Real-time Speech to Text to Speech: Using OpenAl TTS

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^{6.} Real-time Speech to Text to Speech: Building Your AI-Based Alexa - using OpenAI's TTS for Text-to-Speech

Step 1: Implement and run Real-time Speech to Text to Speech: Building Your AI-Based Alexa - using Google's gTTS for Text-to-Speech

Step 2: Modify <u>Putting Everything Together</u> by <u>replacing Google</u>'s <u>gTTS</u> with <u>OpenAI</u>'s <u>Text-to-Speech</u>

Step 3: Re-run the code?

Step 1: Implement and run Real-time Speech to Text to Speech

For step 1, please refer to the document about <u>Real-time Speech to Text to Speech : Building Your Al-Based Alexa</u> and follow the steps to complete

```
(venv) PS D:\VS CODE\Python\CS589\TextSpeech> python app.py --model base --english --energy 300 --pau
se 0.8 --dynamic energy --wake word "hey computer" --verbose
D:\VS CODE\Python\CS589\TextSpeech\venv\lib\site-packages\pydub\utils.py:170: RuntimeWarning: Couldn'
t find ffmpeg or avconv - defaulting to ffmpeg, but may not work
  warn("Couldn't find ffmpeg or avconv - defaulting to ffmpeg, but may not work", RuntimeWarning)
Listening...
D:\VS CODE\Python\CS589\TextSpeech\venv\lib\site-packages\whisper\transcribe.py:115: UserWarning: FP1
6 is not supported on CPU; using FP32 instead
  warnings.warn("FP16 is not supported on CPU; using FP32 instead")
You said the wake word. Processing ...
You said: who is the founder of OpenAI
The answer content:, a research institute that is dedicated to developing friendly AI. He is also a c
o-founder of Tesla and Neuralink, companies that are focused on developing advanced technologies in t
he fields of electric cars and brain-computer interfaces, respectively.
Musk was born in South Africa in 1971 and showed an early interest in computers and technology. He mo
ved to Canada at the age of 17 to attend Queen's University and then transferred to the University of
 Pennsylvania, where he received dual bachelor's degrees in economics and physics. He then went on to
 pursue a PhD in energy physics at Stanford University, but dropped out after two days to pursue entr
epreneurial ventures.
In 1995, Musk co-founded Zip2, a company that provided online content publishing software for newspap
Transform the answer to mp3 ... Result will be in reply.mp3!
```

Step 2: Replacing Google's gTTS with OpenAl's Text-to-Speech

To implement OpenAl's TTS, we need to change the code in the reply() function to apply the new OpenAl TTS based on the guide from <u>Text to speech - OpenAl API</u>

```
from openai import OpenAI
    client = OpenAI()
3
4
5
    response = client.audio.speech.create(
6
        model="tts-1",
        voice="alloy",
        input="Hello world! This is a streaming test.",
9
10
   response.stream_to_file("output.mp3")
```

```
from pathlib import Path
                                                                                              except Exception as e:
from openai import OpenAI
                                                                                                  choices = [
client = OpenAI()
                                                                                                      "I'm sorry, I don't know the answer to that",
                                                                                                      "I'm not sure I understand",
def reply(result queue, verbose):
                                                                                                      "I'm not sure I can answer that",
   while True:
       question = result queue.get()
                                                                                                      "Please repeat the question in a different way"
       prompt = "Q: {}?\nA:".format(question)
                                                                                                  answer = choices[np.random.randint(0, len(choices))]
                                                                                                  print("The answer content:" + answer)
       data = openai.Completion.create(
                                                                                                  print("Transform the answer to mp3... Result will be in reply.mp3!")
           model="gpt-3.5-turbo-instruct",
           prompt=prompt,
                                                                                                  text to speak = answer
           temperature=0.5,
           max_tokens=100,
           n=1,
           stop=["\n"]
                                                                                                  speech file path = Path(_file__).parent / "speech.mp3"
                                                                                                  response = client.audio.speech.create(
       try:
                                                                                                    model="tts-1", # Adjust model if needed (check OpenAI documentation)
           answer = data["choices"][0]["text"]
           print("The answer content:" + answer)
                                                                                                    voice="nova", # Choose a voice from available options
           print("Transform the answer to mp3... Result will be in reply.mp3!")
                                                                                                    input=text to speak,
                                                                                                    lang="en", # Specify language explicitly (optional)
           text to speak = answer
                                                                                                  response.stream to file(speech file path)
           speech file path = Path( file ).parent / "speech.mp3"
                                                                                                  if verbose:
                                                                                                      print(e)
           response = client.audio.speech.create(
             model="tts-1", # Adjust model if needed (check OpenAI documentation)
             voice="nova", # Choose a voice from available options
                                                                                              audio = AudioSegment.from_file(speech_file_path)
             input=text_to_speak,
                                                                                              play(audio)
             lang="en", # Specify language explicitly (optional)
           response.stream to file(speech file path)
                                                                                      main()
```

