

# What's New in Core Audio for iOS

Session 602

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# Introduction

- Overview of new audio features in iOS 7
- Focus on one exciting new technology

# Audio Input Selection



- Select among available audio inputs
- Choose which microphone on multi-mic devices
- Set microphone polar pattern
  - Achieves directivity through beam forming processing
  - e.g., cardioid, subcardioid
- See `<AVFoundation/AVAudioSession.h>`

# Multichannel Audio Enhancements



- Discover maximum number of input and output channels
- Set preferred number of input and output channels
- Get audio channel labels
- See `<AVFoundation/AVAudioSession.h>`

# Open AL Extensions



- Per-source spatialization rendering quality
  - Improved high-quality rendering algorithm
  - Render to multichannel output hardware
- Output capture
- See `<OpenAL/oa1MacOSX_0ALExtensions.h>`

# Audio Queue Time-Pitch Capabilities



- Set playback rate
- Adjust playback pitch
- See `<AudioToolbox/AudioQueue.h>`

# Audio Recording Permission



- Recording now requires user approval
- One-time approval remembered for each application
- Changeable in Settings
- Silence until permission is granted
- Triggered by use of `AVAudioSession` categories that enable recording
- API for application control

```
{  
    ...  
    [[AVAudioSession sharedInstance] requestRecordPermission];  
    ...  
}
```

# Deprecated Audio Session C API



- Use `AVAudioSession`
- See `<AVFoundation/AVAudioSession.h>`



# New in iOS 7

## Summary



- Audio input selection
- Multichannel audio enhancements
- Open AL extensions
- Audio Queue time-pitch capabilities
- Audio recording permission
- Audio Session C API officially deprecated—Use `AVAudioSession`

# New in iOS 7

## Summary



- Audio input selection
- Multichannel audio enhancements
- Open AL extensions
- Audio Queue time-pitch capabilities
- Audio recording permission
- Audio Session C API officially deprecated—Use `AVAudioSession`
- Inter-app audio

# Inter-App Audio



- Stream audio between apps in real-time
- Built on familiar APIs

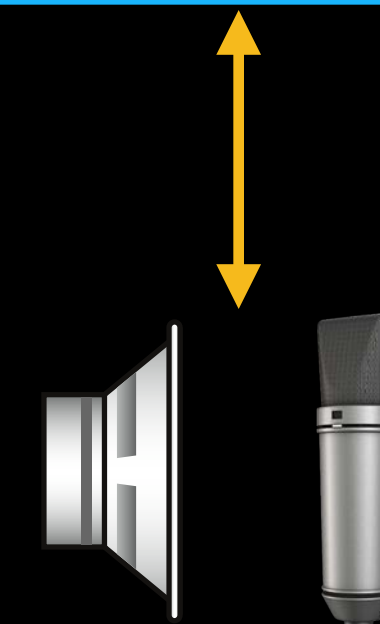
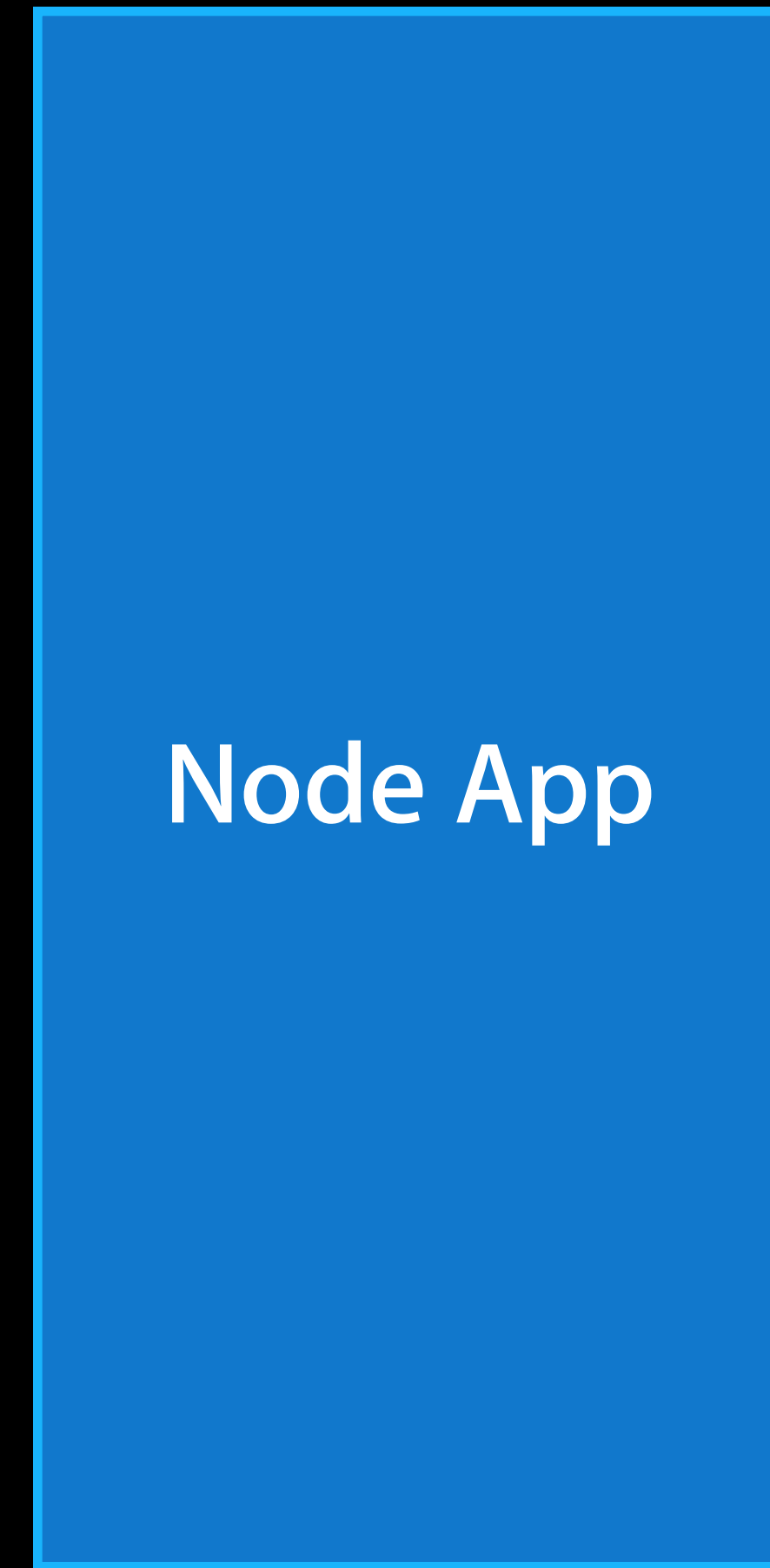
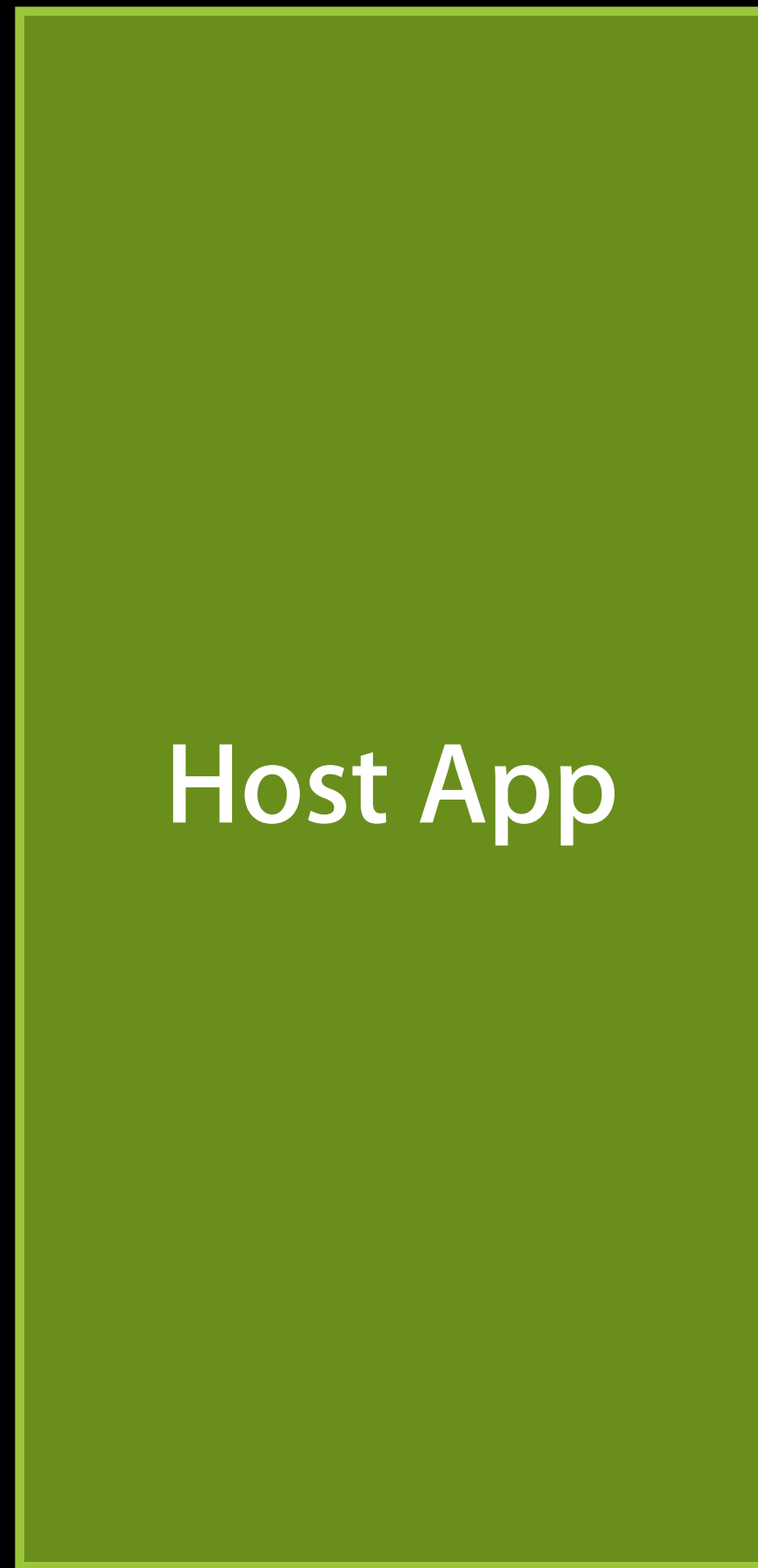
*Demo*

