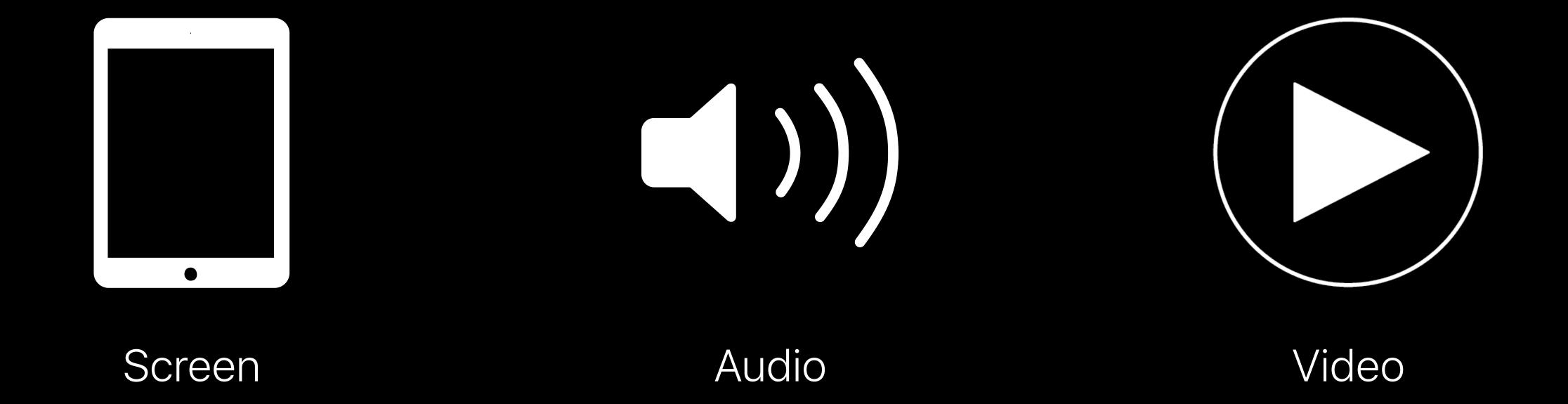
Media #WWDC17

Introducing AirPlay 2 Unlocking multi-room audio

Session 509

David Saracino, AirPlay Engineer

AirPlay Overview

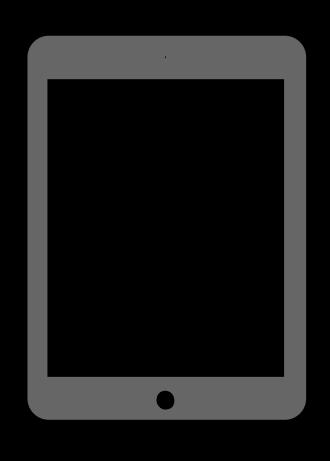


AirPlay Overview

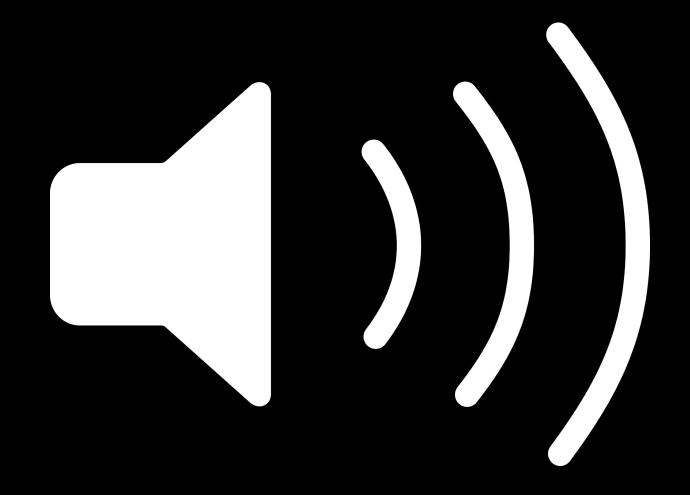


AirPlay Overview

AirPlay 2







Audio

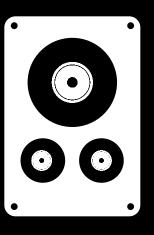


Video

Audio

Wireless audio



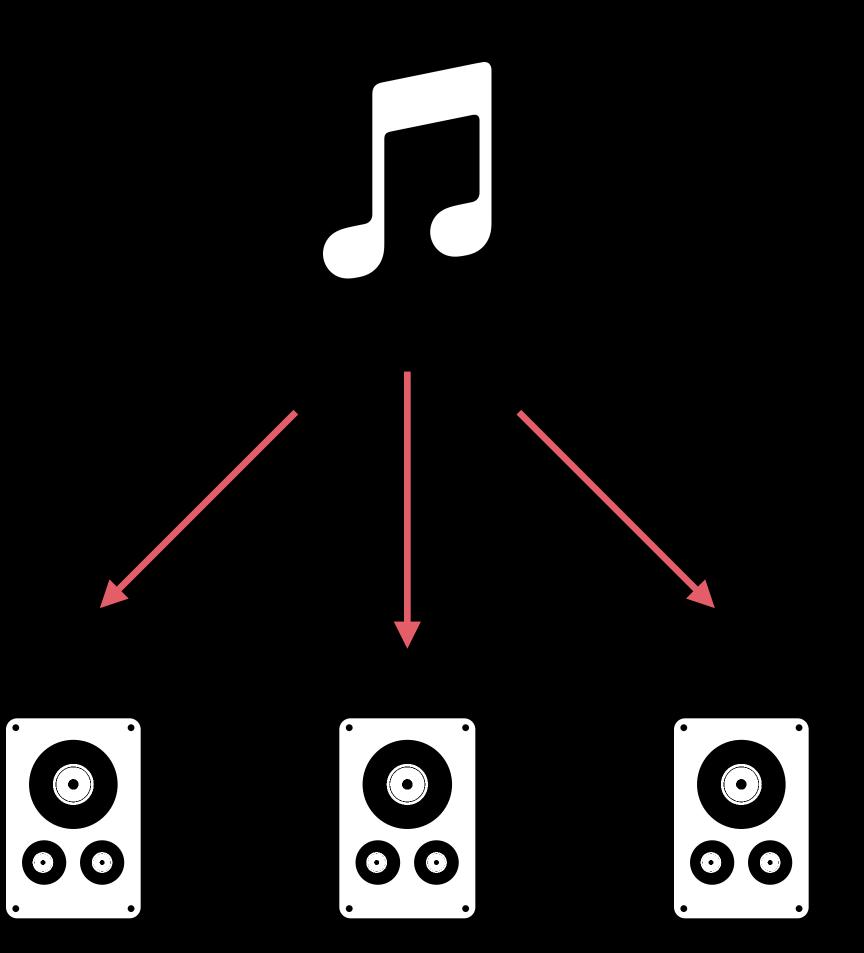


Audio



Wireless audio

Multi-room playback



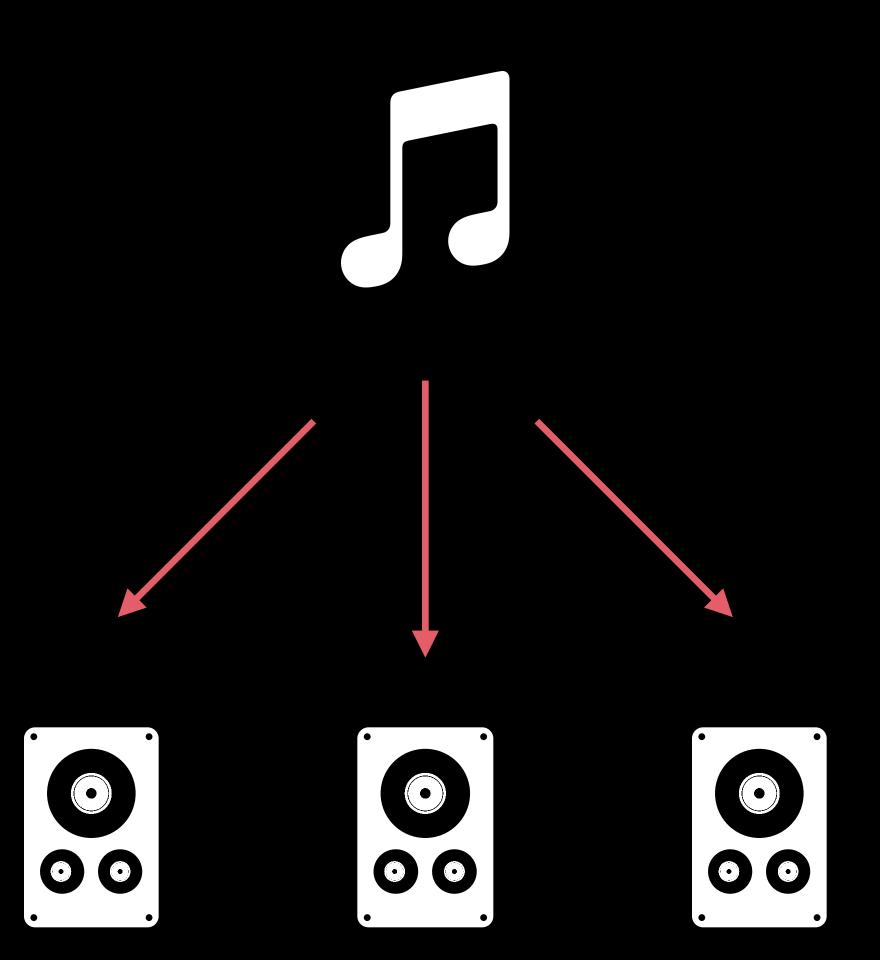
Audio

NEW

Wireless audio

Multi-room playback

Enhanced buffering



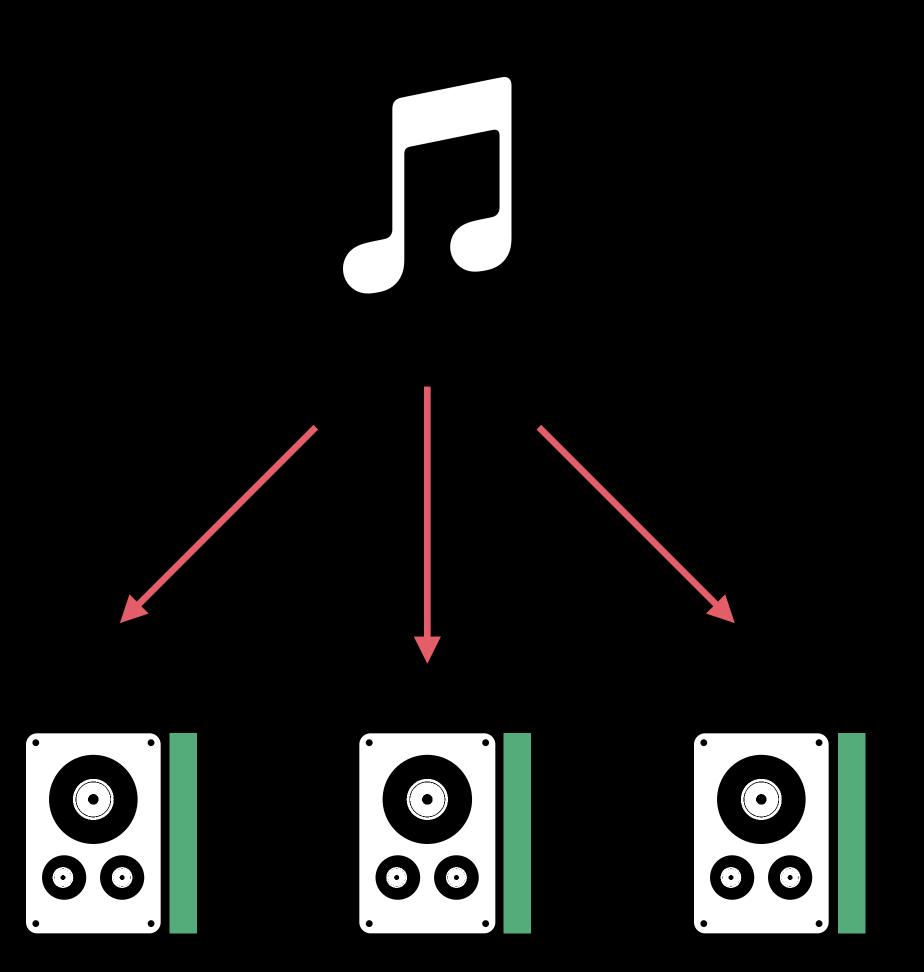
Audio

NEW

Wireless audio

Multi-room playback

Enhanced buffering



Audio

Wireless audio

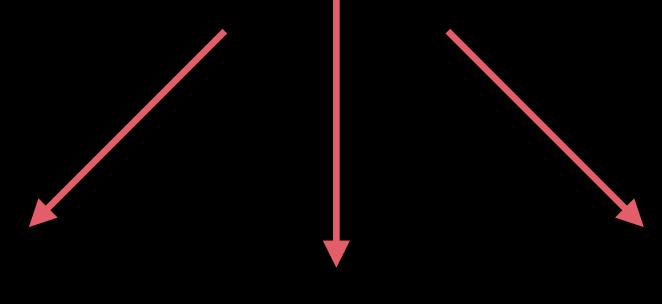
Multi-room playback

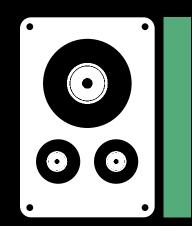
Enhanced buffering

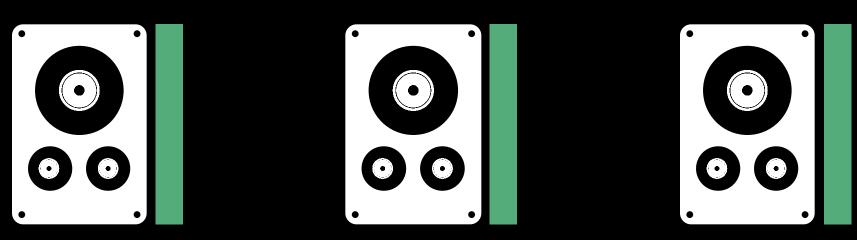
Multi-device control













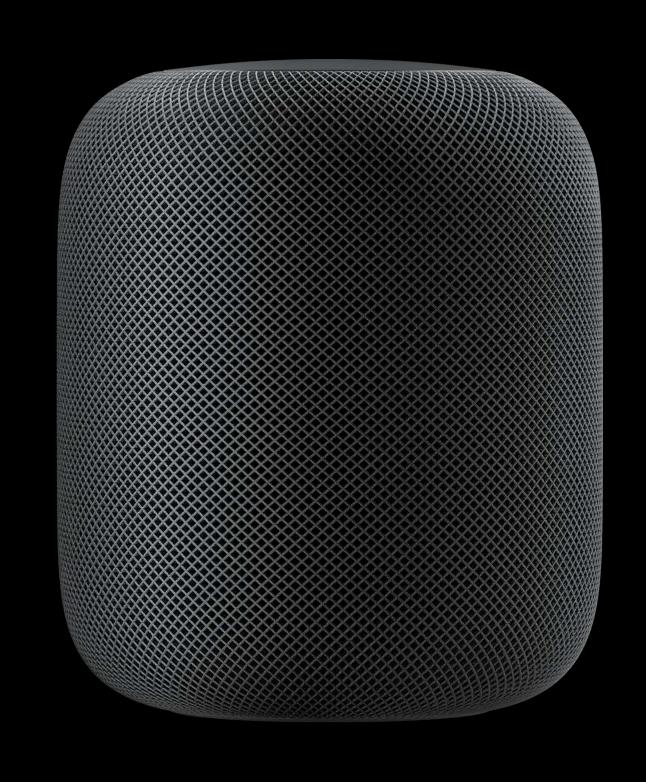


AirPlay 2 Supported platforms

tvOS



AirPlay 2 Supported speakers



HomePod



Apple TV*



3rd Party

^{*} Requires Apple TV (4th generation)

Advanced playback scenarios

Availability

Identify as long-form audio

Identify as long-form audio

Add an AirPlay picker

Identify as long-form audio

Add an AirPlay picker

Integrate with MediaPlayer framework

Identify as long-form audio

Add an AirPlay picker

Integrate with MediaPlayer framework

Adopt an AirPlay 2 playback API





Music, podcasts, or audiobooks



Music, podcasts, or audiobooks

Set AVAudioSession route sharing policy



Music, podcasts, or audiobooks

Set AVAudioSession route sharing policy



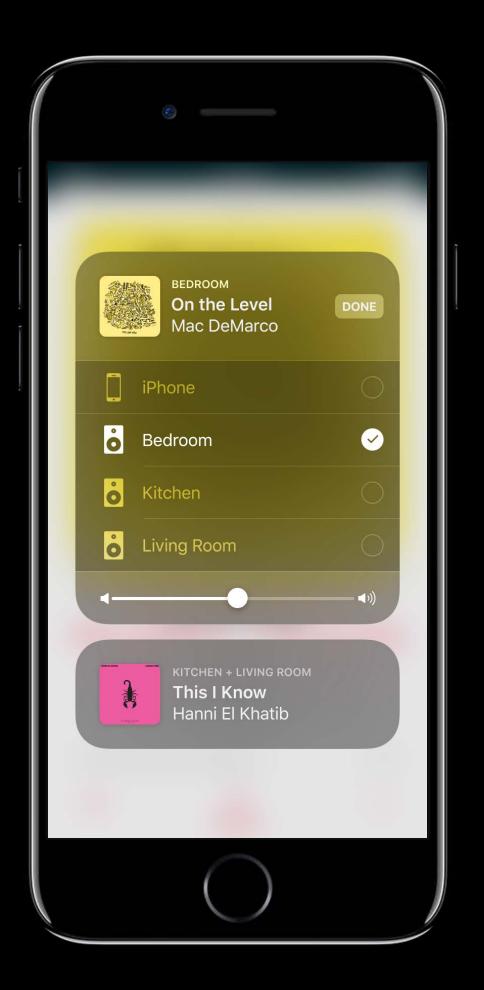
Music, podcasts, or audiobooks

Set AVAudioSession route sharing policy

What's New in Audio WWDC 2017



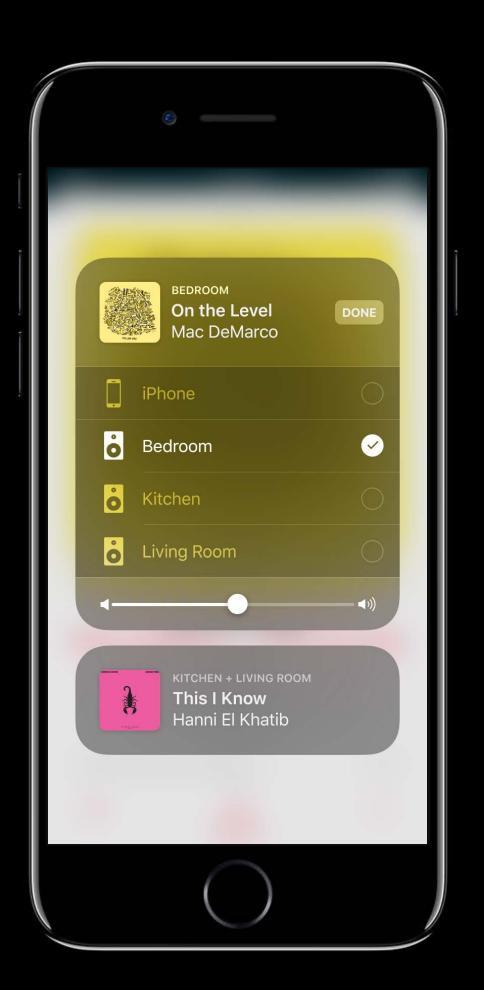
Adopt





Adopt

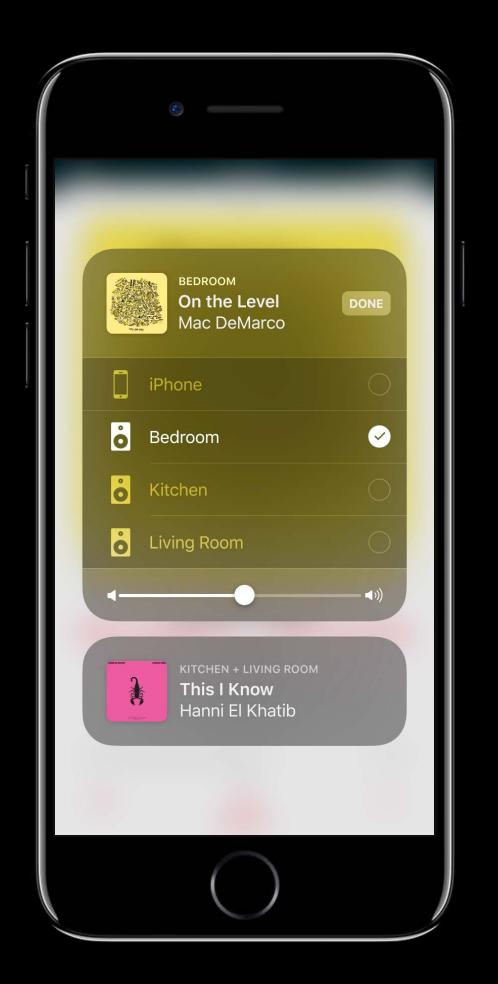
AVRoutePickerView





Adopt

- AVRoutePickerView
- AVRouteDetector



Integrate with Media Player Framework



Integrate with Media Player Framework

Handle remote media commands

MPRemoteCommandCenter



Integrate with Media Player Framework

Handle remote media commands

MPRemoteCommandCenter

Display current track info

MPNowPlayingInfoCenter



AirPlay 2 Playback APIs

Existing AirPlay

Existing AirPlay

Real-time audio stream

Existing AirPlay

Real-time audio stream

Speaker adds a small buffer before output

Existing AirPlay

Real-time audio stream

Speaker adds a small buffer before output

Works fine for streaming to single speaker

Enhanced Audio Buffering

AirPlay 2



AirPlay 2



Large audio buffering capacity on speakers

AirPlay 2



Large audio buffering capacity on speakers

Faster-than-real-time streaming to speakers

AirPlay 2



Large audio buffering capacity on speakers

Faster-than-real-time streaming to speakers

Benefits

AirPlay 2



Large audio buffering capacity on speakers

Faster-than-real-time streaming to speakers

Benefits

Adds robustness

AirPlay 2



Large audio buffering capacity on speakers

Faster-than-real-time streaming to speakers

Benefits

- Adds robustness
- More responsive playback

Adoption

Supported with specific playback APIs

Enhanced Audio Buffering Adoption

Supported with specific playback APIs

Enhanced Audio BufferingAdoption

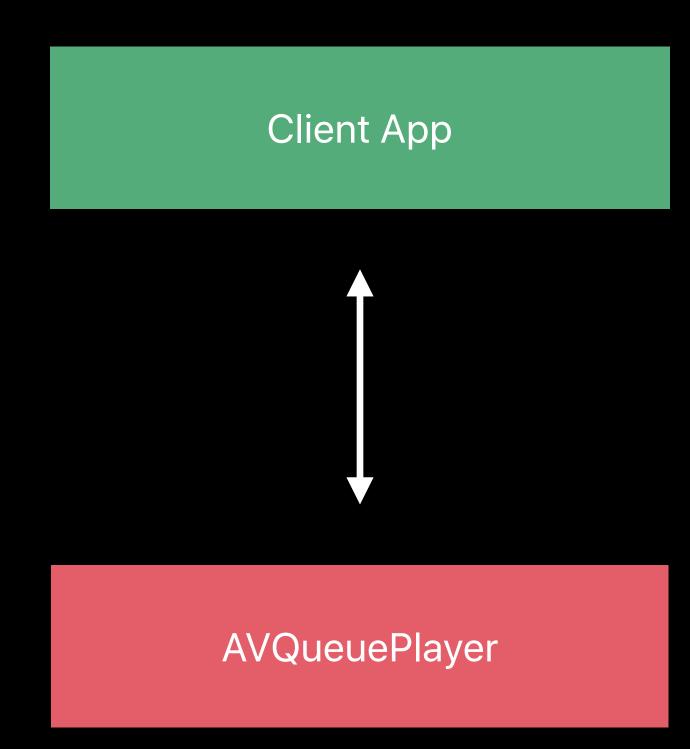


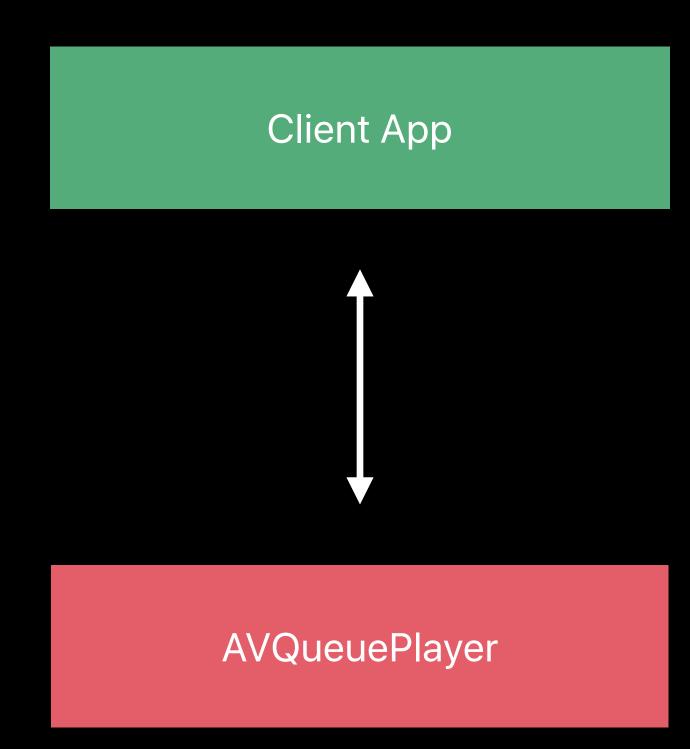
Supported with specific playback APIs

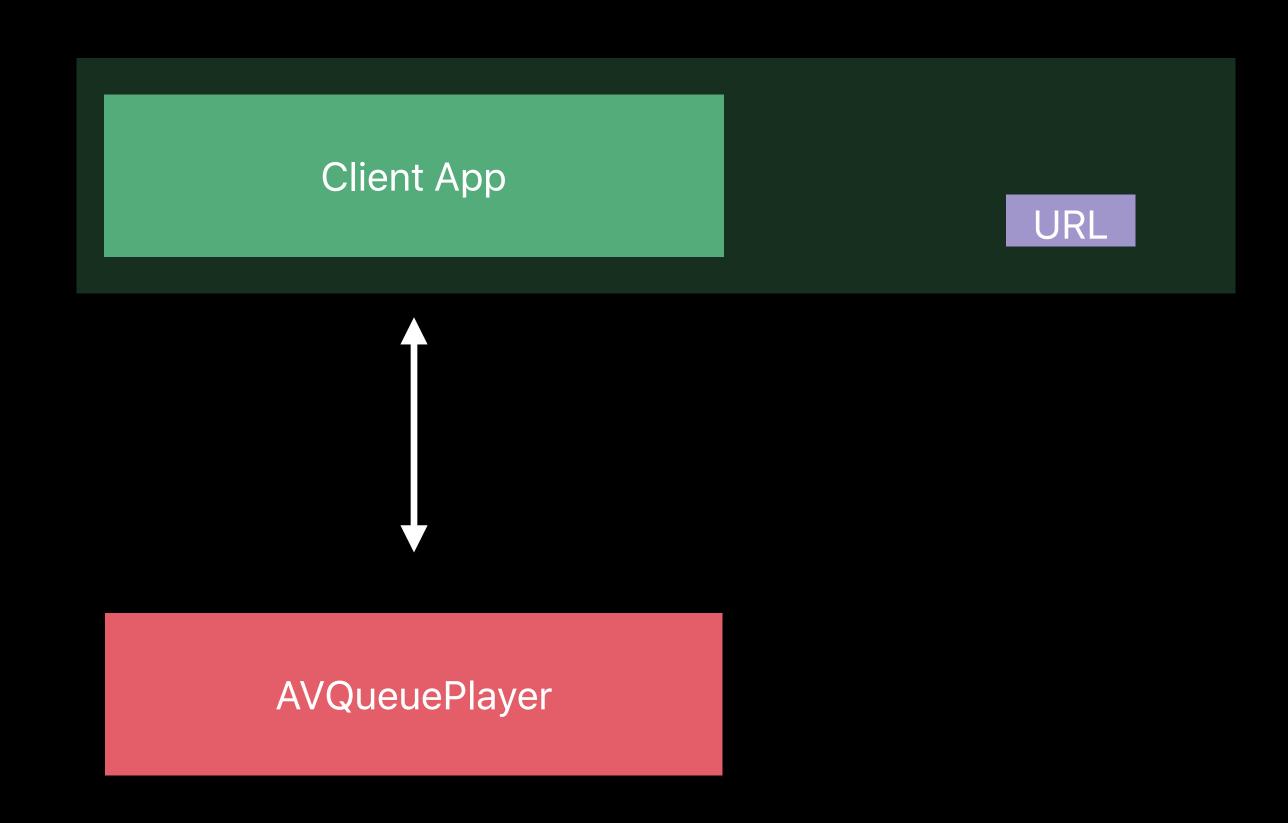
- AVPlayer / AVQueuePlayer
- AVSampleBufferAudioRenderer
 - AVSampleBufferRenderSynchronizer

