

## ALSA and DAHDI Audio Architecture Analysis

### ALSA Audio Driver Logic

The ALSA framework controls the timing of audio data exchange using two core mechanisms: `pointer()` and `snd_pcm_period_elapsed()`.

Mono channel (1 channel), 16-bit samples (S16\_LE)

✓ 1 frame = 2 bytes

#### Step 1: Period Elapsed Notification

The driver periodically calls `snd_pcm_period_elapsed()`. This can be triggered by a timer or hardware interrupt. This tells ALSA that a period (e.g., 20ms of audio) has been processed and requests ALSA to continue.

#### Step 2: ALSA Calls `pointer()`

In response, ALSA invokes `substream->ops->pointer(substream)`, which returns the current frame position (i.e., where the DMA has reached). This is not a period count, but rather a pointer in the ring buffer (unit: frame).

Example:  $pos = (\text{delta\_ms} * \text{runtime->rate}) / 1000$

#### Step 3: Determine if a new period has elapsed

ALSA compares the current and previous pointer values:

$(pos - \text{last\_pos} + \text{buffer\_size}) \% \text{buffer\_size} \geq \text{period\_size}$

If the condition is true, a new period has elapsed.

✓ ALSA then calls `substream->ops->copy_user()`

This pulls audio data from userspace and invokes:

play

`copy_from_user(kbuf, user_buf, bytes);`

`write_pcm_20ms(kbuf, bytes);`

Record

`read_pcm_20ms(kbuf, bytes);`

`copy_to_user(user_buf, kbuf, bytes);`

## ALSA and Hardware Timing and Buffering

In a typical system, timing mechanisms differ across ALSA, hardware, and userspace applications:

Module	Timing Source
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ALSA kernel layer	`snd_pcm_timer_function()`, jiffies, hrtimer
Hardware driver	DMA + Interrupt + Hardware Clock
Userspace program	Relies on `poll()` or `select()` (no timer)

Typical use of ring buffers:

Name	Writer	Reader
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RX ringbuf	Hardware	ALSA
TX ringbuf	ALSA	Hardware

Concepts:

Overrun: Writing too fast, overwriting unread data.

Underrun: Reading too fast, no data available.

RX Ring Buffer:

Overrun → Data loss, frame drop

Underrun → Short reads, I/O errors

TX Ring Buffer:

Underrun → Clicks, hardware faults

Overrun → Frame drop (acceptable in moderation)

⚠ Important Notes:

1. On `SNDRV\_PCM\_TRIGGER\_START/STOP`, ring buffers and pointer state must be reset.

2. For echo cancellation (EC/AEC), ring buffer length must be < EC window:

EC window: 128ms → buffer ≤ 2048B (2KB)

AEC window: 200–1000ms → buffer ≤ 4096B (4KB)

## DAHDI Audio Driver Logic

DAHDI architecture is similar to ALSA in terms of driver design but uses different APIs.

ALSA exchange interface:

`snd_pcm_copy_user(substream, channel, pos, user_buf, count);`

DAHDI exchange interface:

dahdi\_transmit(&d\_span); → pushes writechunk  
 bcm\_pull\_dahdi\_chunk() → writechunk → TX ringbuf  
 bcm\_push\_dahdi\_chunk() → RX ringbuf → readchunk  
 dahdi\_receive(&d\_span); → pull readchunk to DAHDI core

RX direction (capture):  
 Hardware → RX ringbuf → bcm\_push\_dahdi\_chunk() → readchunk → DAHDI

TX direction (playback):  
 DAHDI → writechunk → bcm\_pull\_dahdi\_chunk() → TX ringbuf → Hardware

Comparison between ALSA and DAHDI:

Aspect	ALSA	DAHDI
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User interface	snd_pcm_copy_user()	dahdi_transmit()/receive()
Ring buffer	RX/TX ringbuf	Same
Data flow	user buf ↔ ringbuf ↔ hardware	writechunk/readchunk ↔ ringbuf ↔ hardware

Note: Both use ringbufs for hardware-driver interaction. Only the interface layer differs.