modify the file from sdk\project\realtek\_amebapro2\_v0\_example\inc\sensor.h

```
#define USE_SENSOR SENSOR_GC4653
```

### 18.1.2 The FCS parameter

```
typedef struct video_boot_stream_cfg {
 video_params_t video_params[4];//For different channel parameter
 isp_info_t isp_info;//isp_info
 uint32_t voe_heap_addr;//Heap address for VOE
 uint32_t voe_heap_size;//Heap size for VOE
 uint8_t video_enable[4];//Enable channel
 uint8_t video_snapshot[4];//Enable the snapshot, it only support channel 0 now
 isp_multi_fcs_ld_info_t p_fcs_ld_info;//The info to load iq and sensor firmware
 uint32_t fcs_channel; //channel -> //How many channel to use fcs, the default is 1
 uint32_t fcs_status;// status -> 1:successful 0:fail
 uint32_t fcs_setting_done;// status -> 1:successful 0:fail
 uint32_t fcs_voe_fw_addr;//
 uint32_t fcs_isp_ae_enable;//Enable the AE init function
 uint32_t fcs_isp_ae_init_exposure;//Setup the exposure parameters.
 uint32_t fcs_isp_ae_init_gain;//Setup the gain parameters.
 uint32_t fcs_isp_awb_enable;//Enable the AWB init function.
 uint32_t fcs_isp_awb_init_rgain;//Setup the AWB rgain parameters.
 uint32_t fcs_isp_awb_init_bgain;//Setup the AWB bgain parameters.
} video_boot_stream_t;
```

```
typedef struct video_params_s {
 uint32_t stream_id; //Channel ID
 uint32_t type; //Codec type
 uint32_t resolution;//Resolution
 uint32_t width;
 uint32_t height;
 uint32_t bps;
 uint32_t fps;
 uint32_t gop;
 uint32_t rc_mode;
 uint32_t jpeg_qlevel;
 uint32_t rotation;
 uint32_t out_buf_size;//Reserve buf to encode queue
 uint32_t out_rsvd_size;//Reserve buf to encode
 uint32_t direct_output;//Don't care
 uint32_t use_static_addr;//Don't care
 uint32_t fcs;//Enable the FCS mode
} video_params_t;
```

For detailed parameter settings, please refer to the following file.

```
component\video\driver\RTL8735B\video_user_boot.c
```

### 18.1.3 How to run FCS example?

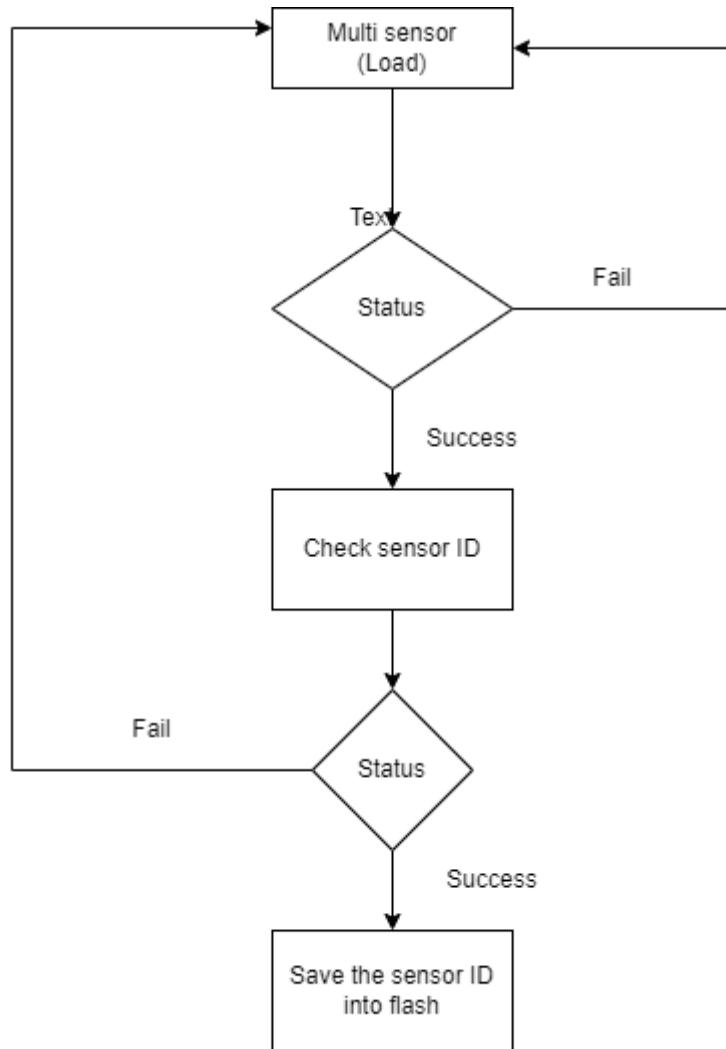
Refer to 18.1.1 to choose your sensor.

