Audio2Art: Transforming Voice Prompts into Visual Creations Using Transformers

Project Documentation

Introduction

Project Title: Audio2Art

Team Members:

- 1. Ayush Prajapati
- 2. Rutu Bhatt
- 3. Dipen Trivedi

Project Overview

Purpose:

Audio2Art bridges the gap between voice and vision by converting spoken language into stunning AI-generated images using transformer models. It is designed to assist artists, educators, and therapists in transforming creative or expressive audio prompts into rich visual artwork.

Features:

- Voice-to-image conversion using state-of-the-art transformer models
- Wav2Vec2-powered speech-to-text conversion
- Stable Diffusion-powered image generation
- Web interface built using Streamlit
- Colab-based deployment with GPU acceleration
- Public access via Localtunnel

Architecture

Model Workflow:

- 1. **Input**: User uploads a .wav audio file.
- 2. **Speech Recognition**: Wav2Vec2 transcribes the audio into a text prompt.
- 3. **Image Generation**: A Stable Diffusion pipeline interprets the prompt and generates a matching image.
- 4. **Output**: The user can view and download the image from the web interface.

Component Flow:

- ImageModel.py handles:
 - Audio processing via promptgen()
 - Image generation via text2image()
- app.py handles:
 - User interface with audio upload and model selection
 - Calling backend functions and displaying results

Setup Instructions

Prerequisites:

- Python ≥ 3.8
- Google Colab (GPU Runtime Recommended)
- Installed Libraries:
 - transformers
 - diffusers
 - torch
 - librosa
 - accelerate
 - streamlit
 - localtunnel

Installation Steps (Colab):

- 1. Clone/prepare the notebook environment.
- 2. Install required libraries:

!pip install transformers diffusers torch librosa accelerate streamlit localtunnel

- 3. Use %writefile to create ImageModel.py and app.py directly in Colab.
- 4. Run the Streamlit app:

```
!streamlit run app.py & !npx localtunnel --port 8501
```

Folder Structure

Since it's Colab-based, there is no local directory. However, virtual structure:

Model and Function Documentation

promptgen(file)

• Input: .wav audio file

• Output: Transcribed text string

• Libraries used: librosa, transformers (Wav2Vec2)

text2image(prompt, repo_id)

• Input: Transcribed text, model ID from Hugging Face

• Output: PIL Image object

• Libraries used: diffusers, torch, time

Testing

Manual Testing (Colab + Streamlit UI):

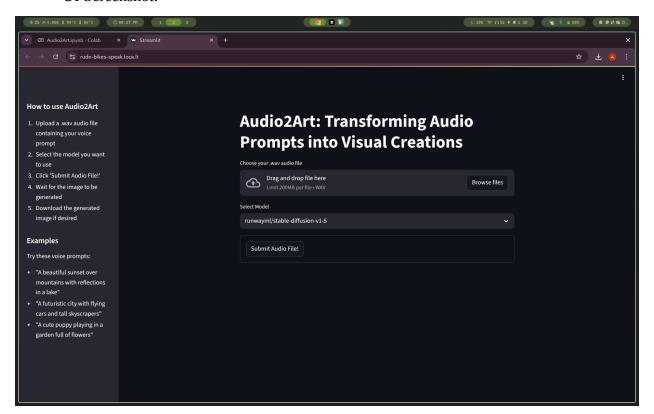
- Audio upload and transcription verified for various accent/pronunciation
- Image generation tested across multiple prompts and models (e.g., runwayml/stable-diffusion-v1-5)

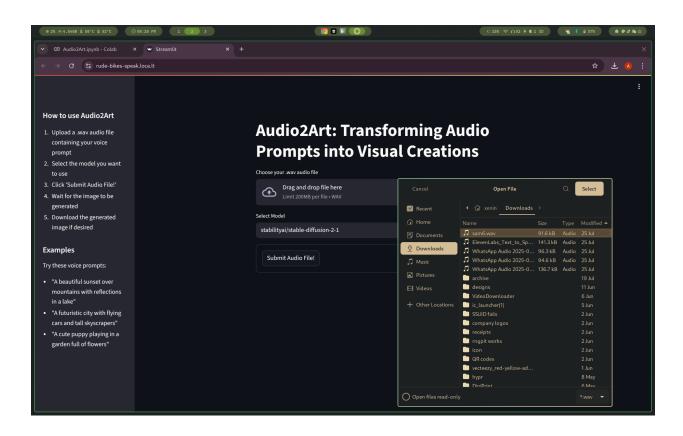
Planned Testing:

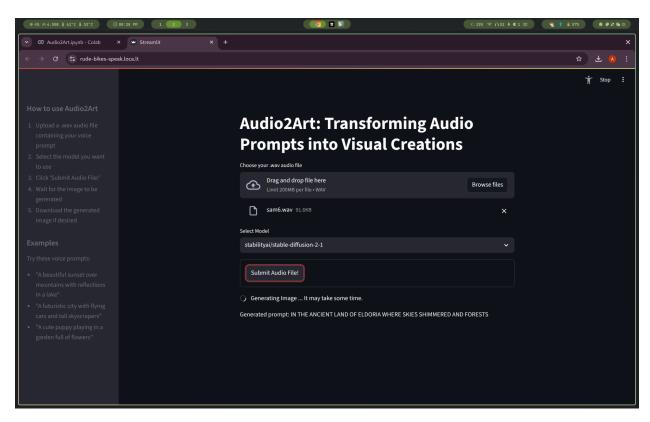
• Unit testing for promptgen and text2image using mock audio/text inputs (future scope)

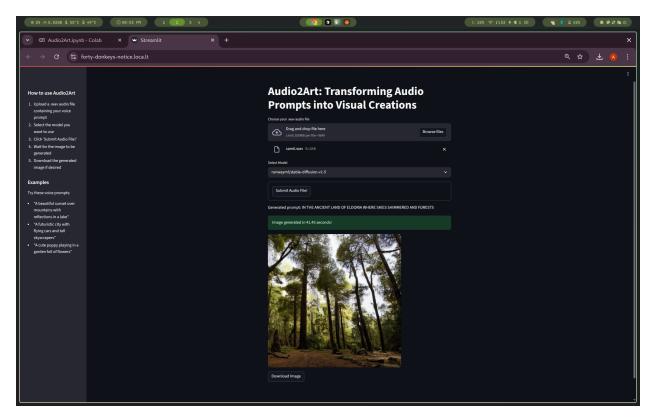
Screenshots or Demo

• UI Screenshot:









• Demo URL (Localtunnel/Colab): [To be generated dynamically during runtime]

Known Issues

- Latency in image generation on CPU-only environment
- Occasional mis-transcription due to unclear audio
- Limited model customization via UI (single model selection)

Future Enhancements

- Add support for MP3/other audio formats
- Multiple image outputs with style variations
- User authentication & image history
- Enhanced audio noise filtering before transcription
- Mobile-responsive UI via Streamlit customization or React frontend