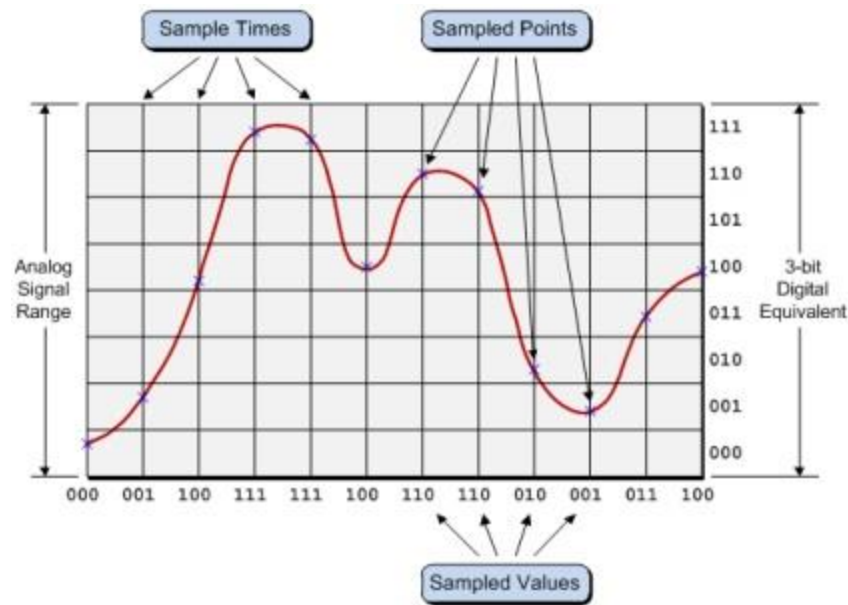


# INSIGHT OF AN AUDIO DRIVER BASED ON ALSA

Chandrasekar R

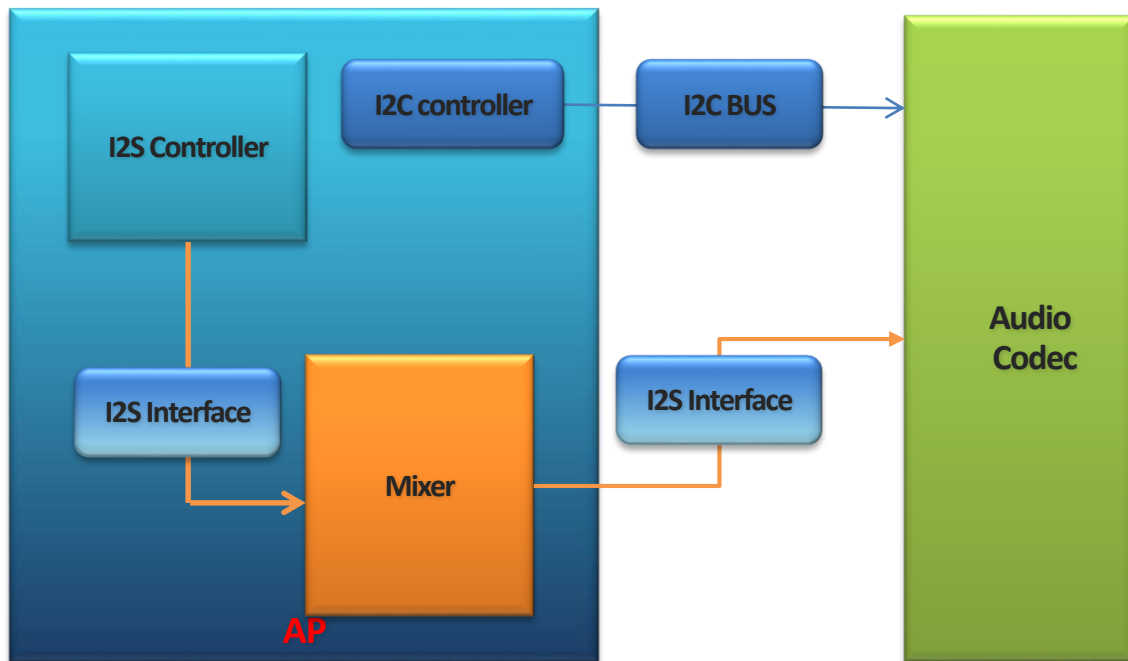
- ❑ Basics of Audio
- ❑ Audio Subsystem Overview
- ❑ Audio software architecture
- ❑ Android Audio System

- ❑ Analog Audio Data vs Digital Audio Data
- ❑ Conversion from Analog Audio Data to Digital Audio Data
  - ❑ Data at each discrete time is called Sample.
  - ❑ Number of samples per second
    - ❑ Sampling Rate, Frame rate (FS)
  - ❑ Number of bits per sample
    - ❑ Bit-Length Count (BLC)
- ❑ Channels
  - ❑ Mono, stereo, 5.1 channel etc

