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Record Audio in .NET Core

Asked 4 years, 7 months ago Modified 3 months ago Viewed 8k times



15



As the title suggests, I'm looking to capture audio from a microphone in .NET Core. It's important for me that it's cross platform. I'm doing a lot of video processing in OpenCV on Windows, Linux and OSX already. Audio is a missing piece of the puzzle for me.

[.net](#)[audio](#)[.net-core](#)[cross-platform](#)

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asked Sep 21, 2018 at 13:24

[Wojtek Turowicz](#)

4,086 ● 3 ● 27 ● 38

OpenCV is a library that eventually calls OS-specific APIs. To use it from .NET you'd have to use interop or a managed wrapper. You should probably look for managed wrappers for the audio library you want to use. – [Panagiotis Kanavos](#) Sep 25, 2018 at 8:58

@PanagiotisKanavos I did find OpenTK.NETCore that does cross platform audio. – [Wojtek Turowicz](#) Jan 17, 2019 at 15:40

OpenTK.NetCore has been deprecated in favour of OpenTK.NetStandard – [bre_dev](#) Sep 20, 2020 at 13:44

4 Answers

Sorted by:

Highest score (default)



OpenTK.NETCore is a really good cross platform port for OpenTK.

4

Using it's `AudioCapture` class I have been able to capture microphone data across different OS platforms.



```
using (var audioCapture = new AudioCapture(_recorders.First(), _sampling_rate,
ALFormat.Mono16, buffer_length_samples))
{
    audioCapture.Start();

    int available_samples = audioCapture.AvailableSamples;

    _buffer = _pool.Rent(MathHelper.NextPowerOfTwo((int)((available_samples *
_sampleToByte / (double)BlittableValueType.StrideOf(_buffer)) + 0.5)));

    if (available_samples > 0)
    {
        audioCapture.ReadSamples(_buffer, available_samples);
    }
    else
    {
        _pool.Return(_buffer);
    }
}
```

Edit: Based on `Parrot` project in OpenTK samples.

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answered Oct 2, 2018 at 13:13

[Wojtek Turowicz](#)

4,086 ● 3 ● 27 ● 38

please send Parrot project in OpenTK samples url! – [Ali Rasouli](#) Aug 2, 2020 at 8:34

I think he is referring to these examples: github.com/mono/opentk/tree/master/Source/Examples/OpenAL/1.1 – Kappacake Aug 28, 2020 at 10:08

After spending some time reading and running the Parrot sample, I realized that it just plays back the audio as it captures it. How would you save that audio to a file? Have you implemented this functionality? – Kappacake Aug 28, 2020 at 10:44

How to save the captured audio in a wav file? Any suggestion about how to save it? – bre_dev Sep 17, 2020 at 20:23

To save it as WAV, check out this answer: stackoverflow.com/a/63962284/1204153 – Andy Sep 18, 2020 at 20:22

I don't know an API or library that does that for you, unfortunately.

If others also don't know, then you have two options:

1. Write a cross-platform api yourself (take a look at the CoreFx or CoreFxLab repos), on Windows you might use P/Invoke to use system dlls, on the other platforms you'll need something else
2. Take a cross-platform recorder executable and interface with that from your .net core functionality

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answered Sep 25, 2018 at 6:35



[hansmbakker](#)

1,089 ● 14 ● 29

1 Thanks I've already managed with OpenTK.NETCore library and it's OpenAL support. – [Wojtek Turowicz](#) Oct 2, 2018 at 13:08

I've written a recorder app before on .NET Core, but its not my property anymore... However, here's [the library](#) that I used while working on it.