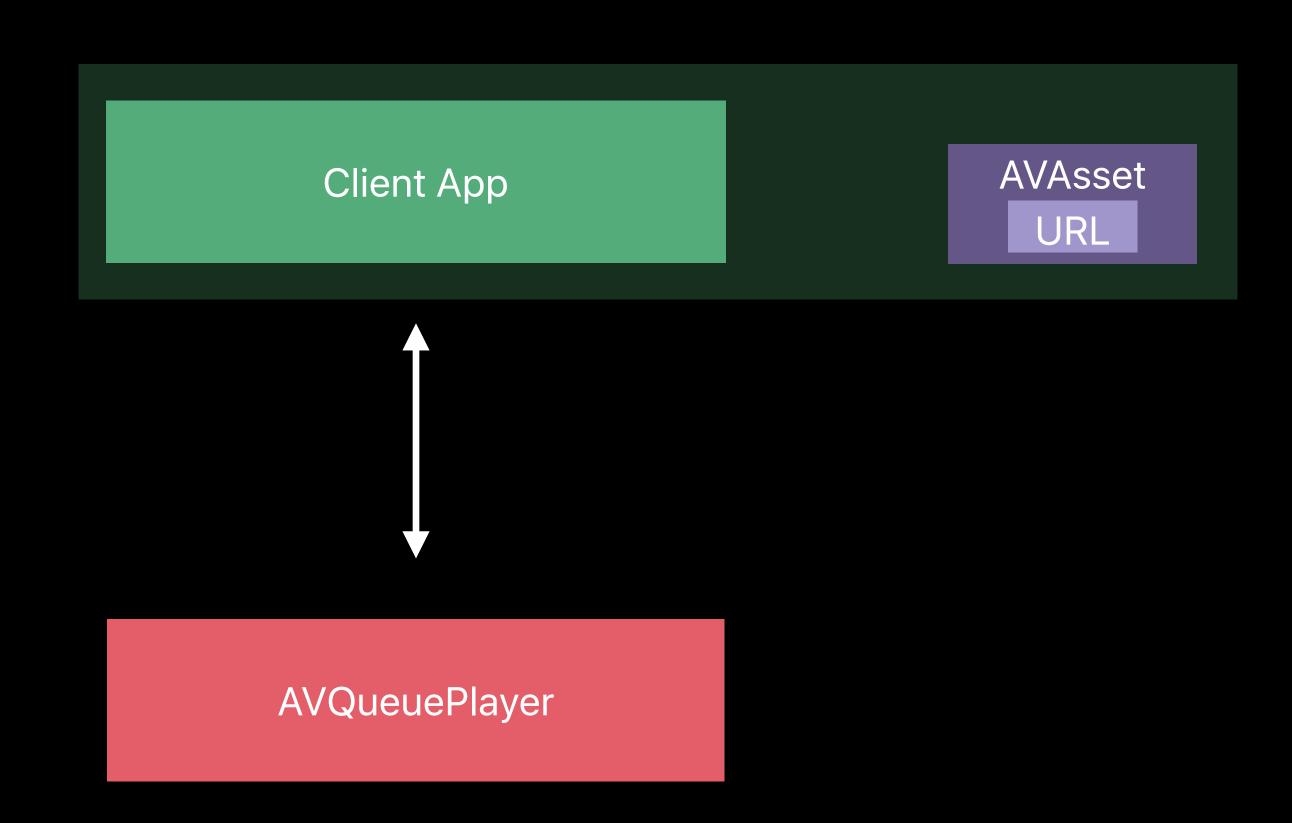
AVPlayer / AVQueuePlayer

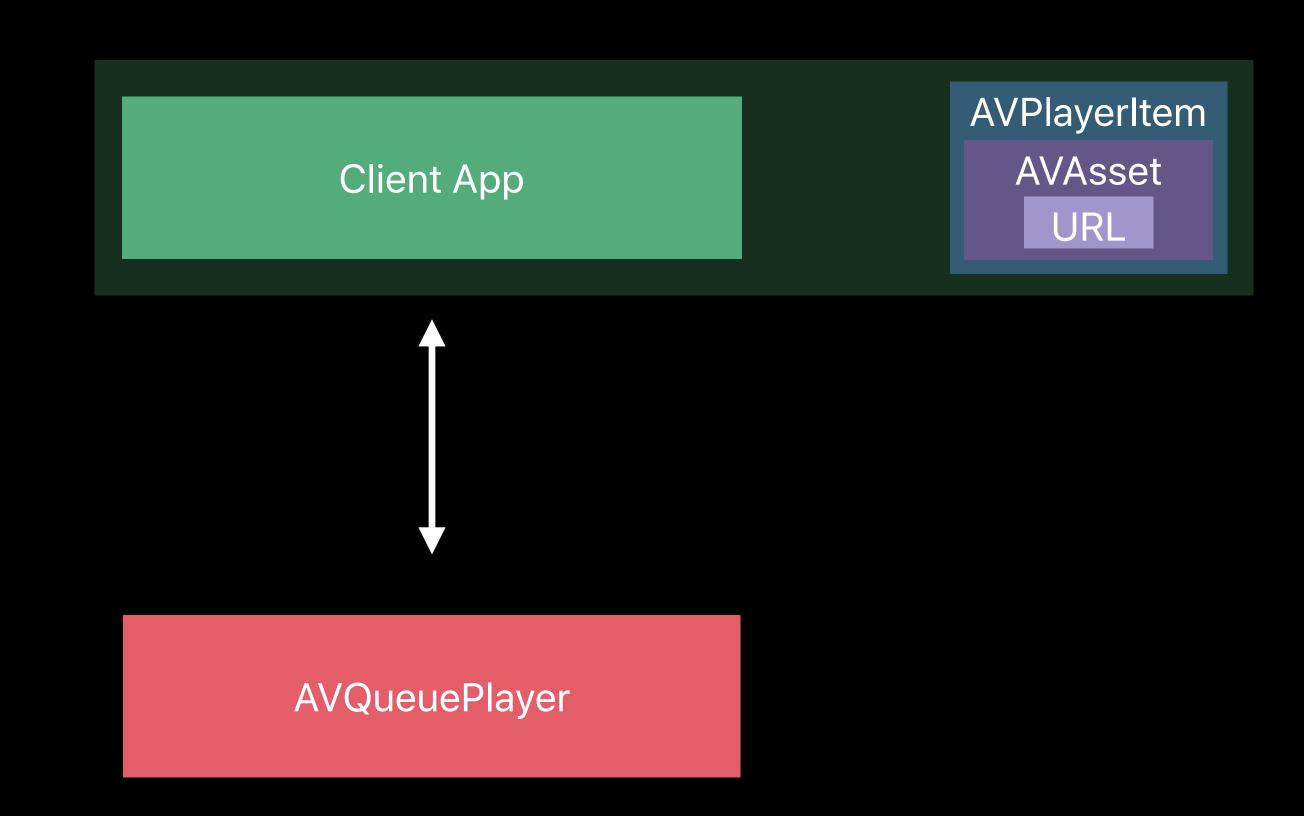
Client App

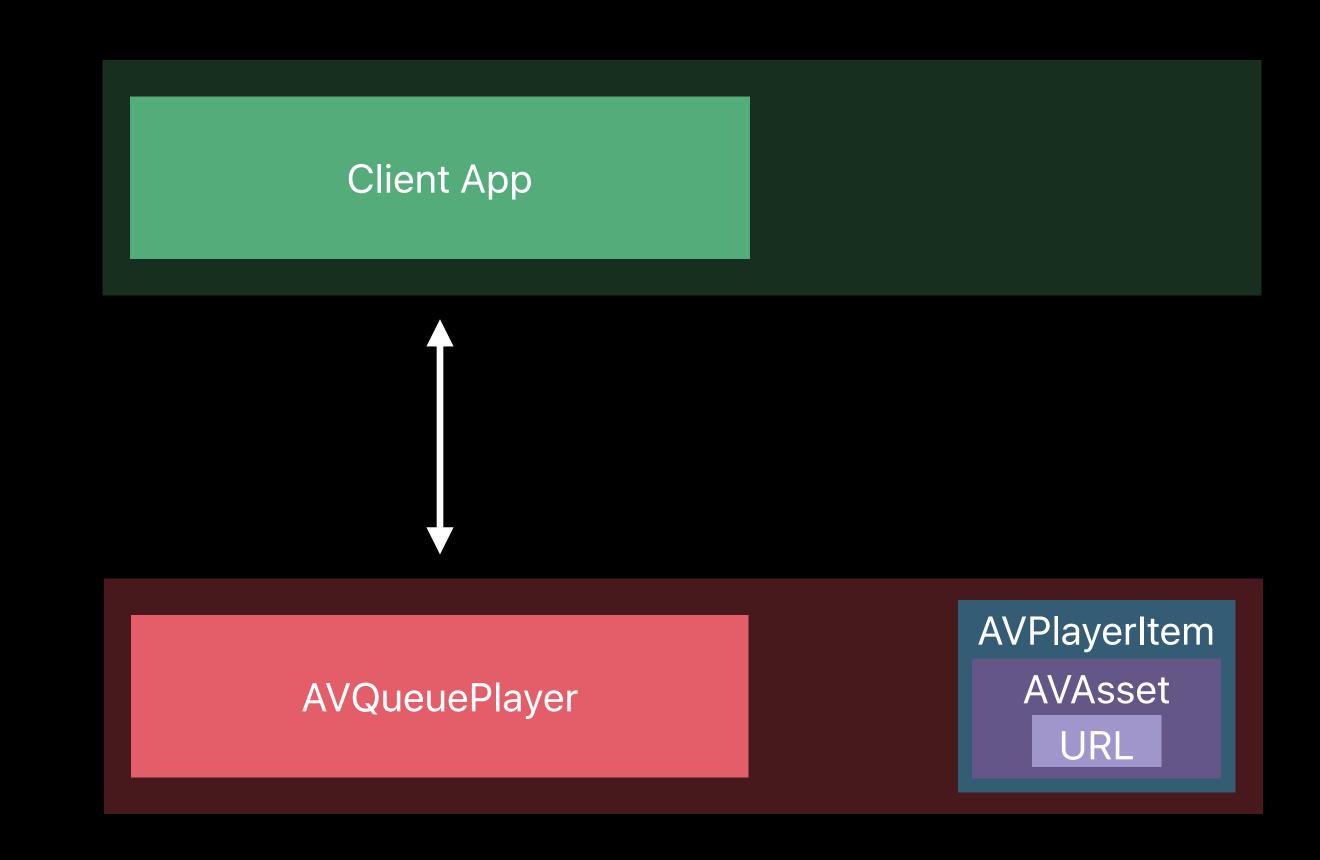


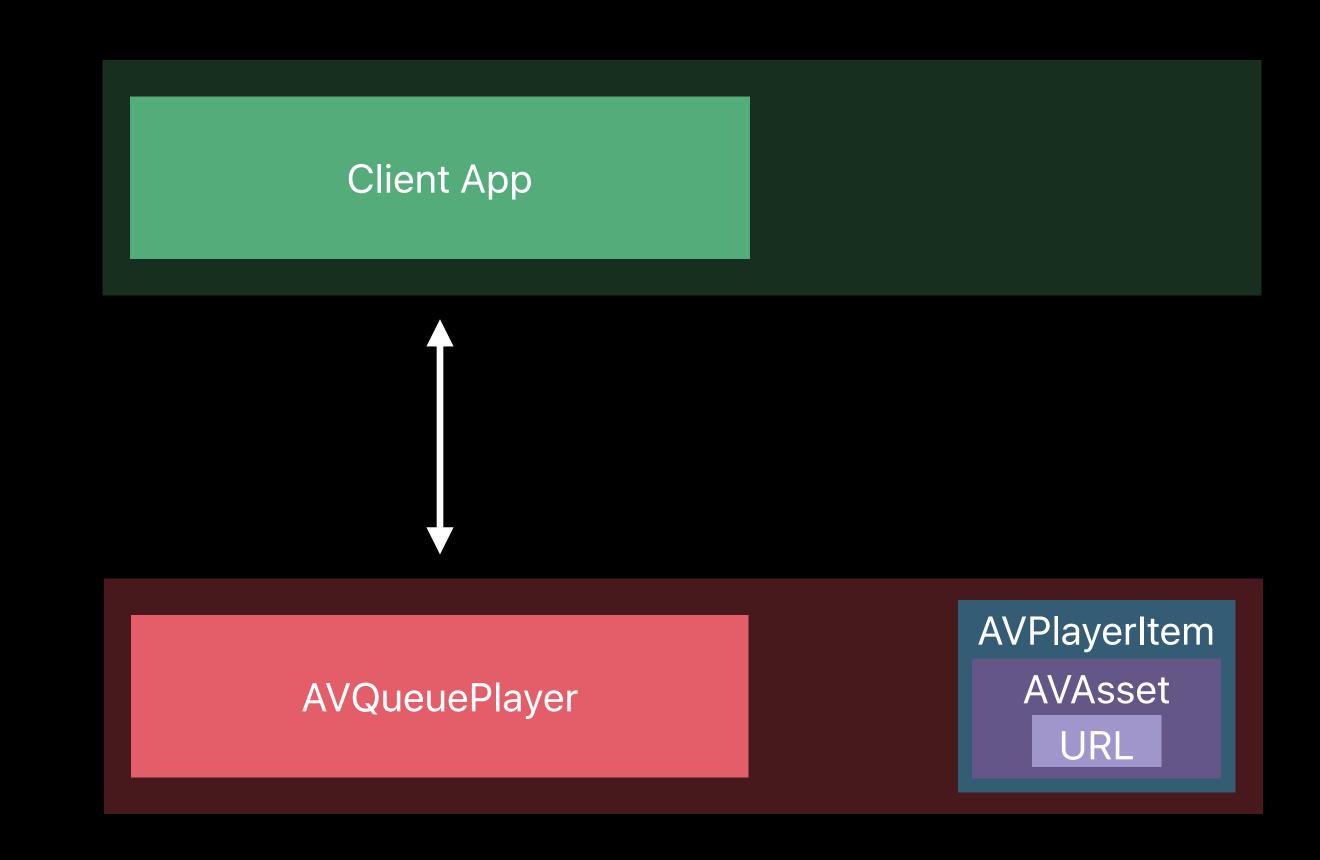


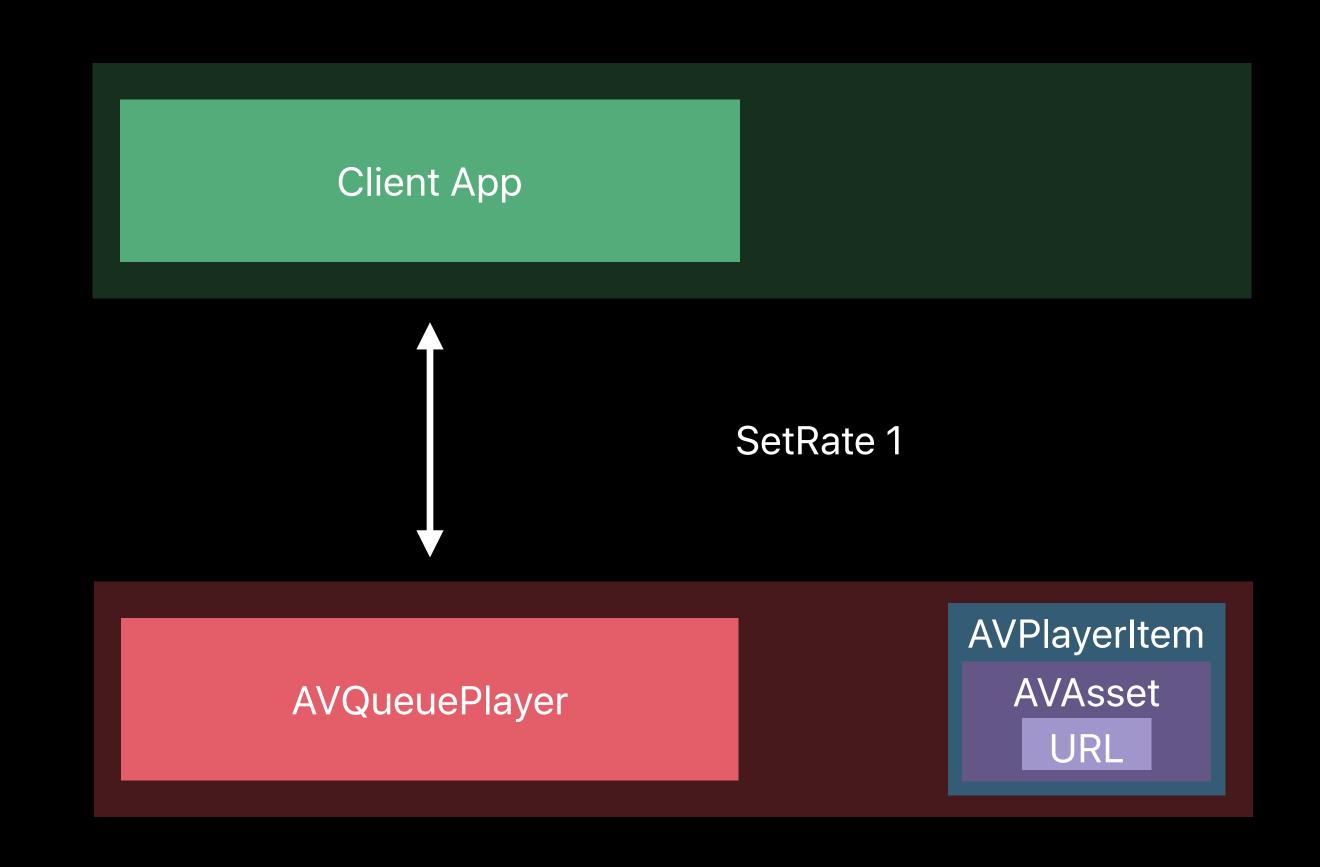


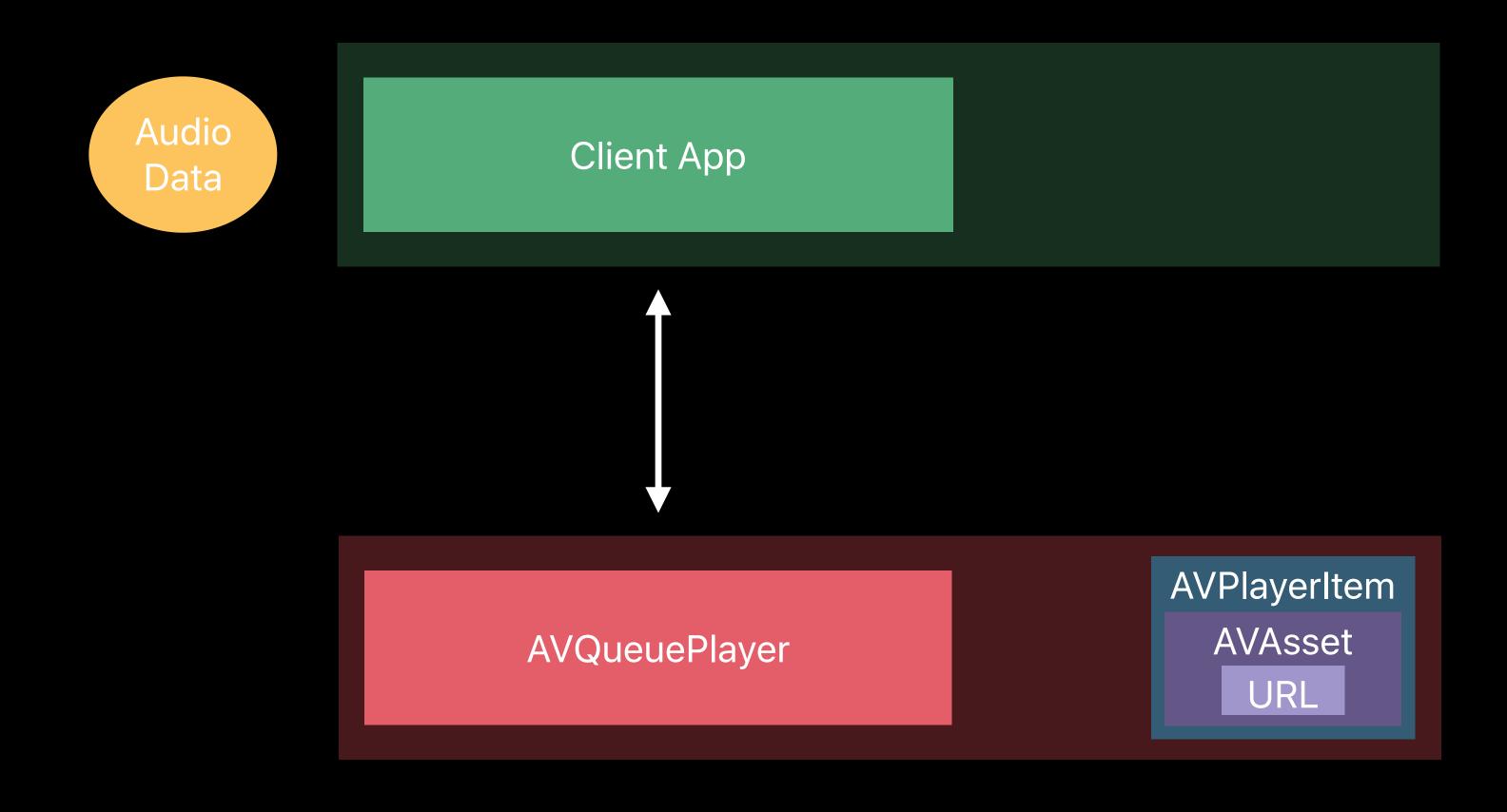


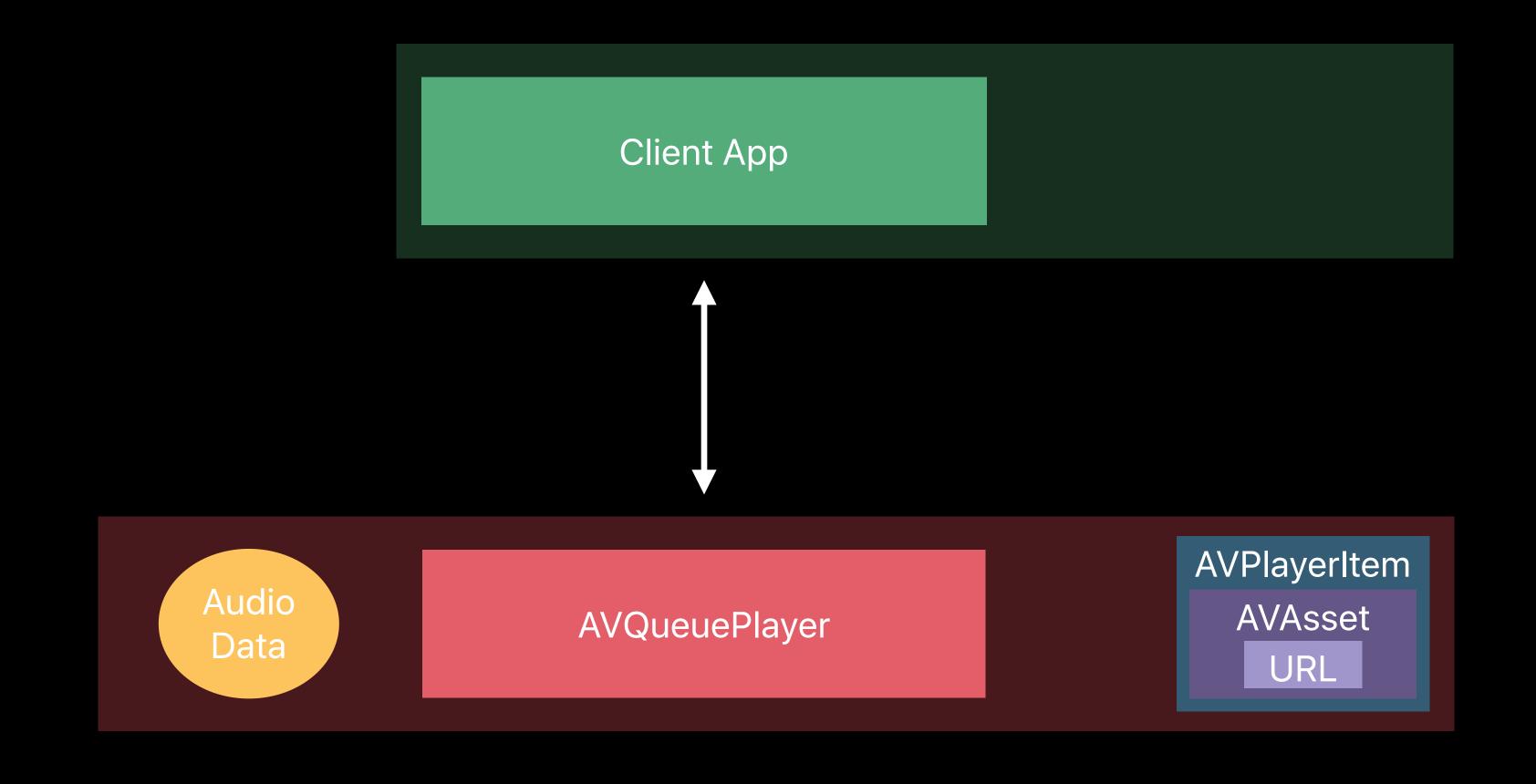


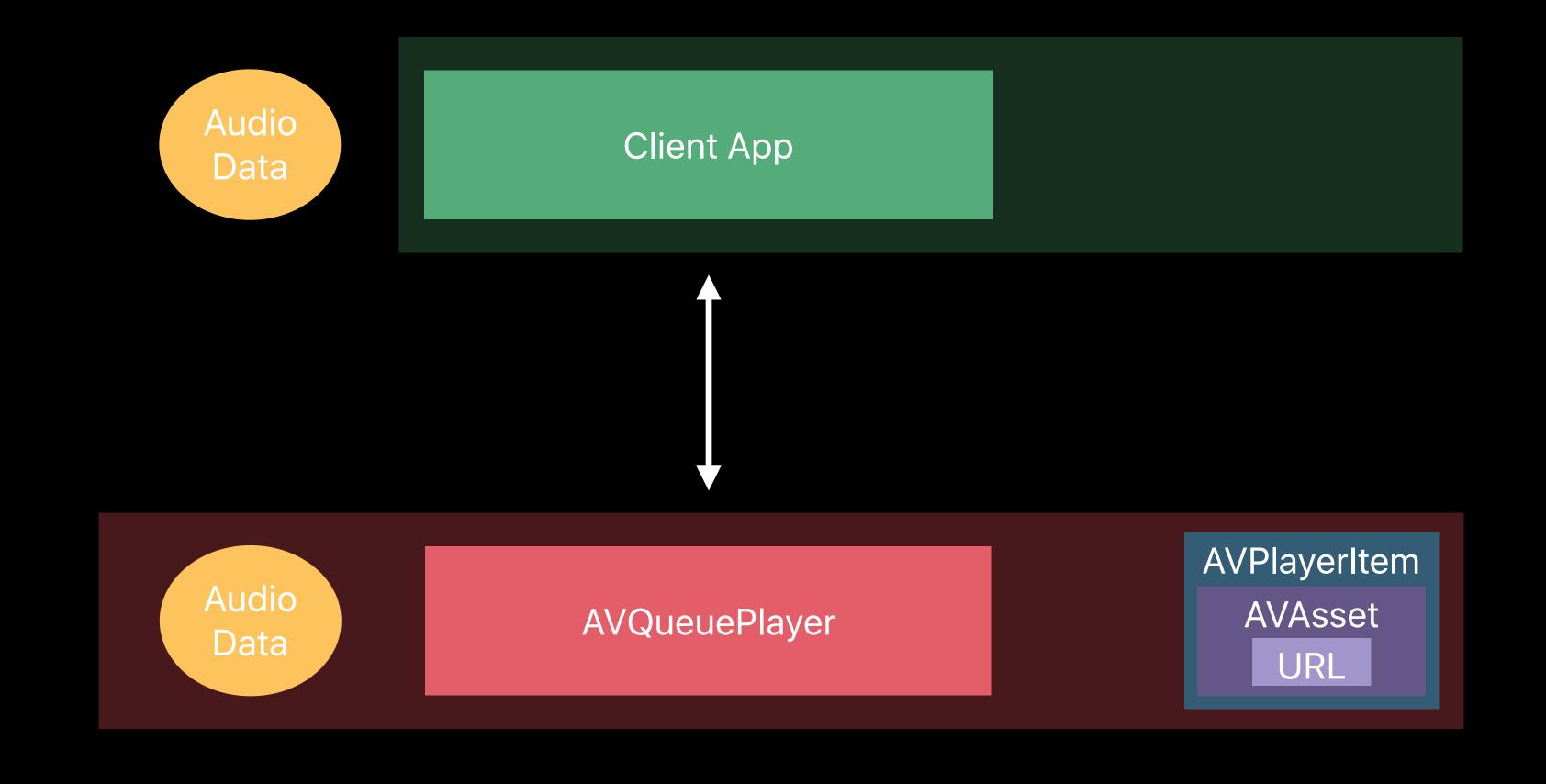


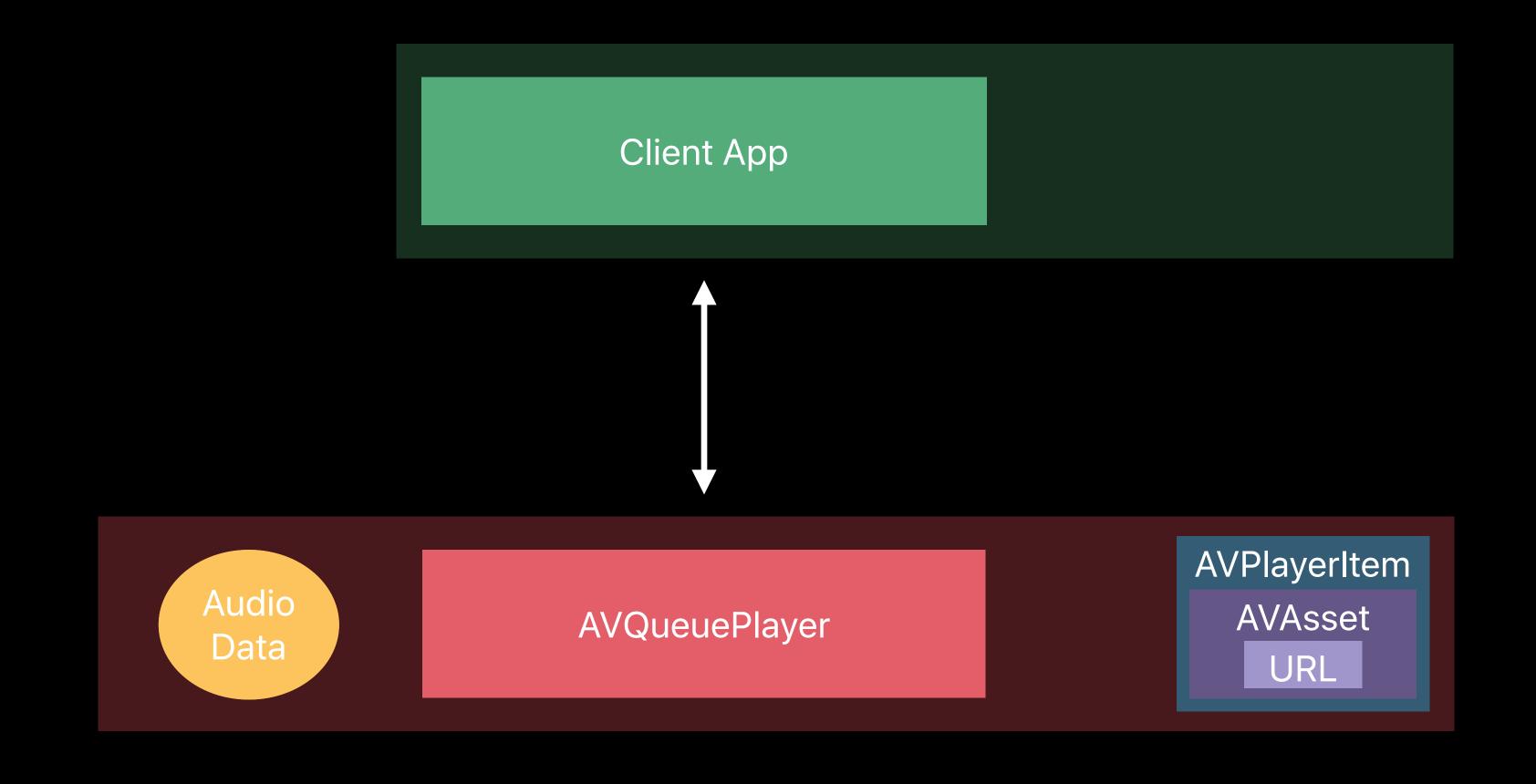


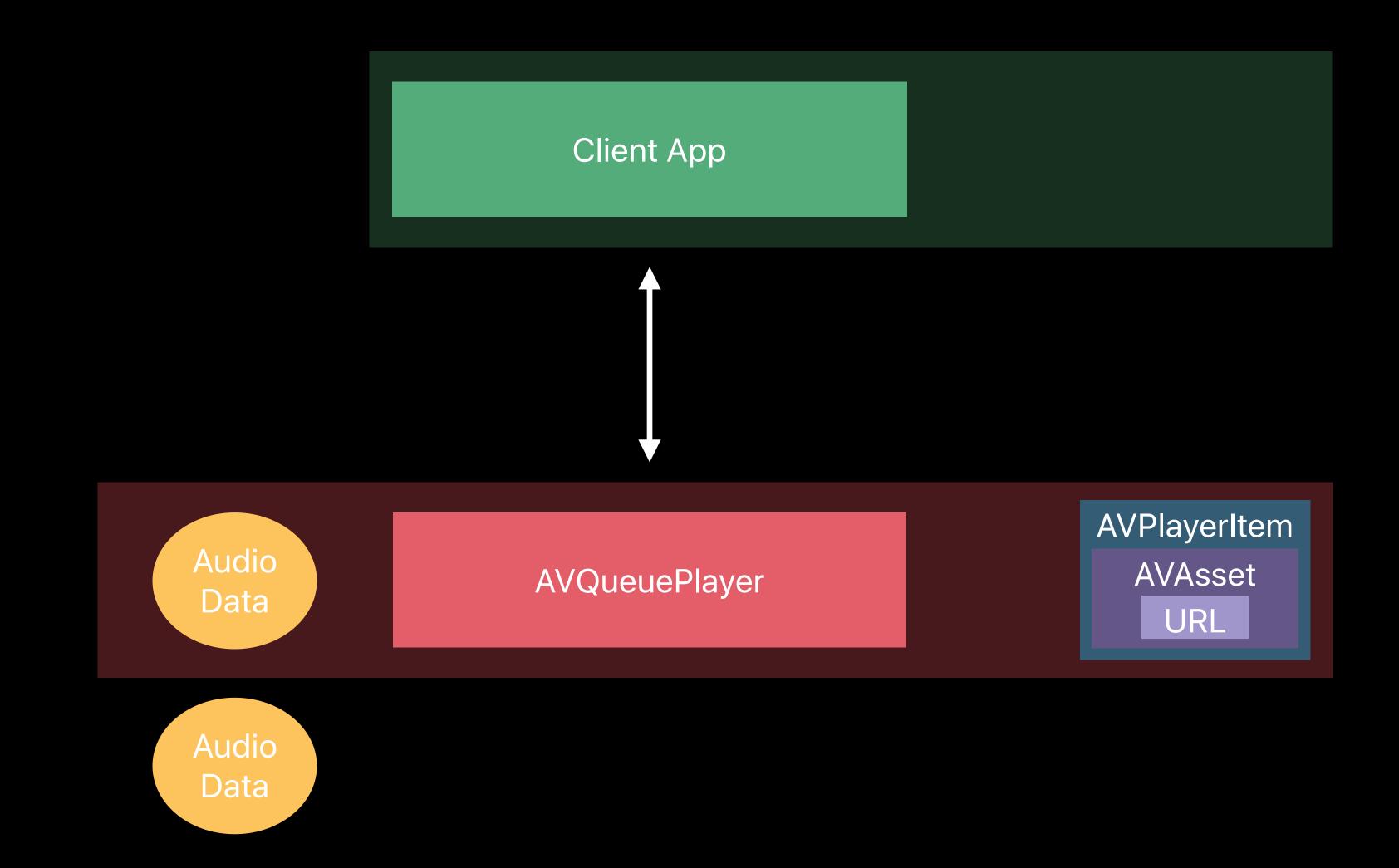


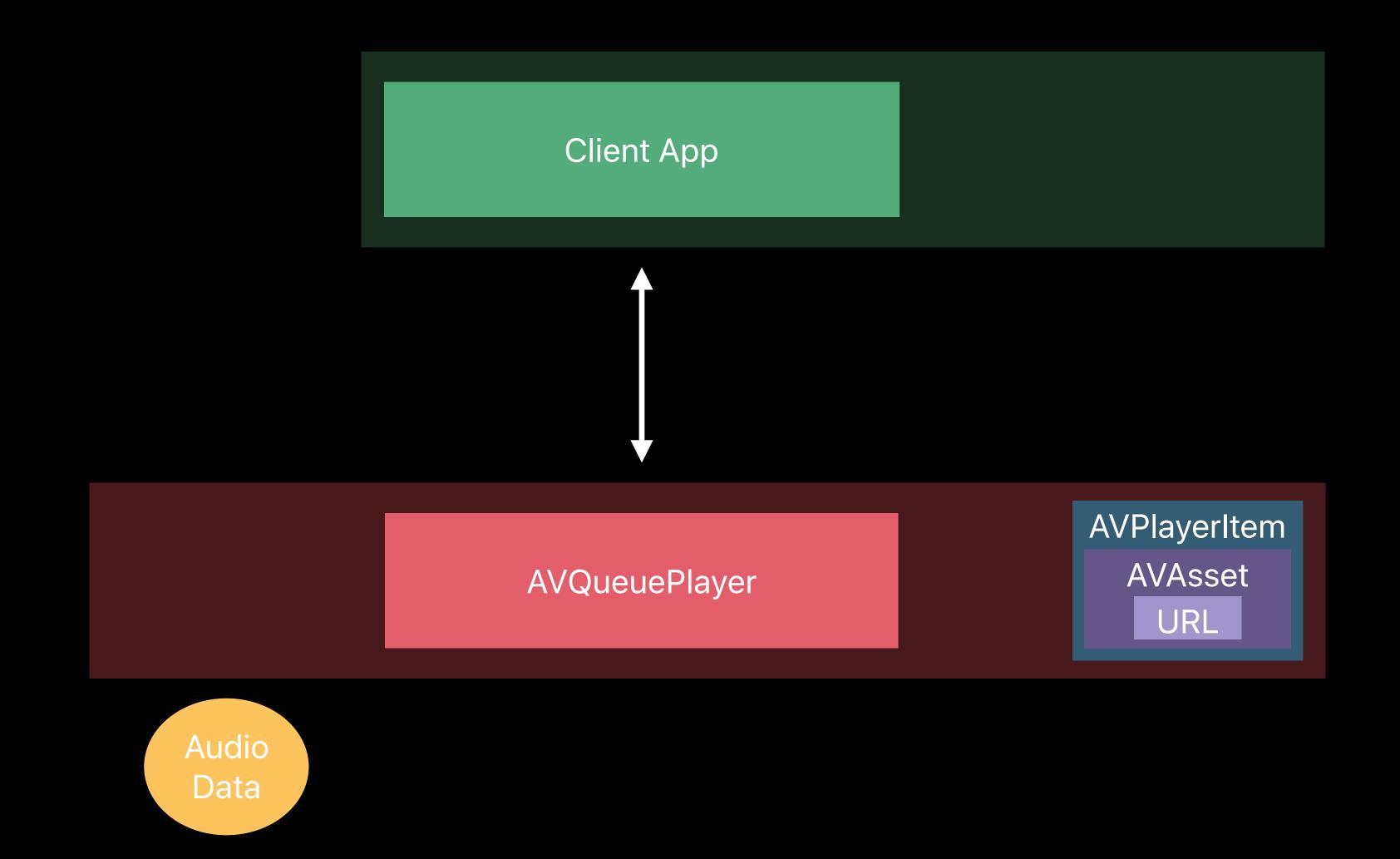














AVSampleBufferAudioRenderer / AVSampleBufferRenderSynchronizer