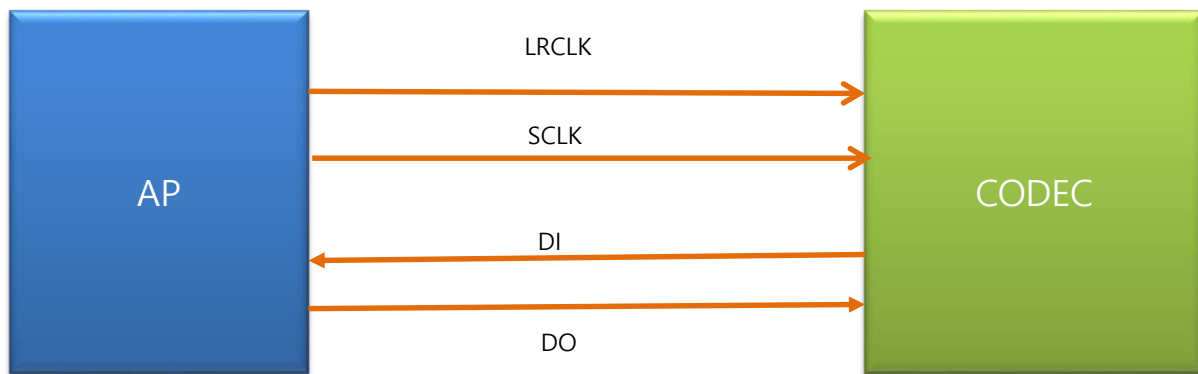


Audio Subsystem comprises of following blocks

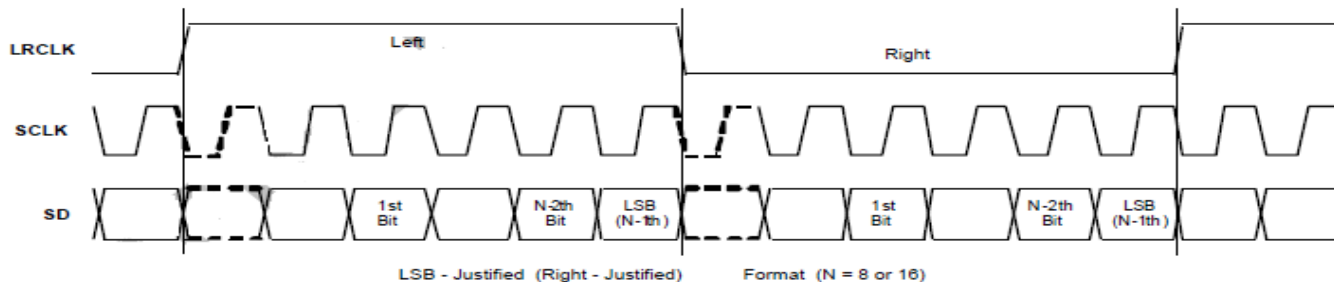
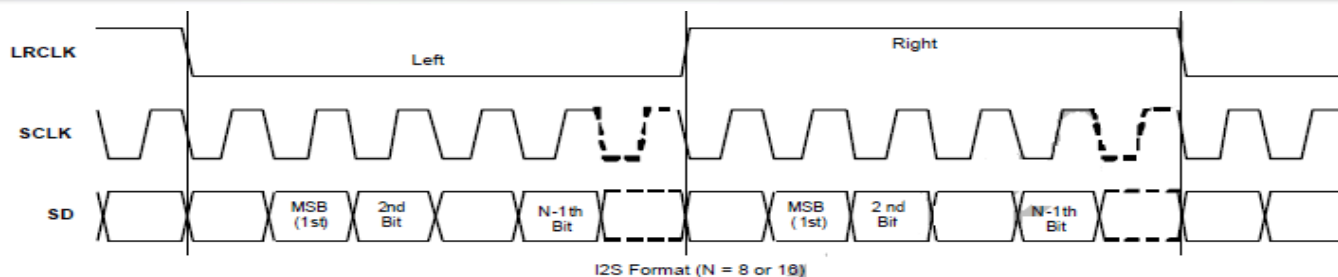
- ❑ I2S Interface
- ❑ Audio Subsystem Clock
- ❑ DMA
- ❑ I2C
- ❑ Codec (Analog to digital and Digital to analog converter)

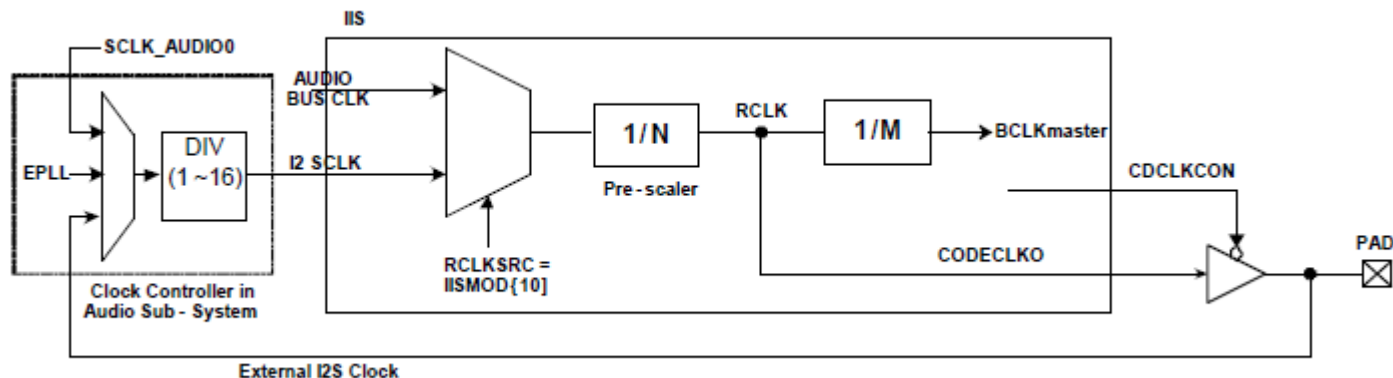


- ❑ I2S is a 3-line serial bus-interface
  - ❑ Word select line – LRCLK line
  - ❑ Clock line – SCLK line, bit-clock line
  - ❑ Data line – One line for two time-multiplexed channels



