DS565 - Generative Al-Driven Intelligent Apps Development

Project: ChatBot For SFBU

Course: DS565

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Project Overview

- Objective: Create a web-based chatbot with a conversational interface to assist users with inquiries and tasks.
- Description: A chatbot powered by natural language processing (NLP) and integrated with external data sources (e.g., documents, URLs) to provide intelligent, context-aware responses.

Project Workflow

Step 1: Data Loading

Import data from PDFs, web URLs, and other sources.

Step 2: Vectorization

• Convert text into vector representations for efficient similarity search.

Step 3: Similarity Search

• Retrieve relevant information based on user gueries.

Step 4: Generate Response/Speech to Text to Speecg

• Use a pre-trained language model (e.g., GPT) to formulate responses.

Step 5: Display Response

Present the information in a user-friendly web interface.

Technical Architecture

Components:

- Frontend: HTML, CSS, and JavaScript for user interaction.
- Backend: Flask framework to handle API requests.
- **Vector Database**: FAISS or other libraries for efficient similarity search.
- NLP Model: OpenAl GPT or similar models for language understanding and generation.

Data Flow:

User → Chat Interface → Backend API → Vector DB → NLP Model →
Chat Interface->Speech to Text to Speech.

Data Sources

- PDF Documents: Uploaded by the user or administrator, processed for text extraction.
- Web URLs: Loaded dynamically for relevant content extraction.
- YouTube Transcripts: Audio content transcribed and used for Q&A.

Features of the Chatbot

- Natural Language Understanding: Recognizes and processes user input in natural language.
- Contextual Responses: Uses context from previous messages to maintain conversation flow.
- Data Integration: Pulls information from multiple sources like PDFs and web URLs.
- Responsive UI: User-friendly interface with real-time responses and typing indicators.
- Loading Spinner: Indicates when the chatbot is processing information.

Code

```
# Function to load documents
def load_documents():
    docs = []
    pdf_loader = PyPDFLoader(file_path="data/sfbu-catalog.pdf")
    docs.extend(pdf_loader.load())
    with open('data/urls.txt', 'r') as file:
        urls = file.read().splitlines()
        url_loader = UnstructuredURLLoader(urls=urls)
        docs.extend(url_loader.load())
    return docs
```

```
@app.route('/ask', methods=['POST'])
def ask():
    global chat_history
    question = request.json.get("question")
    inputs = {
        "question": question,
        "chat_history": chat_history
    }
    response = qa_chain.invoke(inputs)
    chat_history.append((question, response['answer']))
    return jsonify({"answer": response['answer']})
```

Code

```
# Create vectorstore from documents
documents = load documents()
vectorstore = FAISS.from documents(documents, embedding model)
# Use SimpleMemory instead of ConversationMemory
conversation memory = SimpleMemory()
# Create conversational retrieval chain
qa chain = ConversationalRetrievalChain.from llm(
    11m=11m,
    retriever=vectorstore.as retriever(),
    memory=conversation memory
chat history = []
```

Code

```
def generate tts(text, output dir="static/audio"):
    global previous audio file
   os.makedirs(output dir, exist ok=True)
    unique filename = f"response {uuid.uuid4().hex}.mp3"
    filepath = os.path.join(output dir, unique filename)
   if previous audio file and os.path.exists(previous audio file):
       try:
           os.remove(previous audio file)
       except Exception as e:
            print(f"Error deleting file {previous audio file}: {e}")
   previous audio file = filepath
   with client.audio.speech.with streaming response.create(
       model="tts-1",
       voice="onyx",
    ) as response:
       response.stream to file(filepath)
   return filepath # Return the full path of the audio file
```

```
model = whisper.load_model("base")

def transcribe_audio(audio_file):
    try:
        # Resolve the absolute path to the `temp` folder
        base_dir = os.path.dirname(os.path.abspath(__file__))  # Get directory of `utility.py`
    temp_dir = os.path.join(base_dir, "temp")  # Resolve the temp directory path

# Ensure the temp directory exists
    os.makedirs(temp_dir, exist_ok=True)

# Save the audio file to the `temp` directory
    filename = secure_filename(audio_file.filename)
    temp_path = os.path.join(temp_dir, filename)
    audio_file.save(temp_path)
    transcription = model.transcribe(temp_path,fp16=False)
    return transcription['text']

except Exception as e:
    return f"Error during transcription: {str(e)}"
```