Your app has additional responsibilities

AVSampleBufferAudioRenderer / AVSampleBufferRenderSynchronizer



Your app has additional responsibilities

Sourcing and parsing the content

AVSampleBufferAudioRenderer / AVSampleBufferRenderSynchronizer

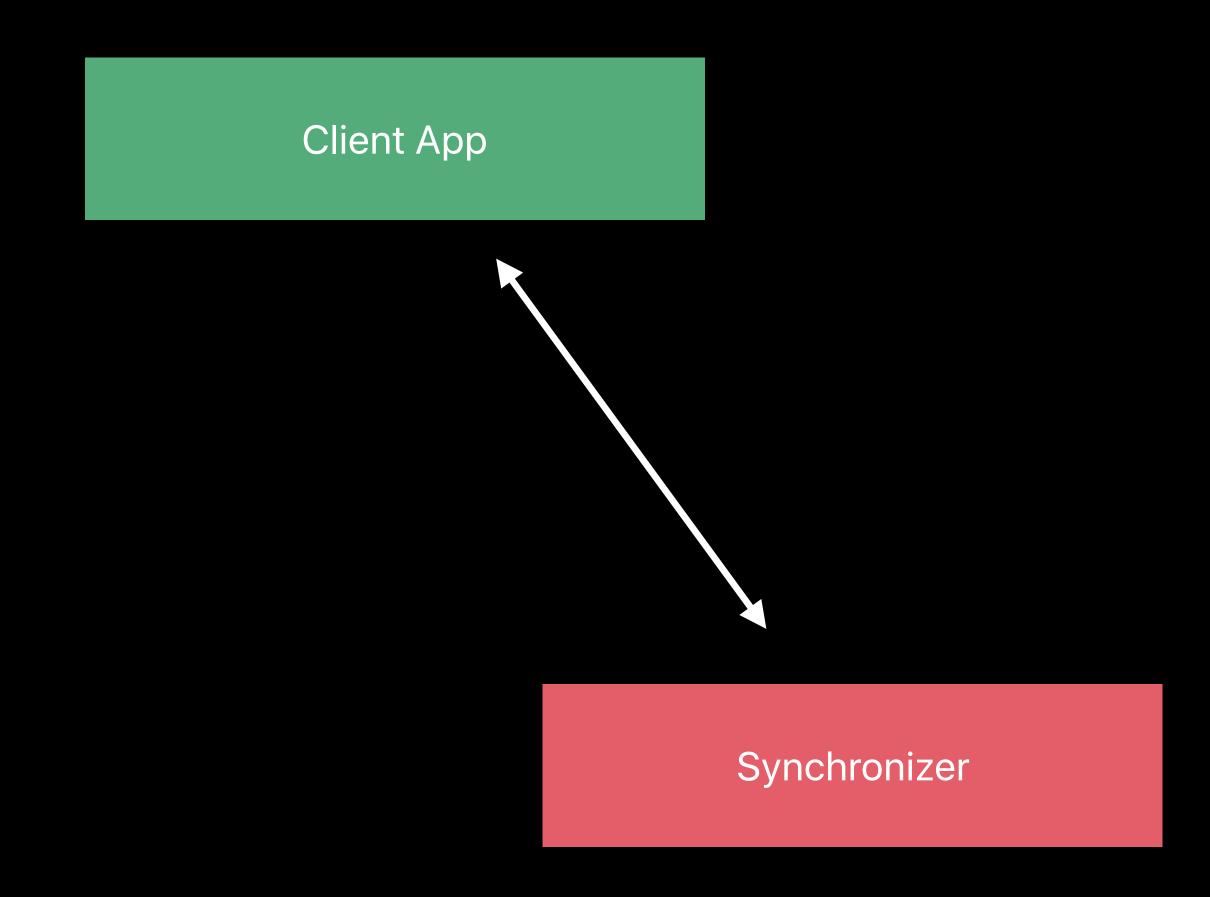


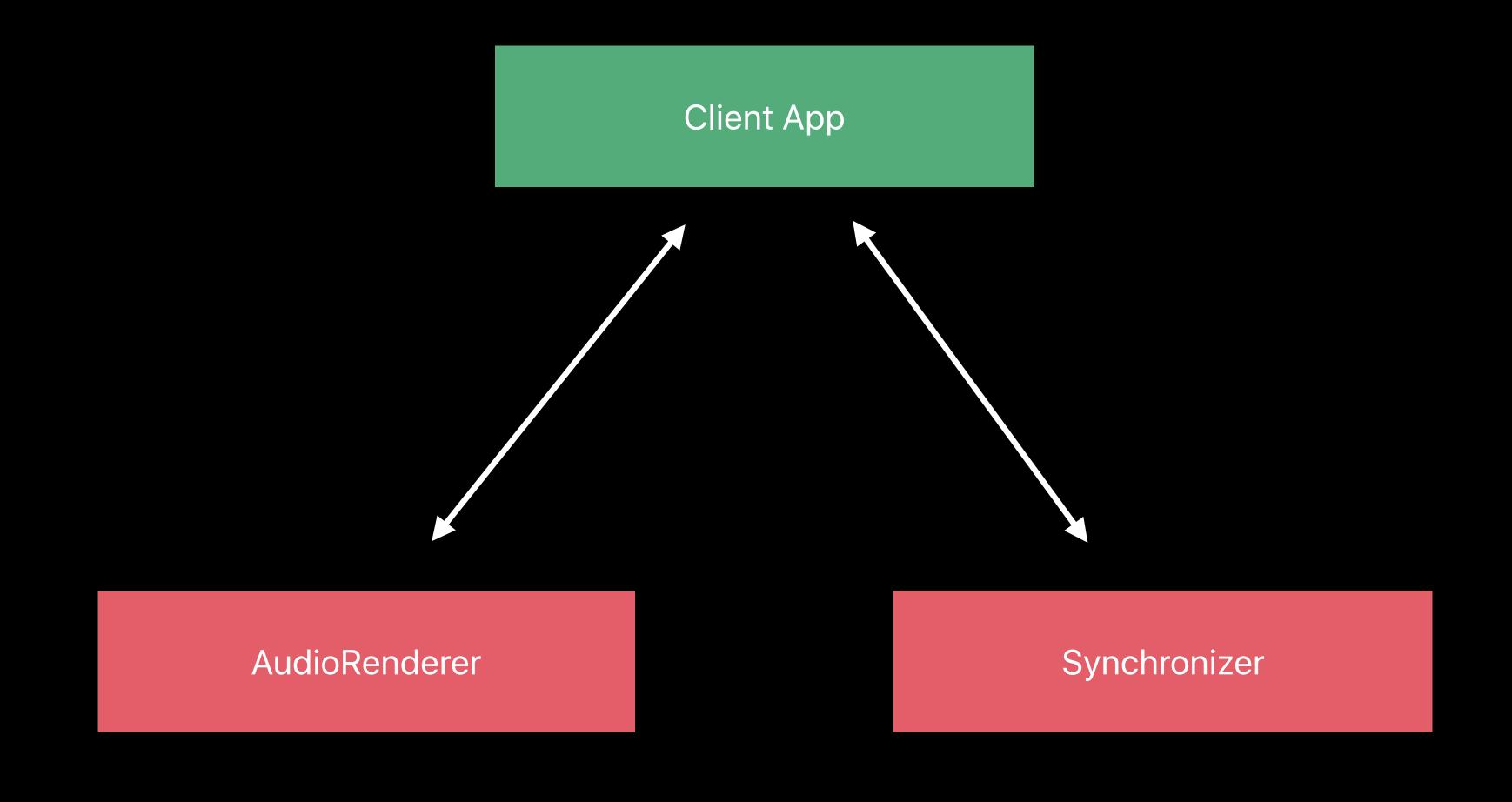
Your app has additional responsibilities

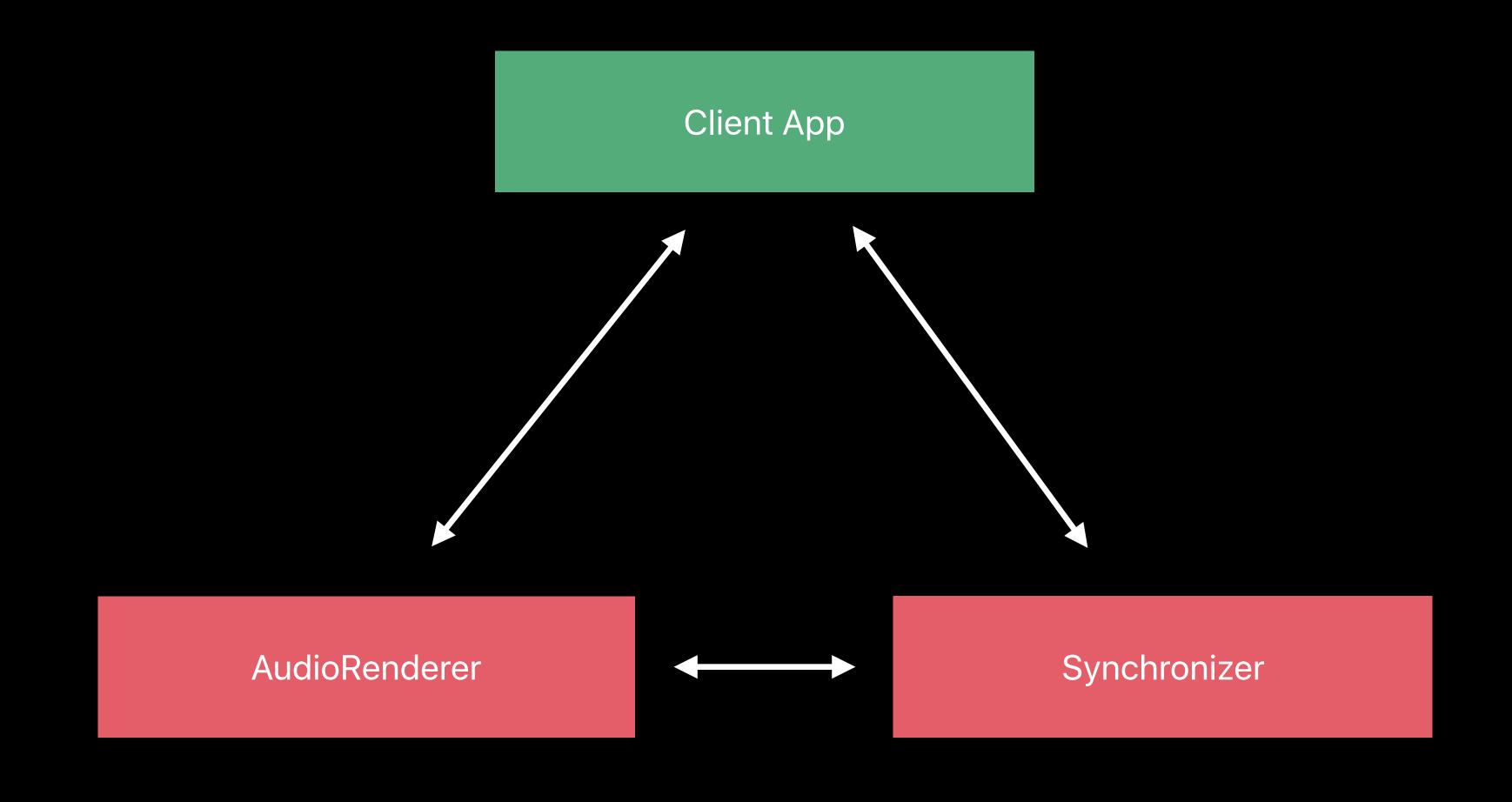
- Sourcing and parsing the content
- Providing raw audio buffers for rendering

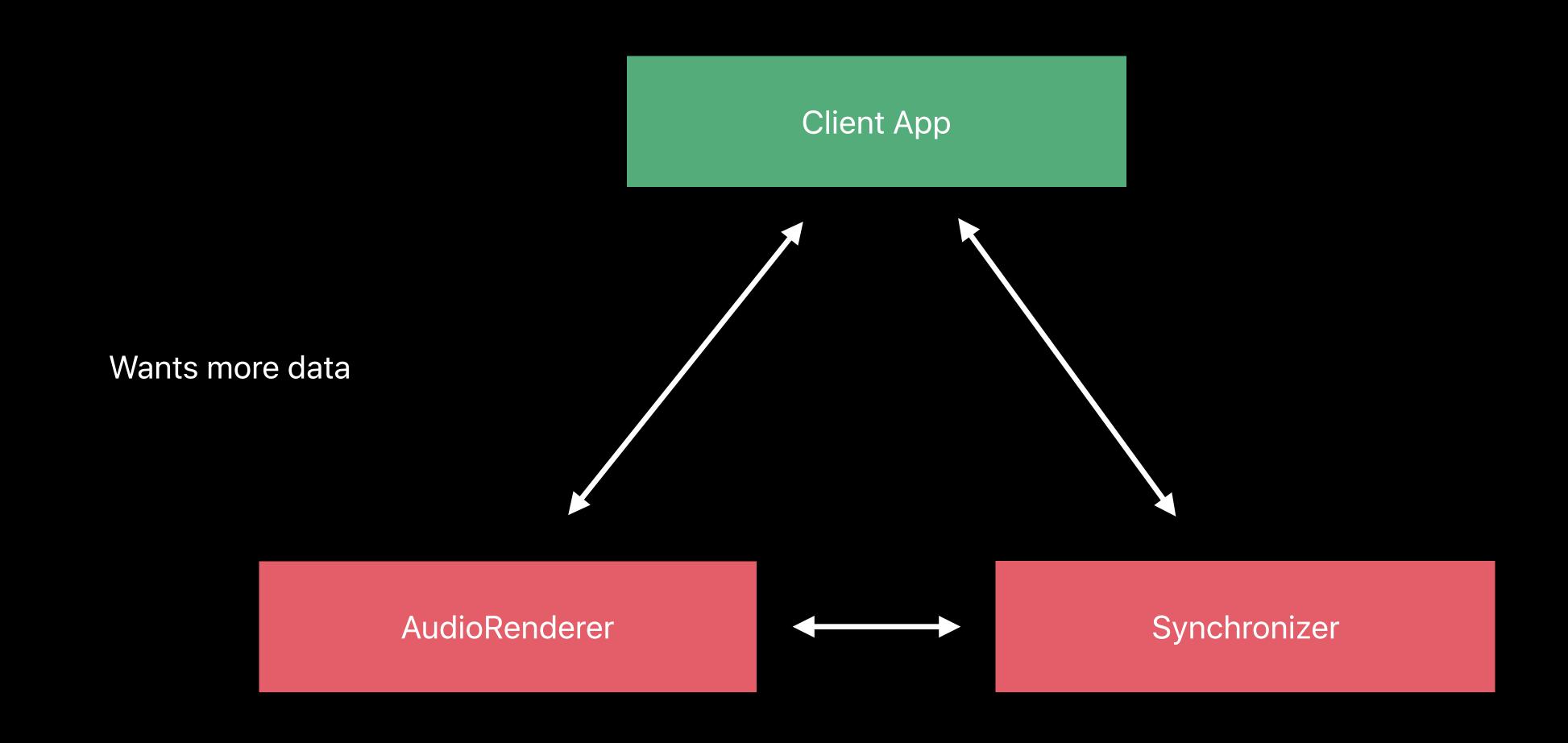
AVSampleBufferAudioRenderer / AVSampleBufferRenderSynchronizer