**Alec Little**  
GarageBand

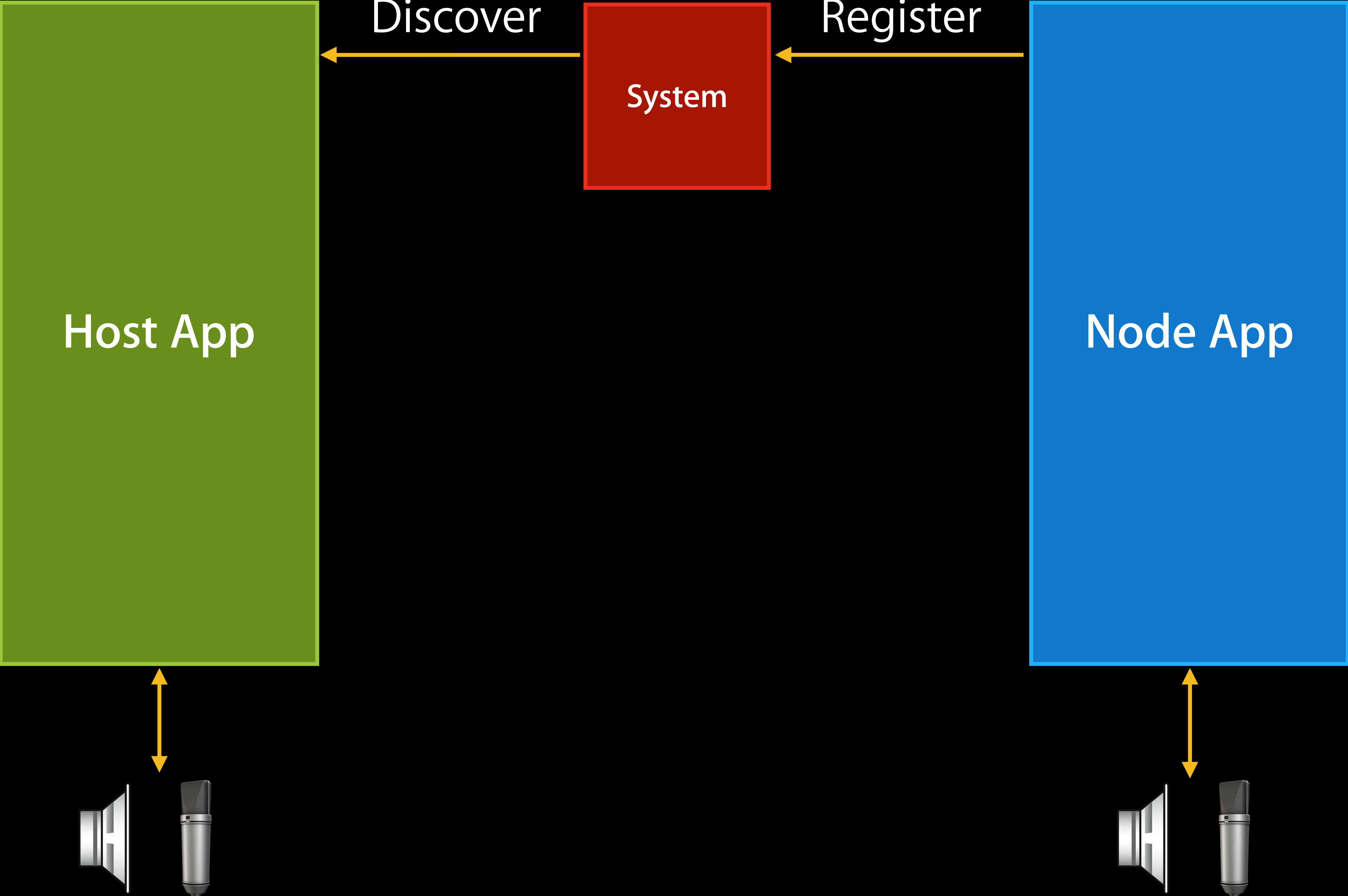
# Inter-App Audio in Detail

**Doug Wyatt**  
Core Audio Plumber

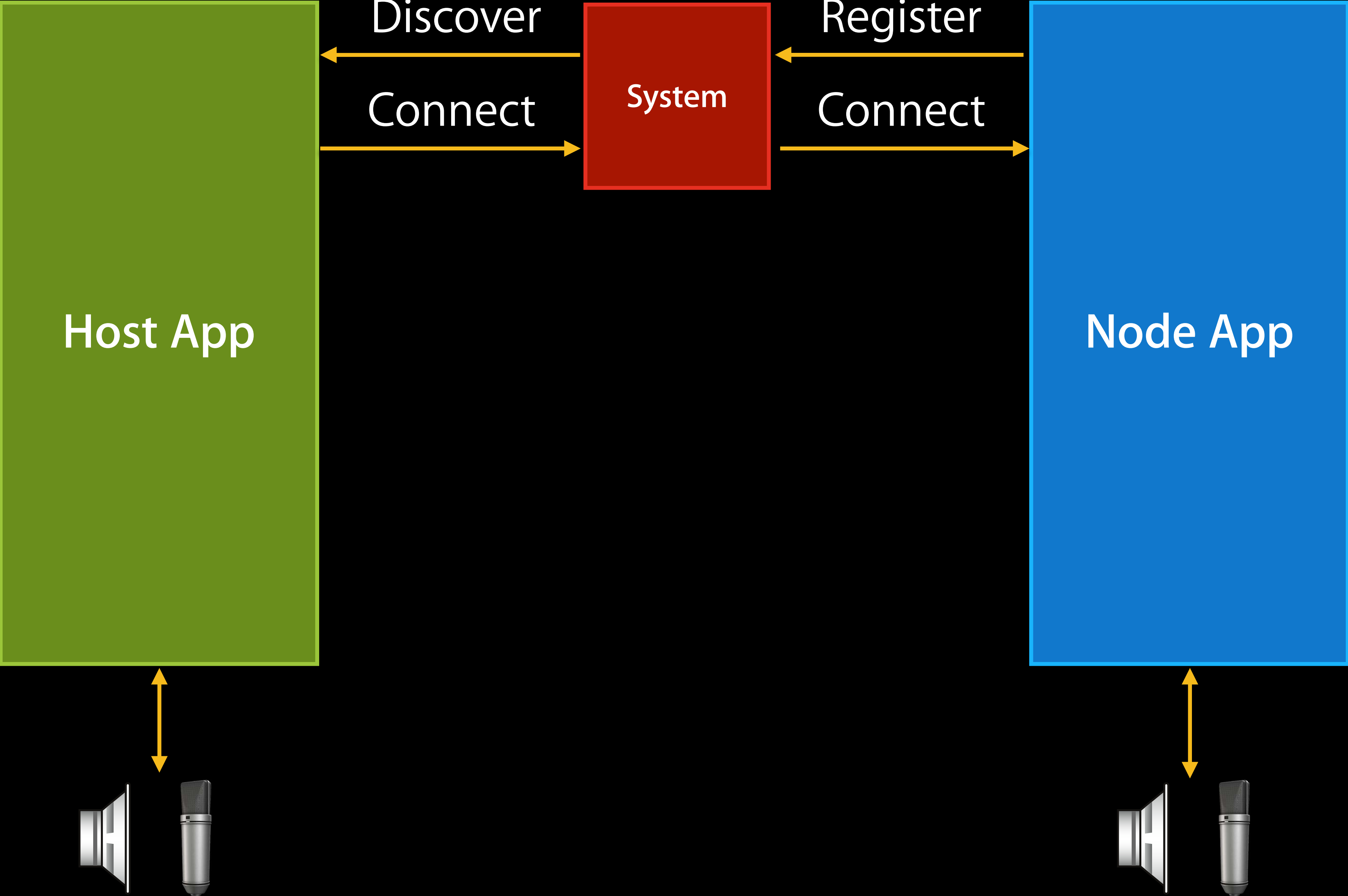
# API Overview



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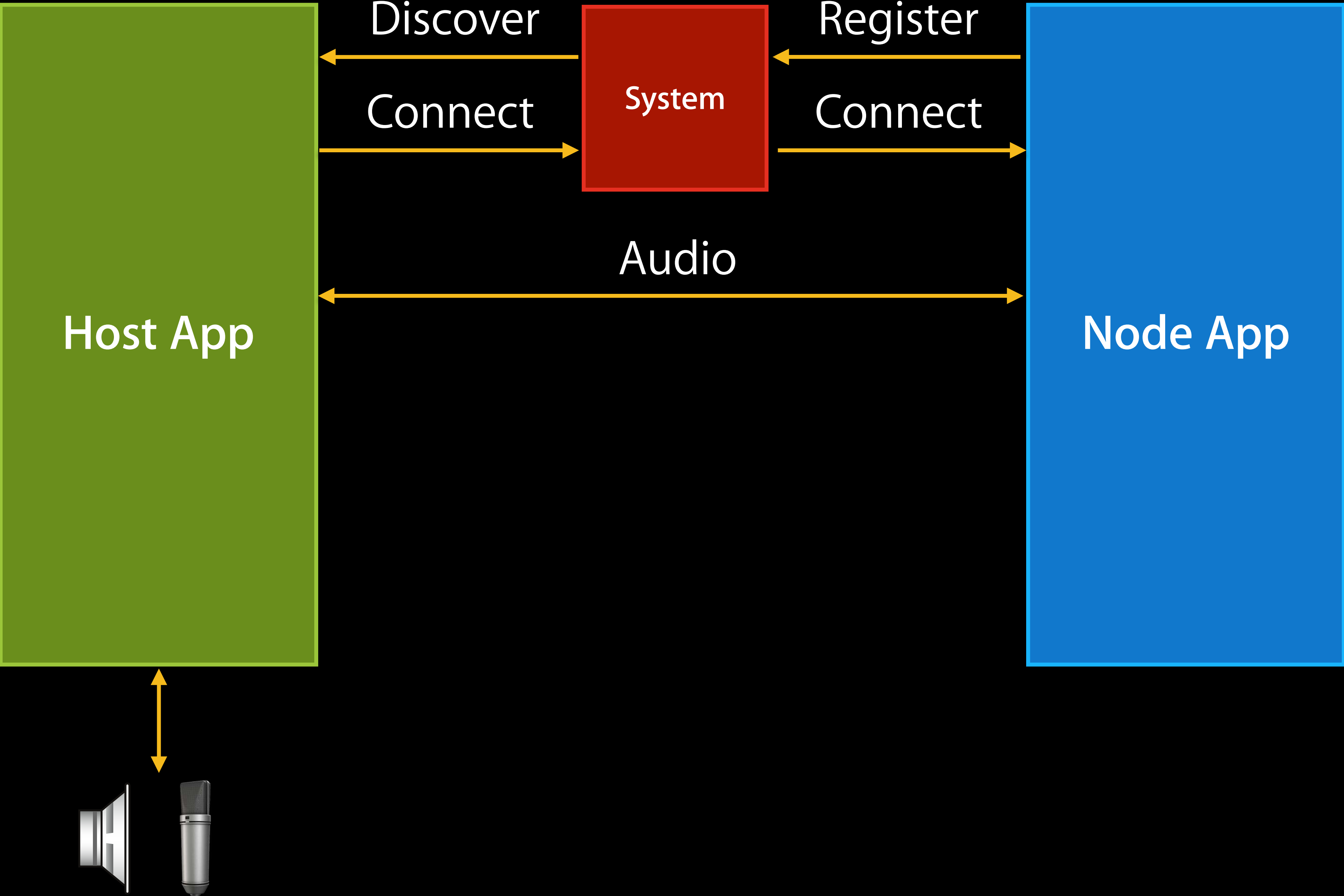


# API Overview

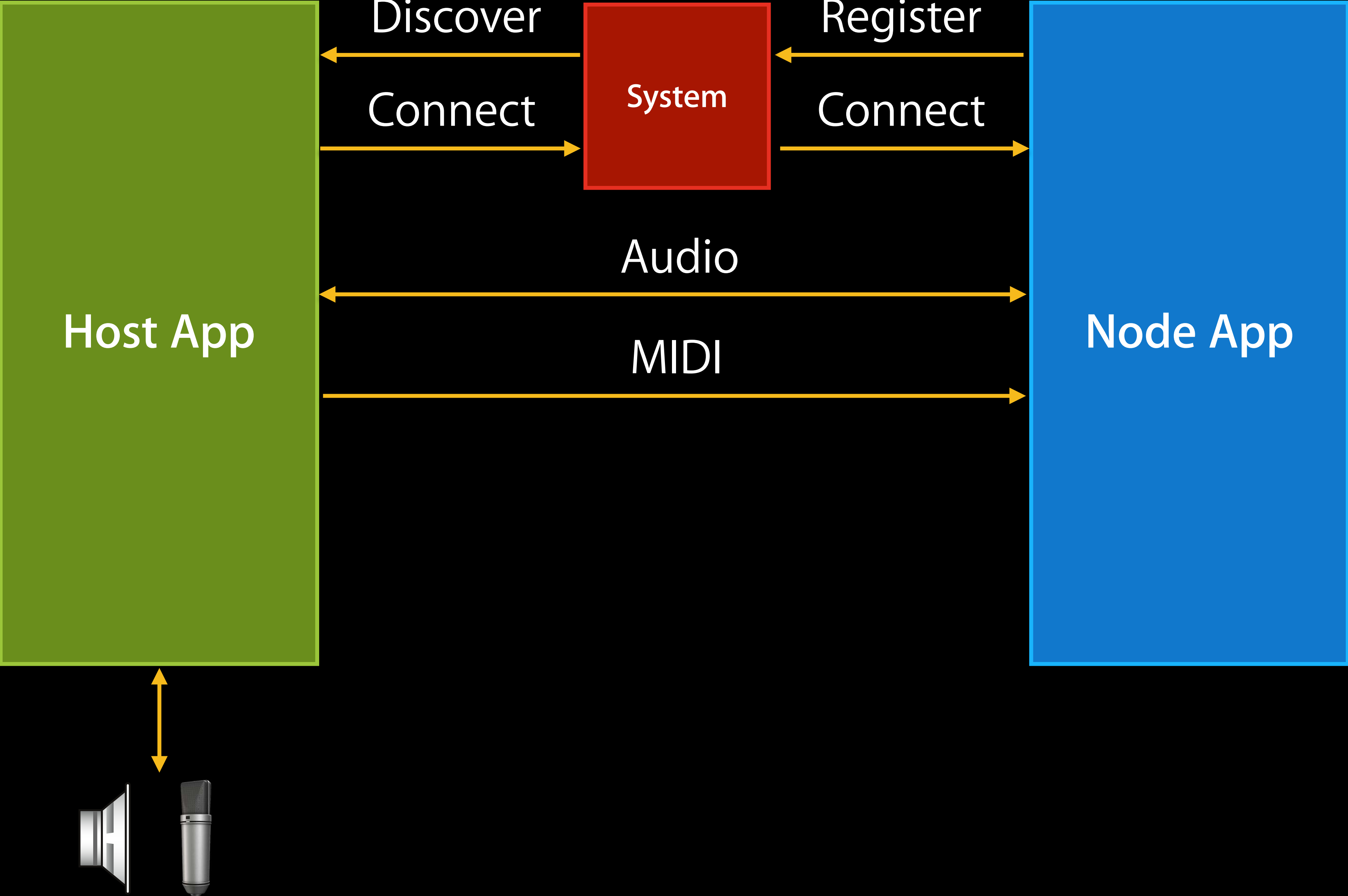




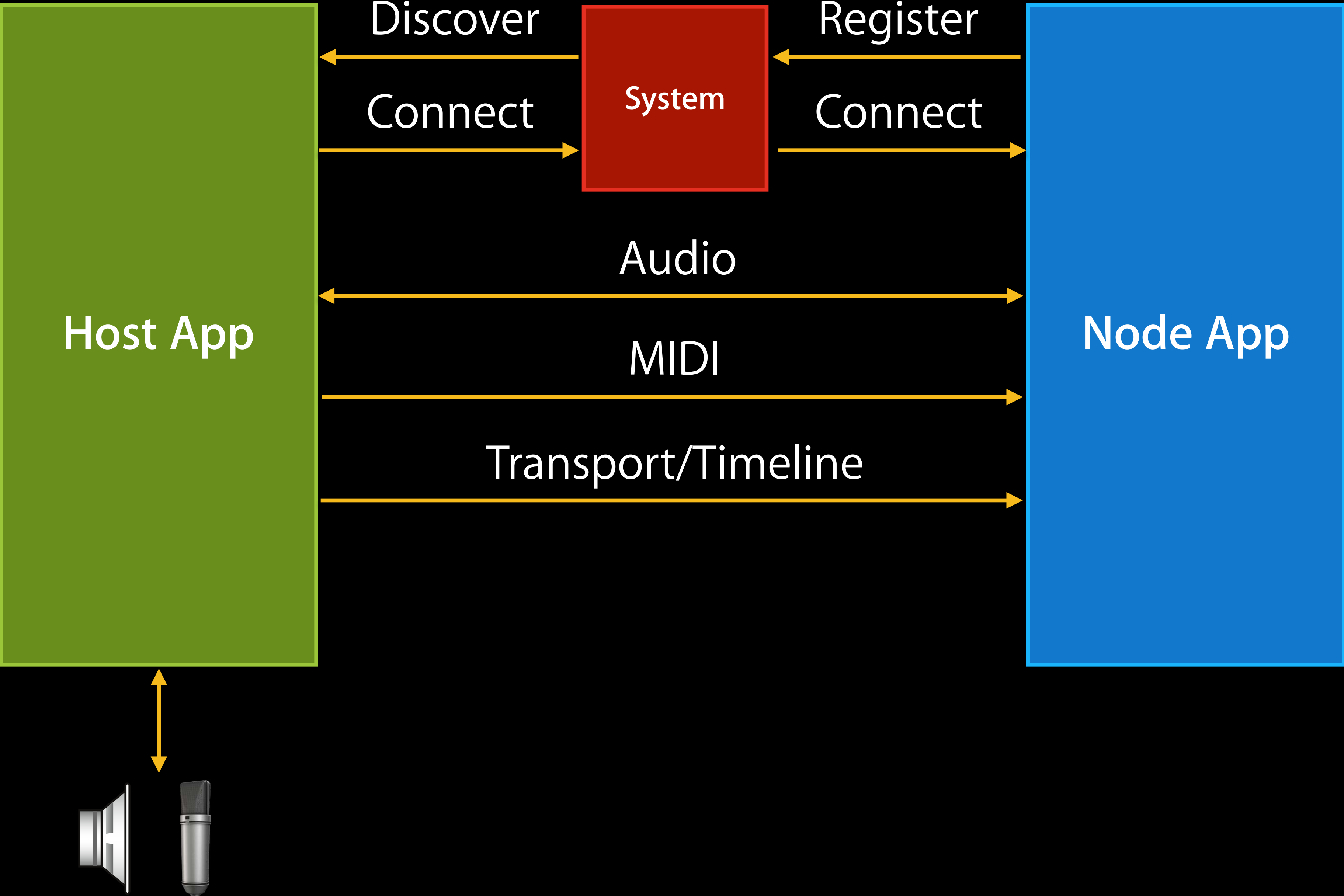
# API Overview



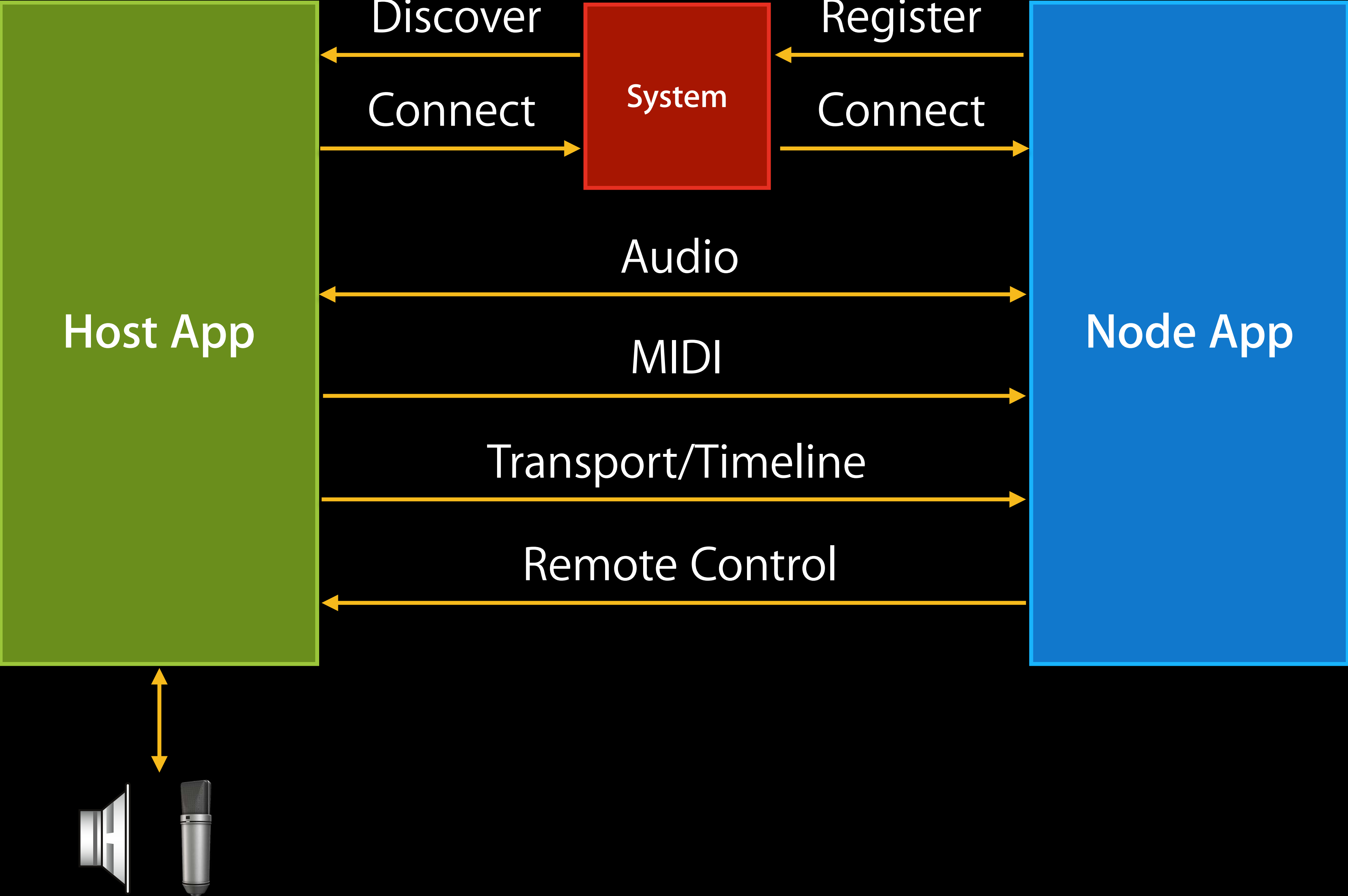
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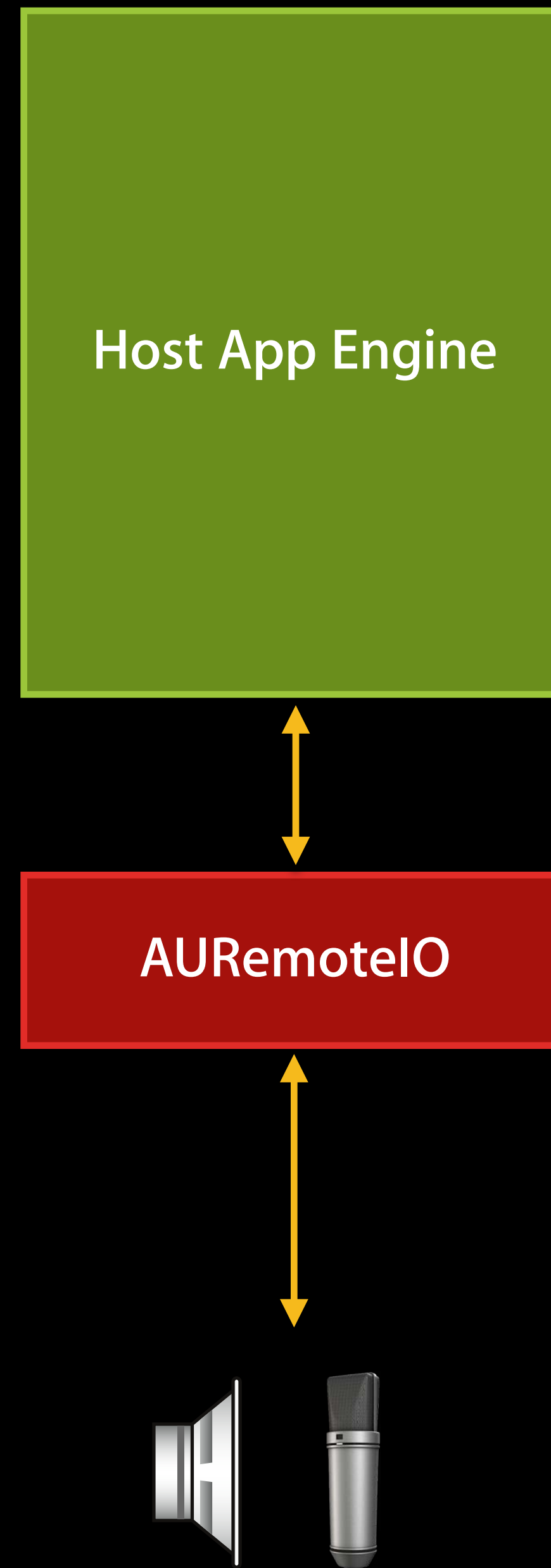


