

对话音频生成

對話範例

```
[  
  {  
    "speaker": "A",  
    "line": " 嘿，我最近經常聽到關於機器學習（ML）、深度學習（DL）和 GPT 的討論。你能幫我解釋一下嗎？"  
  },  
  {  
    "speaker": "B",  
    "line": " 當然！我們從基礎開始講起。機器學習是計算機科學的一個領域，系統通過數據學習來提高性能，而  
  }  
]
```

程式碼

```
import os  
import json  
import random  
import subprocess  
from google.cloud import texttospeech  
import tempfile  
import time  
import argparse  
  
# 固定的對話輸出目錄  
OUTPUT_DIRECTORY = "assets/conversations"  
INPUT_DIRECTORY = "scripts/conversation"  
  
def text_to_speech(text, output_filename, voice_name=None):  
    print(f" 正在生成音頻：{output_filename}")  
    try:  
        client = texttospeech.TextToSpeechClient()  
        synthesis_input = texttospeech.SynthesisInput(text=text)  
        if not voice_name:  
            voice_name = random.choice(["en-US-Journey-D", "en-US-Journey-F", "en-US-Journey-O"])  
        voice = texttospeech.VoiceSelectionParams(language_code="en-US", name=voice_name)  
        audio_config = texttospeech.AudioConfig(  
            audio_encoding=texttospeech.AudioEncoding.MP3,
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    effects_profile_id=["small-bluetooth-speaker-class-device"]
)

retries = 5
for attempt in range(1, retries + 1):
    try:
        response = client.synthesize_speech(input=synthesis_input, voice=voice, audio_config=audio_config)
        with open(output_filename, 'wb') as out:
            out.write(response.audio_content)
        print(f" 音頻內容已寫入 {output_filename}")
    return True
except Exception as e:
    print(f" 第 {attempt} 次嘗試失敗 : {e}")
    if attempt == retries:
        print(f" 經過 {retries} 次嘗試後仍無法生成音頻。")
        return False
    wait_time = 2 ** attempt
    print(f" 等待 {wait_time} 秒後重試...")
    time.sleep(wait_time)
except Exception as e:
    print(f" 生成音頻時發生錯誤 : {e}")
    return False

def process_conversation(filename):
    filepath = os.path.join(INPUT_DIRECTORY, filename)
    output_filename = os.path.join(OUTPUT_DIRECTORY, os.path.splitext(filename)[0] + ".mp3")

    if os.path.exists(output_filename):
        print(f" 音頻文件已存在 : {output_filename}")
        return

    try:
        with open(filepath, 'r', encoding='utf-8') as f:
            conversation = json.load(f)
    except Exception as e:
        print(f" 加載對話文件 {filename} 時發生錯誤 : {e}")
        return

    temp_files = []

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voice_name_A = random.choice(["en-US-Wavenet-D", "en-US-Wavenet-E", "en-US-Wavenet-F"])
voice_name_B = random.choice(["en-US-Studio-O", "en-US-Studio-M", "en-US-Studio-Q"])

for idx, line_data in enumerate(conversation):
    speaker = line_data.get("speaker")
    line = line_data.get("line")
    if not line:
        continue
    temp_file = os.path.join(OUTPUT_DIRECTORY, f"temp_{idx}.mp3")
    temp_files.append(temp_file)

    voice_name = None
    if speaker == "A":
        voice_name = voice_name_A
    elif speaker == "B":
        voice_name = voice_name_B

    if not text_to_speech(line, temp_file, voice_name=voice_name):
        print(f"生成 {filename} 的第 {idx+1} 行音頻失敗")
        # 清理臨時文件
        for temp_file_to_remove in temp_files:
            if os.path.exists(temp_file_to_remove):
                os.remove(temp_file_to_remove)
        return

    if not temp_files:
        print(f"未生成 {filename} 的音頻")
        return

# 使用 ffmpeg 進行合併
concat_file = os.path.join(OUTPUT_DIRECTORY, "concat.txt")
with open(concat_file, 'w') as f:
    for temp_file in temp_files:
        f.write(f"file '{os.path.abspath(temp_file)}'\n")

try:
    subprocess.run([
        ['ffmpeg', '-f', 'concat', '-safe', '0', '-i', concat_file, '-c', 'copy', output_filename],
        check=True,
        capture_output=True
    ])

```