- ❑ 8/16/24 bits per sample
- ❑ Master/slave mode





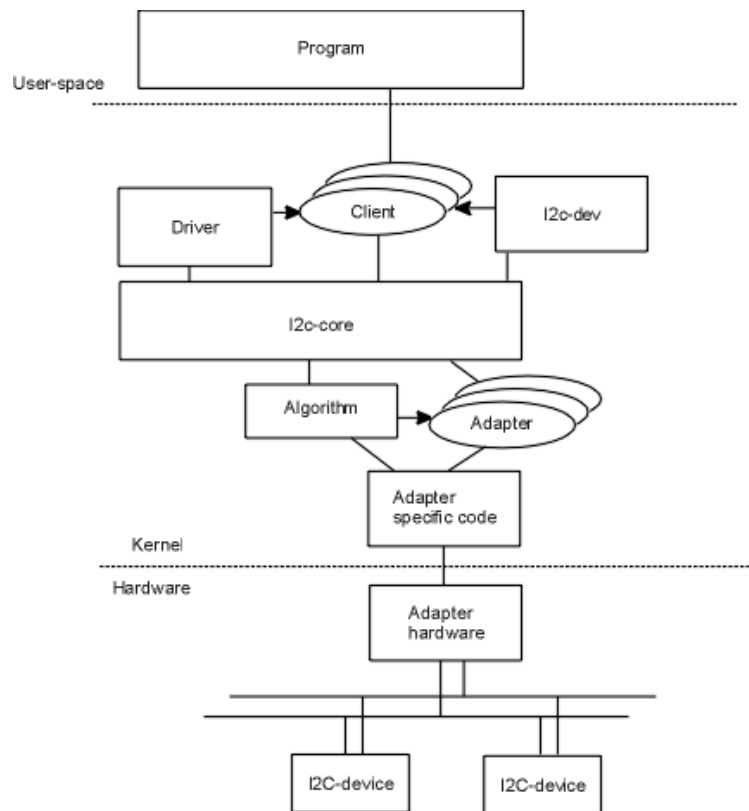
- ❑ Clock configuration is the important element in audio
- ❑ Based on the sampling rate appropriate root clock need to be configured
- ❑ The same clock need to be passed on to the codec if it takes from the external source
- ❑ Some codec generates the clock, in that case i2s block will take the root clock from the external audio codec

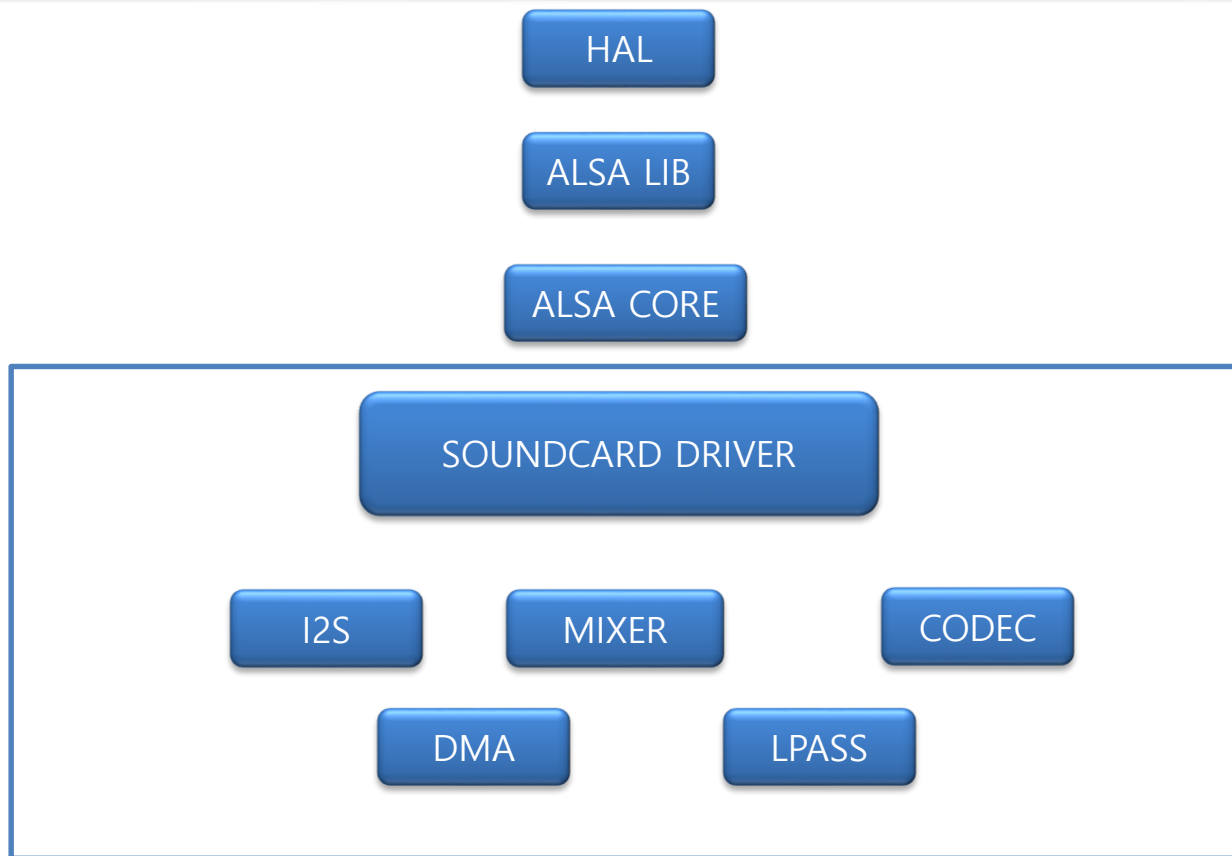


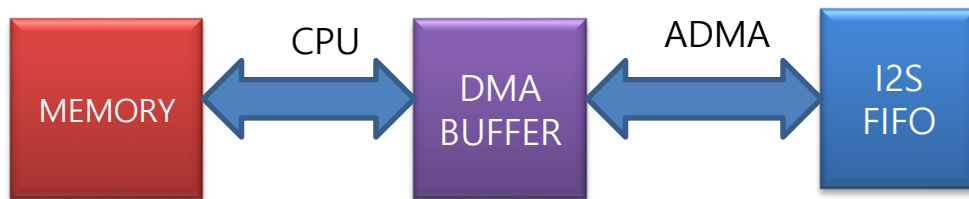
## 3 DMA channels normally used for mobile SoC