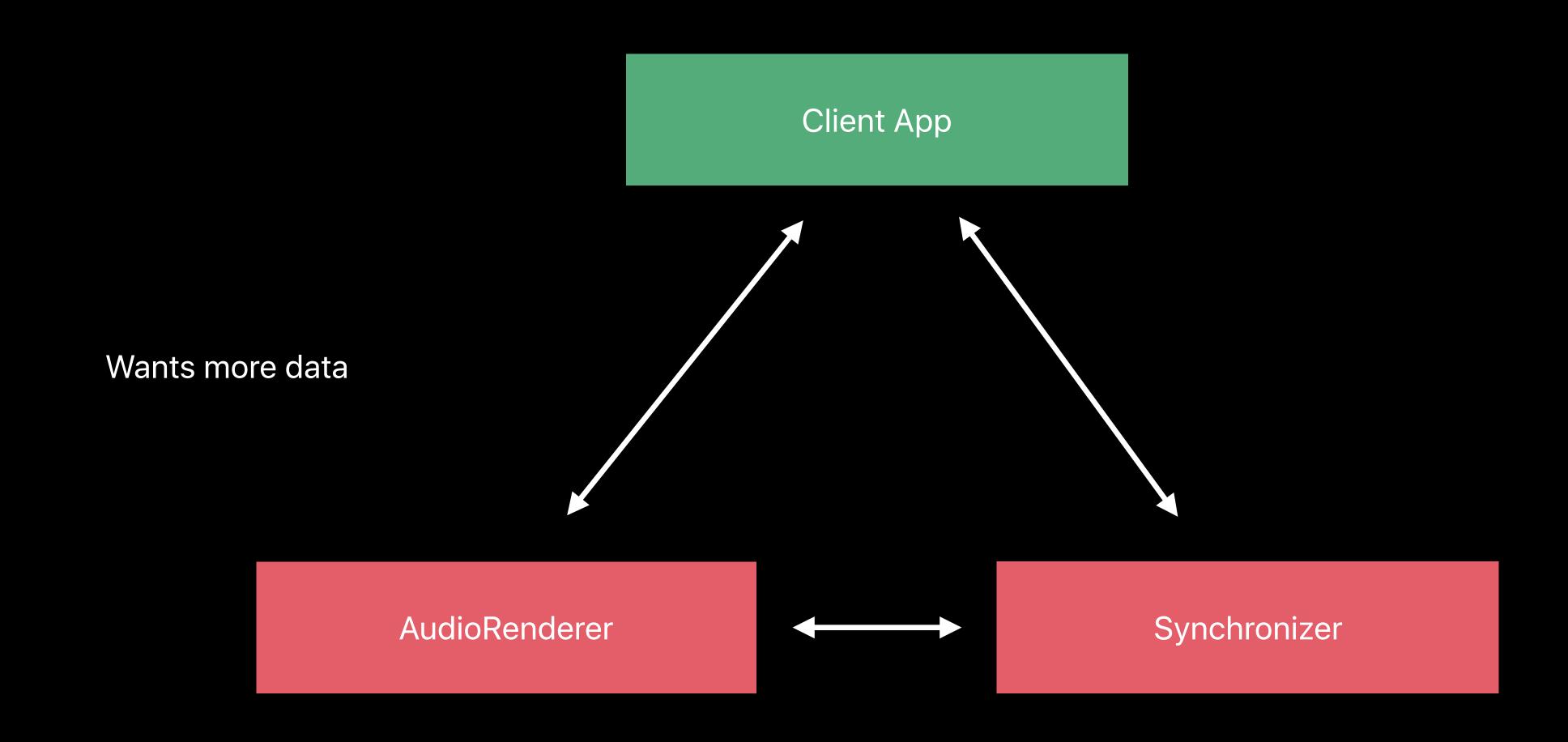
Client App

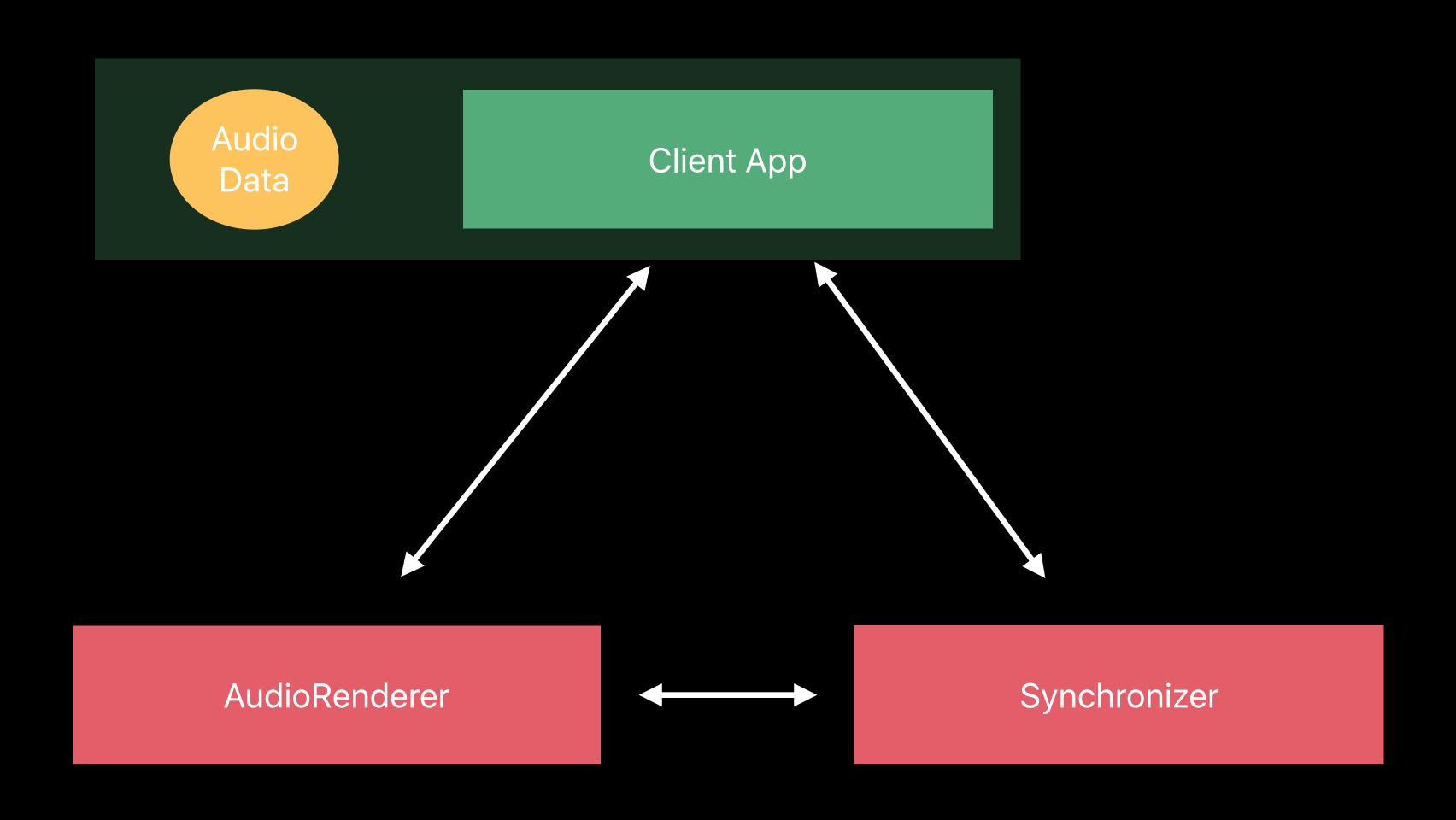


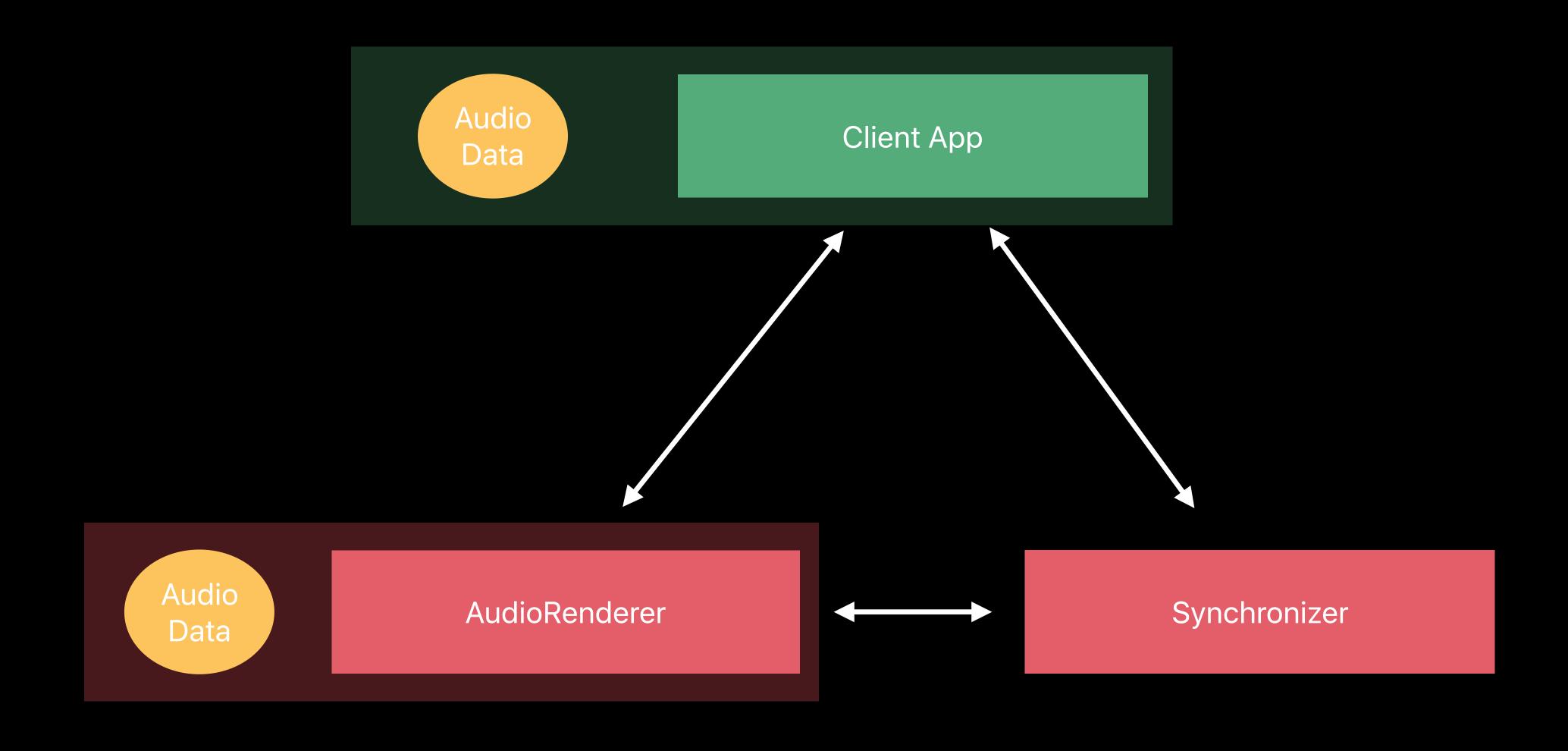


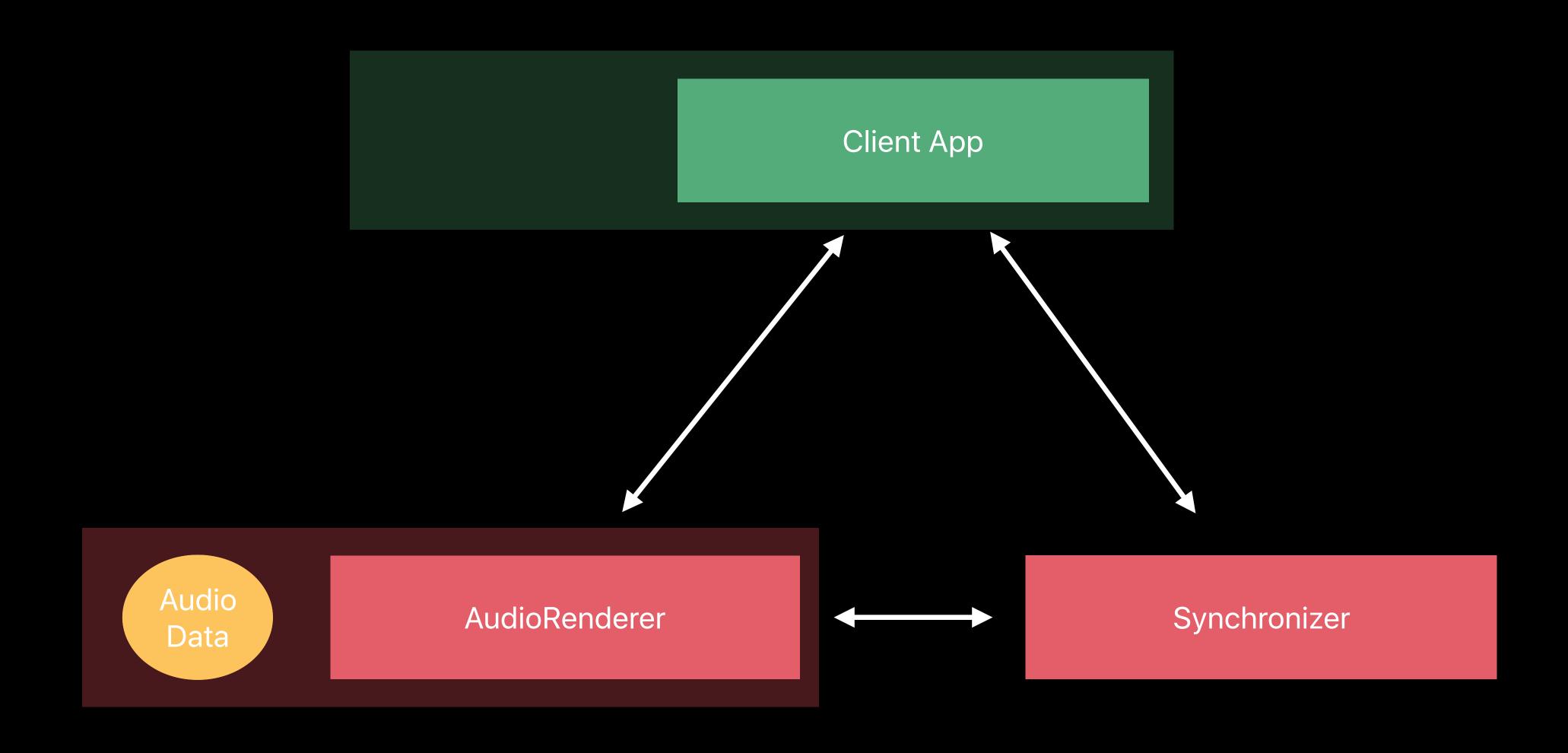


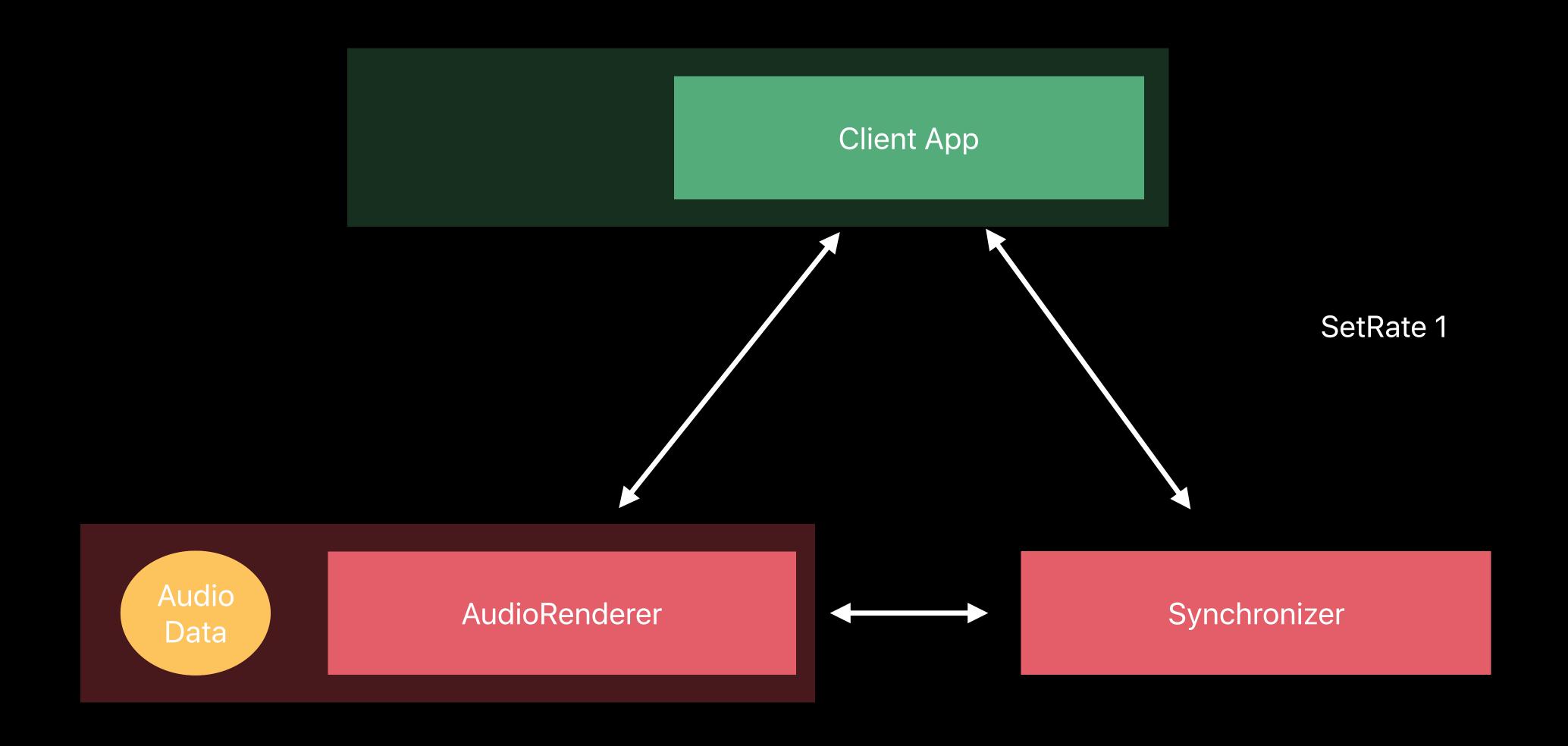


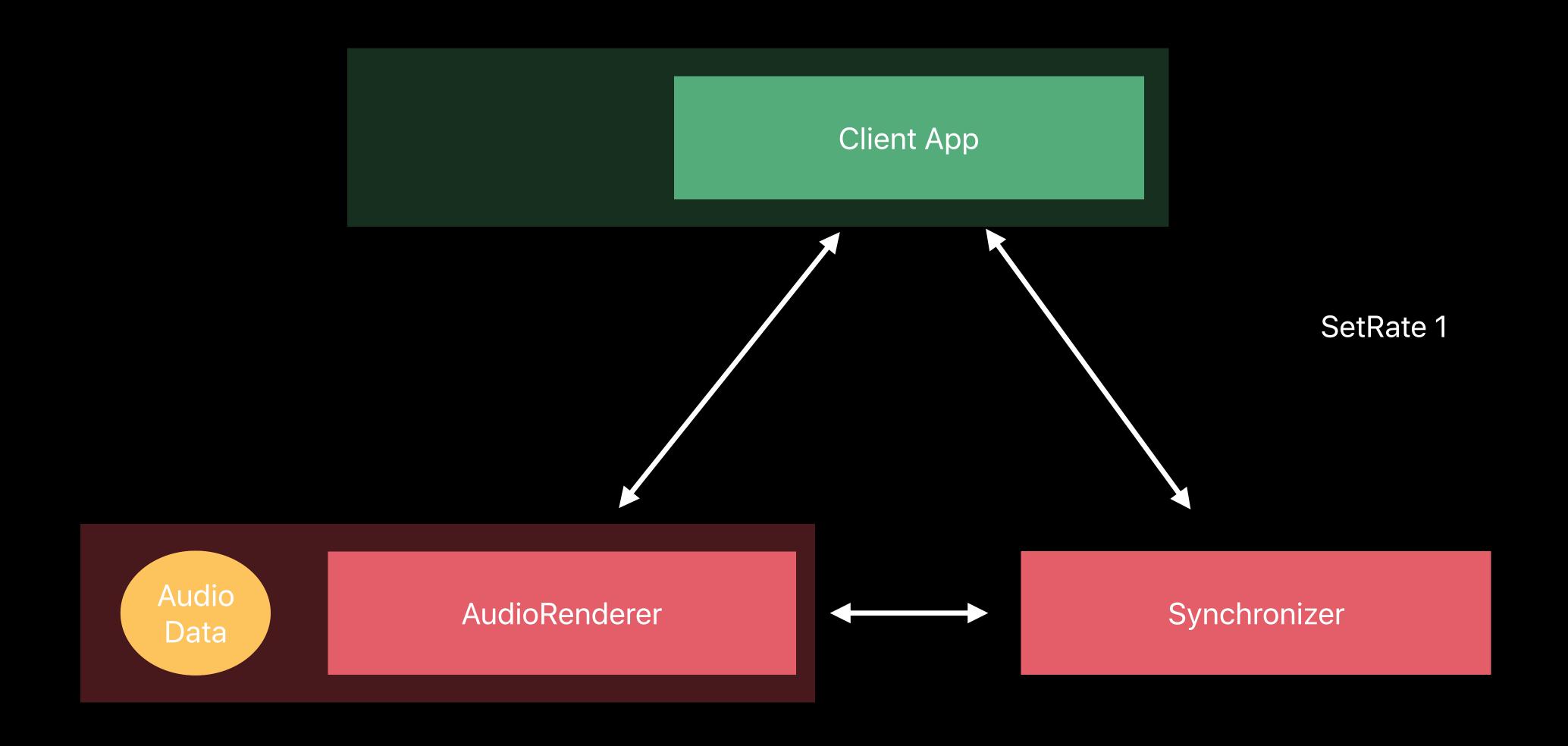


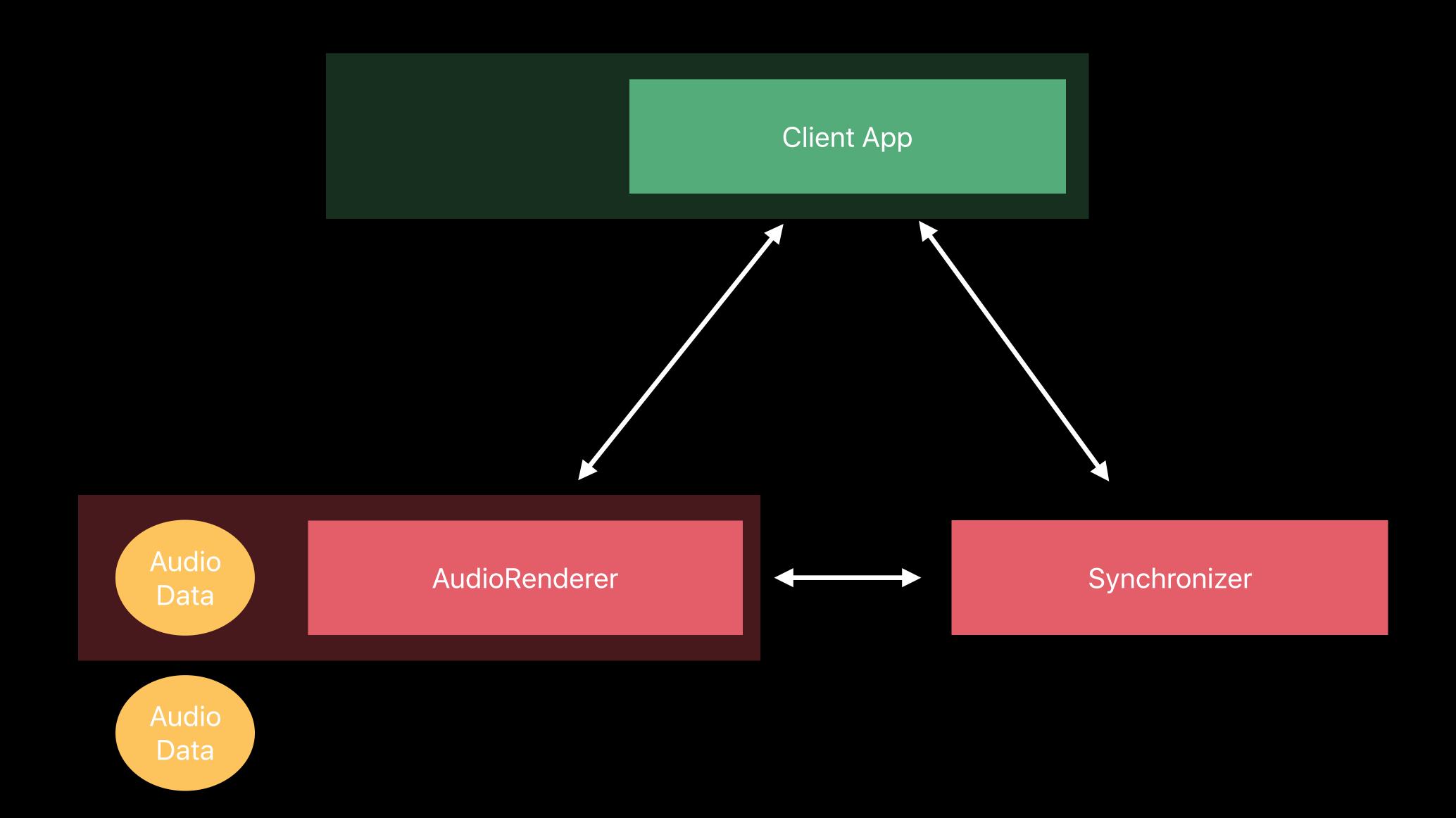


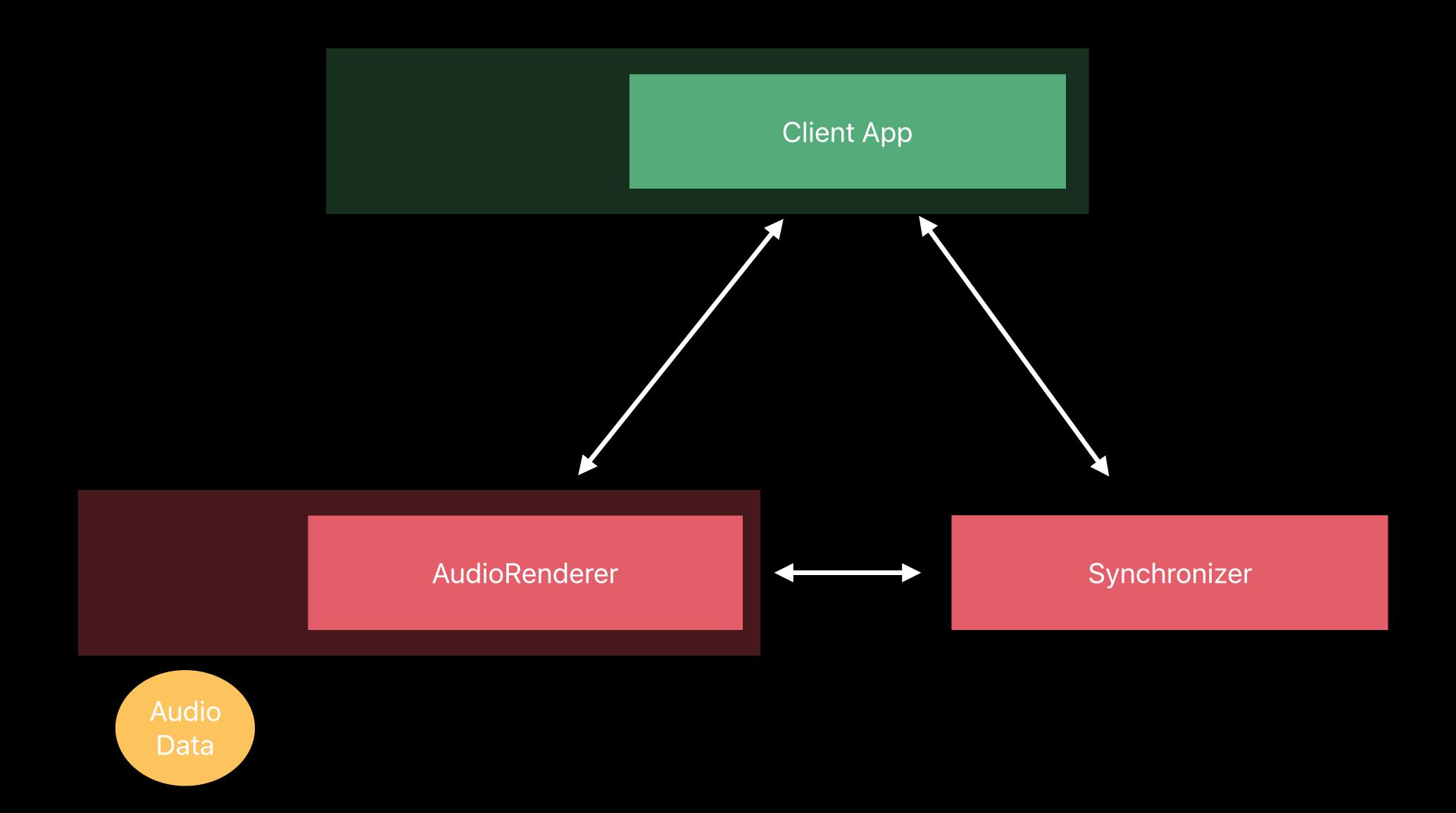












Demo

AirPlay 2 with enhanced buffering

Adam Sonnanstine, AVFoundation Engineer

Advanced Playback Scenarios

AVSampleBufferAudioRenderer

Audio buffer levels

Seek

Play queues

Supported audio formats

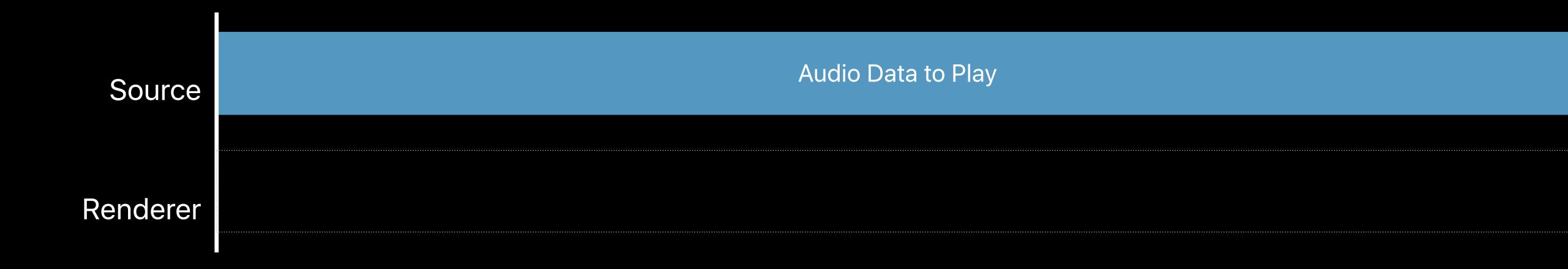
Video synchronization

AVSampleBufferAudioRenderer

The amount of data requested will vary depending on the current route

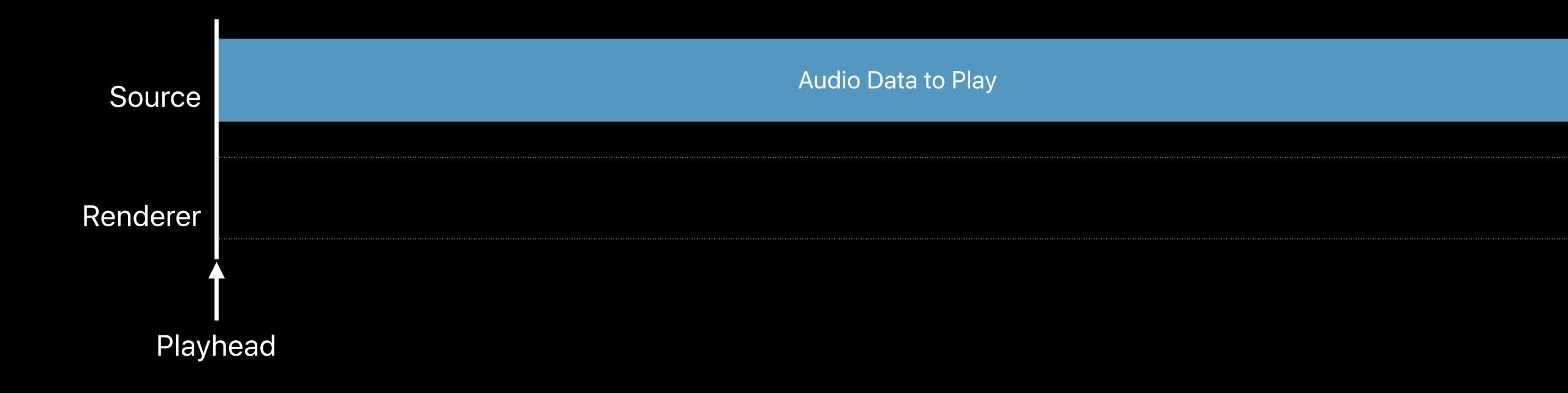
AVSampleBufferAudioRenderer

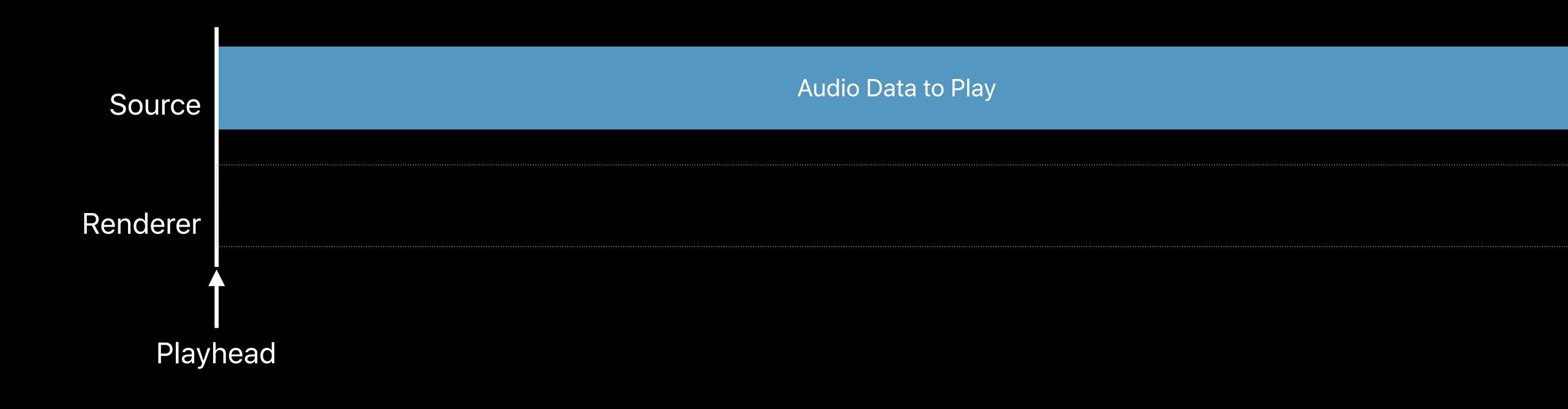
The amount of data requested will vary depending on the current route



AVSampleBufferAudioRenderer

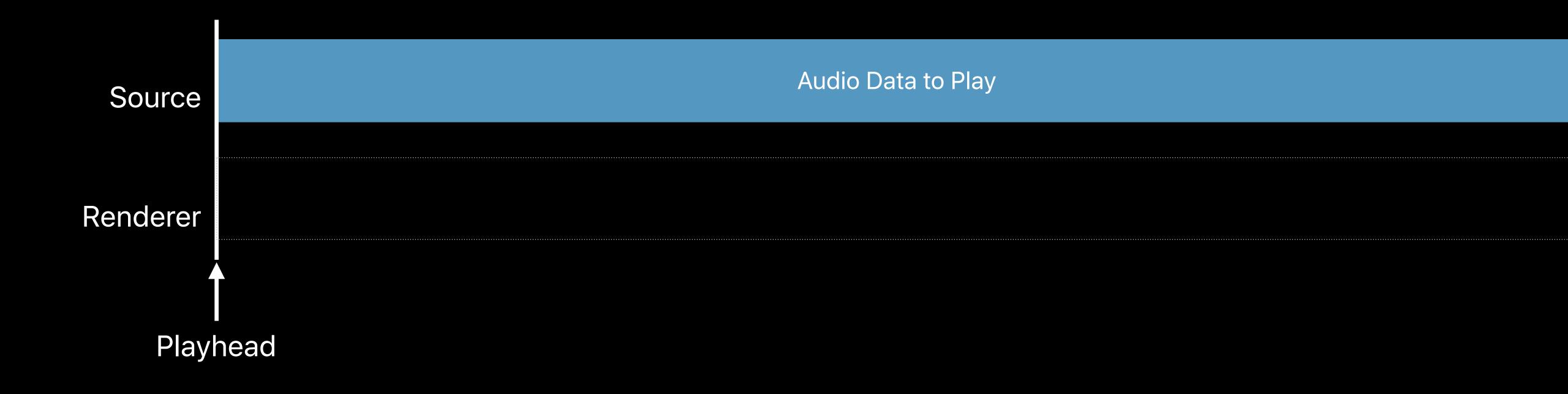
The amount of data requested will vary depending on the current route





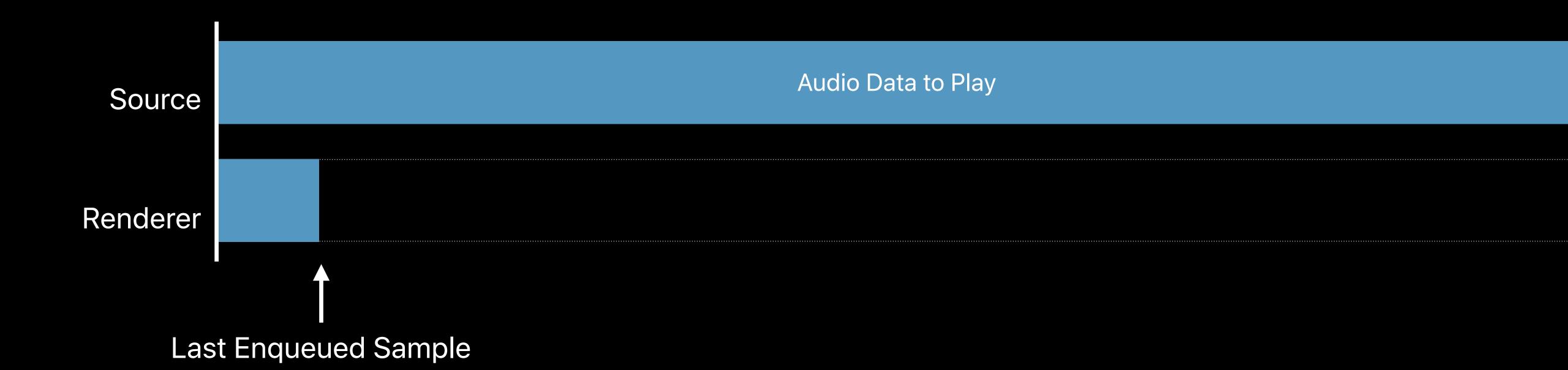
AVSampleBufferAudioRenderer

Local Playback: enqueue just a few seconds ahead of the Playhead



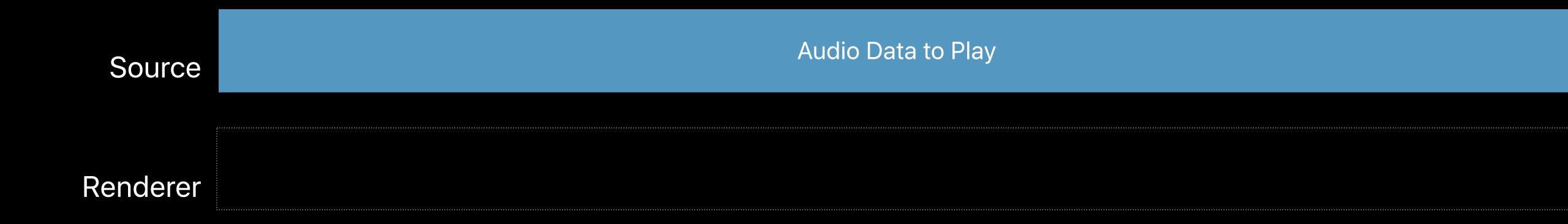