# API Overview



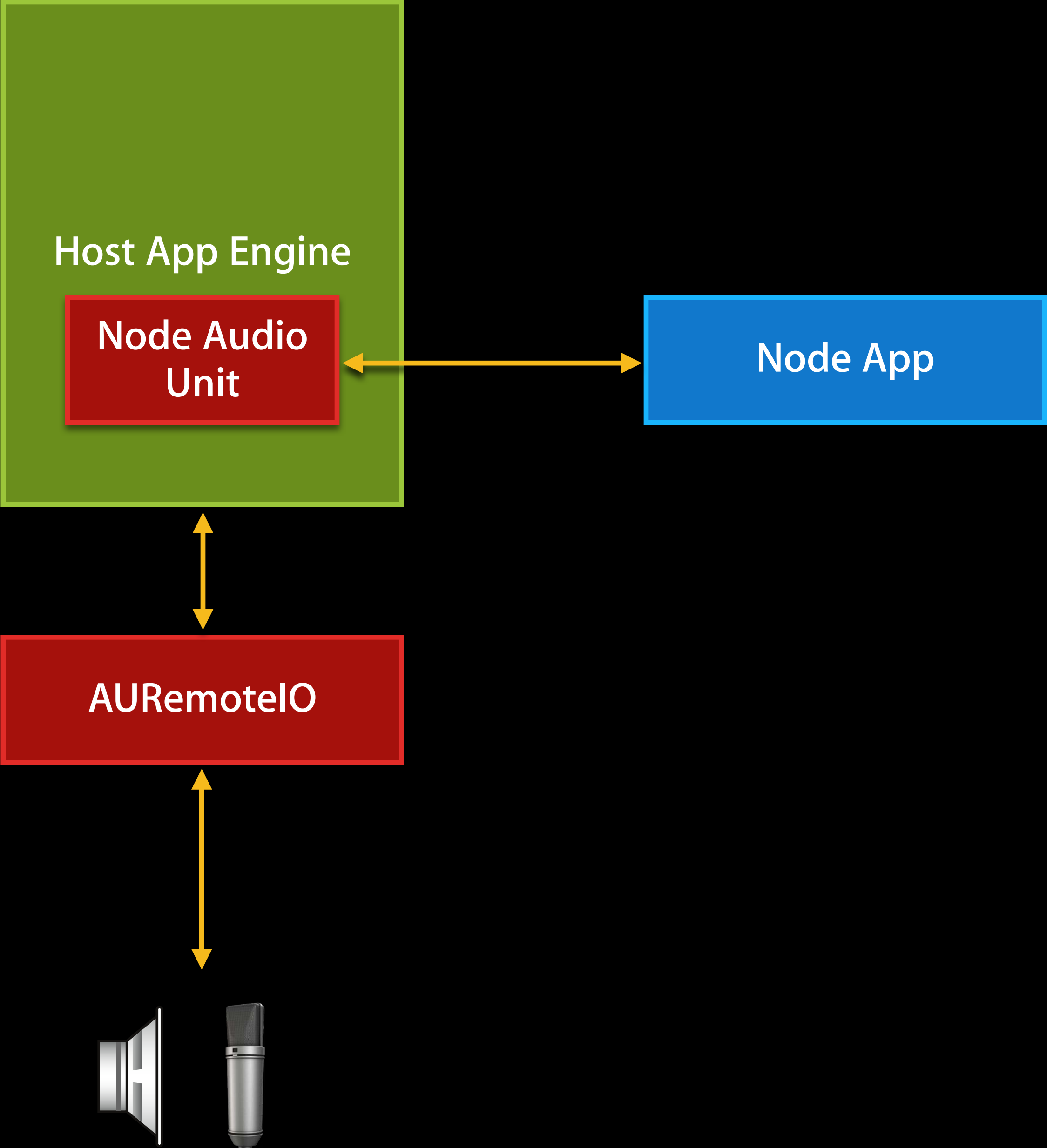
# Introduction

Host app



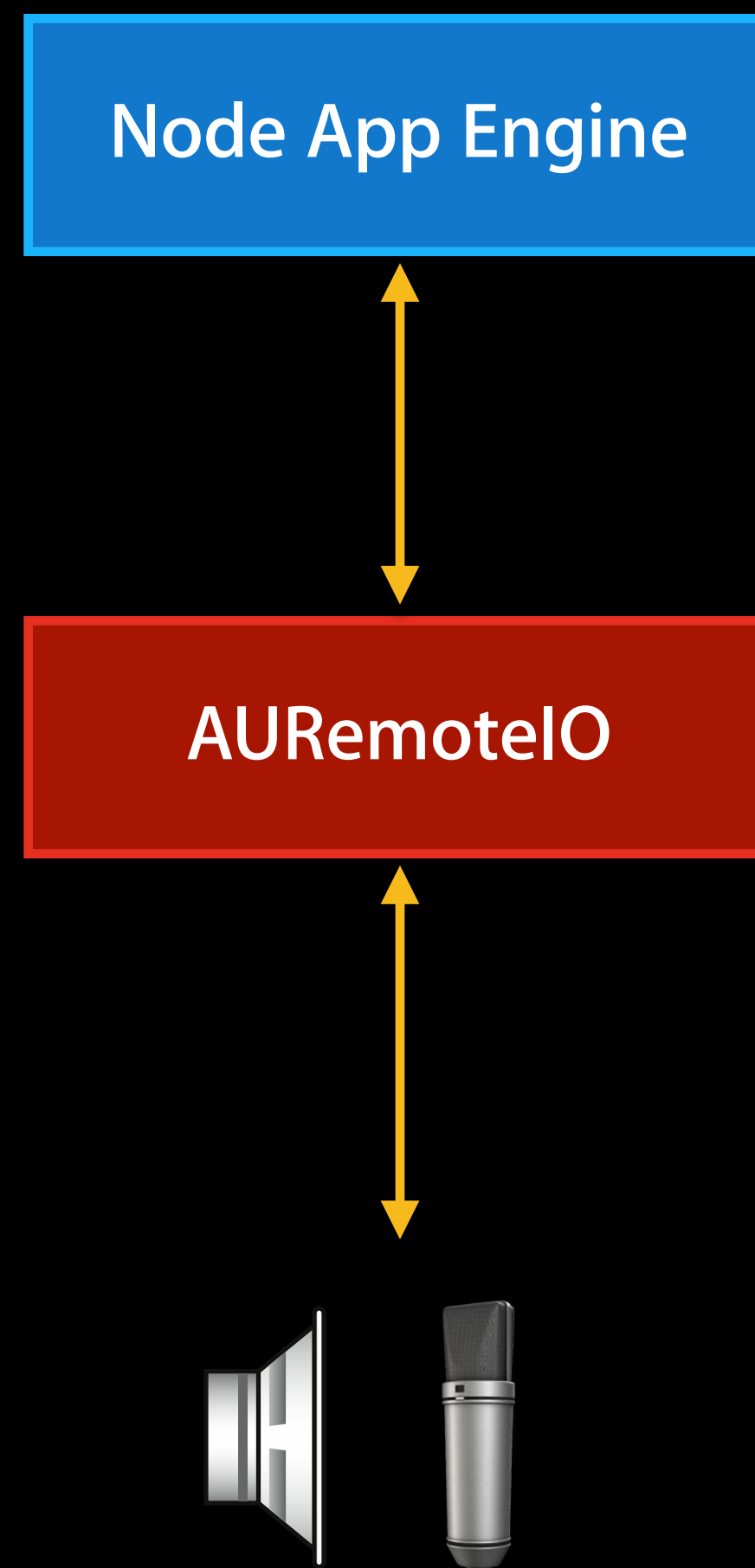
# Introduction

Host app



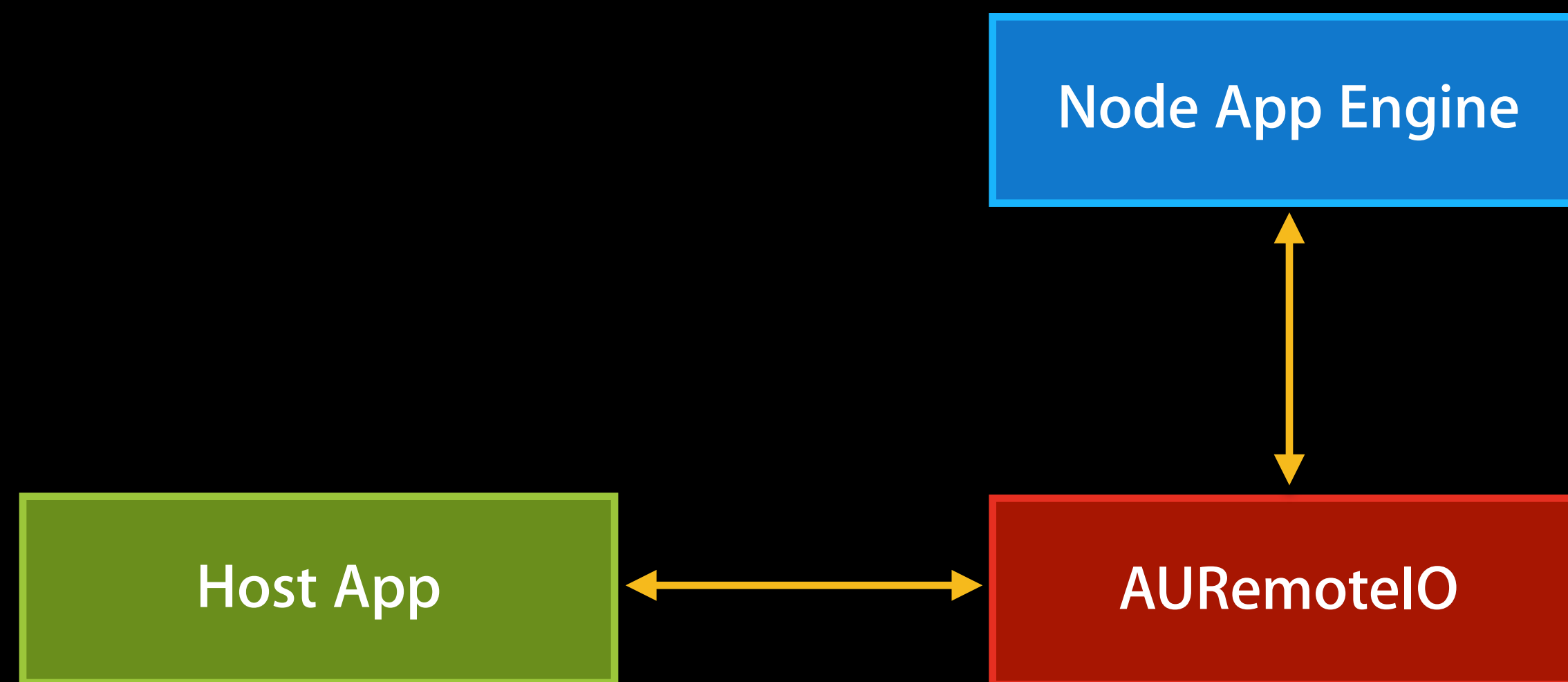
# Introduction

## Node app



# Introduction

## Node app





# Introduction

- Extensions to `AudioUnit.framework` API
- Host sees node as `AudioUnit`
- Node's I/O unit redirected to host

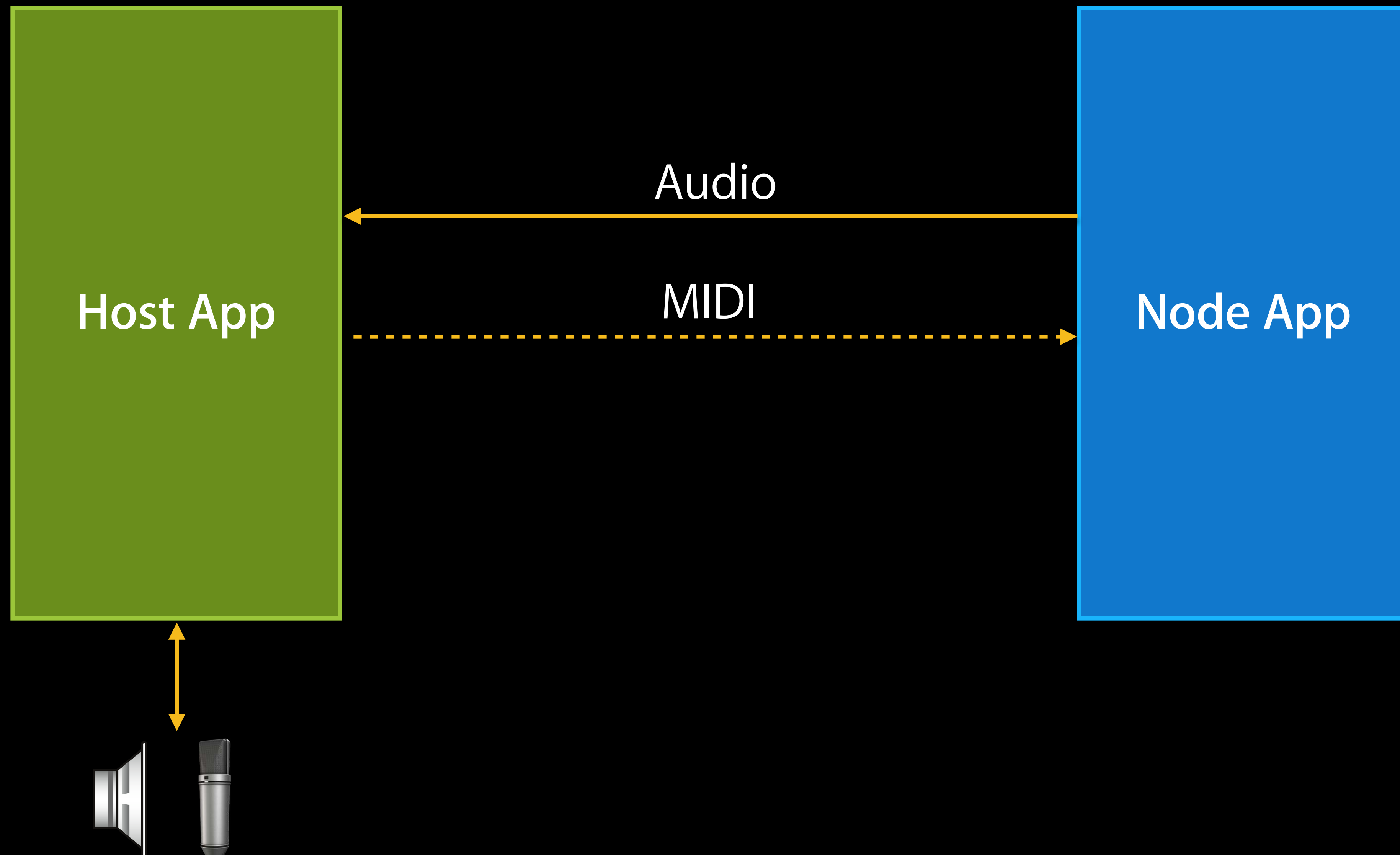
# Introduction

## New AudioUnit types

Audio Component Type	Input (from Host)	Output
kAudioUnitType_RemoteGenerator	-	audio
kAudioUnitType_RemoteInstrument	MIDI	audio
kAudioUnitType_RemoteEffect	audio	audio
kAudioUnitType_RemoteMusicEffect	audio and MIDI	audio