- ☐ TX Primary DMA
- ☐ RX DMA
- ☐ TX-Secondary DMA

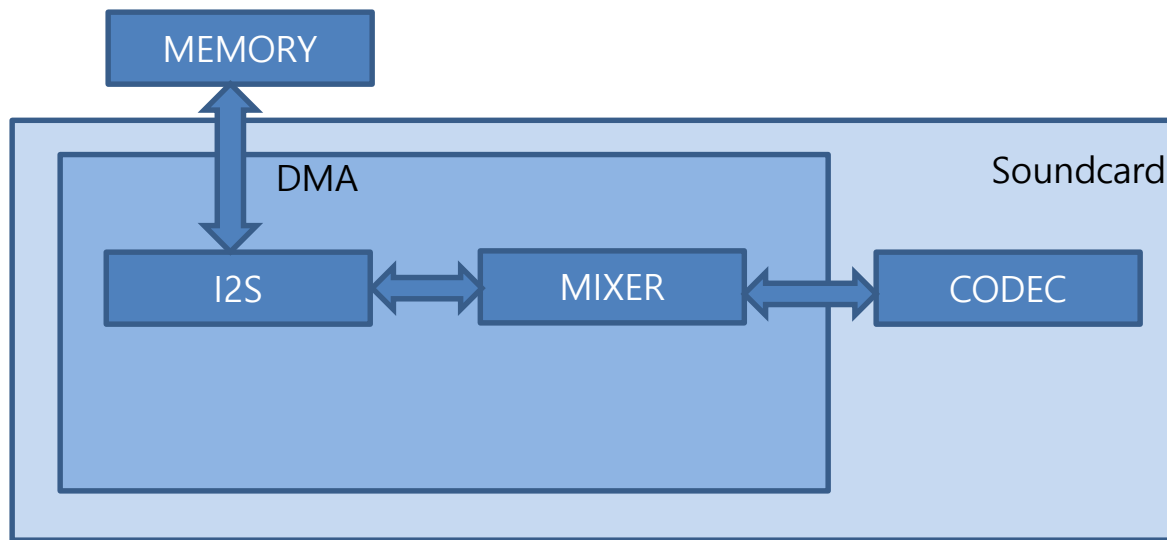
- ❑ Audio codec chip is controlled through I2C interface
- ❑ Audio codec driver registers itself as a client device to the I2C
- ❑ Codec driver used regmap to access the code registers through I2C



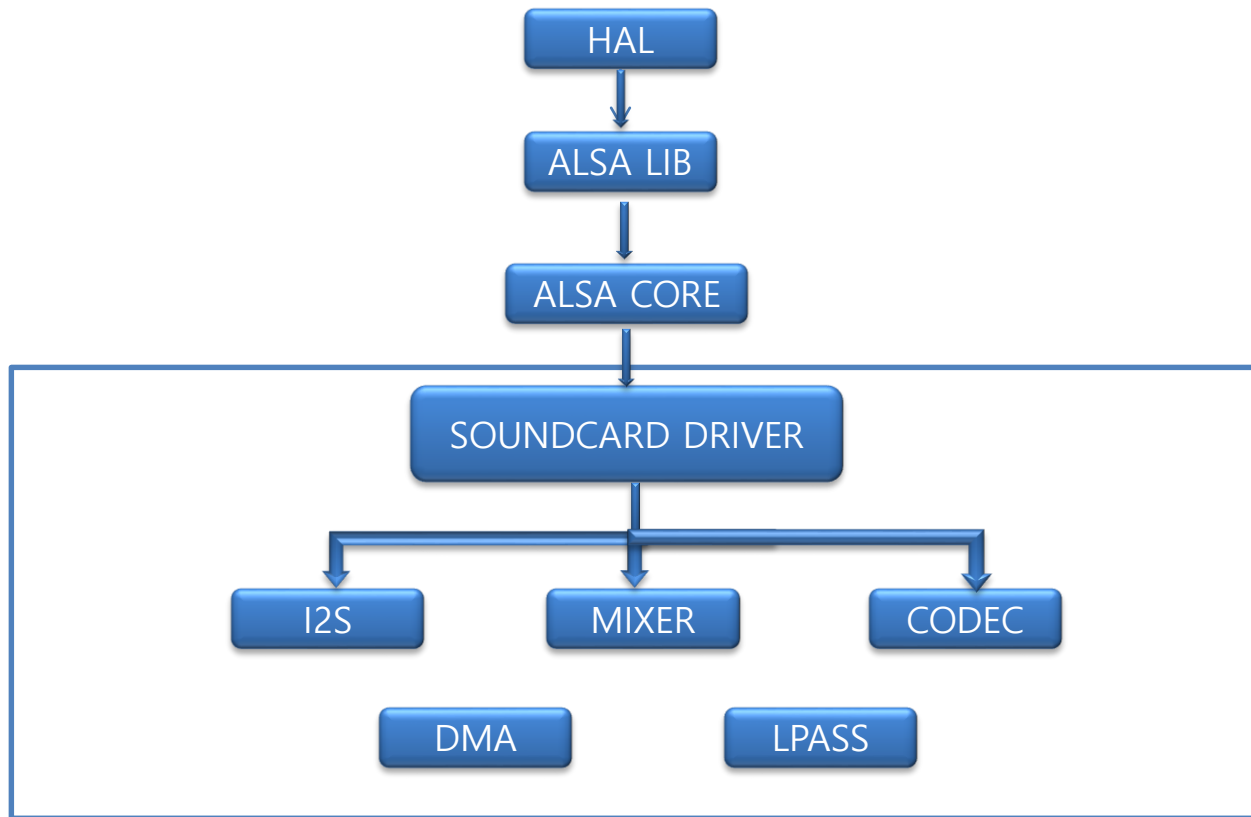




- ❑ CPU copies PCM data from memory in user space to DMA buffers
- ❑ ADMA channels copy from DMA buffer to I2S FIFO
- ❑ From I2S FIFO data is on I2S BUS



