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C# SMS API

Ozeki VOIP SIP SDK

Manual

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How to convert Speech to Text using C# with the help of Google Cloud Platform

Tutorial

Download: google-stt.zip

This example demonstrates how to implement the speech-to-text feature in c#, which is able to convert audio dat to text messages. The conversion is based on the powerful Google Cloud Speech API. The converted data can b an audio file, audio stream or real time human voice as well. Any audio supported by Ozeki VoIP SIP SDK i accepted. To understand this article, please read the following tutorial as well: How to configure Google Cloud Platform to your Ozeki VoIP SDK projects

You can choose from all Google translation API supported languages.

An internet access is required. To try this example, you need to have Ozeki VoIP SIP SDK installed, and a reference t OzekiSDK.dll should be added to your Visual Studio project.

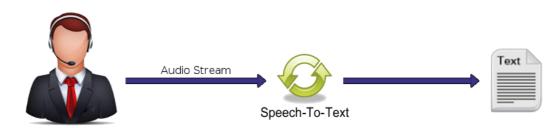


Figure 1 - Speech to Text conversion

What is Speech-to-Text used for?

A speech-to-text (STT) system converts normal speech from multiple languages into text. Users can set speech inputs an save them as text files, so later on the files can be read or analysed. You can use the text results for several purposes. For example you could store phone conversations in written forms. You can also store texts in an SQL database, forward them i e-mail or SMS or search keywords in them.

Speech-to-Text refers to the ability to listen to an audio stream and converting it to a text message. STT engines with different languages, dialects and specialized vocabularies are available through the Google Cloud Speech API. Check if your require ▶ Google SpeechToText language is supported.

How to implement Google speech-to-text feature in your Ozeki VoIP SIP SDK project?

First you will need to register to the Google Cloud Platform, than you need to set the API access credentials on your operatin system and install the Google Cloud Speech SDK. After the installation is finished you will need to reboot your computer t test the example codes in Ozeki VoIP SIP SDK. Here is a detailed tutorial on how you can set up and try you examples.

The sample projects can be downloaded from here (google-stt.zip). Each project contains a basic example that combines th functionality of our SDK and the features provided by the Google Cloud Speech API, presented in a simple C# class GoogleSTT. The GoogleSTT class demonstrates how to implement Speech-to-Text functionality with OzekiSDK powered b the robust Google Cloud Speech API. A wide variety of languages can be given as parameter (e.g. an instance is created a line 25 to 27 in the C# example below). This instance can be attached to the call through the correct sender object (line 29 The instance in the current example can recognise United Kingdom English speech arriving through the microphone an converts it to text messages.

Microphone signals are converted to text in C# using the Google Cloud Speech API

'Program.cs'

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SMS Gateway software Ozeki brings you outstanding SMS Gateway technology. SMS Use Server our products on Windows, Linux, or Android

```
using Ozeki.Media;
   using System;
4
    namespace Google_Speech_To_Text_V1
6
        class Program
            static MediaConnector connector;
```

```
Developers can use our C#
                            10
                                          static Microphone microphone;
SMS API to send SMS from C#.Net. The C# SMS API
                            11
                            12
                                          static GoogleSTT googleSTT;
                            13
comes with full source code
                                          public static void Main(string[] args)
                            14
                            15
PHP SMS API
                                              Console.OutputEncoding = System.Text.Encoding.UTF8;
The ozeki PHP SMS gateway
                            17
software can be used to send
                            18
                                              connector = new MediaConnector();
SMS from PHP and to receive
                                              microphone = Microphone.GetDefaultDevice();
                            19
SMS usig PHP on your
                            20
website
                            21
                                              var format = new WaveFormat(48000, 16, 1);
                            22
SMPP SMS Gateway
                            23
                                              microphone.ChangeFormat(format);
SMS service providers use
                            24
                                              googleSTT =
our SMPP gateway solution,
                                              that
       offers
                     high
performance SMPP server and SMPP client gateway
                            28
                                              connector.Connect(microphone, googleSTT);
       amazing
with
                   routing
                            30
capabilities
                                              microphone.Start();
                                              googleSTT.Start();
                            34
                                              Console.WriteLine("Speak !!");
                            35
                            36
                            37
                                              Console.ReadLine();
                            38
                            39
                                              Console.WriteLine("Disconnect");
                            40
                            41
                                              connector.Disconnect(microphone, googleSTT);
                            42
                            43
                                              Console.WriteLine("Google dispose");
                            44
                                              googleSTT.Dispose();
                            45
                            46
                                              googleSTT = null;
                            47
                            48
                                              Console.WriteLine("microphone dispose");
                            49
                            50
                                              microphone.Dispose();
                            51
                                              microphone = null;
                            53
                                              Console.WriteLine("connector dispose");
```

connector.Dispose();

connector = null;

'GoogleSTT.cs'

}

}

54 55

56

57

58

59 }

This 'GoogleSTT.cs' example class is capable to provide Speech-to-Text functionality through the Google Cloud Speech AF You can write classes similar to 'GoogleSTT.cs'.