AVSampleBufferAudioRenderer

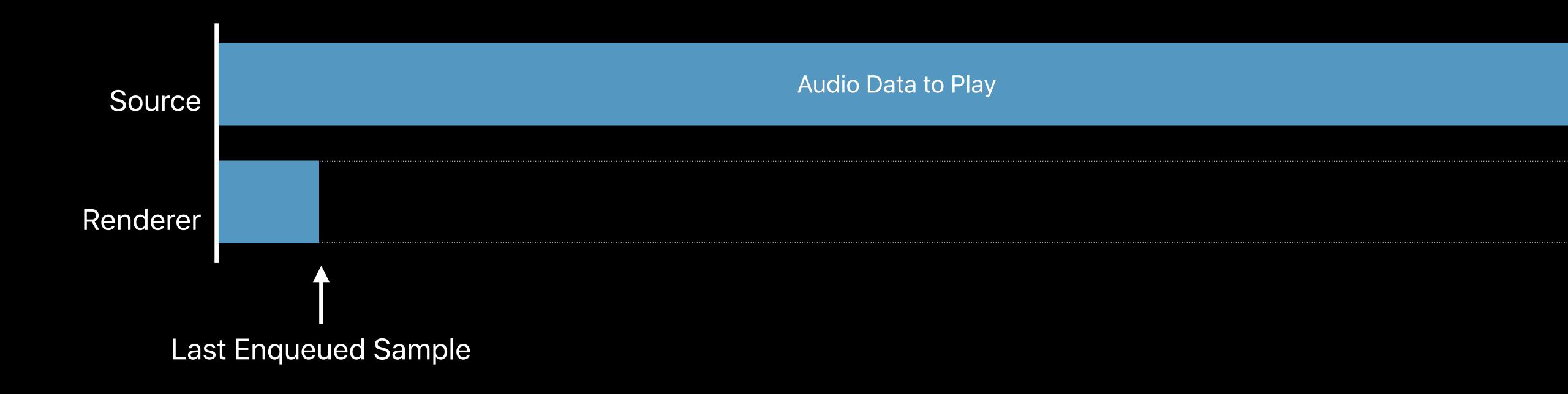
Local Playback: enqueue just a few seconds ahead of the Playhead



AVSampleBufferAudioRenderer

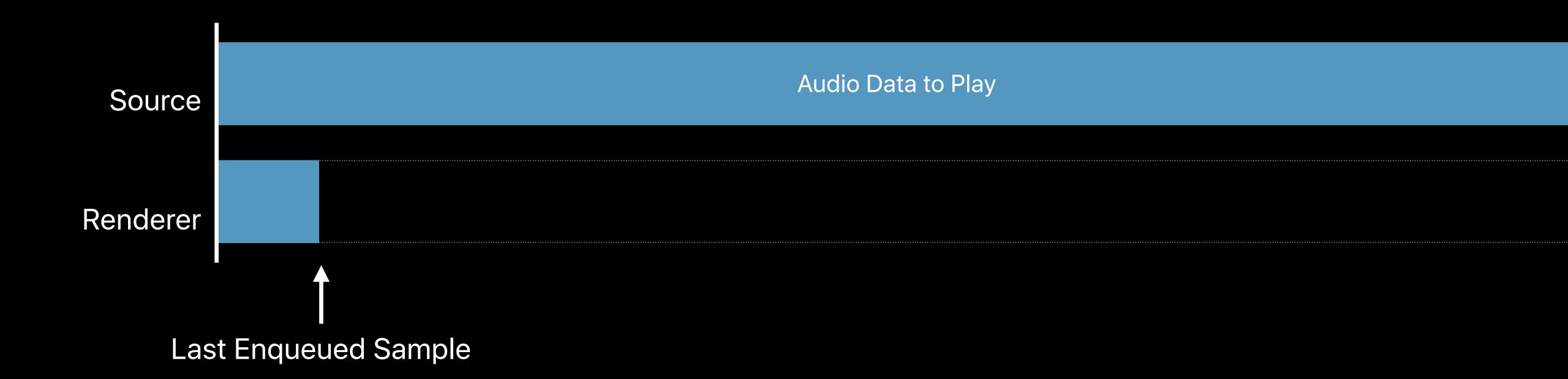
Local Playback: enqueue just a few seconds ahead of the Playhead





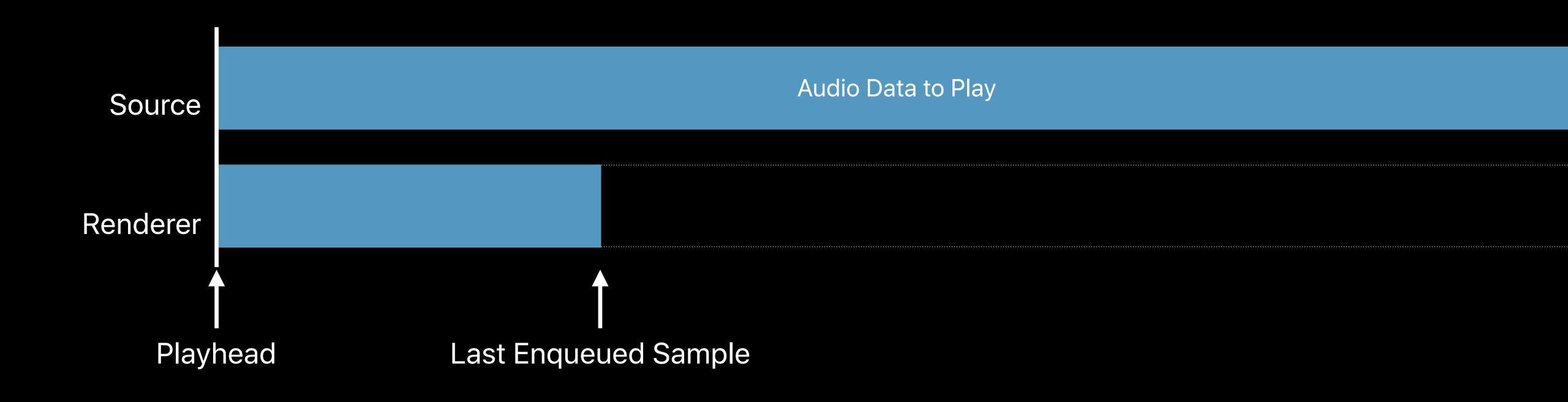
AVSampleBufferAudioRenderer

AirPlay 2 Speaker: enqueue up to multiple minutes of the Playhead



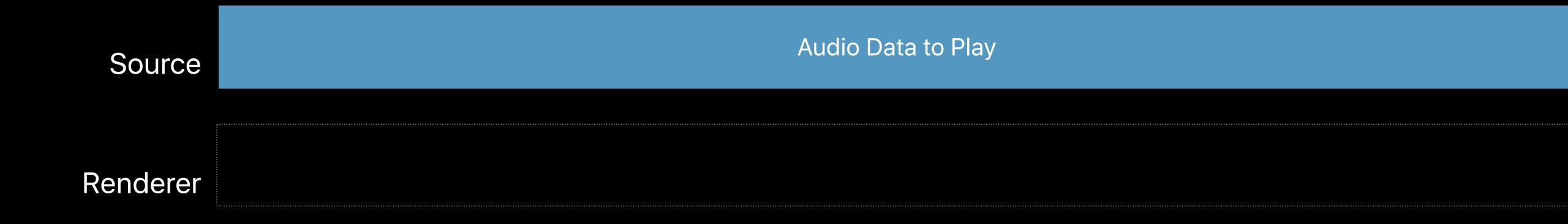
AVSampleBufferAudioRenderer

AirPlay 2 Speaker: enqueue up to multiple minutes of the Playhead



AVSampleBufferAudioRenderer

AirPlay 2 Speaker: enqueue up to multiple minutes of the Playhead



AVSampleBufferAudioRenderer

AVSampleBufferAudioRenderer

AVSampleBufferAudioRenderer

Requested data amount varies by audio route

Local Seconds

AVSampleBufferAudioRenderer

Local	Seconds
Bluetooth	Seconds

AVSampleBufferAudioRenderer

Local	Seconds
Bluetooth	Seconds
AirPlay	Seconds

AVSampleBufferAudioRenderer

Local	Seconds
Bluetooth	Seconds
AirPlay	Seconds
AirPlay 2	Minutes

AVSampleBufferAudioRenderer

Requested data amount varies by audio route

Local	Seconds
Bluetooth	Seconds
AirPlay	Seconds
AirPlay 2	Minutes

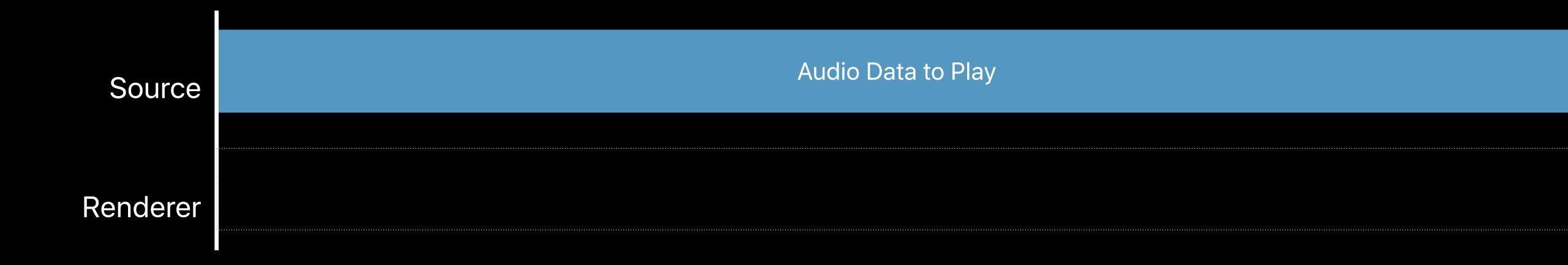
Your app should handle these request changes