Data flow from memory to CODEC



- ❑ Each Interface functionality uses H/W as separate entity
- ❑ A separate device for each functionality is created

For Example:

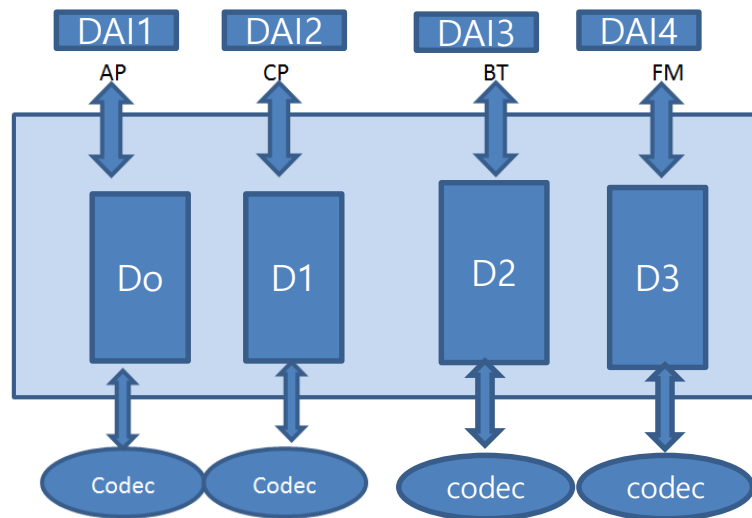
Like pcmC0D0c and pcmc0d0p –

c0 –card0

D0- device 0

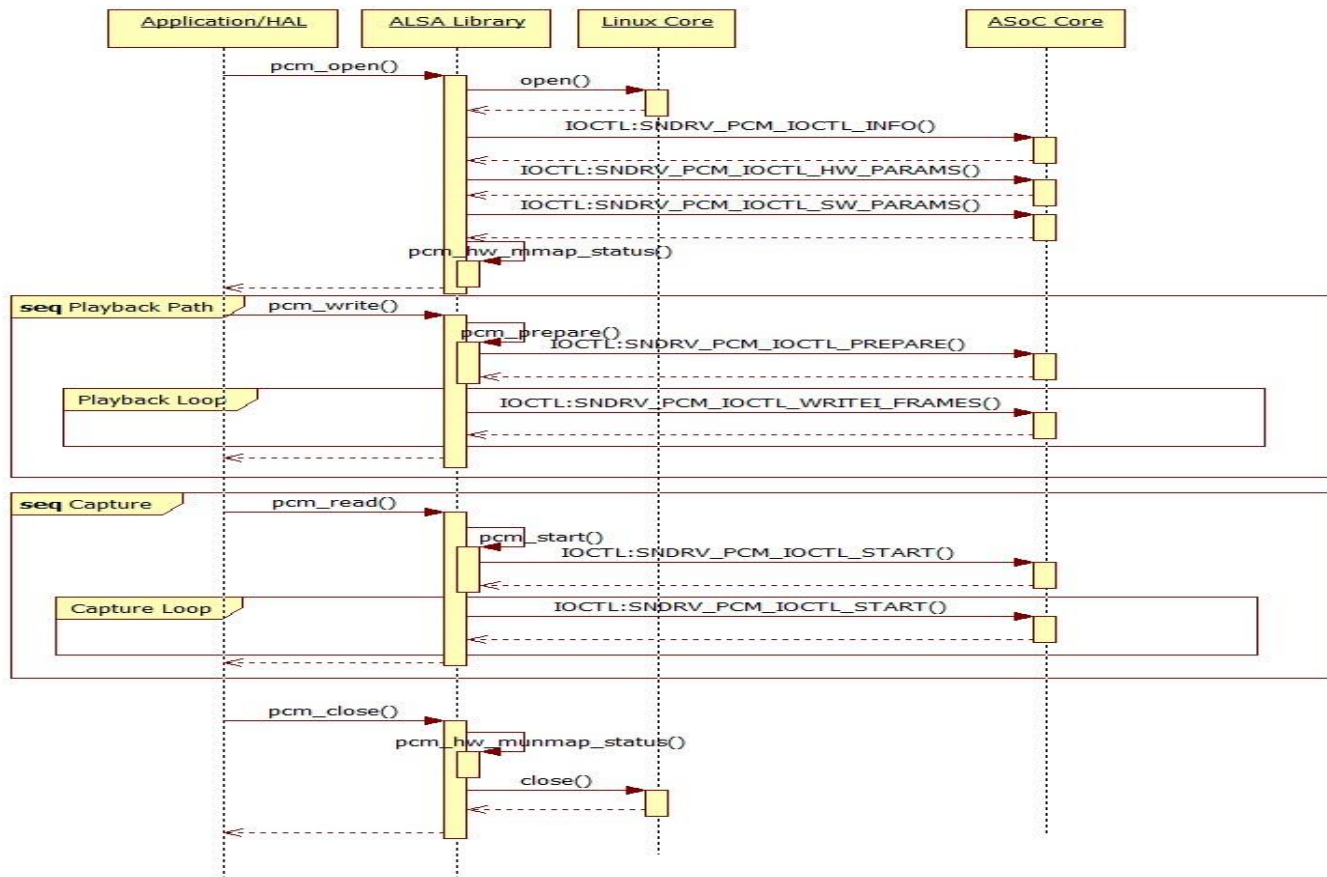
P for playback and c for capture