Step 3: Rerun the code

When executing the program, we have the error below:

```
(venv) PS D:\VS CODE\Python\CS589\TextSpeech> python app_openaiTTS.py --model base --english --energy 300 --pause 0.8 --dynamic_energy --wake_word "hey computer" --verbose
D:\VS CODE\Python\CS589\TextSpeech\venv\lib\site-packages\pydub\utils.py:170: RuntimeWarning: Couldn't find ffmpeg or avconv - defaulting to ffmpeg, but may not work
warn("Couldn't find ffmpeg or avconv - defaulting to ffmpeg, but may not work", RuntimeWarning)
Traceback (most recent call last):
File "app_openaiTTS.py", line 89, in <module>
from openai import OpenAI
ImportError: cannot import name 'OpenAI' from 'openai' (D:\VS CODE\Python\CS589\TextSpeech\venv\lib\site-packages\openai\__init__.py)
(venv) PS D:\VS CODE\Python\CS589\TextSpeech>
```

The error happens because we use the version 0.28 for the previous assignments (CustomerEmail, Agents, DocumentLoad, etc.)

Refer to the document below to fix the problem.

Cannot import name 'OpenAl' from 'openai' - API

```
Requirement already satisfied: openai in d:\vs code\python\cs589\textspeech\venv\lib\site-packages (0.28.0)
Collecting openai
 Downloading openai-1.14.1-pv3-none-anv.whl (257 kB)
                                     257 kB 1.6 MB/s
Requirement already satisfied: sniffio in d:\vs code\python\cs589\textspeech\venv\lib\site-packages (from openai) (1.3.1)
Requirement already satisfied: pydantic<3,>=1.9.0 in d:\vs code\python\cs589\textspeech\venv\lib\site-packages (from openai) (2.6.4)
Requirement already satisfied: tqdm>4 in d:\vs code\python\cs589\textspeech\venv\lib\site-packages (from openai) (4.66.2)
Requirement already satisfied: typing-extensions<5,>=4.7 in d:\vs code\pvthon\cs589\textspeech\venv\lib\site-packages (from openai) (4.10.0)
Requirement already satisfied: distro<2,>=1.7.0 in d:\vs code\python\cs589\textspeech\venv\lib\site-packages (from openai) (1.9.0)
Requirement already satisfied: httpx<1,>=0.23.0 in d:\vs code\python\cs589\textspeech\venv\lib\site-packages (from openai) (0.27.0)
Requirement already satisfied: anyio<5,>=3.5.0 in d:\vs code\python\cs589\textspeech\venv\lib\site-packages (from openai) (4.3.0)
Requirement already satisfied: exceptiongroup>=1.0.2 in d:\vs code\python\cs589\textspeech\venv\lib\site-packages (from anyio<5,>=3.5.0->openai) (1.2.0)
Requirement already satisfied: idna>=2.8 in d:\vs code\python\cs589\textspeech\venv\lib\site-packages (from anyio<5,>=3.5.0->openai) (3.6)
Requirement already satisfied: certifi in d:\vs code\python\cs589\textspeech\venv\lib\site-packages (from httpx<1,>=0.23.0->openai) (2024.2.2)
Requirement already satisfied: httpcore==1.* in d:\vs code\python\cs589\textspeech\venv\lib\site-packages (from httpx<1,>=0.23.0->openai) (1.0.4)
Requirement already satisfied: h11<0.15,>=0.13 in d:\vs code\python\cs589\textspeech\venv\lib\site-packages (from httpcore==1.*->httpx<1,>=0.23.0->openai) (0.14.0)
Requirement already satisfied: pydantic-core==2.16.3 in d:\vs code\python\cs589\textspeech\venv\lib\site-packages (from pydantic<3,>=1.9.0->openai) (2.16.3)
Requirement already satisfied: annotated-types>=0.4.0 in d:\vs code\python\cs589\textspeech\venv\lib\site-packages (from pydantic<3,>=1.9.0->openai) (0.6.0)
Requirement already satisfied: colorama in d:\vs code\python\cs589\textspeech\venv\lib\site-packages (from tqdm>4->openai) (0.4.6)
Installing collected packages: openai
  Attempting uninstall: openai
    Found existing installation: openai 0.28.0
   Uninstalling openai-0.28.0:
```