LibsoundIO Sharp is able to extend configurations to things such as the sampling rate of the Input or output and has sweet examples like that:

```
class Record
{
    static SoundIORingBuffer ring_buffer = null;

    static SoundIOFormat [] prioritized_formats = {
        SoundIODevice.Float32NE,
        SoundIODevice.Float32FE,
        SoundIODevice.S32NE,
        SoundIODevice.S32FE,
        SoundIODevice.S24NE,
        SoundIODevice.S24FE,
        SoundIODevice.S16NE,
        SoundIODevice.S16FE,
        SoundIODevice.Float64NE,
        SoundIODevice.Float64FE,
        SoundIODevice.U32NE,
        SoundIODevice.U32FE,
        SoundIODevice.U24NE,
        SoundIODevice.U24FE,
        SoundIODevice.U16NE,
        SoundIODevice.U16FE,
        SoundIOFormat.S8,
        SoundIOFormat.U8,
        SoundIOFormat.Invalid,
    };

    static readonly int [] prioritized_sample_rates = {
        48000,
        44100,
        96000,
        24000,
        0,
    };

    public static int Main (string [] args)
```

```

{
    string device_id = null;
    string backend_name = null;
    bool raw = false;
    string outfile = null;

    foreach (var arg in args) {
        switch (arg) {
            case "--raw":
                raw = true;
                continue;
            default:
                if (arg.StartsWith ("--backend:"))
                    backend_name = arg.Substring (arg.IndexOf (':') + 1);
                else if (arg.StartsWith ("--device:"))
                    device_id = arg.Substring (arg.IndexOf (':') + 1);
                else
                    outfile = arg;
                continue;
        }
    }

    var api = new SoundIO ();

    var backend = backend_name == null ? SoundIOBackend.None :
    (SoundIOBackend)Enum.Parse (typeof (SoundIOBackend), backend_name);
    if (backend == SoundIOBackend.None)
        api.Connect ();
    else
        api.ConnectBackend (backend);
    Console.WriteLine ("backend: " + api.CurrentBackend);

    api.FlushEvents ();

    var device = device_id == null ? api.GetInputDevice
    (api.DefaultInputDeviceIndex) :
        Enumerable.Range (0, api.InputDeviceCount)
            .Select (i => api.GetInputDevice (i))
            .FirstOrDefault (d => d.Id == device_id && d.IsRaw == raw);
    if (device == null) {
        Console.Error.WriteLine ("device " + device_id + " not found.");
        return 1;
    }
    Console.WriteLine ("device: " + device.Name);
    if (device.ProbeError != 0) {
        Console.Error.WriteLine ("Cannot probe device " + device_id + ".");
        return 1;
    }

    var sample_rate = prioritized_sample_rates.First (sr =>
    device.SupportsSampleRate (sr));

    var fmt = prioritized_formats.First (f => device.SupportsFormat (f));

    var instream = device.CreateInStream ();
    instream.Format = fmt;
    instream.SampleRate = sample_rate;
    instream.ReadCallback = (fmin, fmax) => read_callback (instream, fmin,
    fmax);

    instream.OverflowCallback = () => overflow_callback (instream);

    instream.Open ();

    const int ring_buffer_duration_seconds = 30;
    int capacity = (int)(ring_buffer_duration_seconds * instream.SampleRate *
    instream.BytesPerFrame);
    ring_buffer = api.CreateRingBuffer (capacity);
    var buf = ring_buffer.WritePointer;

    instream.Start ();

    Console.WriteLine ("Type CTRL+C to quit by killing process...");
    using (var fs = File.OpenWrite (outfile)) {
        var arr = new byte [capacity];
        unsafe {
            fixed (void* arrptr = arr) {
                for (; ) {
                    api.FlushEvents ();