AVSampleBufferAudioRenderer

Manually changing Playhead location

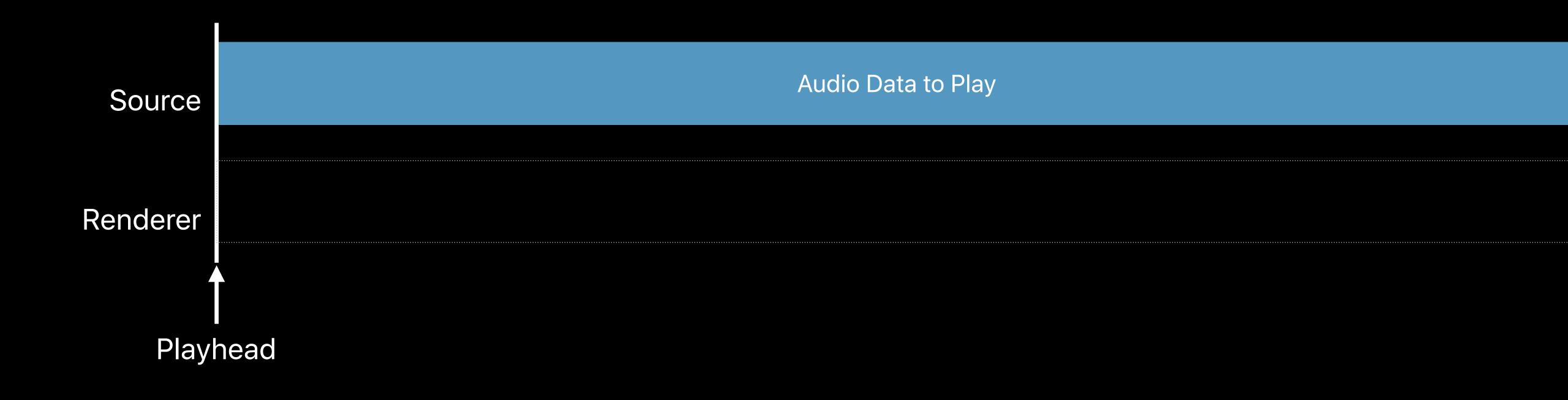
AVSampleBufferAudioRenderer

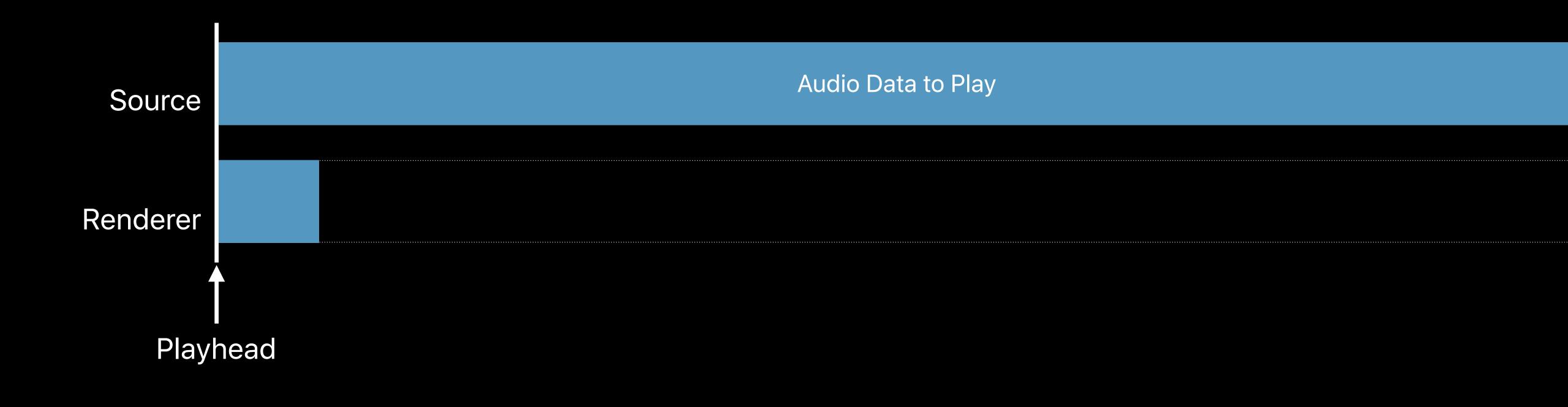
Manually changing Playhead location

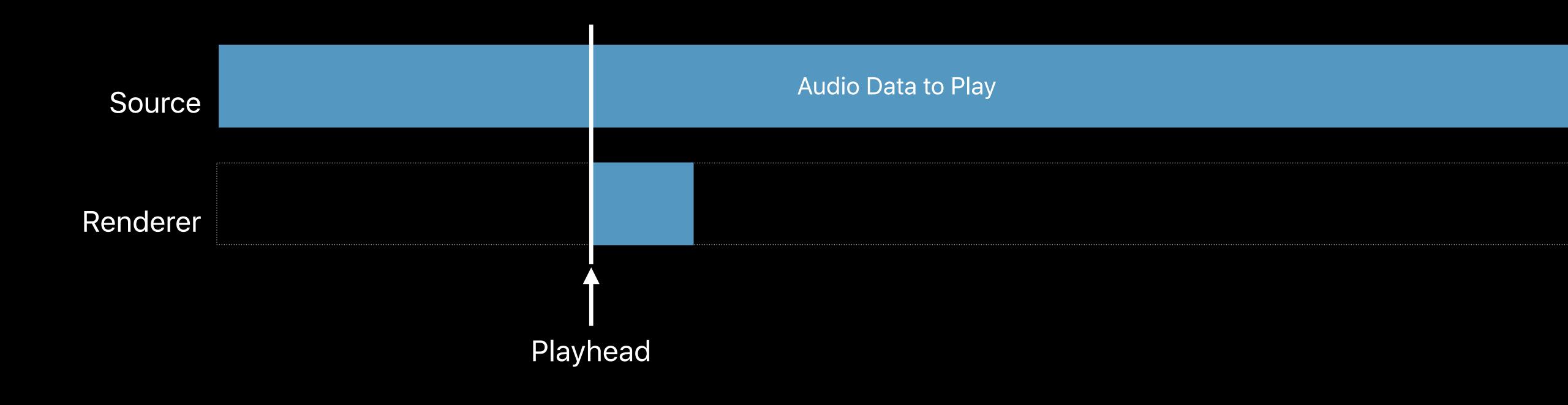


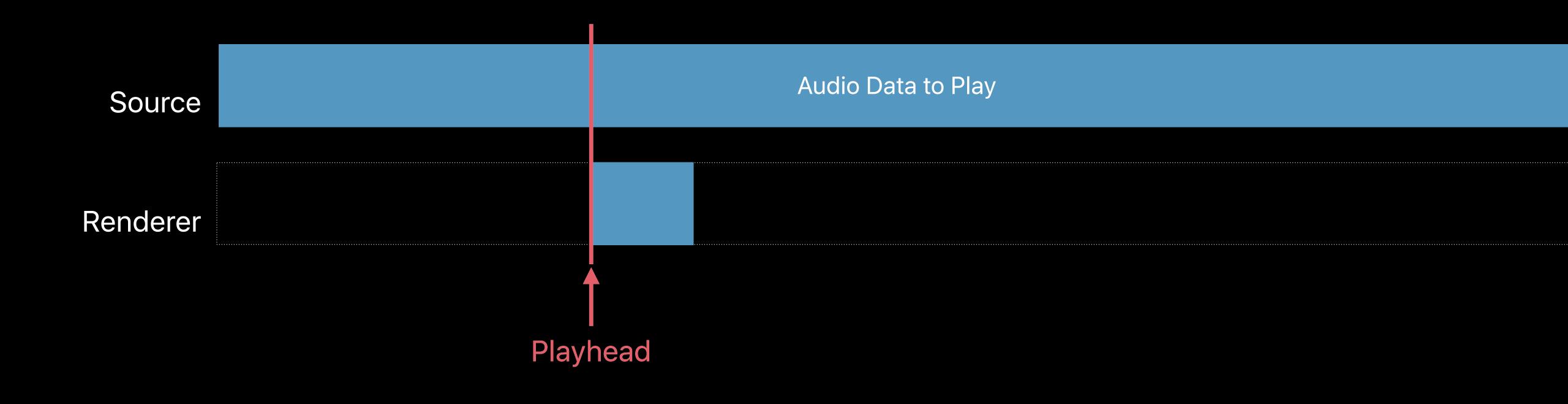
AVSampleBufferAudioRenderer

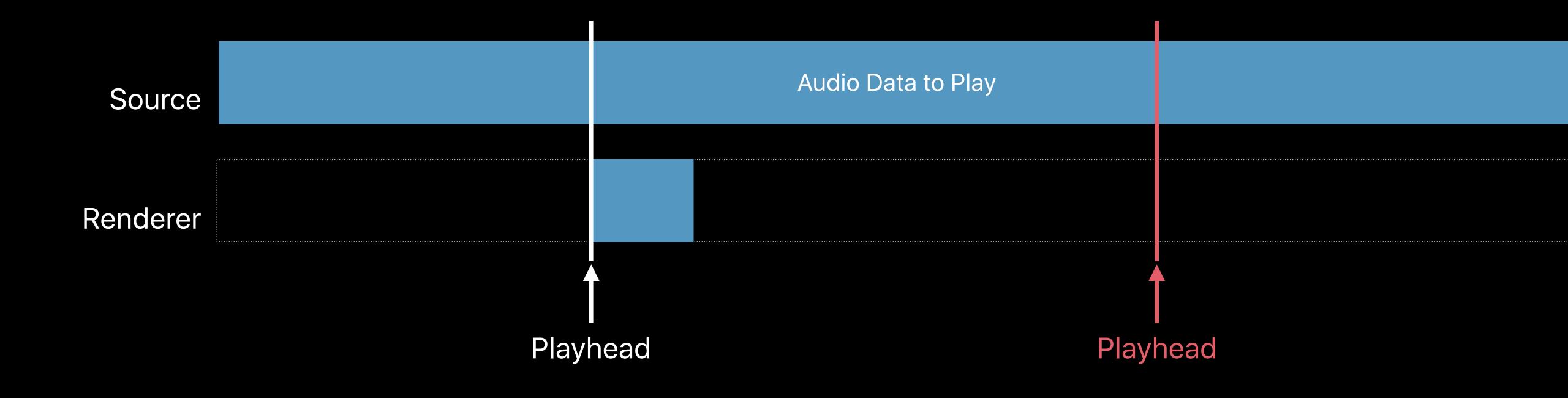
Manually changing Playhead location

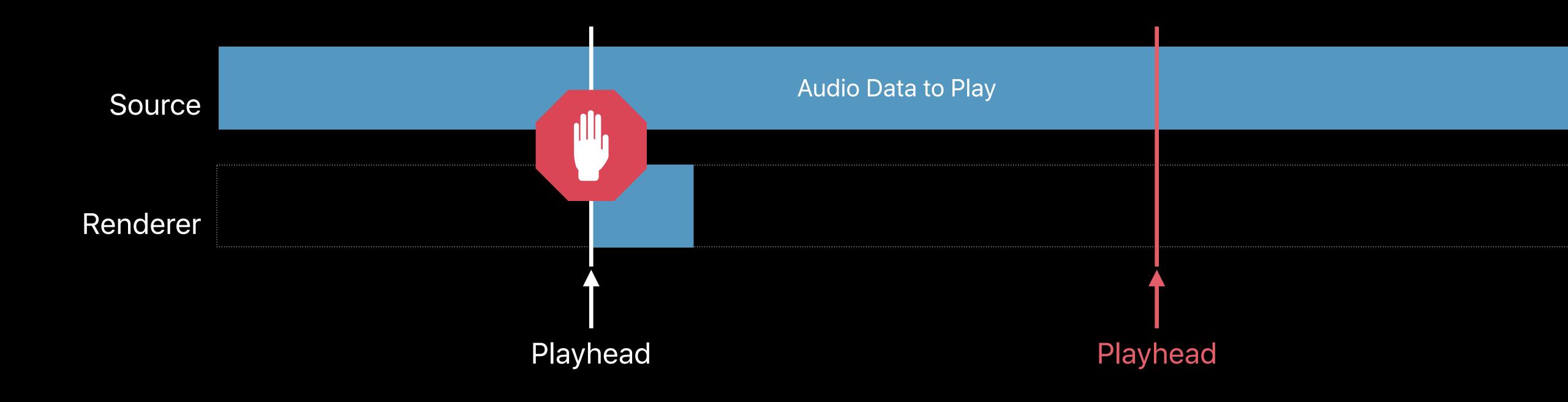


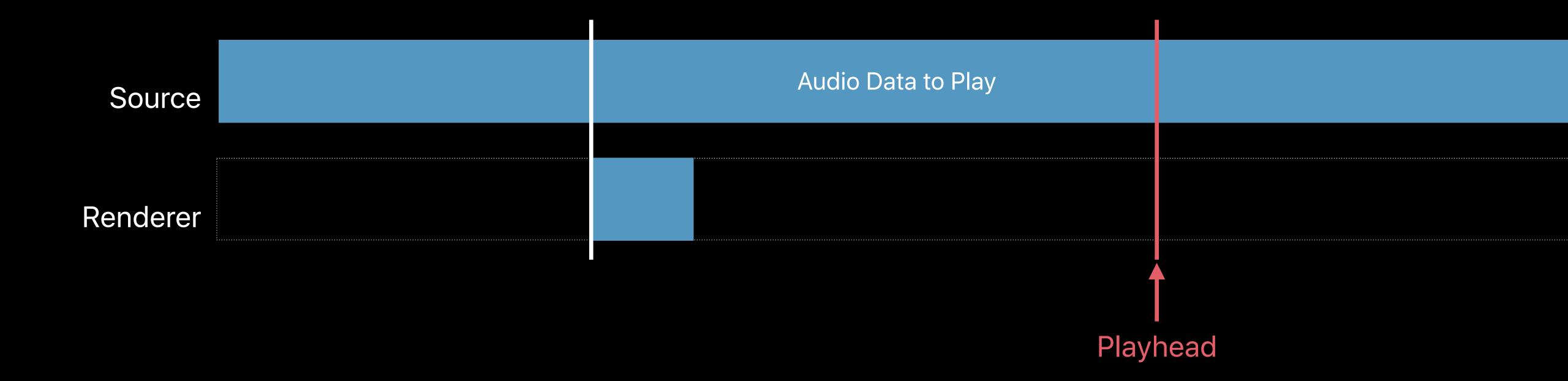


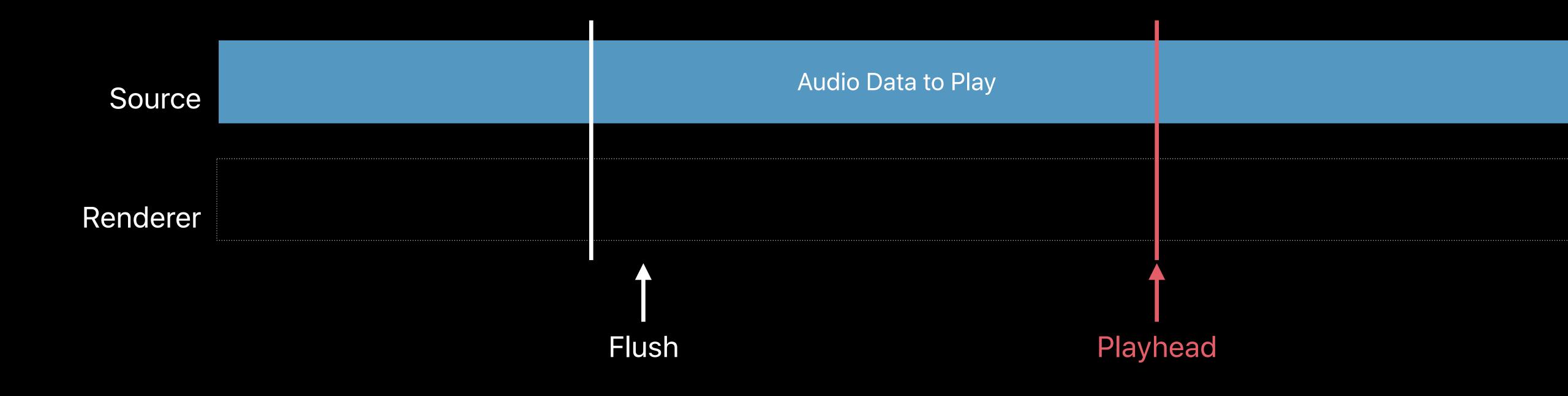


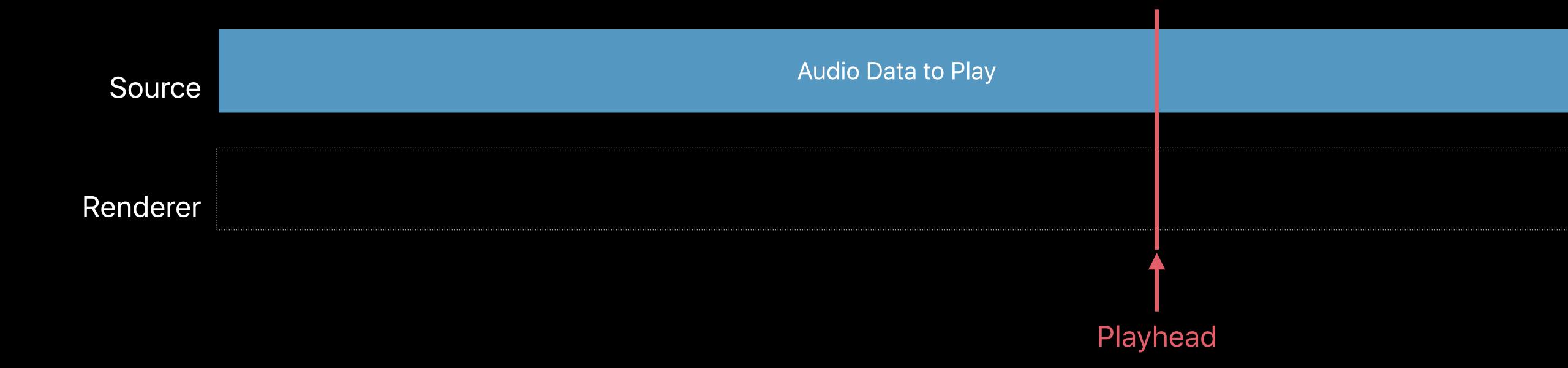


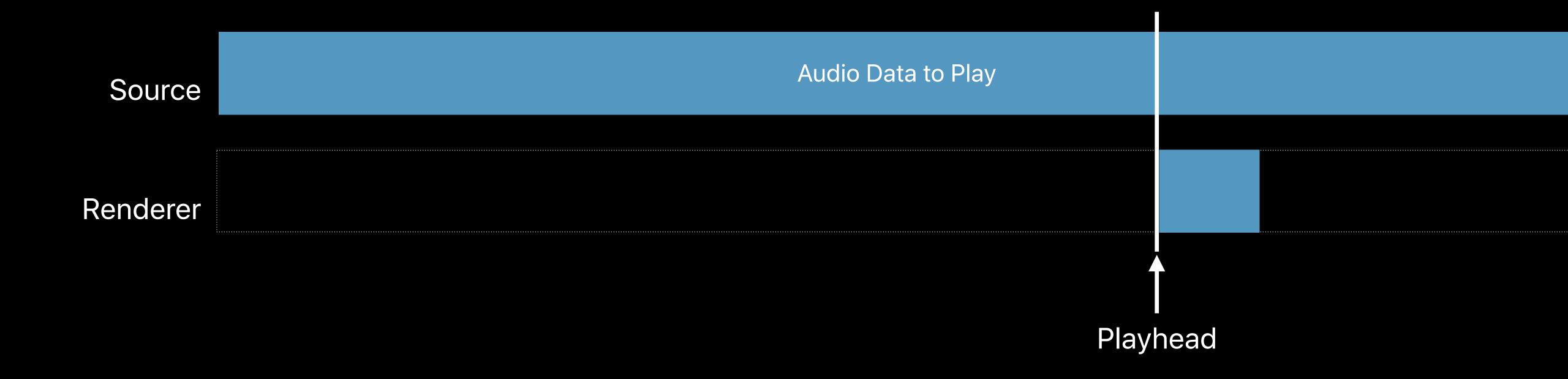


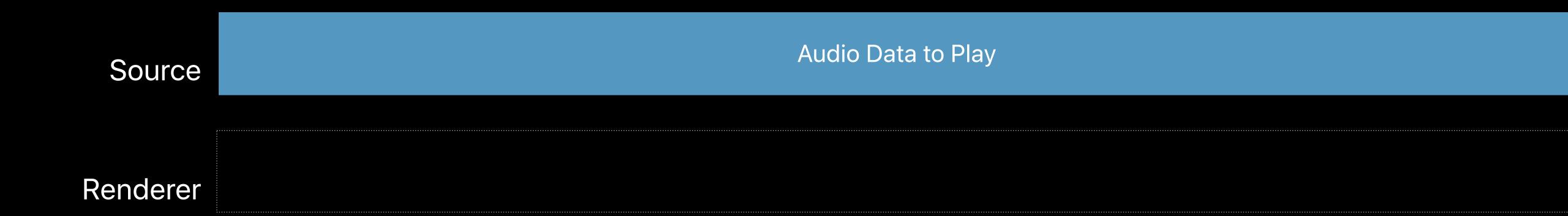












```
// Seek - AVSampleBufferAudioRenderer
func seek(toMediaTime mediaTime: CMTime) {
   renderSynchronizer.setRate(0.0, time: kCMTimeZero)
   audioRenderer.stopRequestingMediaData()
   audioRenderer.flush()
   myPrepareSampleGenerationForMediaTime(mediaTime)
   audioRenderer.requestMediaDataWhenReady(on: mySerializationQueue) {
       // ...
   renderSynchronizer.setRate(1.0, time: mediaTime)
```

```
// Seek - AVSampleBufferAudioRenderer
func seek(toMediaTime mediaTime: CMTime) {
   renderSynchronizer.setRate(0.0, time: kCMTimeZero)
   audioRenderer.stopRequestingMediaData()
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       // ...
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```

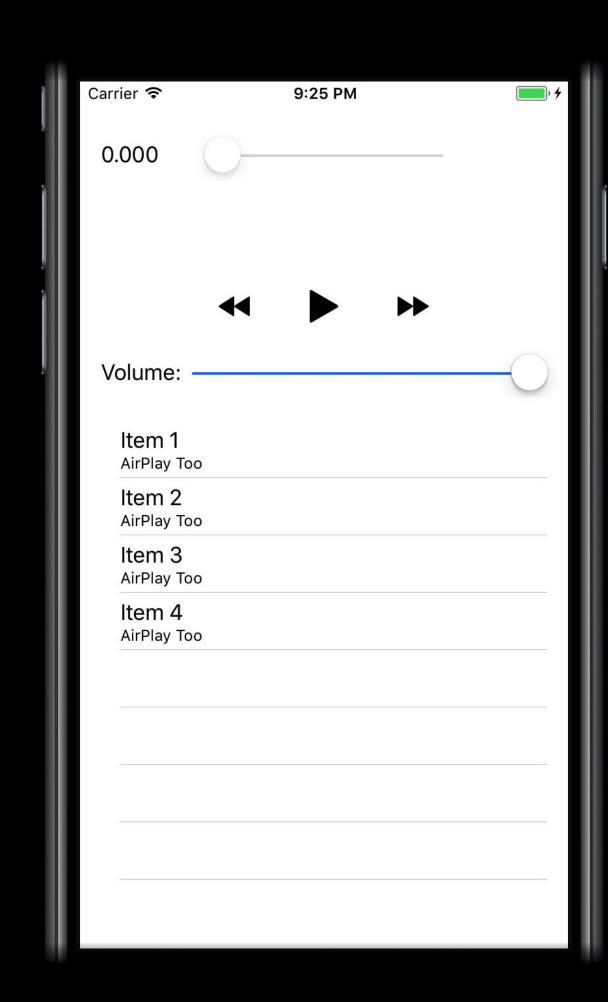
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   renderSynchronizer.setRate(0.0, time: kCMTimeZero)
   audioRenderer.stopRequestingMediaData()
   audioRenderer.flush()
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   audioRenderer.requestMediaDataWhenReady(on: mySerializationQueue) {
       // ...
   renderSynchronizer.setRate(1.0, time: mediaTime)
```

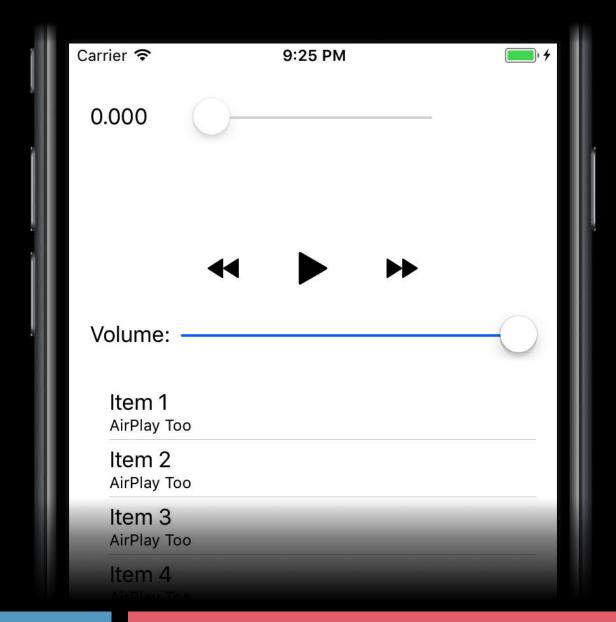
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   audioRenderer.stopRequestingMediaData()
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   audioRenderer.requestMediaDataWhenReady(on: mySerializationQueue) {
        // ...
   renderSynchronizer.setRate(1.0, time: mediaTime)
```

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// Seek - AVSampleBufferAudioRenderer
func seek(toMediaTime mediaTime: CMTime) {
   renderSynchronizer.setRate(0.0, time: kCMTimeZero)
   audioRenderer.stopRequestingMediaData()
   audioRenderer.flush()
   myPrepareSampleGenerationForMediaTime(mediaTime)
   audioRenderer.requestMediaDataWhenReady(on: mySerializationQueue) {
       // ...
   renderSynchronizer.setRate(1.0, time: mediaTime)
```

Carrier 🖘	9:25 PM	• +
0.000)	
	↔ ▶ ₩	
Volume: —		$\overline{}$
Item 1 AirPlay Too		
Item 2 AirPlay Too		
Item 3 AirPlay Too		
Item 4 AirPlay Too		