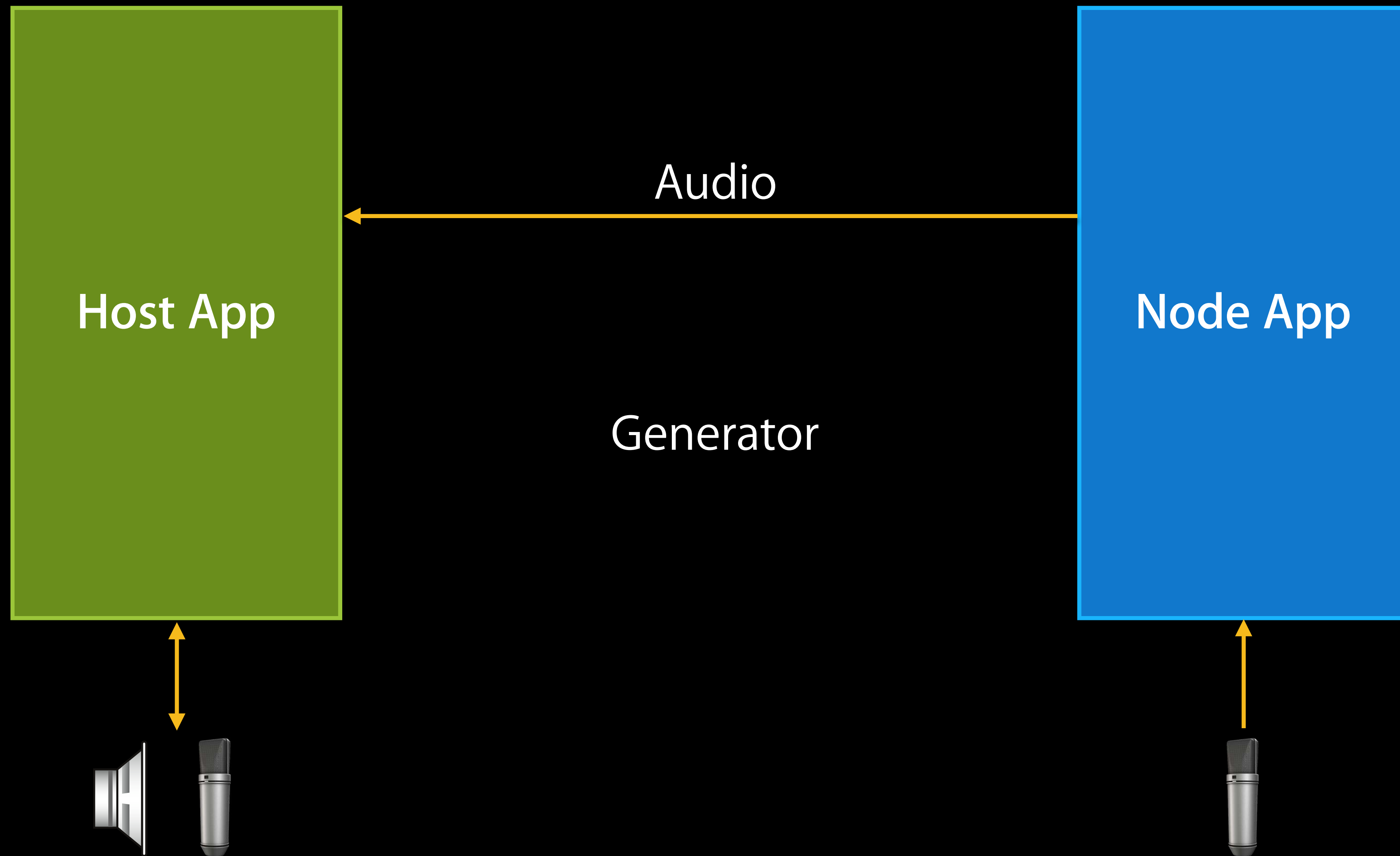
# One Node App, Multiple Components

Generator or instrument



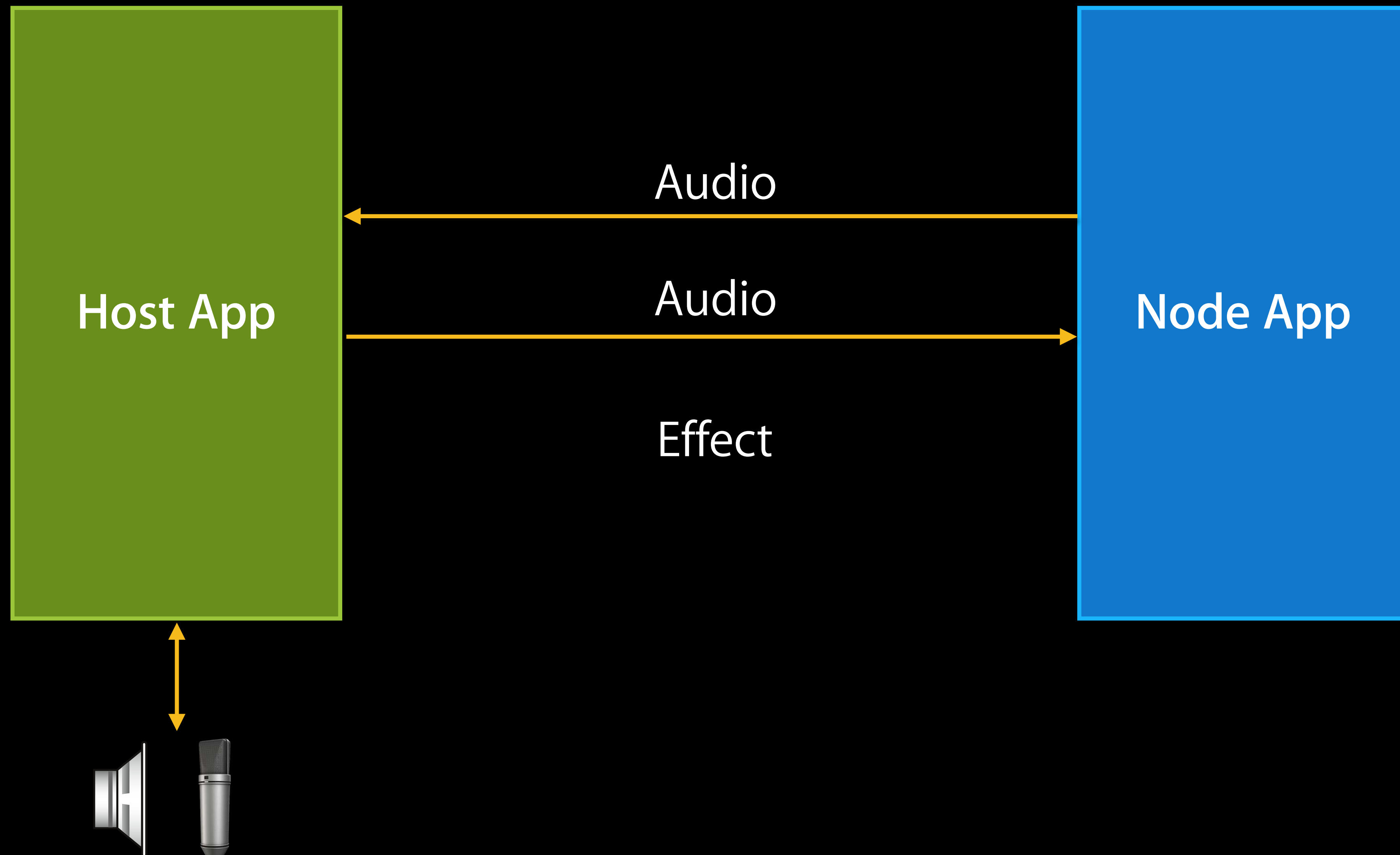
# One Node App, Multiple Components

Generator or effect



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Generator or effect

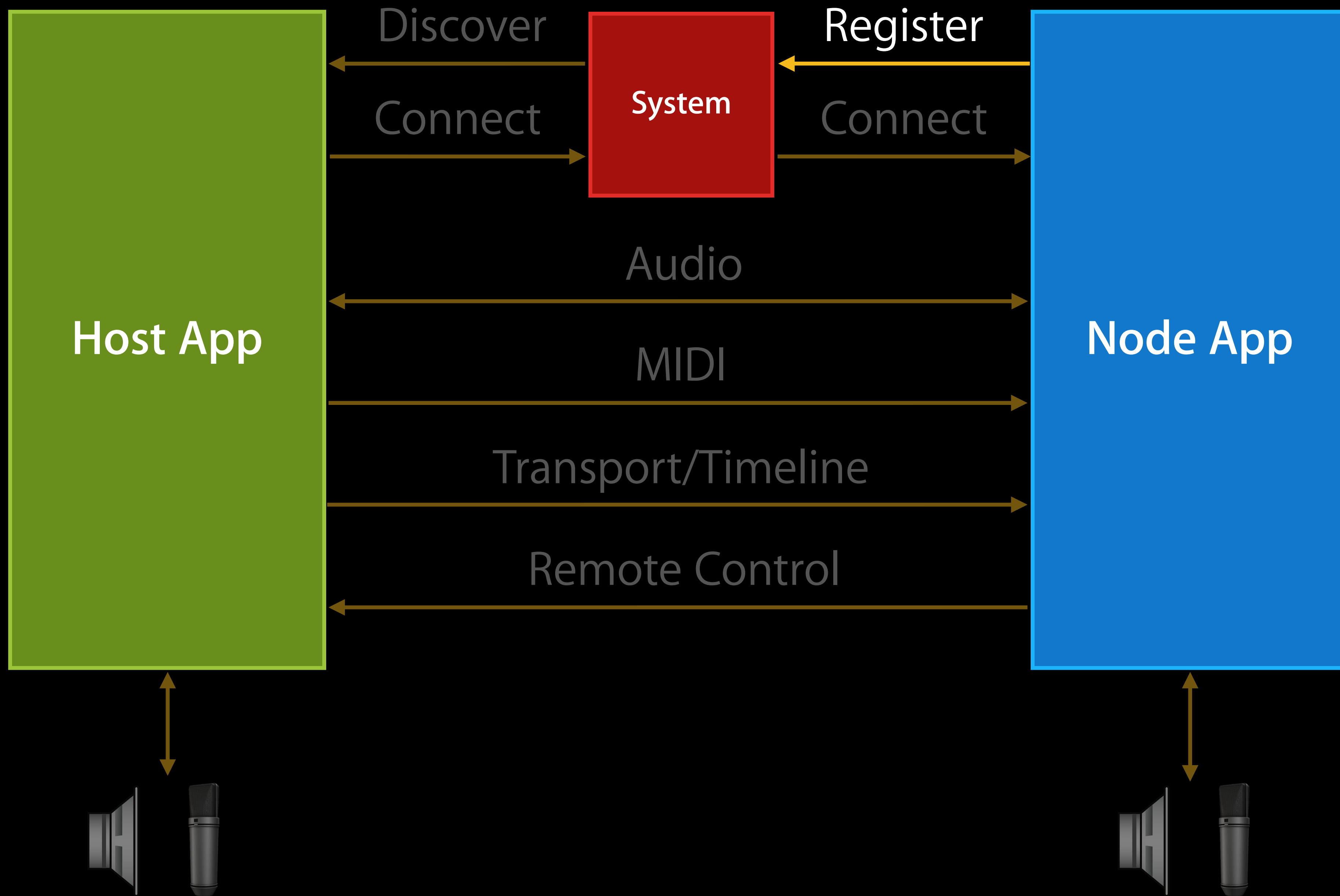


# Introduction

## Requirements

- Available on most iOS 7-compatible devices
  - (Except iPhone 4: API's fail quietly)
- "inter-app-audio" entitlement (all)
- "audio" in UIBackgroundModes (hosts, nodes that access mic)
- AVAudioSessionCategoryOptionMixWithOthers (nodes)

# Registering a Node App



# Registering a Node App

- Register via Info.plist "AudioComponents"
  - Makes app launchable
- AudioOutputUnitPublish
  - "Checks in" the registration



# Registering a Node App

## AudioComponents entry in Info.plist

```
<key>AudioComponents</key>
<array>
  <dict>
    <key>type</key>
    <string>aurg</string>
    <key>subtype</key>
    <string>ACgn</string>
    <key>manufacturer</key>
    <string>ACME</string>
    <key>name</key>
    <string>Acme: SineGenerator</string>
    <key>version</key>
    <integer>1</integer>
  </dict>
</array>
```

# Registering a Node App

## Create and publish the I/O unit

```
// create the AURemoteIO I/O unit

AudioComponentDescription ioUnitDesc = { kAudioUnitType_Output,
kAudioUnitSubType_RemoteIO, kAudioUnitManufacturer_Apple, 0, 0 };

AudioComponent comp = AudioComponentFindNext(NULL, &ioUnitDesc);

err = AudioComponentInstanceNew(comp, &_ioUnit);

// publish the AURemoteIO

AudioComponentDescription generatorDesc = { kAudioUnitType_RemoteGenerator,
'ACgn', 'ACME', 0, 0 };

CFStringRef name = CFSTR("Acme: SineGenerator");

const UInt32 version = 1;

err = AudioOutputUnitPublish(&generatorDesc, name, version, _ioUnit);
```

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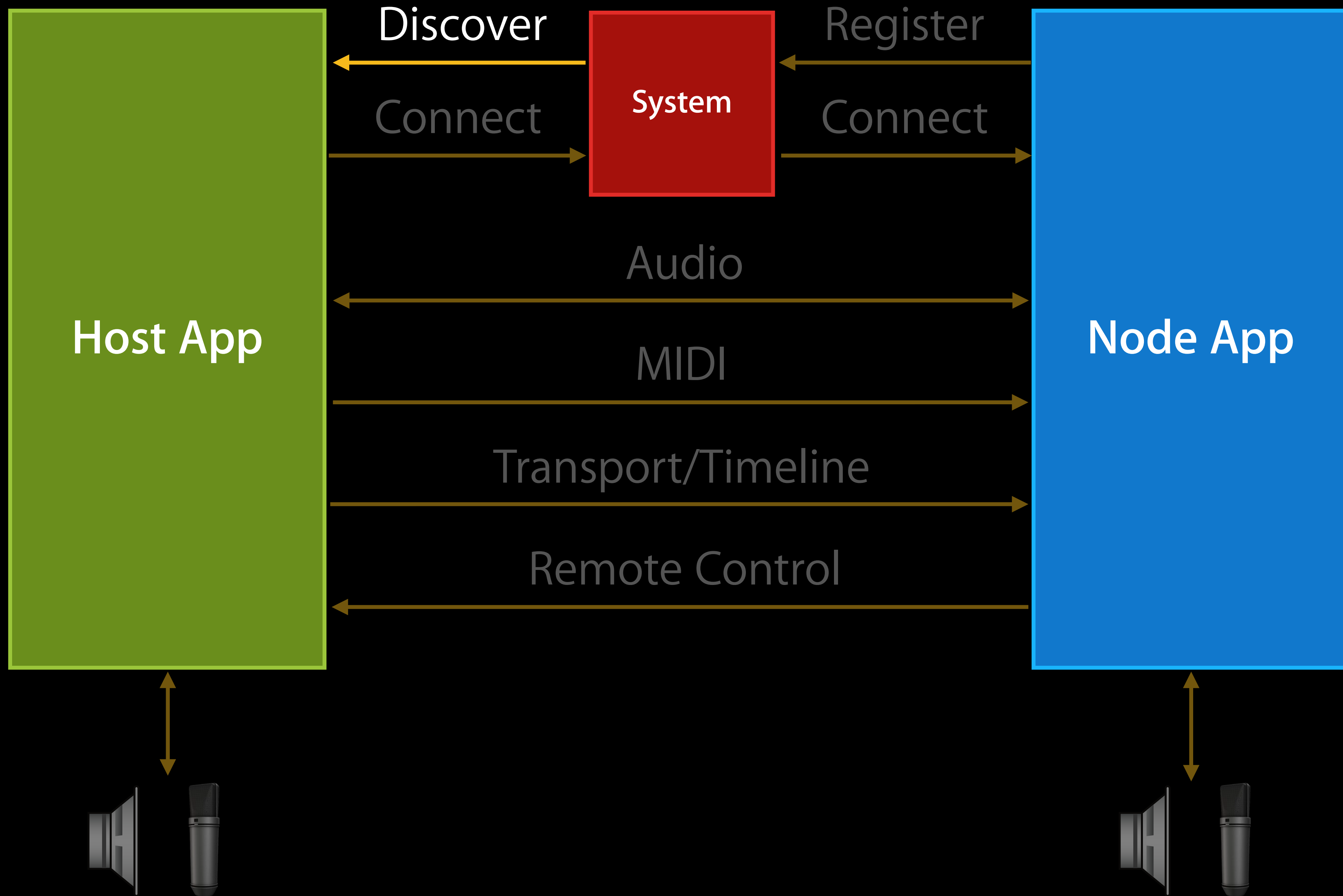
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```