You should consider upgrading via the 'd:\vs code\python\cs589\textspeech\venv\scripts\python.exe -m pip install --upgrade pip' command.

(venv) PS D:\VS CODE\Python\CS589\TextSpeech>

(venv) PS D:\VS CODE\Python\CS589\TextSpeech> pip install openai --upgrade

Successfully uninstalled openai-0.28.0

WARNING: You are using pip version 21.1.1; however, version 24.0 is available.

Successfully installed openai-1.14.1

Execute the program

Retry again with the new OpenAl version, we experience an old error related to *openai.Completion.create* that is no longer supported in the new version (1.14) of OpenAl

(The image of error is in the next slide)

For now, please refer to the blog about openai.Completion.create degradation to solve the issue.



```
'''example completion with openai > 1.1'''
from openai import OpenAI
client = OpenAI()
prompt = "True or false: a banana is smaller than a lemon.\n\n"
response = client.completions.create(
    prompt=prompt,
    model="gpt-3.5-turbo-instruct",
    top_p=0.5, max_tokens=50,
    stream=True)
for part in response:
    print(part.choices[0].text or "")
```

We just need to change openai.Completion.create into client.completions.create and try to run the code again.

```
(venv) PS D:\VS CODE\Python\CS589\TextSpeech> python app openaiTTS.py --model base --english --energy 300 --pause 0.8 --d
 ynamic energy --wake word "hey computer" --verbose
 D:\VS CODE\Python\CS589\TextSpeech\venv\lib\site-packages\pydub\utils.py:170: RuntimeWarning: Couldn't find ffmpeg or avc
 onv - defaulting to ffmpeg, but may not work
   warn("Couldn't find ffmpeg or avconv - defaulting to ffmpeg, but may not work", RuntimeWarning)
 Listening...
 D:\VS CODE\Python\CS589\TextSpeech\venv\lib\site-packages\whisper\transcribe.py:115: UserWarning: FP16 is not supported o
 n CPU; using FP32 instead
   warnings.warn("FP16 is not supported on CPU; using FP32 instead")
 You did not say the wake word.. Ignoring
 You said the wake word. Processing ...
 You said: where is the San Francisco Bay University
 Exception in thread Thread-3:
 Traceback (most recent call last):
   File "C:\Program Files\WindowsApps\PythonSoftwareFoundation.Python.3.8 3.8.2800.0 x64 qbz5n2kfra8p0\lib\threading.py",
  line 932, in bootstrap inner
     self.run()
   File "C:\Program Files\WindowsApps\PythonSoftwareFoundation.Python.3.8 3.8.2800.0 x64 qbz5n2kfra8p0\lib\threading.py",
  line 870, in run
     self. target(*self. args, **self. kwargs)
   File "app openaiTTS.py", line 97, in reply
     data = openai.Completion.create(
   File "D:\VS CODE\Python\C5589\TextSpeech\venv\lib\site-packages\openai\lib\ old api.py", line 39, in call
     raise APIRemovedInV1(symbol=self. symbol)
 openai.lib. old api.APIRemovedInV1:
 You tried to access openai.Completion, but this is no longer supported in openai>=1.0.0 - see the README at https://githu
 b.com/openai/openai-python for the API.
 You can run `openai migrate` to automatically upgrade your codebase to use the 1.0.0 interface.
 Alternatively, you can pin your installation to the old version, e.g. `pip install openai == 0.28`
```