Modify the media\_framework.c and choose the below example. The default parameter is channel 0.

```
mmf2_video_example_joint_test_rtsp_mp4_init_fcs();
```

Build your code and upgrade your FW.

## 18.2 Multi sensor



### 18.2.1 How to use the multi sensor?

We use the GC2053 for the example, it need to assign the sensor, FCS and IQ data.

Modify the sdk\project\realtek\_amebapro2\_v0\_example\GCC-RELEASE\mp\amebapro2\_isp\_iq.json

```

"VARIABLE": {
 "*type*": "0x00 (INVALID) is reserved for invalid type, 0x01 (SENSOR_ID) is
reserved for sensor id. default offset is 2048 from manifest start",
 "tlv": [
 {"type": "SENSOR_ID", "length": 1, "value": "0"}
]
},

```

The value need to setup as zero. The non-zero value is for FCS mode.

Modify the sdk\project\realtek\_amebapro2\_v0\_example\GCC-RELEASE\mp\ amebapro2\_sensor\_set.json

We support five sensors. The default 0 is not used.

```

"ISP_SENSOR_SETS": {
 "multi_fcs_hdr": "MULTI_FCS_HDR",
 "multi_fcs_info": "MULTI_FCS_INFO",
 "sensor_sets": [
 "SENSOR_SET0", //Dummy setup
 "SENSOR_SET1", //GC2336
 "SENSOR_SET2", //GC2053
 "SENSOR_SET3", //GC4653
 "SENSOR_SET4", //MIS2008
 "SENSOR_SET5" //PS5258
]
},

```

Add your sensor into to the below structure. The ID is two for gc2053. The maximum support sensor size is nine.

```
"SENSOR_SET2": {
 "merge_en": true,
 "fcs_data": {
 "source": "binary",
 "file": "fcs_data_gc2053.bin"
 },
 "iq_data": {
 "source": "binary",
 "file": "iq.bin"
 },
 "sensor_data": {
 "source": "binary",
 "file": "sensor_gc2053.bin"
 }
},
```

Modify the file from sdk\project\realtek\_amebapro2\_v0\_example\inc\sensor.h

```
#define SENSOR_DUMMY 0x00 //For dummy sensor, no support fast camera start
#define SENSOR_SC2336 0x01
#define SENSOR_GC2053 0x02
#define SENSOR_GC4653 0x03
#define SENSOR_MIS2008 0x04
#define SENSOR_PS5258 0x05 //It don't support the multi sensor for PS5258 now.If you
want to use the sensor,please remove it.

#define MULTI_DISABLE 0x00
#define MULTI_ENABLE 0x01

#define MULTI_SENSOR MULTI_ENABLE
#define USE_SENSOR SENSOR_GC2053
```

## Revision History

| Date       | Version | Change        |
|------------|---------|---------------|
| 2021-09-30 | V01     | Initial draft |
| 2021-11-15 | V02     | File System   |
| 2022-03-23 | V03     | NN,BT         |
|            |         |               |
|            |         |               |