# Registering a Node App

## Publishing the I/O unit

- Must publish on launch
- Component descriptions, names, and versions must match
- Component name should be “Manufacturer: App name”

# Node App Discovery (for Hosts)



# Node App Discovery

## For hosts

```
AudioComponentDescription searchDesc = { 0, 0, 0, 0, 0 }, foundDesc;
AudioComponent comp = NULL;

while (true) {
    comp = AudioComponentFindNext(comp, &searchDesc);
    if (comp == NULL) break;

    if (AudioComponentGetDescription(comp, &foundDesc) != noErr) continue;

    switch (foundDesc.componentType) {
        case kAudioUnitType_RemoteEffect:
        case kAudioUnitType_RemoteGenerator:
        case kAudioUnitType_RemoteInstrument:
        case kAudioUnitType_RemoteMusicEffect:
            // found a node
            ...
            break;
    }
}
```

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            // found a node  
            ...  
            break;
```

```
    }
```

```
}
```

# Node App Discovery

## Getting node information

```
RemoteAU *rau = [[RemoteAU alloc] init];
```

```
rau->_desc = foundDesc;
```

```
rau->_comp = comp;
```

```
AudioComponentCopyName(comp, (CFStringRef *)&rau->_name);
```

```
rau->_image = [AudioComponentGetIcon(comp, 48) retain];
```

```
rau->_lastActiveTime = AudioComponentGetLastActiveTime(comp);
```

```
[_audioUnits addObject: rau];
```

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# Node App Discovery

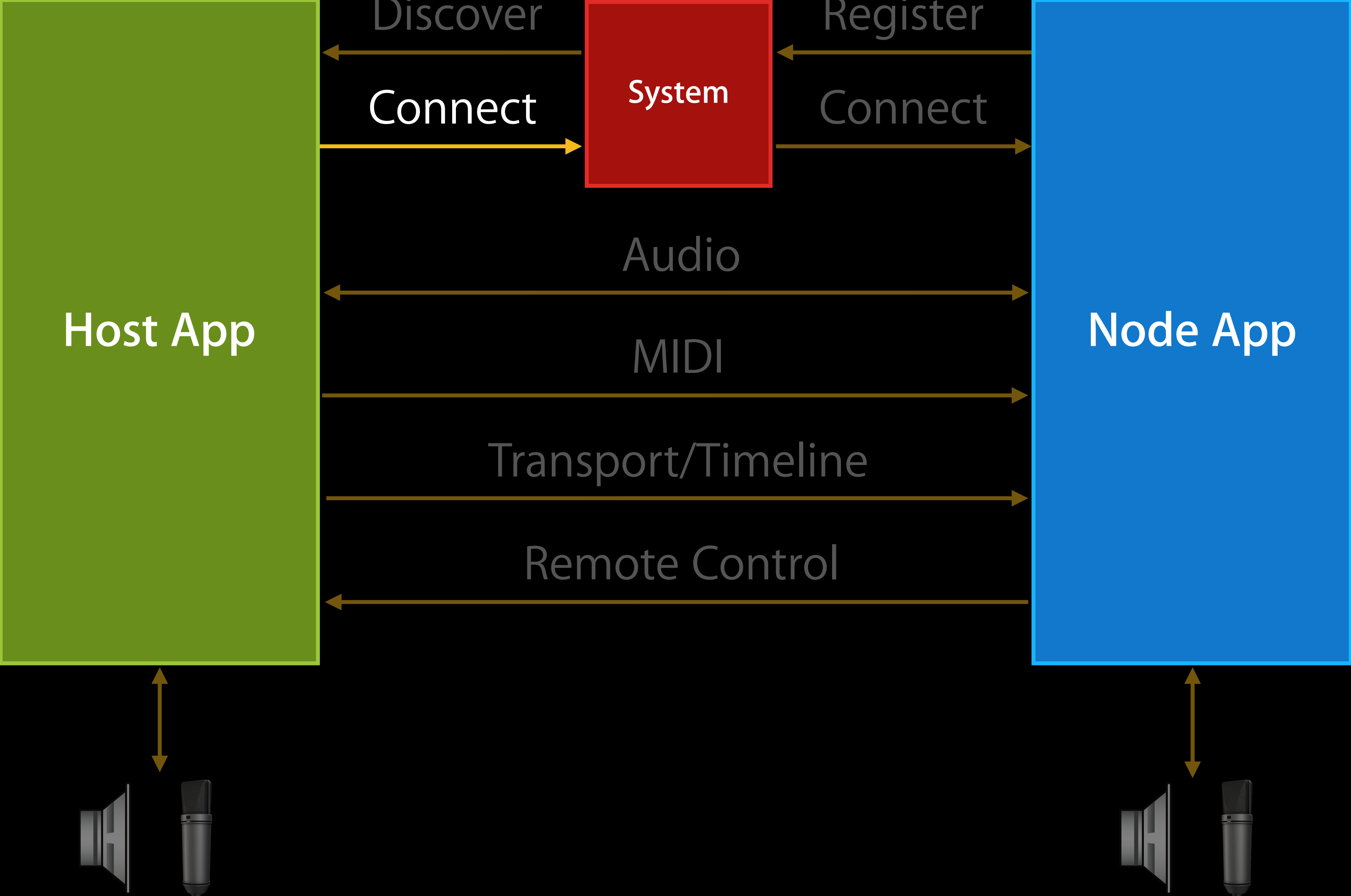
## Observing registration changes

- Registrations change dynamically
  - When apps installed/deleted
  - When media services reset

```
NSNotificationCenter *nc = [NSNotificationCenter defaultCenter];  
[nc addObserverForName:  
    (NSString *) kAudioComponentRegistrationsChangedNotification  
    object: nil queue: nil  
    usingBlock:  
        ^(NSNotification *) {  
            [self refreshAUList];  
        }  
];
```



# Connecting to a Node



# Connecting to a Node

```
AudioUnit myAudioUnit;  
err = AudioComponentInstanceNew(comp, &myAudioUnit);
```

- Node app will be launched into background

# Preparing a Node

## Activate audio session

```
[[AVAudioSession sharedInstance] setActive: YES];
```

- Sample rate, channel count

# Preparing a Node

## Set stream formats

```
AudioStreamBasicDescription format;
```

```
format.mChannelsPerFrame = 2; // stereo  
format.mSampleRate = [AVAudioSession sharedSession].sampleRate;  
format.mFormatID = kAudioFormatLinearPCM;  
format.mFormatFlags = kAudioFormatFlagsNativeFloatPacked |  
                      kAudioFormatFlagIsNonInterleaved;  
format.mBytesPerFrame = format.mBytesPerPacket = sizeof(Float32);  
format.mBitsPerChannel = 32;  
format.mFramesPerPacket = 1;
```

```
AudioUnitSetProperty(myAudioUnit, kAudioUnitProperty_StreamFormat,  
                    kAudioUnitScope_Output, 0, &format, sizeof(format));
```

```
AudioUnitSetProperty(myAudioUnit, kAudioUnitProperty_StreamFormat,  
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```

# Preparing a Node

## Connect input

- For effects
- From AudioUnit:
  - AUGraphConnectNodeInput
  - kAudioUnitProperty\_MakeConnection
- From callback:
  - kAudioUnitProperty\_SetRenderCallback
- Output

# Preparing a Node Disconnection

- Can happen automatically
  - Node terminates
  - Host fails to render
- Instance becomes “zombie”
- `kAudioComponentErr_InstanceInvalidated`

# Preparing a Node

## Disconnection callback

```
AudioUnitAddPropertyListener(myAudioUnit,  
    kAudioUnitProperty_IsInterAppConnected,  
    NodeConnectionListener, self);
```

```
void NodeConnectionListener(void *userData, AudioUnit myAudioUnit,  
    AudioUnitPropertyID, AudioUnitScope, AudioUnitElement)  
{  
    UInt32 connected = 0, size = sizeof(connected);  
    OSStatus err =  
        AudioUnitGetProperty(myAudioUnit, kAudioUnitProperty_IsInterAppConnected,  
            kAudioUnitScope_Global, 0,  
            &connected, &size);  
    if (err != noErr || connected == 0) {  
        // Node disconnected  
    }  
}
```

# Preparing a Node

## Initialization

```
// Prepare for rendering  
AudioUnitInitialize(myAudioUnit);
```

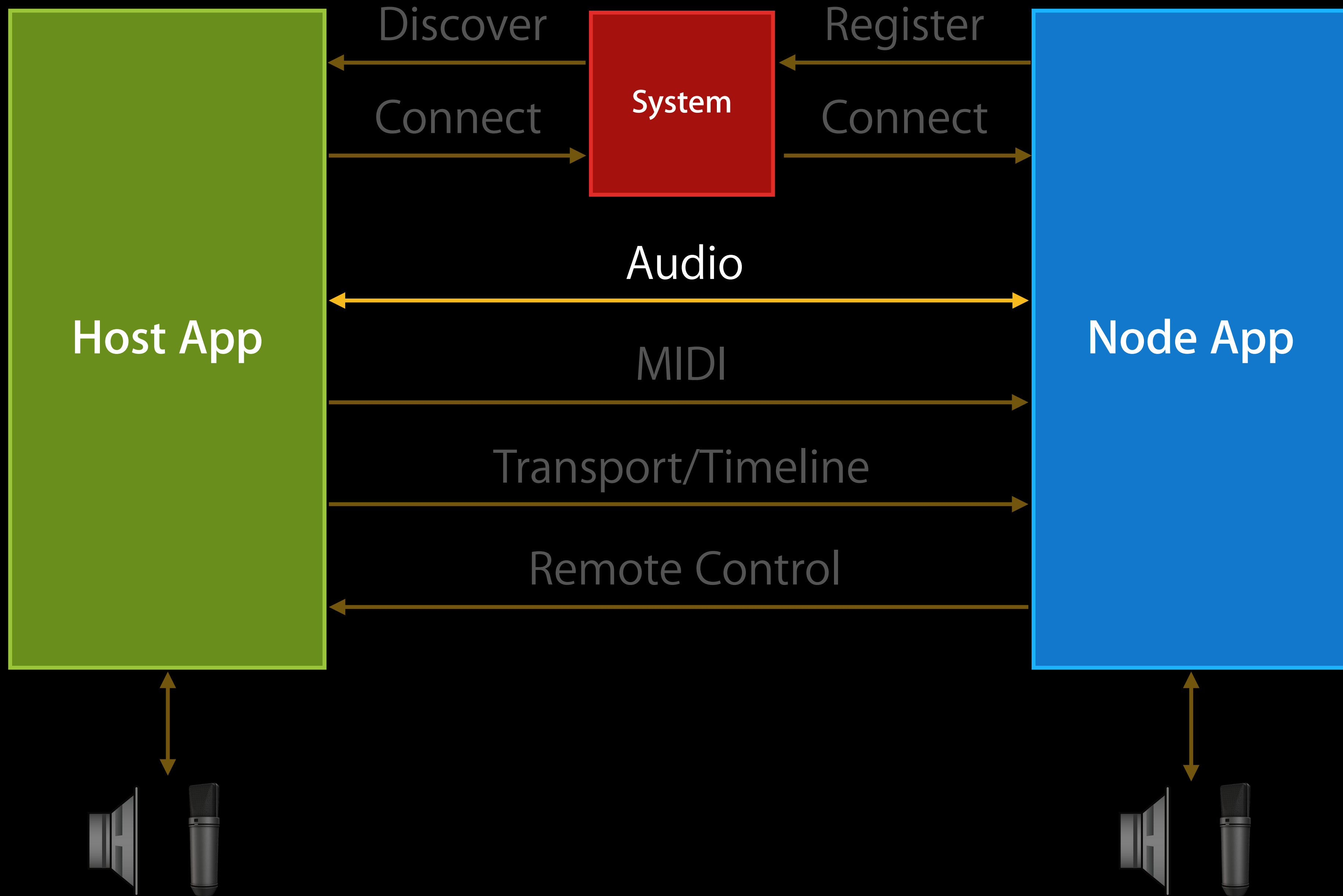
- Host must call `AudioUnitRender` regularly

# Preparing a Node

## Summary

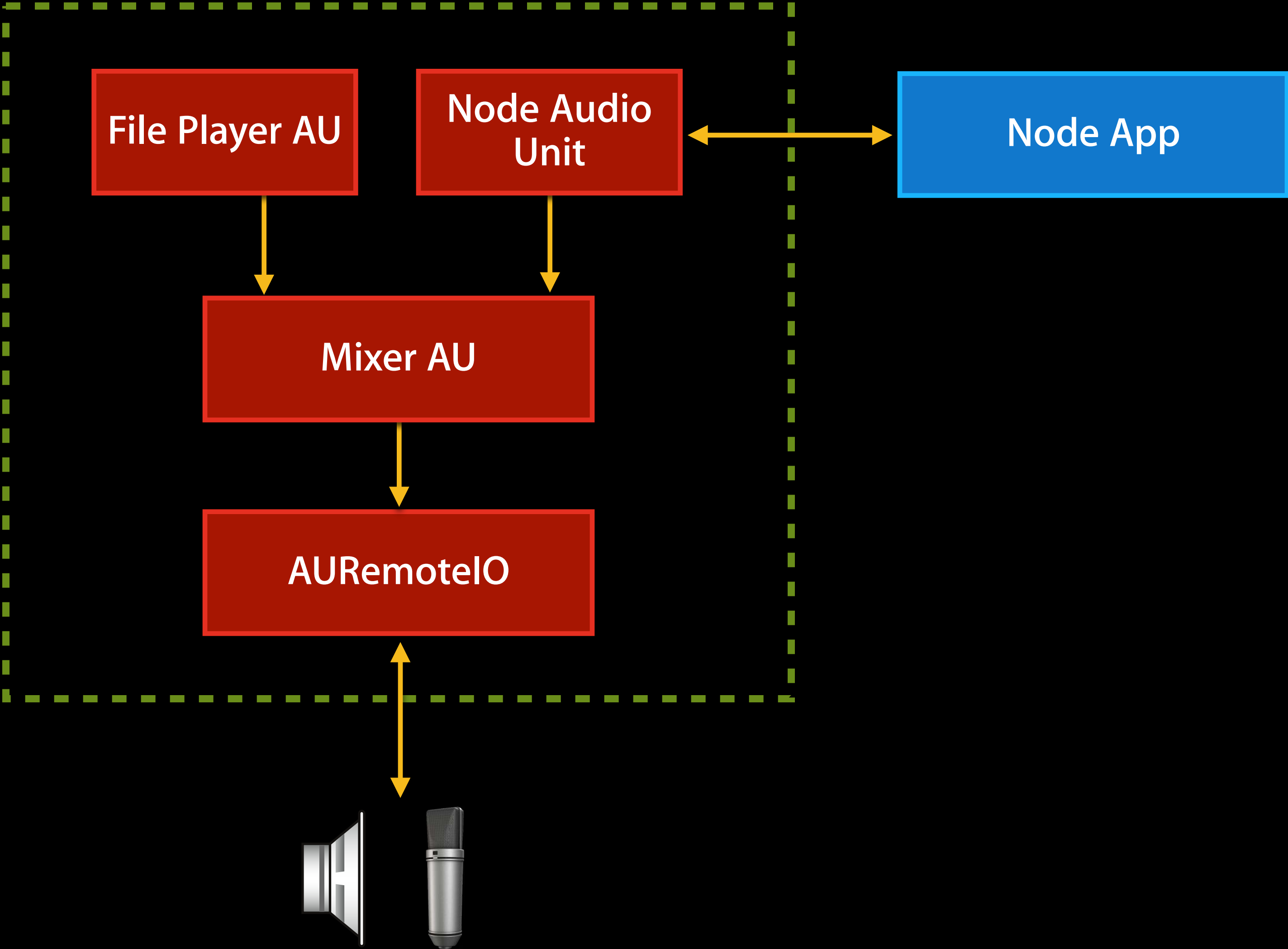
- Activate audio session
- Set stream formats
- Connect audio input
- Add disconnection listener
- Initialize