A detailed migration guide is available here: https://github.com/openai/openai-python/discussions/742

You did not say the wake word. Ignoring You did not say the wake word. Ignoring

'Completion' object is not subscriptable

```
(venv) PS D:\VS CODE\Python\CS589\TextSpeech> python app_openaiTTS.py --model base --english --energy 300
 --pause 0.8 --dynamic energy --wake word "hey computer" --verbose
D:\VS CODE\Python\CS589\TextSpeech\venv\lib\site-packages\pydub\utils.py:170: RuntimeWarning: Couldn't fi
nd ffmpeg or avconv - defaulting to ffmpeg, but may not work
  warn("Couldn't find ffmpeg or avconv - defaulting to ffmpeg, but may not work", RuntimeWarning)
Listening...
D:\VS CODE\Python\CS589\TextSpeech\venv\lib\site-packages\whisper\transcribe.py:115: UserWarning: FP16 is
 not supported on CPU; using FP32 instead
  warnings.warn("FP16 is not supported on CPU; using FP32 instead")
You did not say the wake word.. Ignoring
You did not say the wake word.. Ignoring
You said the wake word.. Processing ...
You said: who is the founder of OpenAI
The answer content: I'm sorry, I don't know the answer to that
Transform the answer to mp3... Result will be in speech.mp3!
app openaiTTS.py:148: DeprecationWarning: Due to a bug, this method doesn't actually stream the response
content, `.with streaming response.method()` should be used instead
  response.stream to file(speech file path)
Exception msg: 'Completion' object is not subscriptable
Verbose here
```

Solution

Please refer to the document <u>Open Al TypeError: 'Completion' object is not subscriptable - Stack Overflow</u> to read about the error.

```
try:
    answer = data["choices"][0]["text"]
    print("The answer content:" + answer)
    print("Transform the answer to mp3... Result will be in reply.mp3!")
    # Replace "answer" with the actual text you want to convert to speech
    text_to_speak = answer
```

We need to change the line into:

<u>answer = data.choices[0].text</u>

Final result:

Retry again and now we have the result we wanted:

```
(venv) PS D:\VS CODE\Python\CS589\TextSpeech> python app openaiTTS.py --model base --english --energy 300
   --pause 0.8 --dynamic energy --wake word "hey computer" --verbose
 D:\VS CODE\Python\CS589\TextSpeech\venv\lib\site-packages\pydub\utils.py:170: RuntimeWarning: Couldn't fi
 nd ffmpeg or avconv - defaulting to ffmpeg, but may not work
   warn("Couldn't find ffmpeg or avconv - defaulting to ffmpeg, but may not work", RuntimeWarning)
 Listening...
 D:\VS CODE\Python\CS589\TextSpeech\venv\lib\site-packages\whisper\transcribe.py:115: UserWarning: FP16 is
  not supported on CPU; using FP32 instead
   warnings.warn("FP16 is not supported on CPU; using FP32 instead")
 You did not say the wake word.. Ignoring
 You said the wake word.. Processing ...
 You said: who is the founder of OpenAI
 The answer content: The founders of OpenAI are Elon Musk, Sam Altman, Greg Brockman, Ilya Sutskever, Wojc
 iech Zaremba, John Schulman, and Chris Olah.
 Transform the answer to mp3... Result will be in speech.mp3!
  app openaiTTS.py:123: DeprecationWarning: Due to a bug, this method doesn't actually stream the response
 content, `.with streaming response.method()` should be used instead
    response.stream to file(speech file path)
```

References

- Real-time Speech to Text to Speech : Building Your Al-Based Alexa
- Text to speech OpenAl API

Original repo: https://github.com/MynameisKoi/CS589/tree/main/TextSpeech

Source code: app_openaiTTS.py

Result: speech.mp3