- ❑ After probe separate PCM DEVICES registered for each DAI-LINK

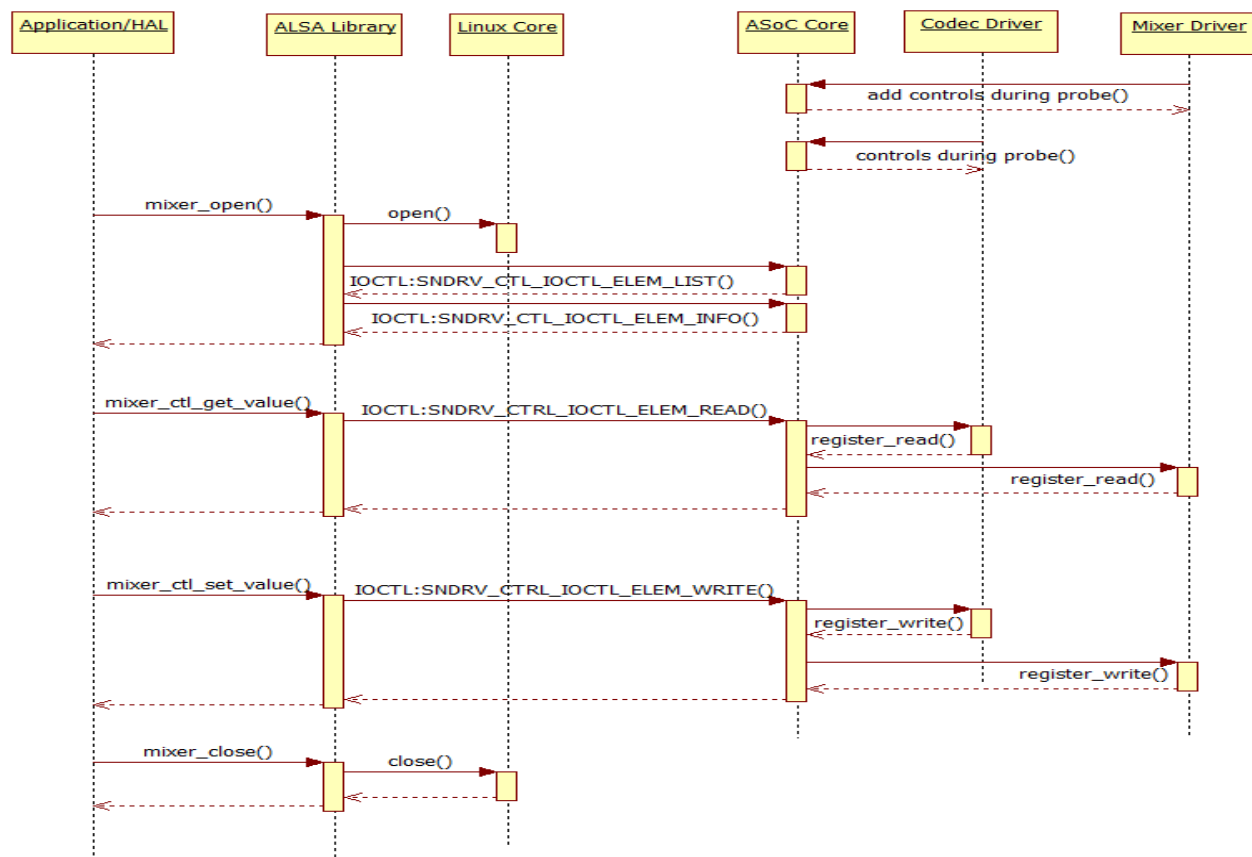




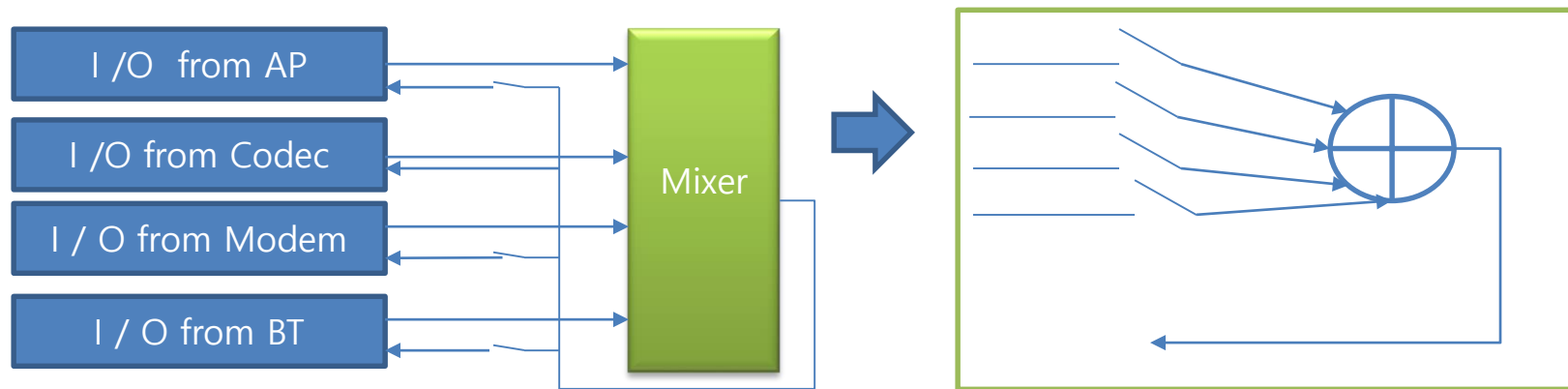
# Sequence Diagram – Playback/Capture



# Sequence Diagrams – Mixer OPs

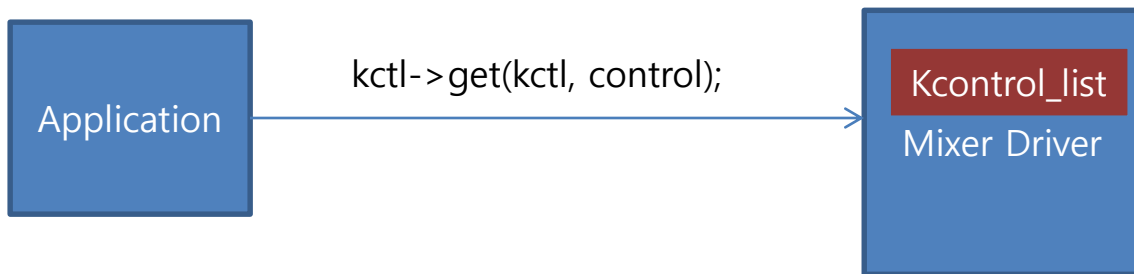


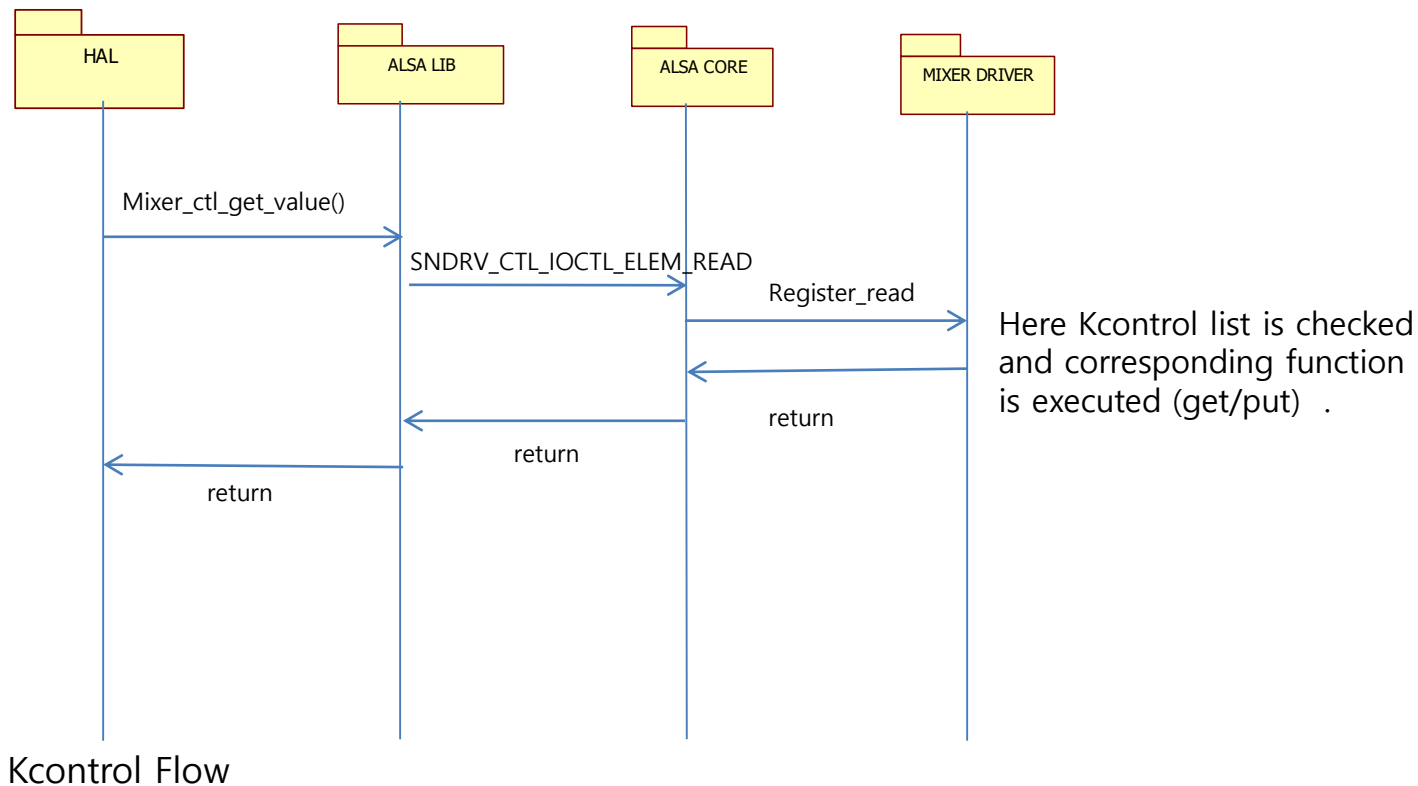
- ❑ Mixer is a hardware device mixes the digital audio data from different sources and send backs to them based on the selection



## Kcontrols in Mixer

- Mixer driver exposes some controls for upper application layer though **Kcontrols**
- These controls are used to change SFRs of mixer which is done through regmap





Power ON/ OFF sequence of respective paths gets when playback or capture stream opened