# Host Rendering Node Audio



# Host Rendering Node Audio

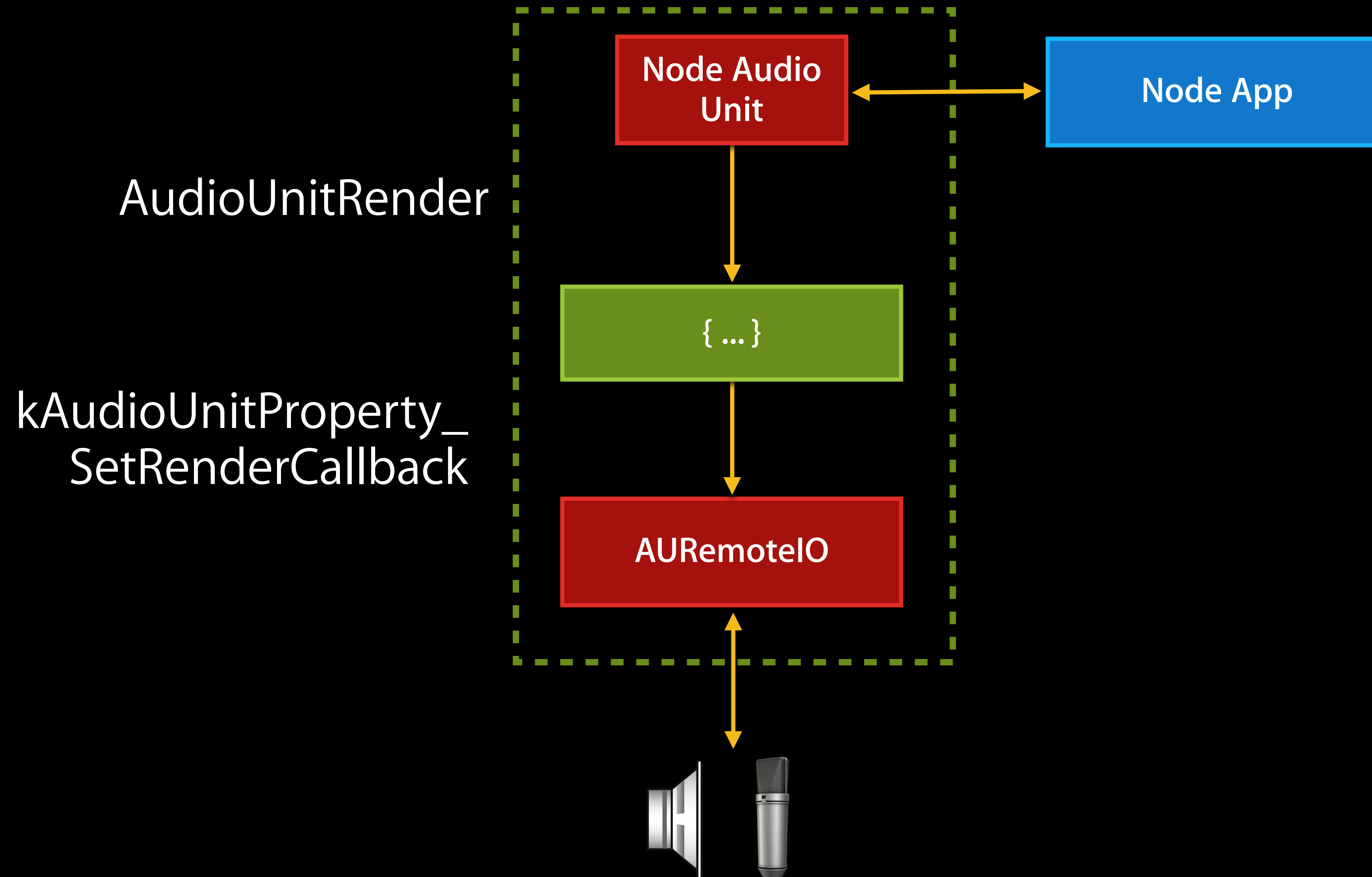
Using AUGraph





# Host Rendering Node Audio

Using AudioUnits directly



# Switching to a Node

- Do this after user opens the node
- Icon; tap to switch

```
void SwitchToNode(AudioUnit node)
{
    NSURL *url = NULL;
    UInt32 propertySize = sizeof(url);
    // property is CFURLRef but NSURL is toll-free-bridged.
    OSStatus err = AudioUnitGetProperty(peerAudioUnit,
    kAudioUnitProperty_PeerURL, kAudioUnitScope_Global, 0, &url,
    &propertySize);
    if (err) return;
    [[UIApplication sharedApplication] openUrl:url];
    [url release];
}
```

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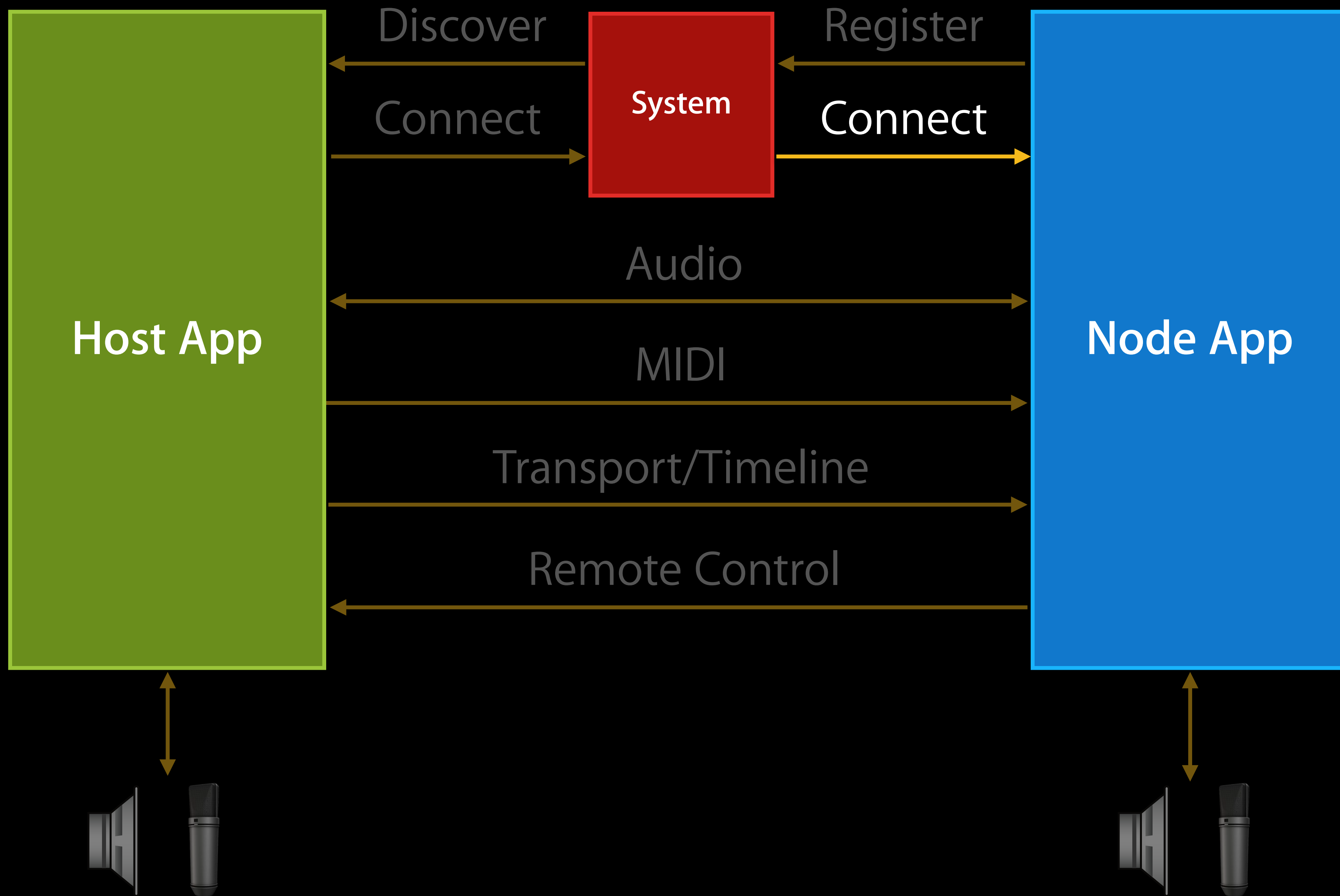
```
}
```

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}
```

# Being Connected (for Nodes)



# Being Connected (for Nodes)

## On launch

- Can be launched into the background
- Can't start running from background
- Must publish I/O unit