From line 80 to 86 you can see the results of the speech-to-text conversation. The 'result.Alternatives' is a list of object containing every possible result and the confidence level of each result. When speech is converted by the Google Clou servers, the servers can understand speech as multiple texts and render a confidence level to each them from 0.0 to 1.0.

This example selects the text with the biggest confidence value and writes it on the console.

```
using Ozeki.Media;
     using Google.Cloud.Speech.V1Beta1;
     using System;
using System.Threading.Tasks;
using System.Threading;
 Δ
 6
7
     using System.Linq;
8
     namespace Google_Speech_To_Text_V1
10
11
          class GoogleSTT : AudioReceiver
12
          {
13
               SpeechClient speech;
              SpeechClient.StreamingRecognizeStream streamingCall;
14
15
              Task printResponses;
16
17
18
              AudioFormat format:
19
20
              private string _languageCode;
21
              private GoogleLanguage _language;
public GoogleLanguage Language
22
23
24
25
                   get { return _language; }
26
                   set
27
                   {
28
                         language = value;
29
                        _languageCode = _language.GetCode();
30
                   }
              }
32
              public GoogleSTT(string languageCode)
                   : this(GoogleLanguageExt.GetGoogleLanguageFromCode(languageCode),
35
                            new AudioFormat())
36
              { }
              public GoogleSTT(GoogleLanguage languageCode, AudioFormat format)
```

```
39
 40
                    Language = languageCode;
 41
 42
                    SetReceiveFormats(format);
 43
 44
                    _format = format;
 45
                    Init();
 46
 47
               }
 48
               private void Init()
 49
 50
 51
                    speech = SpeechClient.Create();
 52
 53
                    streamingCall = speech.StreamingRecognize();
 54
                    streamingCall.WriteAsync(
 55
                       new StreamingRecognizeRequest()
 56
 57
                            StreamingConfig = new StreamingRecognitionConfig()
 58
 59
 60
                                Config = new RecognitionConfig()
 61
 62
                                     Encoding =
 63
                                     RecognitionConfig.Types.AudioEncoding.Linear16,
                                     SampleRate = _format.SampleRate,
LanguageCode = _languageCode,
MaxAlternatives = 5
 64
 65
 66
 67
 68
                                InterimResults = true,
 69
 70
71
                       });
 72
                    printResponses = Task.Run(async () =>
 73
 74
                        while (await streamingCall.ResponseStream.MoveNext(
 75
                             default(CancellationToken)))
 76
 77
                             foreach (var result in streamingCall.ResponseStream
 78
                                  .Current.Results)
 79
 80
                                 if (result.IsFinal)
 81
 82
                                      var top =
                                      result.Alternatives.OrderBy(x => x.Confidence).First();
 83
 85
                                      Console.WriteLine(top.Transcript);
 86
                                 }
                             }
 88
                        }
 89
                    });
 90
               }
 91
               object writeLock = new object();
 92
 93
 94
               public bool IsRunning { get; private set; }
 95
 96
               public void Stop()
 97
 98
                    IsRunning = false;
 99
               }
100
               public void Start()
101
102
103
                    IsRunning = true;
104
105
106
               protected override void OnDataReceived(object sender, AudioData data)
107
108
                    if (!IsRunning) return;
109
110
                    lock (writeLock)
111
112
                        var request = new StreamingRecognizeRequest();
                        request.AudioContent = Google.Protobuf.ByteString
.CopyFrom(data.Data, 0, data.Data.Length);
113
114
115
116
                        try
117
118
                             streamingCall.WriteAsync(request).Wait();
119
                        catch (Exception e)
120
121
122
                             streamingCall.WriteCompleteAsync();
123
                             Init();
124
125
                    }
               }
126
127
128
               protected override void Dispose(bool disposing)
129
130
                    Stop();
131
                    if (printResponses != null)
132
133
134
                        printResponses = null;
                    }
135
136
137
                    if (streamingCall != null)
138
139
                        streamingCall = null;
140
141
                    if (speech != null)
142
```

```
143
                       speech = null;
144
145
146
                   base.Dispose(disposing);
147
148
              }
149
          }
     }
150
```

Related Pages

How to convert Text to Speech through Google

How to ring a SIP extension

How to send voice using microphone

How to make a voice call

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Ozeki VoIP SIP SDK - Training guides and examples

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