Queue





Queue Item 1 Item 3

Timelines

Queue Item 1 Item 3

Timelines



Timelines



Timelines



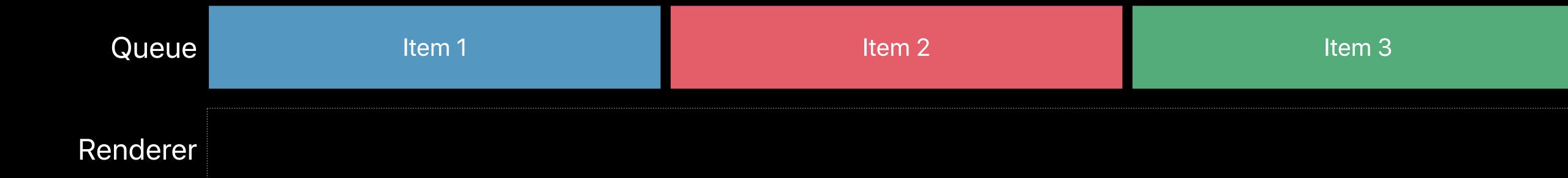
Enqueuing Audio Renderer



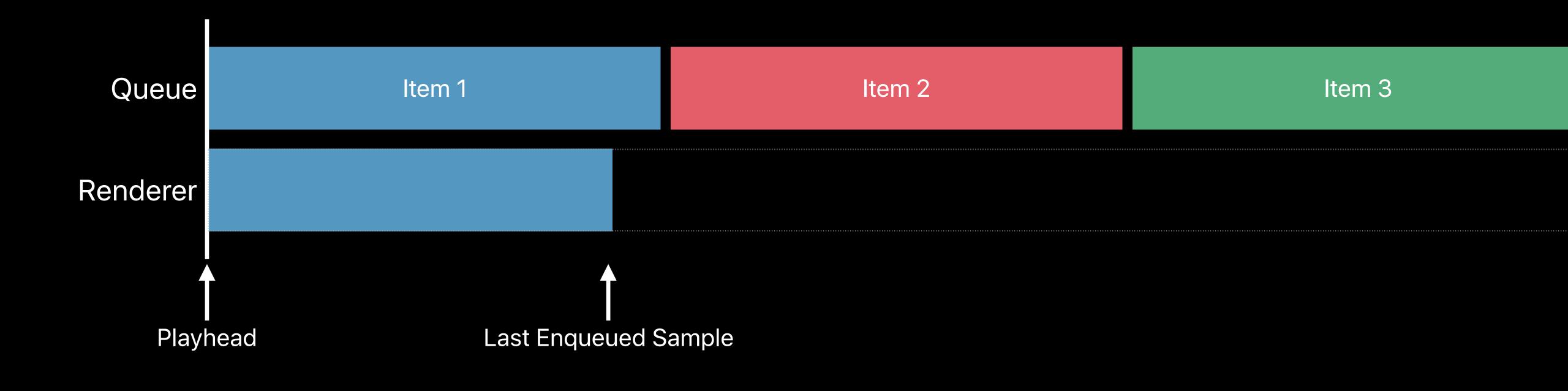
Enqueuing Audio Renderer



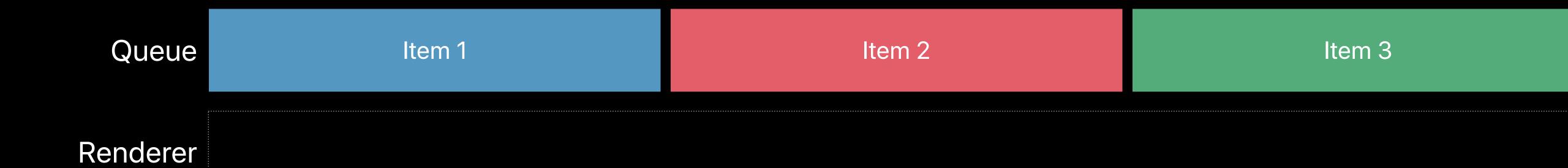
Enqueuing Audio Renderer



Deep Audio Buffer Levels



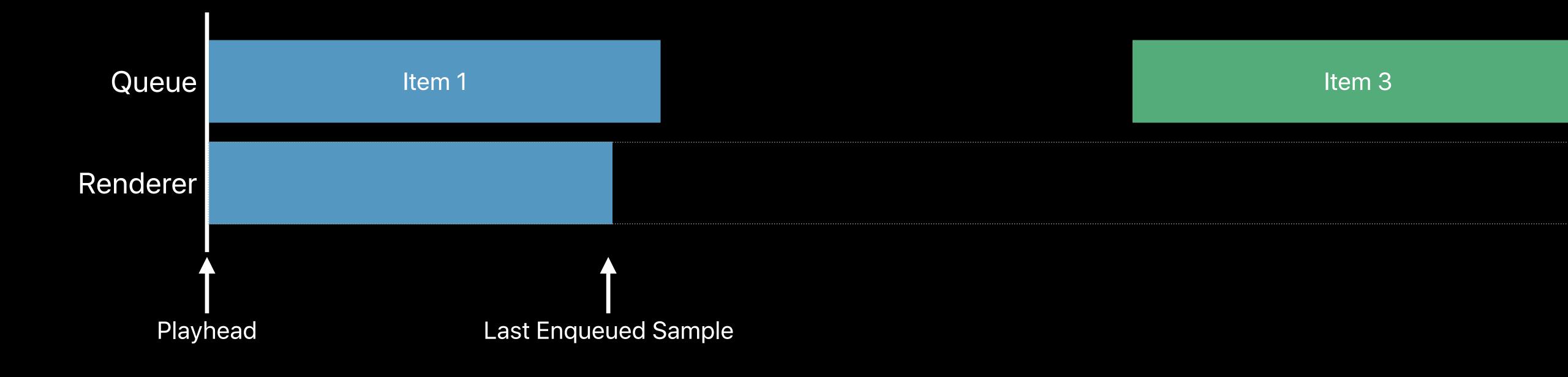
Deep Audio Buffer Levels



Editing



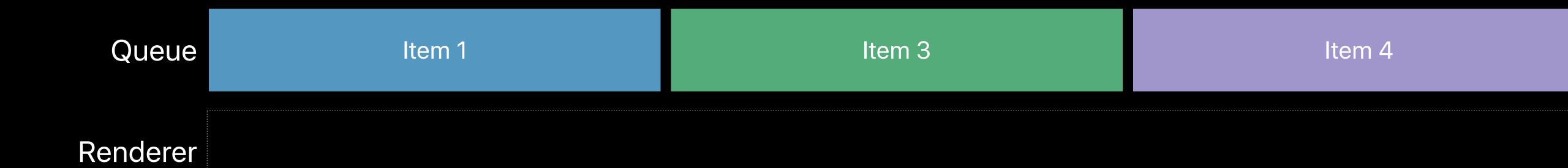
Editing



Editing

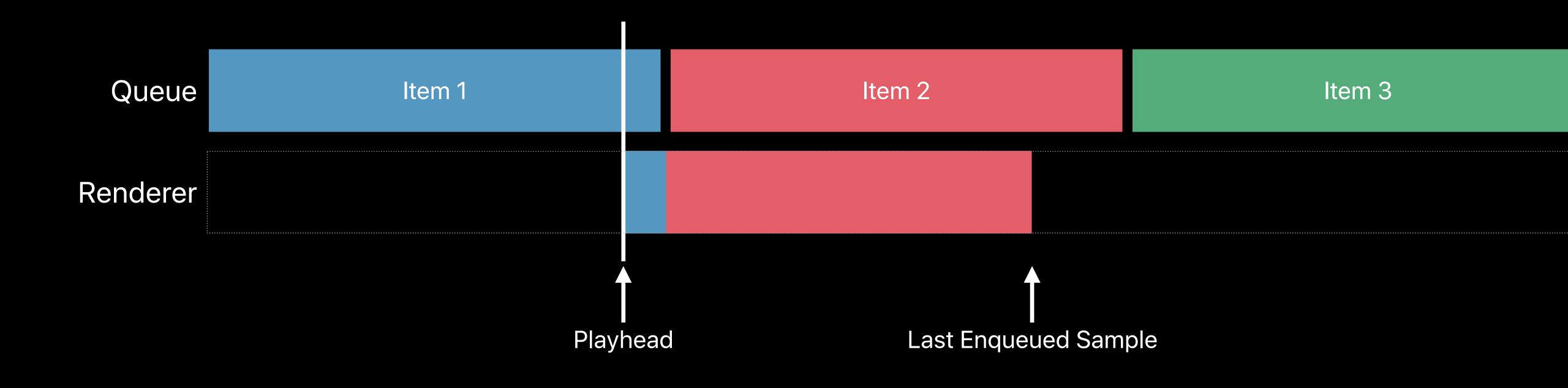


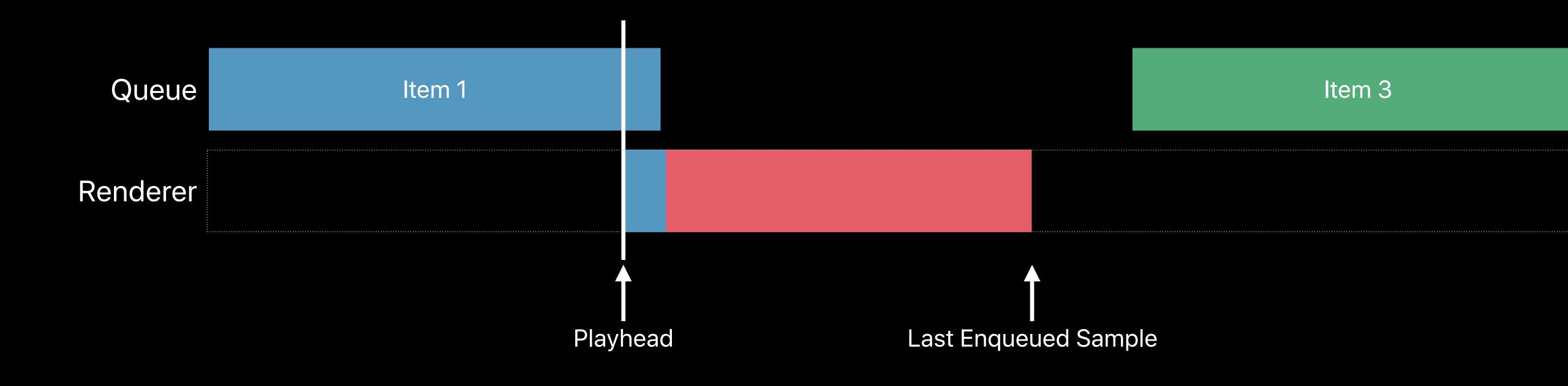
Play Queues Editing

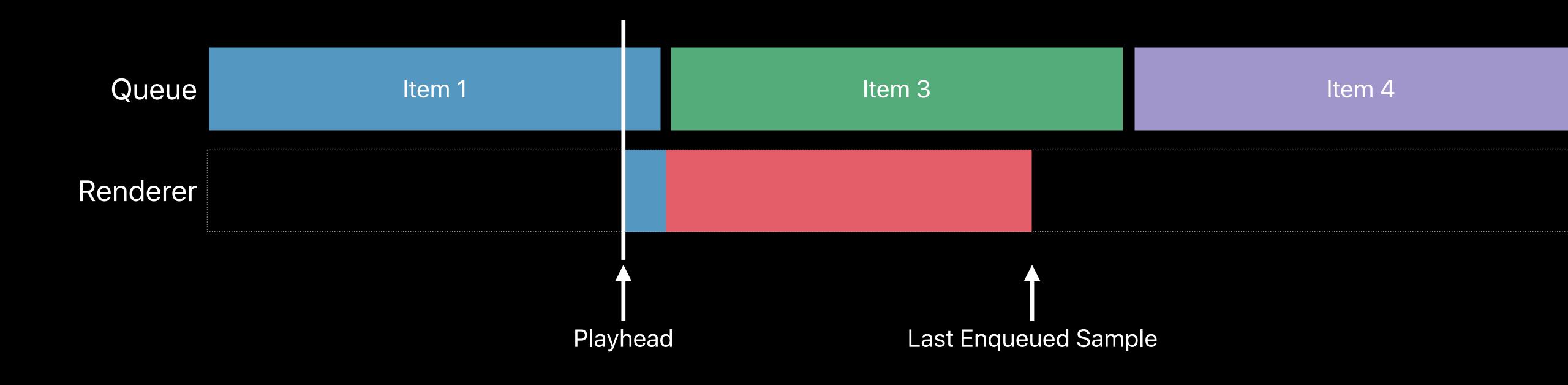






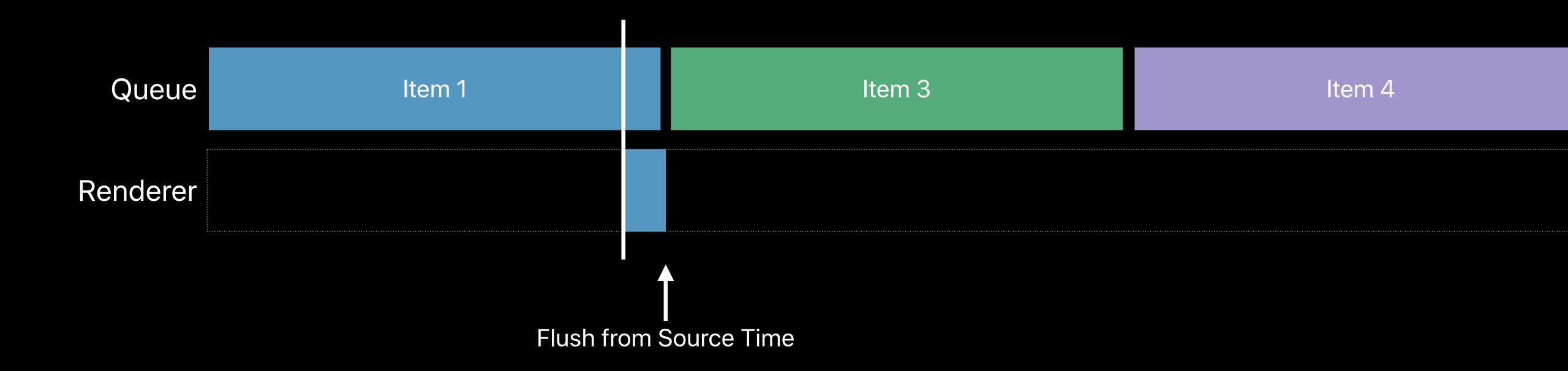


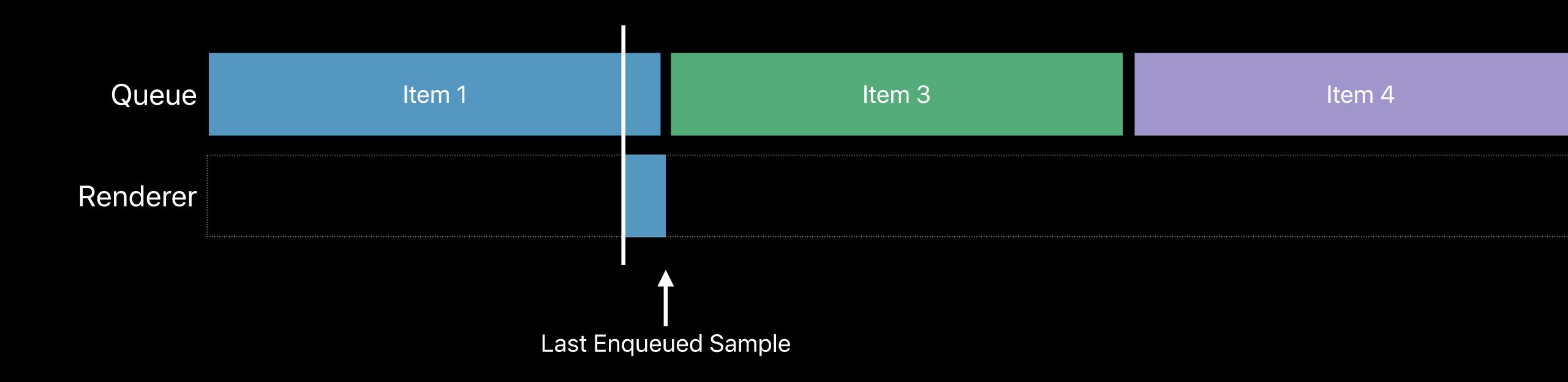


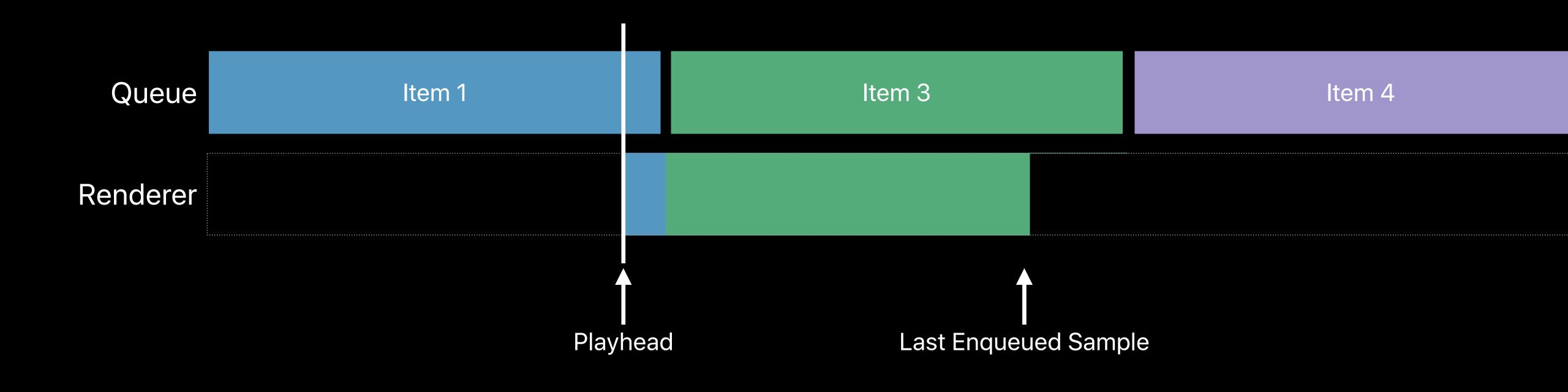




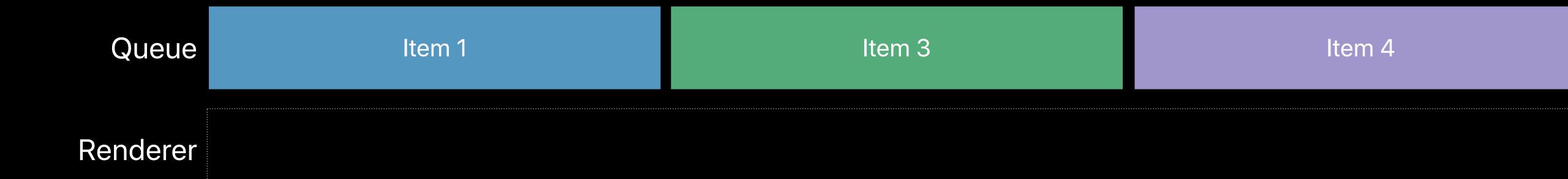












1. Stop enqueueing audio data

- 1. Stop enqueueing audio data
- 2. flush(fromSourceTime:)

- 1. Stop enqueueing audio data
- 2. flush(fromSourceTime:)
- 3. Wait for the callback

Gotchas

Gotchas

Flush may fail!

Gotchas

Flush may fail!

Source time is too close to playhead

Gotchas

Flush may fail!