```
UIApplicationState appstate =  
    [UIApplication sharedApplication].applicationState;  
_inForeground = (appstate != UIApplicationStateBackground);
```

# Being Connected (for Nodes)

## Handling connection

- `kAudioUnitProperty_IsInterAppConnected`
- When value becomes 1:
  - Output unit has been initialized
  - Set audio session active (if accessing mic)
  - Start running
  - Even if in the background
- `AudioOutputUnitGetHostIcon`

# Being Connected (for Nodes)

## Handling disconnection

- When `kAudioUnitProperty_IsInterAppConnected` becomes 0:
  - Output unit has been uninitialized and stopped
  - Set audio session inactive (if using mic)
  - May set session active and resume running, but only if in the foreground

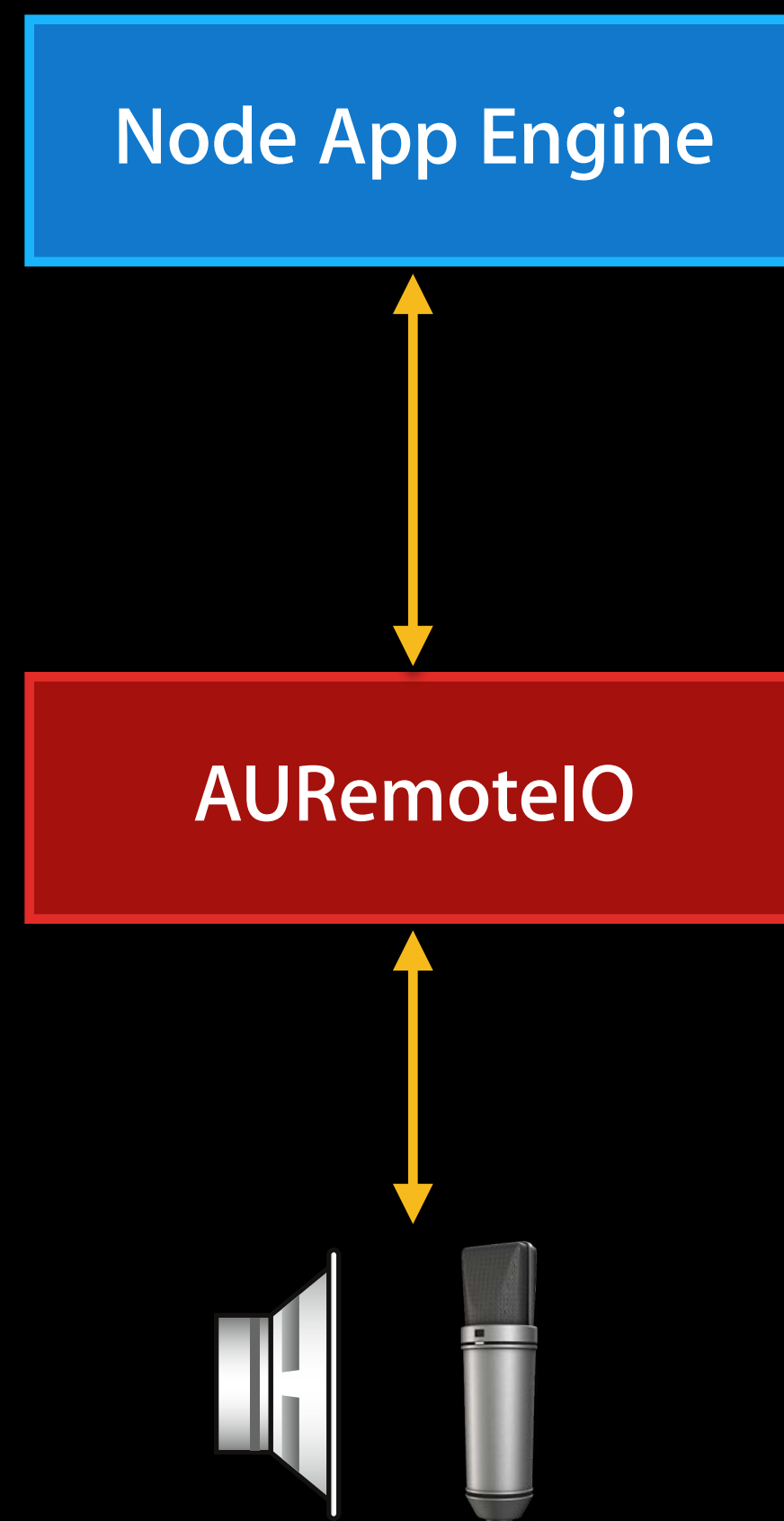


# Running State (for Nodes)

- CanStart = Connected or InForeground
- Running = Connected or AppSpecificConditions

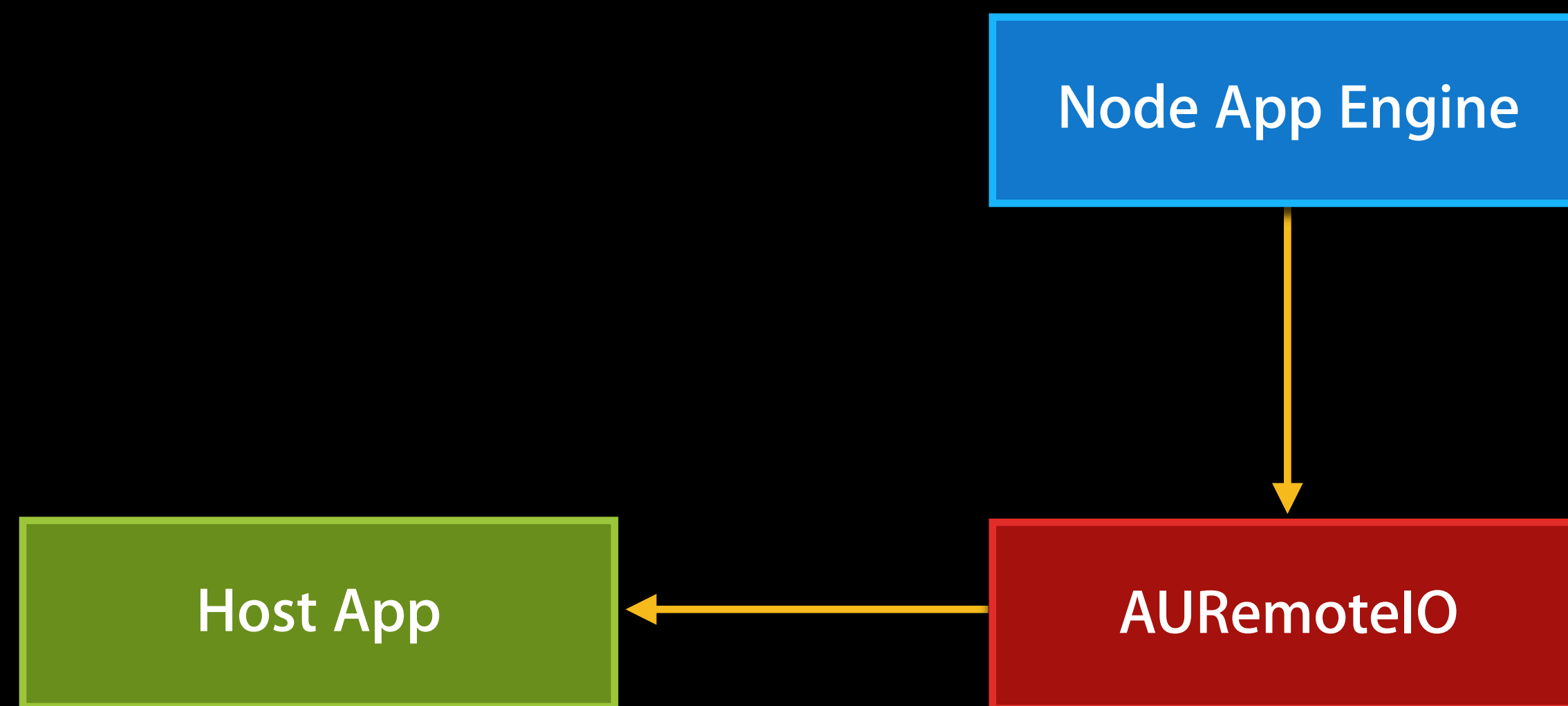
# Node Rendering

## Standalone



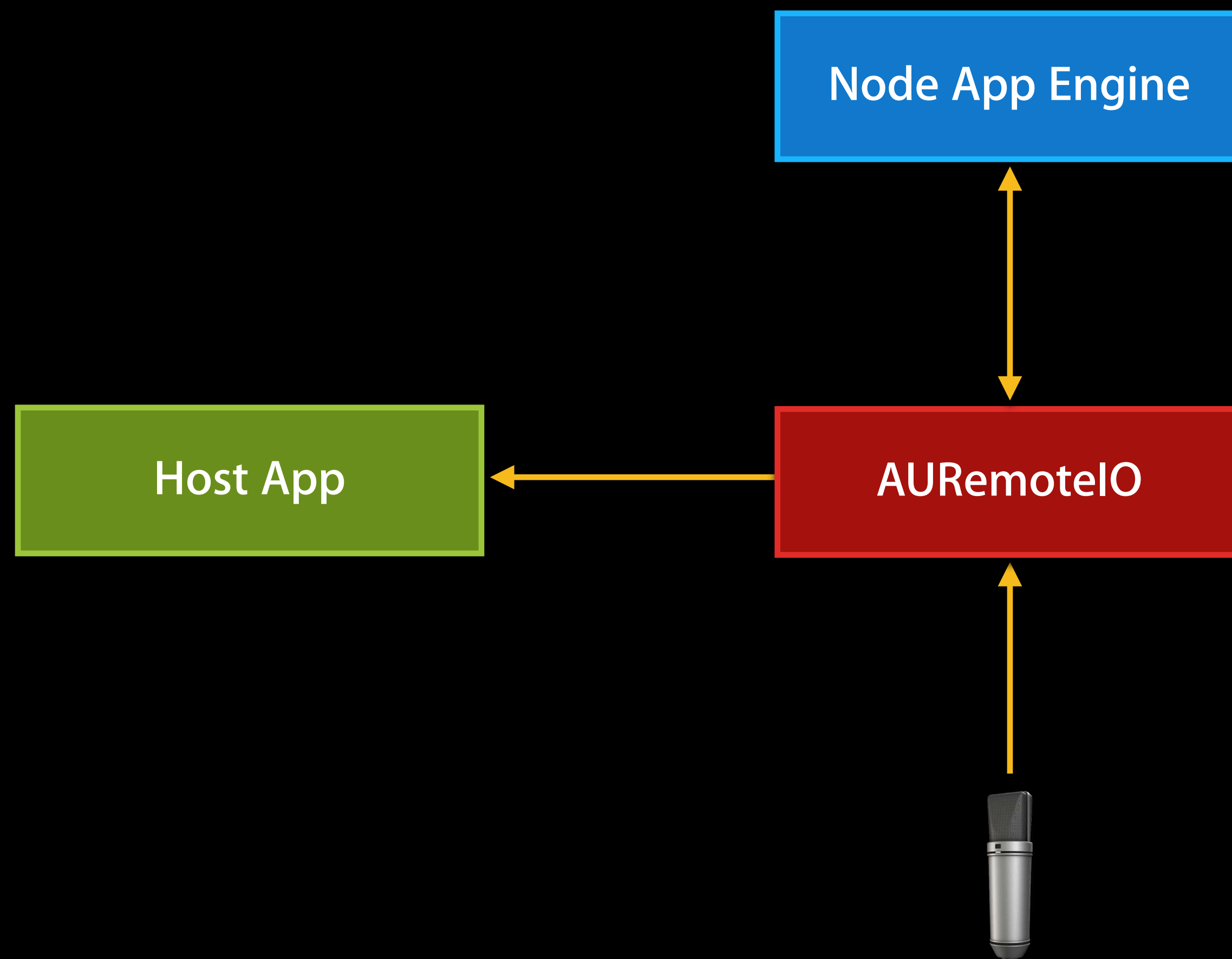
# Node Rendering

Generator or instrument



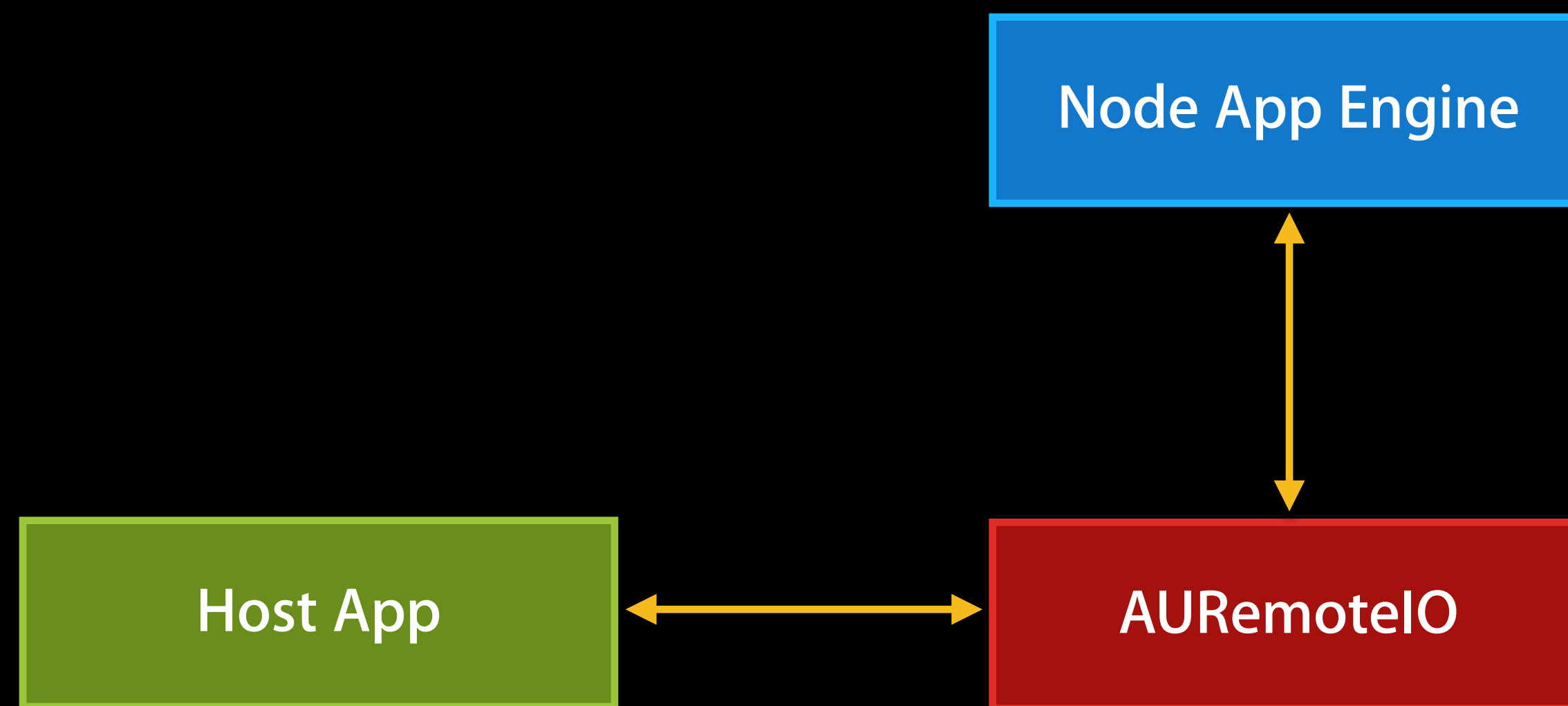
# Node Rendering

Generator or instrument with audio input



# Node Rendering

## Effect



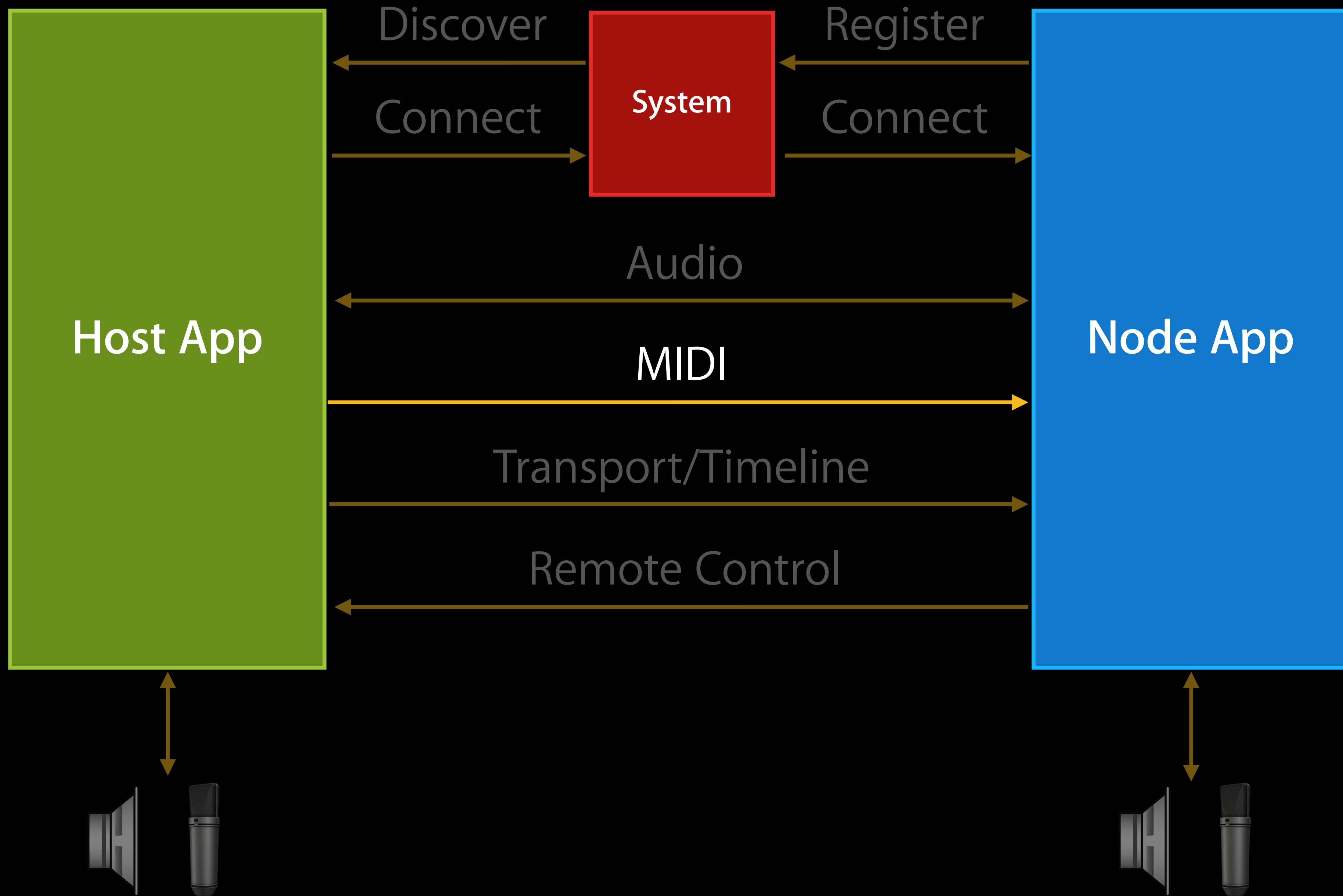
# Switching from Node to Host

- Same as with host: Use `kAudioUnitProperty_PeerURL`

# Stopping Audio Rendering (Host)

- `AudioOutputUnitStop/AUGraphStop`
- `AudioUnitUninitialize`
  - Release resources
  - Can re-initialize
- `AudioComponentInstanceDispose`
  - When completely finished
  - After node invalidated

# MIDI





# MIDI

- For RemoteInstrument and RemoteMusicEffect nodes
  - When MIDI events tied to audio
  - Sample-accurate scheduling
  - Not for sync (clock/timecode)
- Complements CoreMIDI
  - USB/network
  - Apps that don't support inter-app audio

# Sending MIDI Events (Host)

## Immediate, unscheduled

```
UInt32 offsetSampleFrames = 0;
```

```
const UInt8 kMIDINoteOn = 0x90;
```

```
const UInt8 kMiddleC = 60;
```

```
const UInt8 kVelocity = 64;
```

```
MusicDeviceMIDIEvent(myAudioUnit, kMIDINoteOn, kMiddleC, kVelocity,  
    offsetSampleFrames);
```

# Sending MIDI Events (Host)

## Scheduled

```
double offsetSeconds = ...;
UInt32 offsetSampleFrames = offsetSeconds * sampleRate;
const UInt8 kMIDIINoteOn = 0x90;
MusicDeviceMIDIEvent(myAudioUnit, kMIDIINoteOn, 60 /* middle C */, 64,
    offsetSampleFrames);
AudioUnitRender(myAudioUnit, ...);
```

# Sending MIDI Events (Host)

## Scheduled, with AUGraph

```
AUGraphAddRenderNotify(myGraph, MyRenderNotify, self);
```

```
...
```

```
OSStatus MyRenderNotify(...  
    const AudioTimeStamp *inTimeStamp
```

```
    ...
```

```
    UInt32 inNumberFrames ...)
```

```
{
```

```
    double offsetSeconds = ...;
```

```
    UInt32 offsetSampleFrames = offsetSeconds * sampleRate;
```

```
    const UInt8 kMIDINoteOn = 0x90;
```

```
    MusicDeviceMIDIEvent(myAudioUnit, kMIDINoteOn, 60 /* middle C */, 64,  
        offsetSampleFrames);
```

```
    ...
```

```
}
```

# Receiving MIDI Events (Node)

## MIDI Callbacks

```
typedef struct {
    void *userData;
    // see MusicDeviceMIDIEvent, MusicDeviceSysEx
    void (*MIDIEventProc)(void *userData, UInt32 inStatus,
                          UInt32 inData1, UInt32 inData2, UInt32 inOffsetSampleFrame);
    void (*MIDISysExProc)(void *userData, const UInt8* inData,
                          UInt32 inLength);
} AudioOutputUnitMIDICallbacks;
```

# Receiving MIDI Events (Node)

```
void MyMIDIEventProc(void *userData, UInt32 inStatus,
    UInt32 inData1, UInt32 inData2, UInt32 inOffsetSampleFrame)
{
    MyEngine *engine = (MyEngine *)userData;
    ...
}

void InstallMIDICallbacks(MyEngine *engine)
{
    AudioOutputUnitMIDICallbacks callbacks;
    callbacks.userData = engine;
    callbacks.MIDIEventProc = MyMIDIEventProc;
    callbacks.MIDISysExProc = NULL;
    AudioUnitSetProperty(myOutputUnit, kAudioOutputUnitProperty_MIDICallbacks,
        kAudioUnitScope_Global, 0,
        &callbacks, sizeof(callbacks));
}
```

# Receiving MIDI Events (Node)

```
void MyMIDIEventProc(void *userData, UInt32 inStatus,  
    UInt32 inData1, UInt32 inData2, UInt32 inOffsetSampleFrame)  
{  
    MyEngine *engine = (MyEngine *)userData;  
    ...  
}
```

```
void InstallMIDICallbacks(MyEngine *engine)  
{  
    AudioOutputUnitMIDICallbacks callbacks;  
    callbacks.userData = engine;  
    callbacks.MIDIEventProc = MyMIDIEventProc;  
    callbacks.MIDISysExProc = NULL;  
    AudioUnitSetProperty(myOutputUnit, kAudioOutputUnitProperty_MIDICallbacks,  
        kAudioUnitScope_Global, 0,  
        &callbacks, sizeof(callbacks));  
}
```

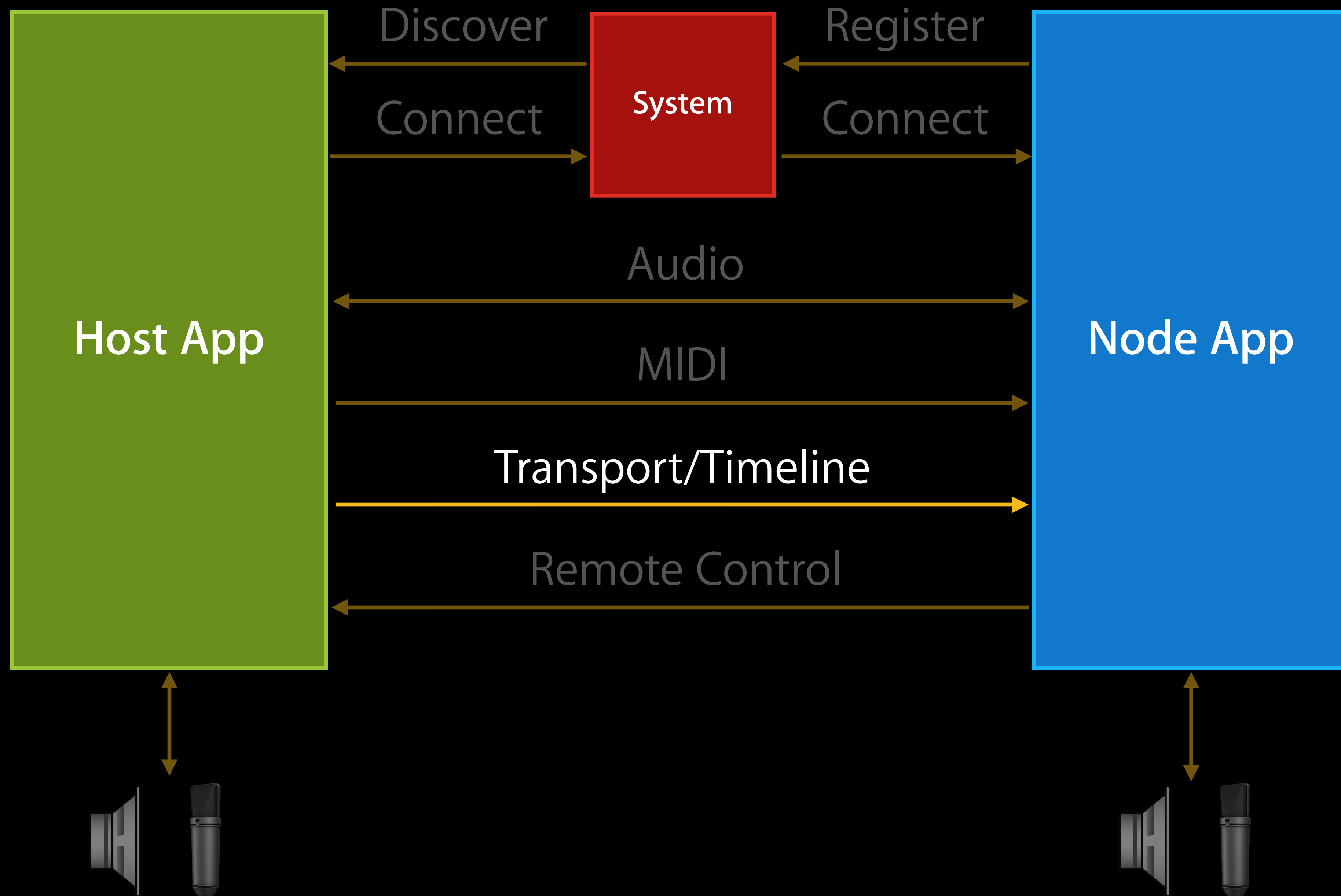
# Receiving MIDI Events (Node)