- ❑ Playback Paths

  - Speaker On /Off**

  - Headphone On /Off**

  - Earpiece(receiver) On /Off**

- ❑ Capture paths

  - MIC1 On /Off**

  - MIC2 On /Off**

- ❑ Gain controls

  - DAC, ADC Digital gain control**

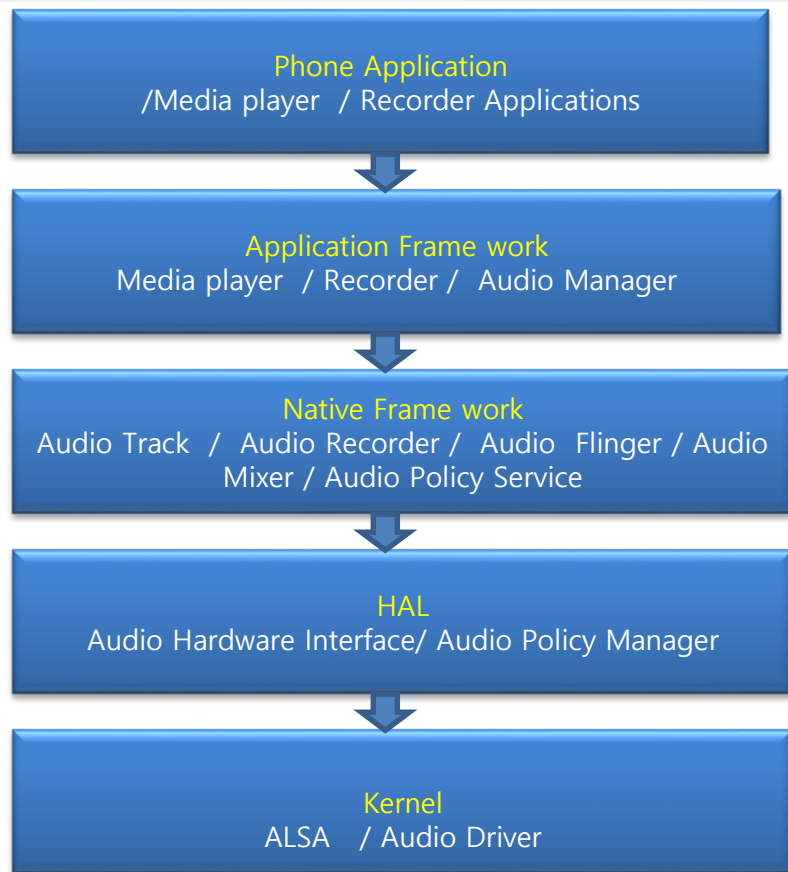
  - MIC1, MIC2, analog gain control**

  - SPK,EP, HP analog gain controls**

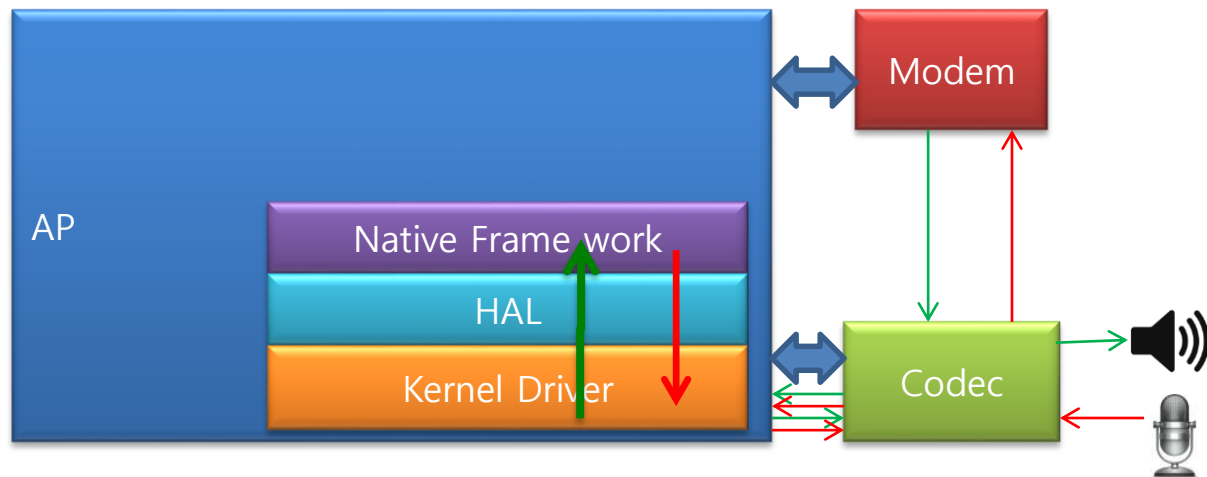
- ❑ Mixer path selection

  - MIC input data mixing can be selected through ADC mixer Controls**

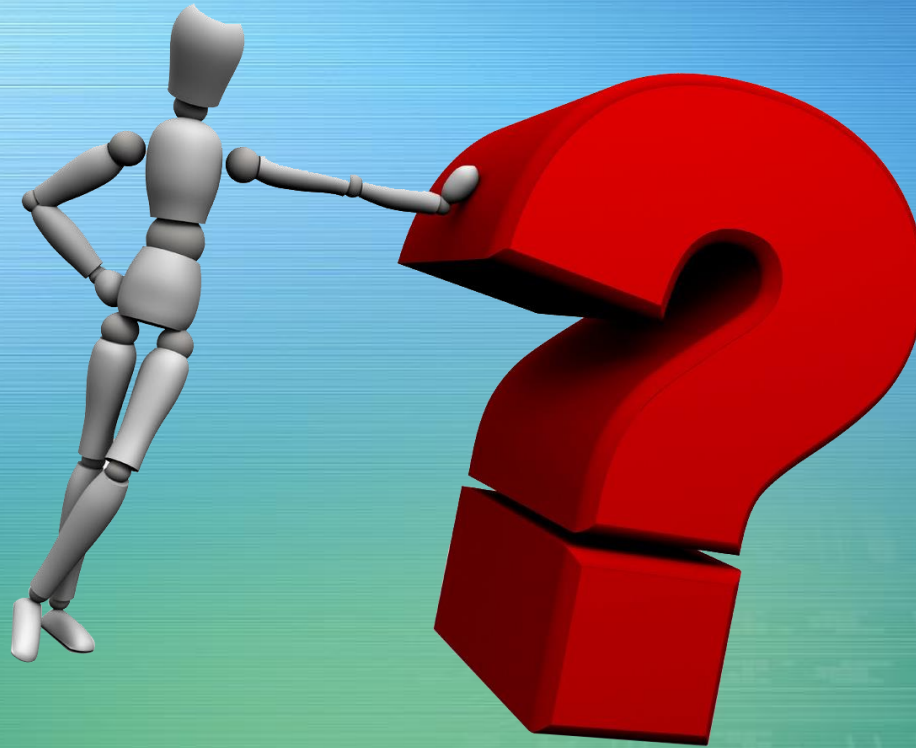
## MOBILE USECASE of AUDIO DRIVER







# Any Questions ?



# THANK YOU