```

```

        Thread.Sleep (1000);
        int fill_bytes = ring_buffer.FillCount;
        var read_buf = ring_buffer.ReadPointer;

        Buffer.MemoryCopy ((void*)read_buf, arrptr, fill_bytes,
fill_bytes);

        fs.Write (arr, 0, fill_bytes);
        ring_buffer.AdvanceReadPointer (fill_bytes);
    }
}
}
}
instream.Dispose ();
device.RemoveReference ();
api.Dispose ();
return 0;
}

static void read_callback (SoundIOInStream instream, int frame_count_min, int
frame_count_max)
{
    var write_ptr = ring_buffer.WritePointer;
    int free_bytes = ring_buffer.FreeCount;
    int free_count = free_bytes / instream.BytesPerFrame;

    if (frame_count_min > free_count)
        throw new InvalidOperationException ("ring buffer overflow"); //
panic()

    int write_frames = Math.Min (free_count, frame_count_max);
    int frames_left = write_frames;

    for (; ; ) {
        int frame_count = frames_left;

        var areas = instream.BeginRead (ref frame_count);

        if (frame_count == 0)
            break;

        if (areas.IsEmpty) {
            // Due to an overflow there is a hole. Fill the ring buffer with
            // silence for the size of the hole.
            for (int i = 0; i < frame_count * instream.BytesPerFrame; i++)
                Marshal.WriteByte (write_ptr + i, 0);
            Console.Error.WriteLine ("Dropped {0} frames due to internal
overflow", frame_count);
        } else {
            for (int frame = 0; frame < frame_count; frame += 1) {
                int chCount = instream.Layout.ChannelCount;
                int copySize = instream.BytesPerSample;
                unsafe {
                    for (int ch = 0; ch < chCount; ch += 1) {
                        var area = areas.GetArea (ch);
                        Buffer.MemoryCopy ((void*)area.Pointer,
(void*)write_ptr, copySize, copySize);
                        area.Pointer += area.Step;
                        write_ptr += copySize;
                    }
                }
            }

            instream.EndRead ();

            frames_left -= frame_count;
            if (frames_left <= 0)
                break;
        }

        int advance_bytes = write_frames * instream.BytesPerFrame;
        ring_buffer.AdvanceWritePointer (advance_bytes);
    }

    static int overflow_callback_count = 0;
    static void overflow_callback (SoundIOInStream instream)
    {
        Console.Error.WriteLine ("overflow {0}", overflow_callback_count++);
    }
}

```

```
}  
}
```

[Adapted from sio-record/Program.cs](#)

As far as .NET Core is concerned, you can leave it in a singleton class as a standalone project or if you're working on a project that has a DI container, simply toss it in as a singleton.

That being said, you need [libsoundio](#) to make things work because afterall, its like a "java" for audio and libsoundio-sharp is the wrapper for C#. Cheers!

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edited Sep 26, 2018 at 10:15

answered Sep 26, 2018 at 2:58



Nicholas

1,873 ● 21 ● 39

Thanks I've already managed with OpenTK.NETCore library and it's OpenAL support. – [Wojtek Turowicz](#) Oct 2, 2018 at 13:08

I wrote a simple program that prints the current volume using the OpenTK.OpenAL nuGet-package.

```
using OpenTK.Audio.OpenAL;  
using System;  
  
Console.WriteLine("Default Mic:\n"+ ALC.GetString(ALDevice.Null,  
ALCGetString.CaptureDefaultDeviceSpecifier));  
Console.WriteLine("Mic List:\n"+ string.Join("\n", ALC.GetString(ALDevice.Null,  
ALCGetStringList.CaptureDeviceSpecifier)));  
  
int bufferLength = 10 * 16000;//10 sec  
ALCaptureDevice mic= ALC.CaptureOpenDevice(null,16000,ALFormat.Mono8,  
bufferLength);//opens default mic //null specifies default  
Console.WriteLine("Using:");  
Console.WriteLine(ALC.GetString(new ALDevice(mic.Handle),  
ALCGetString.DeviceSpecifier));  
  
ALC.CaptureStart(mic);  
byte[] buffer = new byte[bufferLength];  
for (int i = 0; i < 1000; ++i)  
{  
    Thread.Sleep(100);  
    int samplesAvailable = ALC.GetAvailableSamples(mic);  
    ALC.CaptureSamples(mic, buffer, samplesAvailable);  
  
    if(samplesAvailable>0)  
        //Console.WriteLine(new string('|', (buffer[..samplesAvailable].Max())*80/256  
));  
    Console.WriteLine(new string('|', (buffer[..samplesAvailable].Select(x =>  
Math.Abs((int)(x)-128)).Max()) * 80/ 128));  
}  
ALC.CaptureStop(mic);  
  
ALC.CaptureCloseDevice(mic);
```

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edited Feb 2 at 22:07

answered Feb 2 at 21:17



K1521

11 ● 3