```
void MyMIDIEventProc(void *userData, UInt32 inStatus,  
    UInt32 inData1, UInt32 inData2, UInt32 inOffsetSampleFrame)  
{  
    MyEngine *engine = (MyEngine *)userData;  
    ...  
}
```

```
void InstallMIDICallbacks(MyEngine *engine)  
{  
    AudioOutputUnitMIDICallbacks callbacks;  
    callbacks.userData = engine;  
    callbacks.MIDIEventProc = MyMIDIEventProc;  
    callbacks.MIDISysExProc = NULL;  
    AudioUnitSetProperty(myOutputUnit, kAudioOutputUnitProperty_MIDICallbacks,  
        kAudioUnitScope_Global, 0,  
        &callbacks, sizeof(callbacks));  
}
```



# Transport and Timeline Information



# Transport and Timeline Information

- Host is master; node can synchronize
- Musical position
- Transport state
- Called at render time

# Transport and Timeline Information

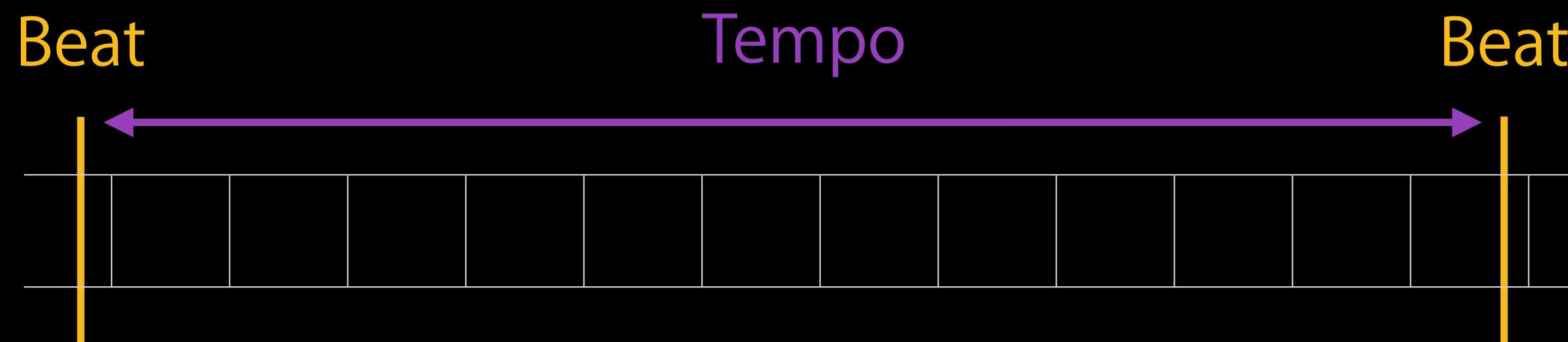
## HostCallbackInfo

```
typedef struct HostCallbackInfo {  
    void *                                hostUserData;  
    HostCallback_GetBeatAndTempo         beatAndTempoProc;  
    HostCallback_GetMusicalTimeLocation  musicalTimeLocationProc;  
    HostCallback_GetTransportState       transportStateProc;  
    HostCallback_GetTransportState2     transportStateProc2;  
} HostCallbackInfo;
```

# Providing Host Callbacks

## Beat and tempo

```
OSStatus MyBeatAndTempo(...)  
{  
    *outCurrentBeat = ...; // position in track (float)  
    *outCurrentTempo = ...; // beats per minute  
    return noErr;  
}
```



# Providing Host Callbacks

## Musical time location

```
OSStatus MyMusicalTimeLocation(...)
{
    *outDeltaSampleOffsetToNextBeat = ...;
    *outTimeSigNumerator = ...;
    *outTimeSigDenominator = ...;
    *outCurrentMeasureDownbeat = ...;
    return noErr;
}
```

# Providing Host Callbacks

## Transport state

```
OSStatus MyGetTransportState(...)
{
    *outIsPlaying = ...;
    *outIsRecording = ...;
    *outTransportStateChanged = ...;
    *outCurrentSampleInTimeline = ...;
    *outIsCycling = ...;
    *outCycleStartBeat = ...;
    *outCycleEndBeat = ...;
    return noErr;
}
```

# Installing Host Callbacks

```
void InstallHostCallbacks(MyEngine *engine, AudioUnit myAudioUnit)
{
    HostCallbackInfo hci;

    hci.hostUserData = engine;
    hci.beatAndTempoProc = MyBeatAndTempo;
    hci.musicalTimeLocationProc = MyMusicalTimeLocation;
    hci.transportStateProc = NULL;
    hci.transportStateProc2 = MyGetTransportState;

    AudioUnitSetProperty(myAudioUnit, kAudioUnitProperty_kHostCallbacks,
        kAudioUnitScope_Global, 0,
        &hci, sizeof(hci));
}
```

# Using Host Callbacks (Node)

- Get `kAudioUnitProperty_HostCallbacks` at connection time

```
HostCallbackInfo myHostCallbacks;  
UInt32 propertySize = sizeof(myHostCallbacks);  
AudioUnitGetProperty(myIOUnit, kAudioUnitProperty_kHostCallbacks,  
    kAudioUnitScope_Global, 0,  
    &myHostCallbacks, &propertySize);
```

- Call the desired host callback(s) at render time
- Caution: Thread-safety



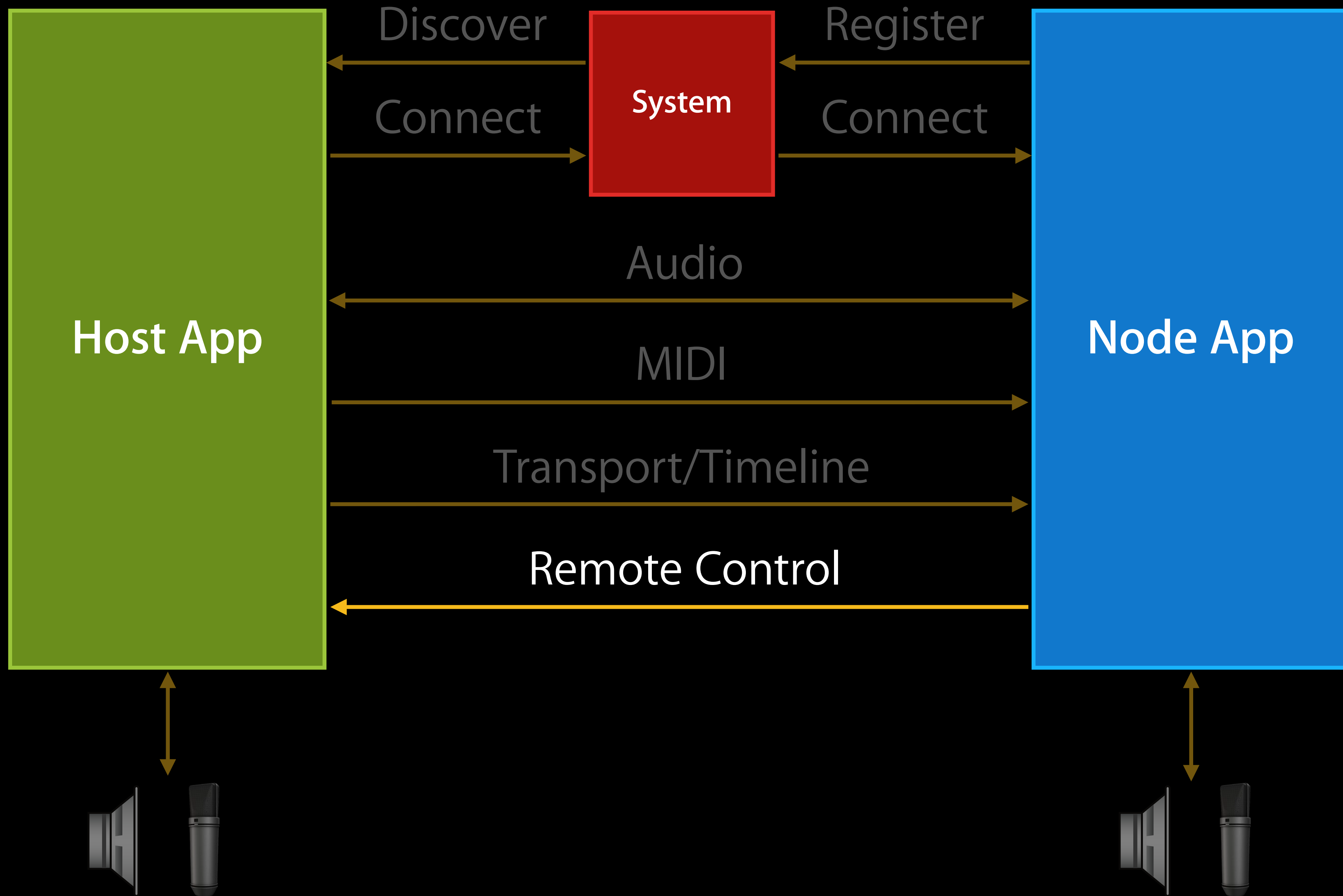
# Using Host Callbacks (Node)

## Observing transport state changes

- Transport state listener
- Called on non-render thread

```
AudioUnitAddPropertyListener(myIOUnit,  
    kAudioOutputUnitProperty_HostTransportState,  
    MyTransportStateListener, self);
```

# Remote Control



# Audio Unit Remote Control Events

- Distinct from UIKit remote control events
- Node can control host's transport

```
enum {  
    kAudioUnitRemoteControlEvent_TogglePlayPause = 1,  
    kAudioUnitRemoteControlEvent_ToggleRecord    = 2,  
    kAudioUnitRemoteControlEvent_Rewind          = 3  
};  
typedef UInt32 AudioUnitRemoteControlEvent;
```

- Node: Use the recommended transport controls in the sample app

# Audio Unit Remote Control Events

## From node

- Query whether host supports them:

```
UInt32 eventsSupported = 0, propertySize = sizeof(eventsSupported);
AudioUnitGetProperty(myIOUnit,
    kAudioOutputUnitProperty_HostReceivesRemoteControlEvents,
    kAudioUnitScope_Global, 0,
    &eventsSupported, &propertySize);
```

- Send an event:

```
AudioUnitRemoteControlEvent theControl =
    kAudioUnitRemoteControlEvent_ToggleRecord;
AudioUnitSetProperty(myIOUnit,
    kAudioOutputUnitProperty_RemoteControlToHost,
    kAudioUnitScope_Global, 0,
    &theControl, sizeof(theControl));
```

# Audio Unit Remote Control Events

## In host

- Set kAudioUnitProperty\_RemoteControlEventListener

```
AudioUnitRemoteControlEventListener listenerBlock =  
    ^(AudioUnitRemoteControlEvent event) {  
        [self handleRemoteEvent: event];  
    };  
AudioUnitSetProperty(myAudioUnit,  
    kAudioOutputUnitProperty_RemoteControlEventListener,  
    kAudioUnitScope_Global, 0,  
    &listenerBlock, sizeof(listenerBlock));
```

# *Demo*

Inter-app host, instrument and effect, with remote control

Harry Tormey

# Audio Session Interruptions

- The usual rules apply
  - Your remote I/O has already been stopped underneath you
- In hosts, system has uninitialized your node audio units

# Handling Media Services Reset

- AVAudioSessionMediaServicesWereResetNotification
- All inter-app audio connections are broken (component instances invalidated)
- Host: Dispose node AudioUnit and AURemotelO
- Node: Dispose AURemotelO
- Proceed as if app has been launched



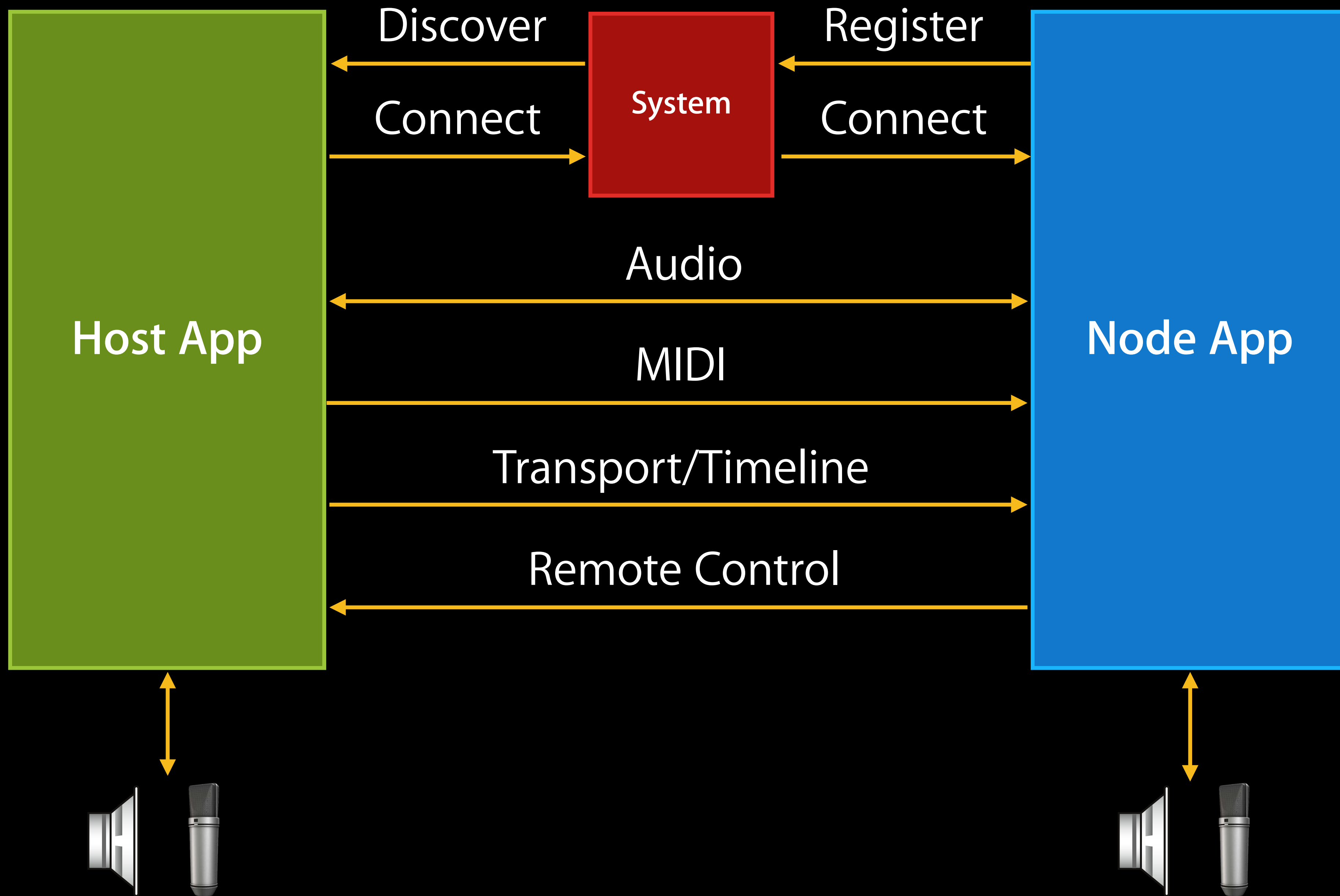
# Odds and Ends

- Multiple hosts?
  - If all mixable, yes
- Multiple nodes?
  - Yes

# Tips

- Debug node registration
  - Watch console log
- Error -12985
  - “Operation denied”
  - Can't start playing from the background

# Review



# Conclusion

- Existing apps can be converted to nodes fairly simply
- Creating a host is a bit more work, but use AudioUnit APIs for power and flexibility
- Use sample code from WWDC library
- Make great music apps!

# Apple Evangelists

## Contact information

### John Geleynse

Director, Technology Evangelist  
[geleynse@apple.com](mailto:geleynse@apple.com)

### Documentation

AV Foundation Programming Guide

<http://developer.apple.com/library/ios/#documentation/AudioVideo/Conceptual/AVFoundationPG/>

### Apple Developer Forums

<http://devforums.apple.com>

# Related Sessions

Moving to AV Kit and AV Foundation	Pacific Heights Tuesday 4:30PM	
Preparing and Presenting Media for Accessibility	Nob Hill Wednesday 10:15AM	
What's New in Camera Capture	Nob Hill Wednesday 11:30AM	
Advanced Editing with AV Foundation	Marina Thursday 9:00AM	

# Labs

Audio Lab	Media Lab B Tuesday 2:00PM	
Audio Lab	Media Lab B Wednesday 9:00AM	

 WWDC2013