Source time is too close to playhead

Wait for the callback!

```
// FlushFromSourceTime - AVSampleBufferAudioRenderer
func performFlush(fromSourceTime sourceTime: CMTime) {
   audioRenderer.stopRequestingMediaData()
    // App-specific logic to ensure no more media data is enqueued
   audioRenderer.flush(fromSourceTime: sourceTime) { (flushSucceeded) in
       if flushSucceeded {
            self.myPrepareSampleGenerationForMediaTime(sourceTime)
            audioRenderer.requestMediaDataWhenReady(on: mySerializationQueue) { /*...*/ }
        else {
            // Flush and interrupt playback
```

```
// FlushFromSourceTime - AVSampleBufferAudioRenderer
func performFlush(fromSourceTime sourceTime: CMTime) {
   audioRenderer.stopRequestingMediaData()
    // App-specific logic to ensure no more media data is enqueued
   audioRenderer.flush(fromSourceTime: sourceTime) { (flushSucceeded) in
       if flushSucceeded {
            self.myPrepareSampleGenerationForMediaTime(sourceTime)
            audioRenderer.requestMediaDataWhenReady(on: mySerializationQueue) { /*...*/ }
        else {
            // Flush and interrupt playback
```

```
// FlushFromSourceTime - AVSampleBufferAudioRenderer
func performFlush(fromSourceTime sourceTime: CMTime) {
   audioRenderer.stopRequestingMediaData()
    // App-specific logic to ensure no more media data is enqueued
   audioRenderer.flush(fromSourceTime: sourceTime) { (flushSucceeded) in
       if flushSucceeded {
            self.myPrepareSampleGenerationForMediaTime(sourceTime)
            audioRenderer.requestMediaDataWhenReady(on: mySerializationQueue) { /*...*/ }
        else {
            // Flush and interrupt playback
```

```
// FlushFromSourceTime - AVSampleBufferAudioRenderer
func performFlush(fromSourceTime sourceTime: CMTime) {
   audioRenderer.stopRequestingMediaData()
    // App-specific logic to ensure no more media data is enqueued
   audioRenderer.flush(fromSourceTime: sourceTime) { (flushSucceeded) in
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```

Audio Format Support

AVSampleBufferAudioRenderer

AVSampleBufferAudioRenderer

AVSampleBufferAudioRenderer

All platform-supported audio formats

AVSampleBufferAudioRenderer

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• e.g. LPCM, AAC, mp3, or ALAC

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- e.g. 44.1 kHZ or 48 kHZ

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Mixed formats may be enqueued

AVSampleBufferAudioRenderer

All platform-supported audio formats

- e.g. LPCM, AAC, mp3, or ALAC
- e.g. 44.1 kHZ or 48 kHZ
- various bit depths

Mixed formats may be enqueued

Renderer AAC @ 44.1kHZ

MP3 @ 48kHz

16bit ALAC @ 48kHz

AVSampleBufferAudioRenderer

AVSampleBufferAudioRenderer

Original audio format (decrypted)

AVSampleBufferAudioRenderer

Original audio format (decrypted)

Interleaved channel formats

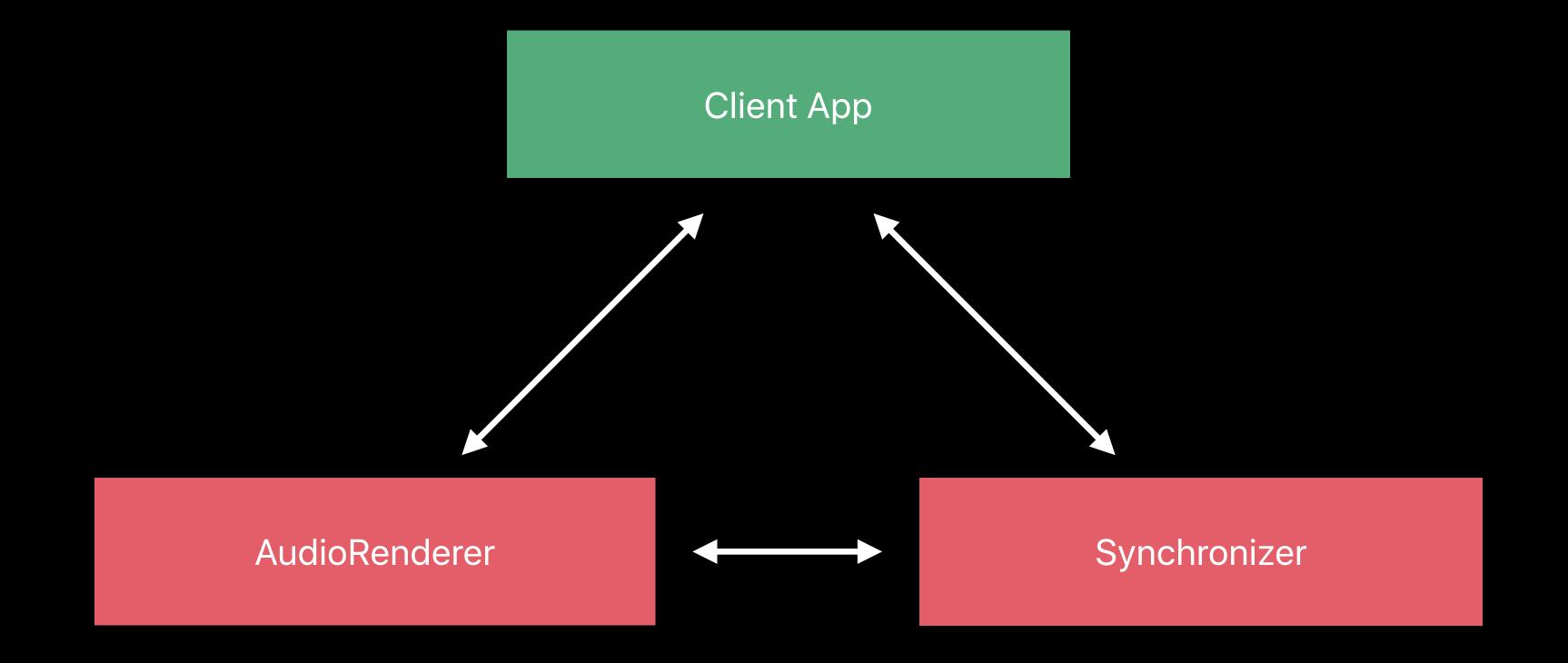
AVSampleBufferAudioRenderer

Original audio format (decrypted)

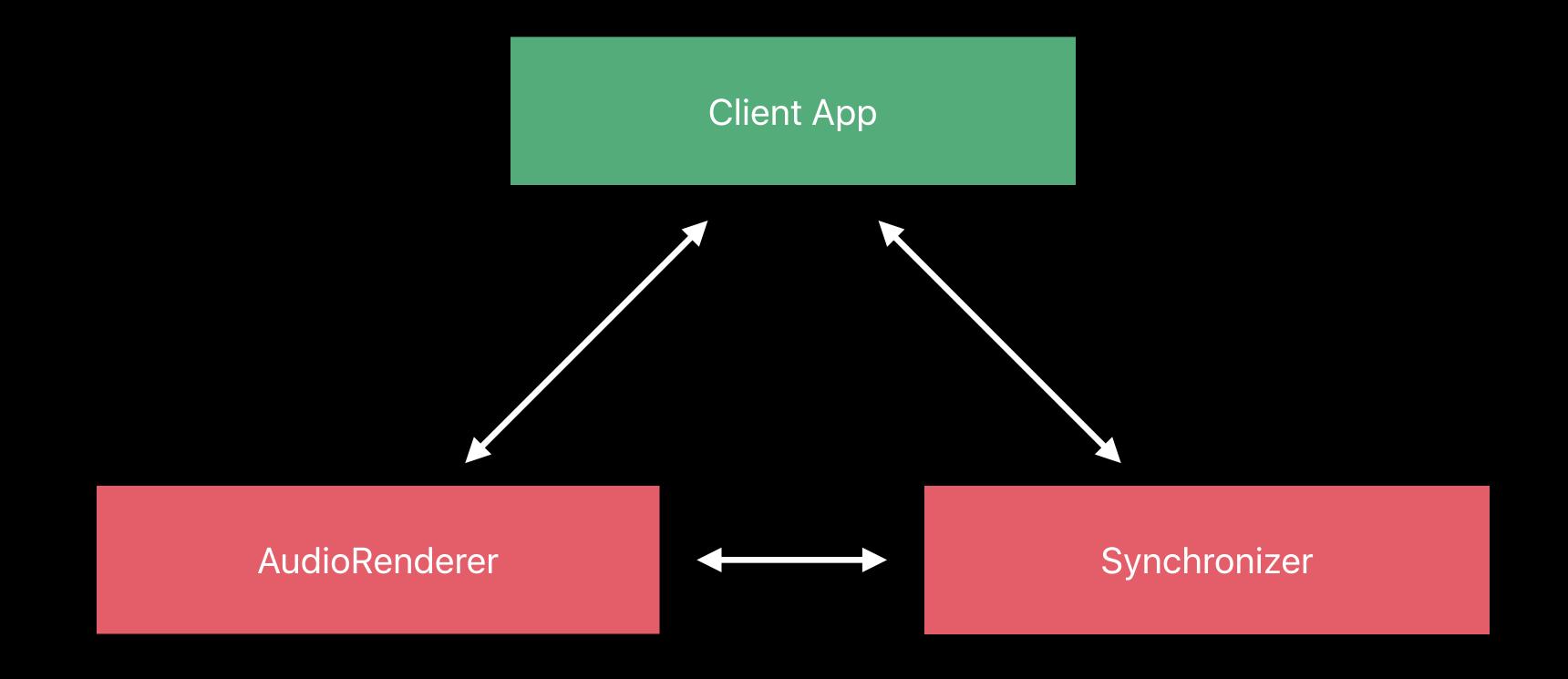
Interleaved channel formats

1-2 seconds of audio per CMSampleBuffer

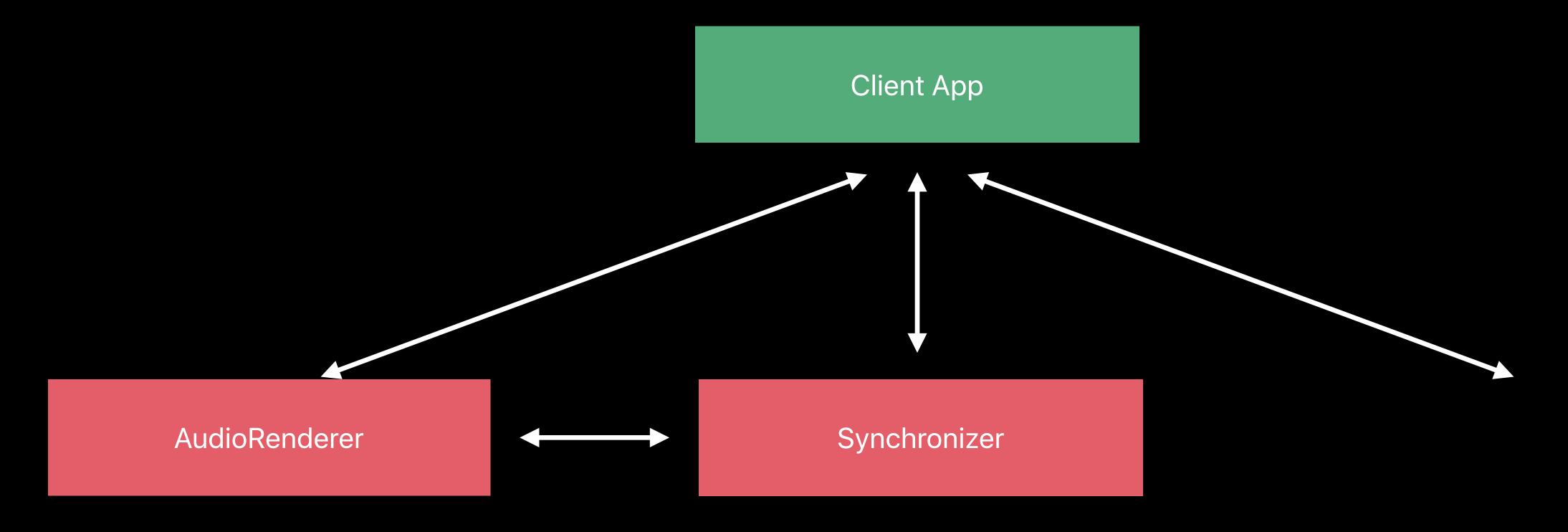
AVSampleBufferDisplayLayer

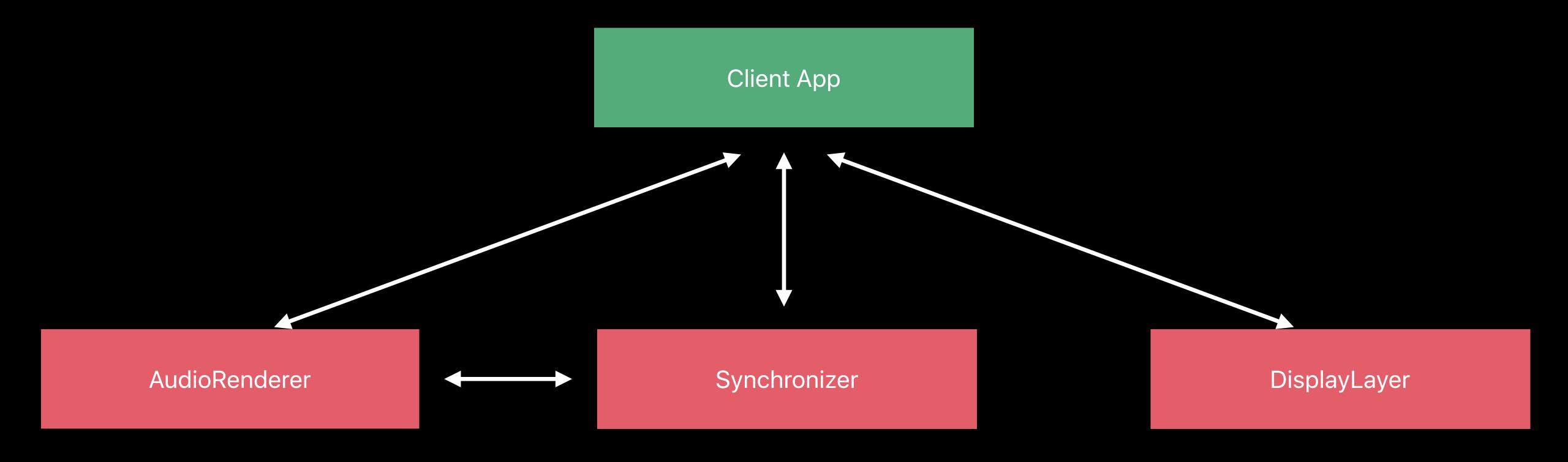


AVSampleBufferDisplayLayer

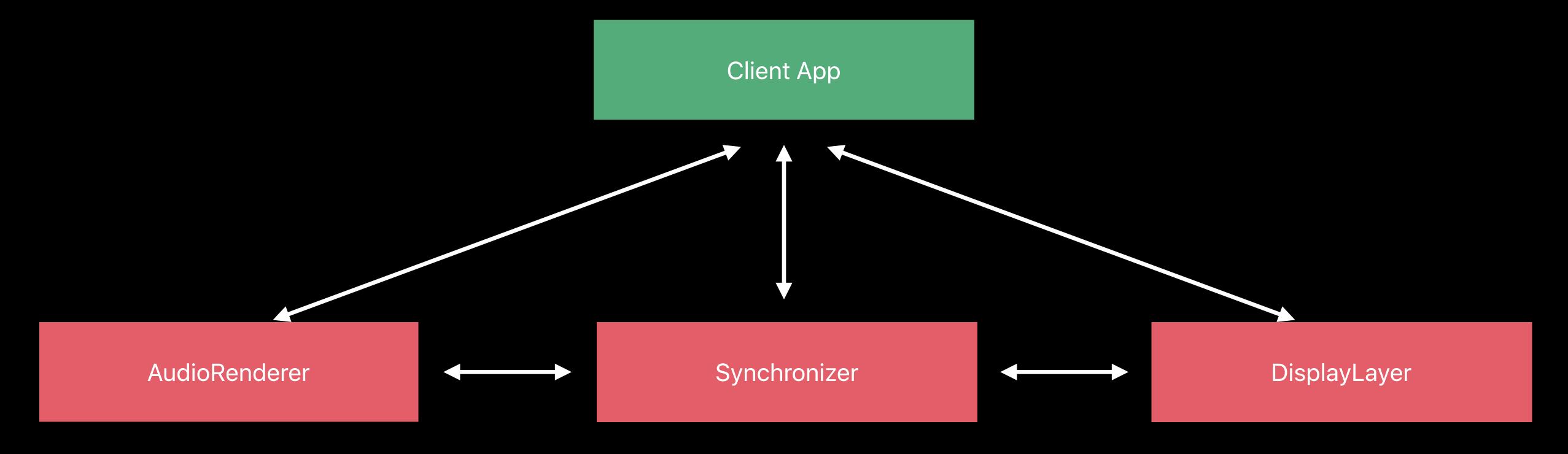


AVSampleBufferDisplayLayer

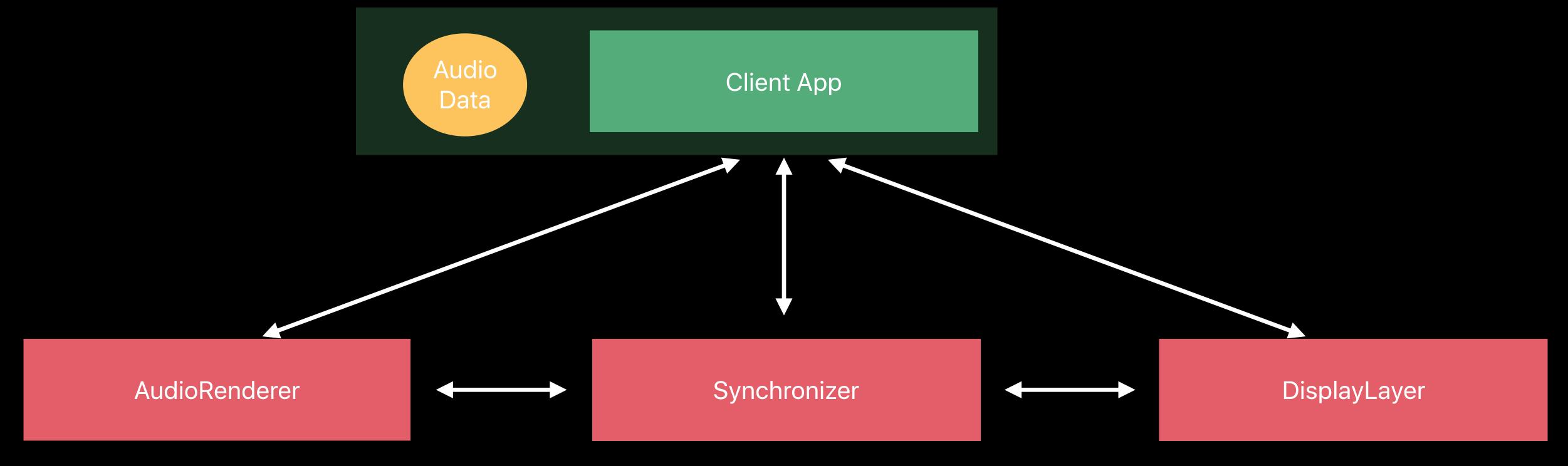




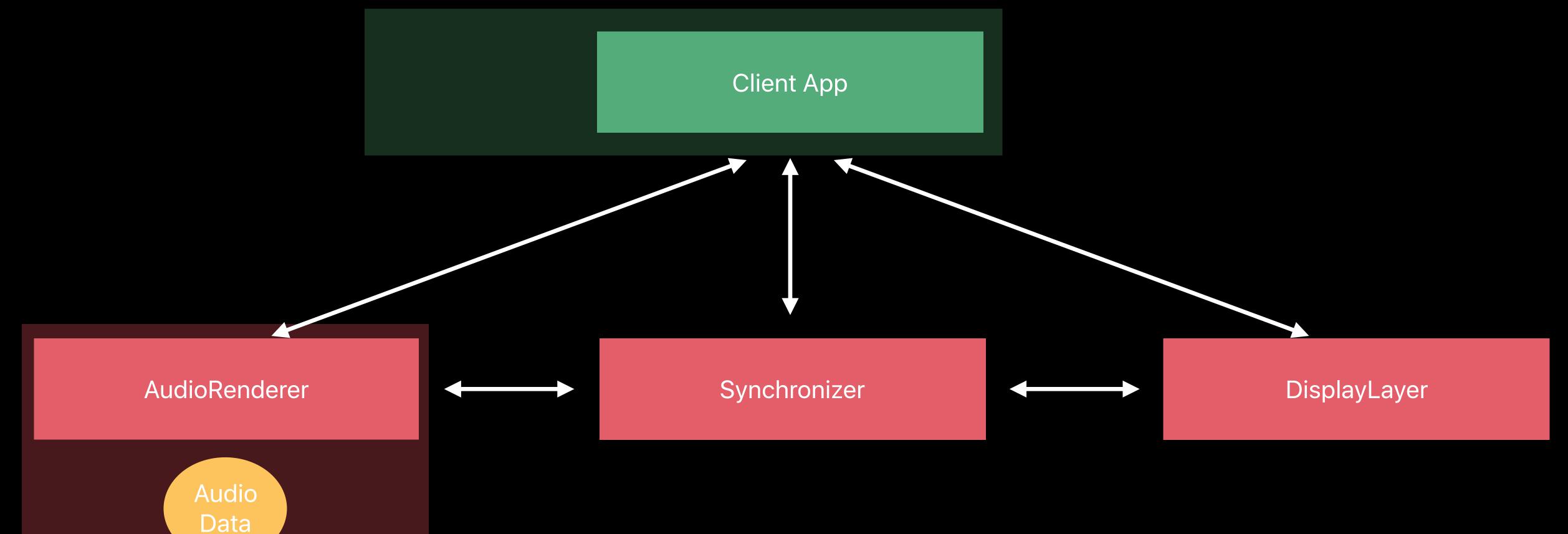








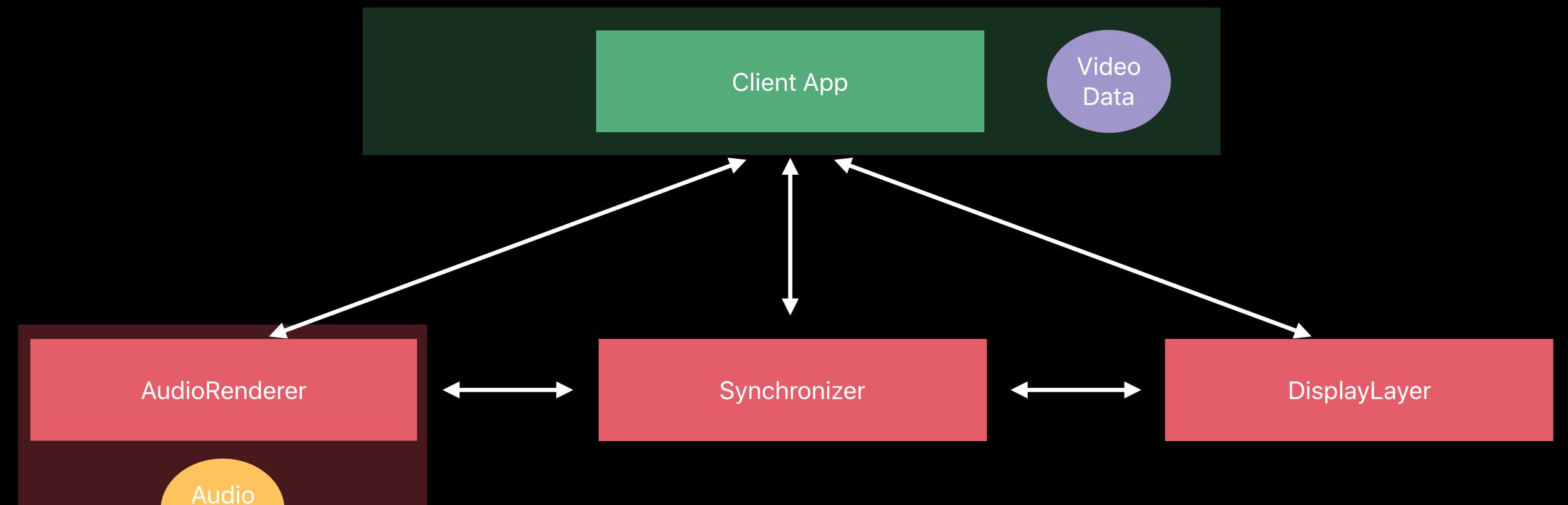




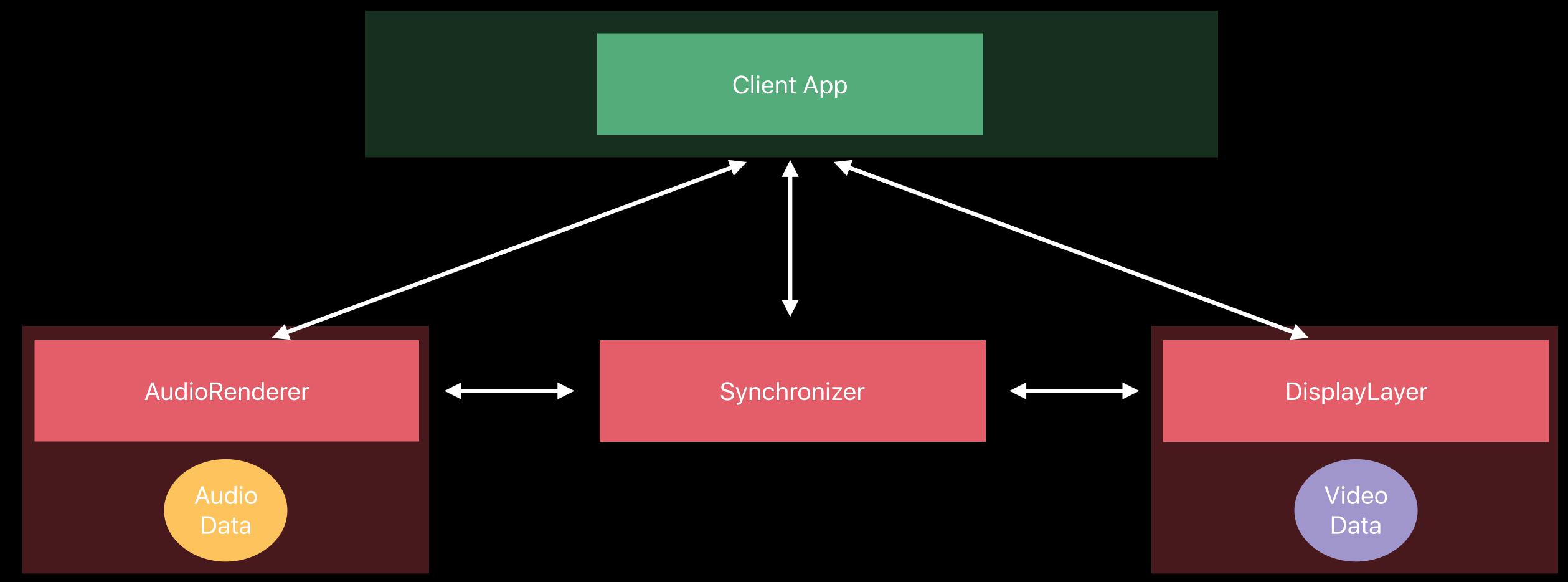
AVSampleBufferDisplayLayer

Data

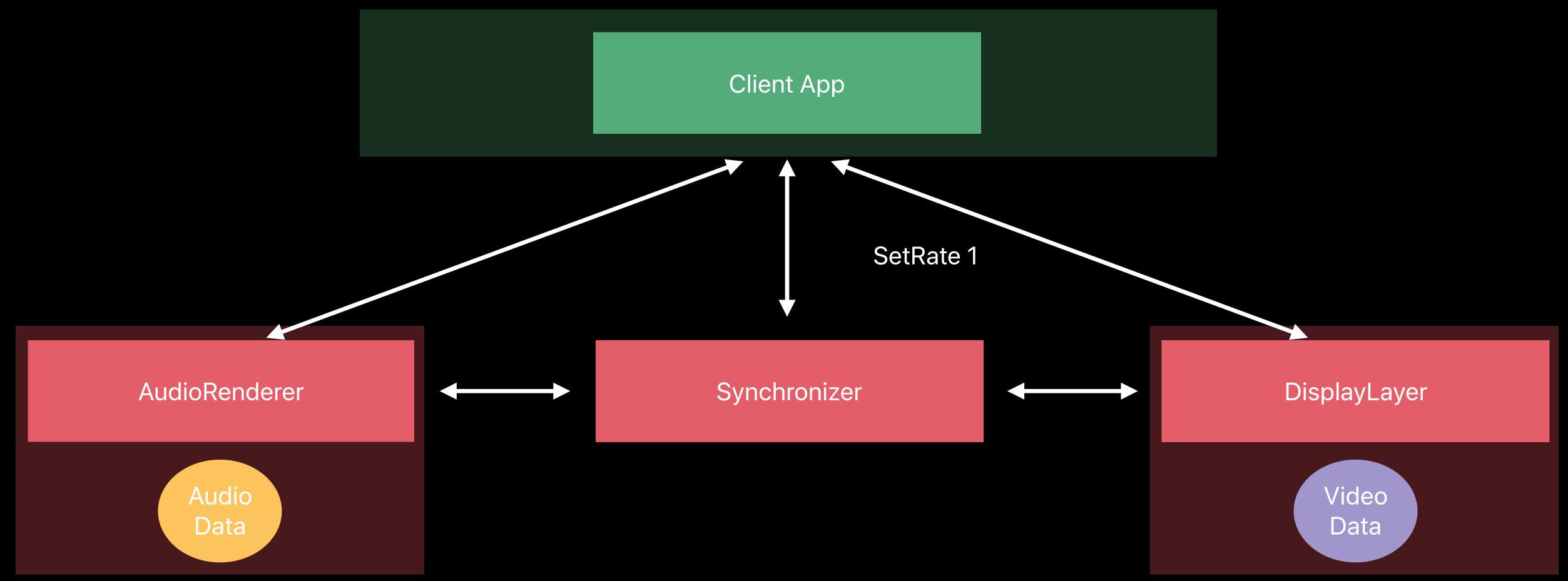




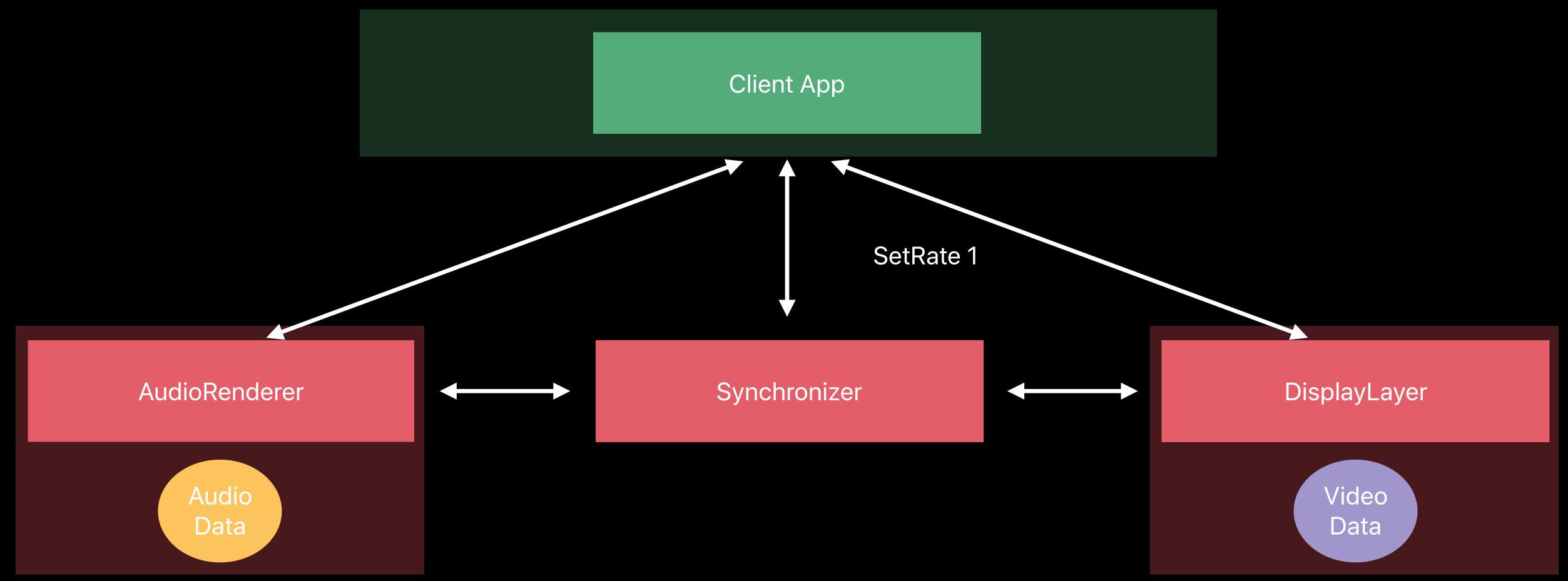






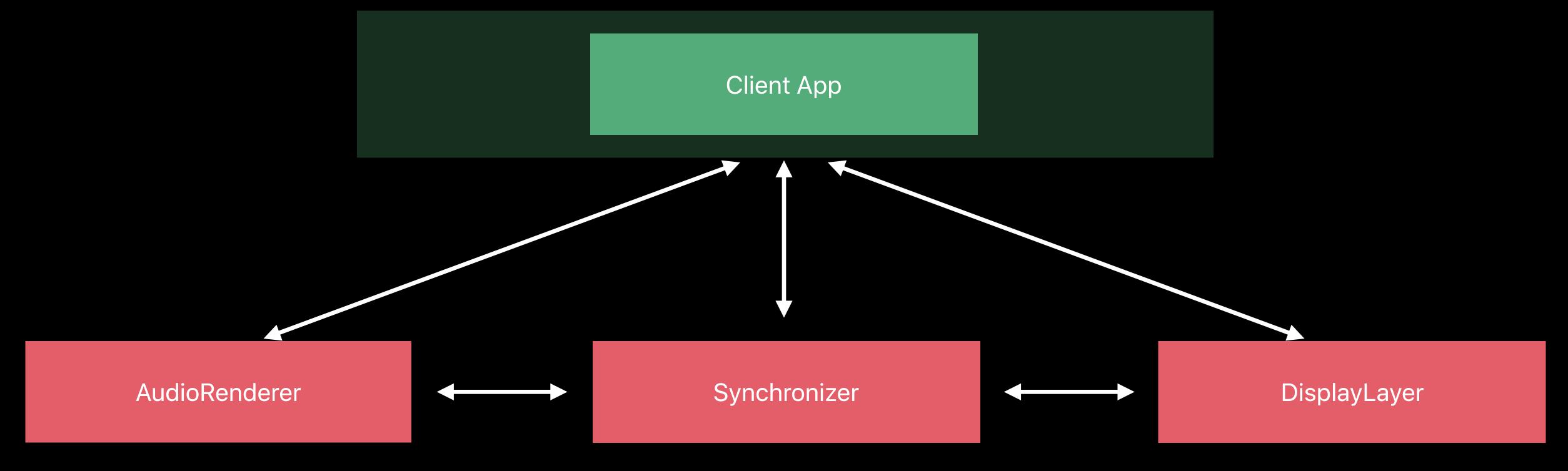






AVSampleBufferDisplayLayer





Audio Data



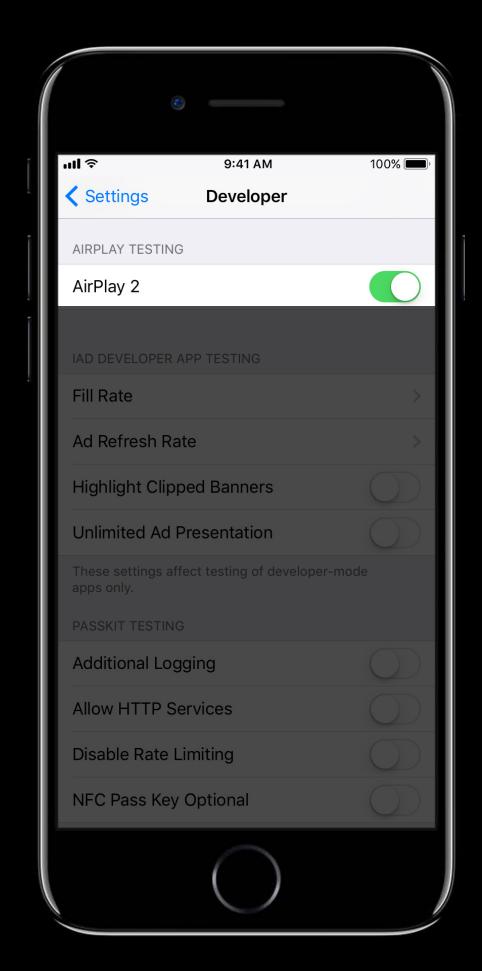
Availability

APIs and enhanced buffering

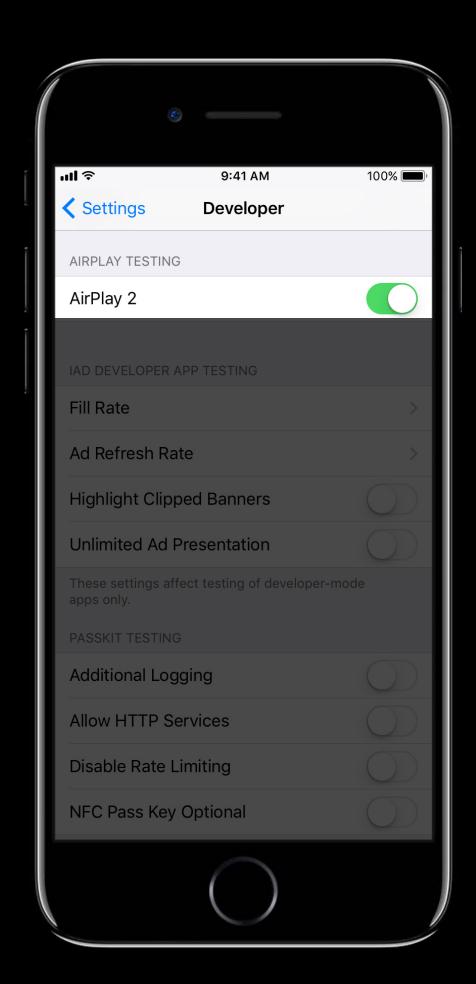
Beta 1

APIs and enhanced buffering

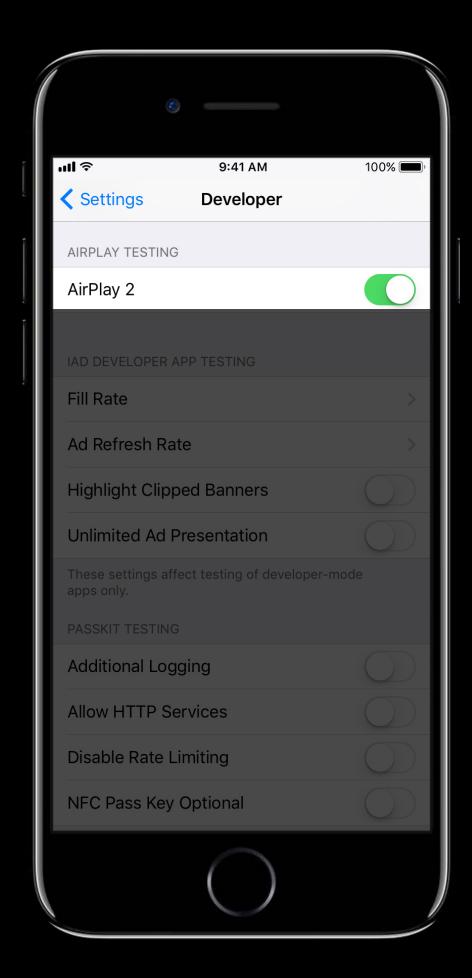
Beta 1



APIs and enhanced buffering	Beta 1
Multi-room audio	Upcoming Beta



APIs and enhanced buffering	Beta 1
Multi-room audio	Upcoming Beta
Available to users	Customer Release



AirPlay 2 introduces many new features for audio

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Long-form audio applications can enable AirPlay 2 with a few steps

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AirPlay 2 adoption can begin today

More Information

https://developer.apple.com/wwdc17/509

Related Sessions

What's New in Audio	WWDC 2017
Introducing MusicKit	WWDC 2017

Labs

AirPlay Lab Technology Lab A Fri 9:00AM-11:00AM

SWWDC17