CODE:

```
# -*- coding: utf-8 -*-
Pre-process-3: Combined and Refined Multimodal Deception Detection Script
import numpy as np
import pandas as pd
import torch
import torch.nn as nn
from sklearn.model_selection import LeaveOneGroupOut
from transformers import BertTokenizer, BertModel
from sklearn.metrics import accuracy_score, f1_score
import os
import re
import cv2 # OpenCV for video processing
import face recognition # Mandatory for subject identification
import librosa # For audio analysis
from moviepy.editor import VideoFileClip # For extracting audio
import torchvision.models as models
import torchvision.transforms as transforms
from PIL import Image
import time
import warnings
import torch.nn.functional as F # Needed for interpolate if using HSTA, but not
used in this simpler model
# Suppress warnings from libraries like moviepy/librosa if needed
warnings.filterwarnings("ignore")
# --- Configuration ---
BERT MODEL NAME = 'bert-base-uncased'
VISUAL_MODEL_NAME = 'resnet18' # Using ResNet18 for visual features
AUDIO N MFCC = 13 # Number of MFCCs for audio features
VISUAL FRAMES_TO_SAMPLE = 70 # Increased frames back to version 1's value [cite:
491]
DEVICE = torch.device("cuda" if torch.cuda.is available() else "cpu")
print(f"Using device: {DEVICE}")
# --- 1. Subject Identification (Mandatory Facial Recognition) ---
# Using the robust function from pre-process-2 [cite: 3, 495]
def identify subjects facial recognition(data):
    Identifies subjects MANDATORILY using facial recognition from the first frame.
    Maps trial id to a subject label.
        data (list): List of dictionaries from load_data, containing
                      'video_path' and 'video_id' (trial_id). [cite: 4]
        dict: A dictionary mapping trial_ids (video_ids) to subject labels
              (e.g., 'subject_1', 'unknown_trial_xyz'). [cite: 5, 494]
    print("Starting mandatory facial recognition for subject identification...")
    subject mapping = {}
    known faces = {} # Store known face encodings and labels {label: encoding}
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subject counter = 1
    unknown counter = 1
    for item in data:
        video_path = item['video_path']
        trial_id = item['video_id'] # Use the actual trial_id
        subject label = None
        trv:
            cap = cv2.VideoCapture(video path)
            if not cap.isOpened():
                print(f"Warning: Could not open video file {video path} for trial
{trial_id}. Assigning unknown subject.")
                subject_label = f'unknown_video_open_error_{unknown_counter}'
                unknown_counter += 1
                subject_mapping[trial_id] = subject_label
                continue
            # Read the first frame
            ret, frame = cap.read()
            if not ret:
                print(f"Warning: Could not read frame from video {video path} for
trial {trial_id}. Assigning unknown subject.")
                subject_label = f'unknown_frame_read_error_{unknown_counter}'
                unknown counter += 1
                subject_mapping[trial_id] = subject_label
                cap.release()
                continue
            # Convert the frame to RGB (face recognition uses RGB)
            rgb_frame = cv2.cvtColor(frame, cv2.COLOR_BGR2RGB)
            # Find face locations and encodings in the frame [cite: 7, 496]
            face_locations = face_recognition.face_locations(rgb_frame)
            face encodings = face recognition.face encodings(rgb frame,
face locations)
            if not face encodings:
                #print(f"Warning: No faces found in the first frame of video
{trial id}. Assigning unique unknown subject.")
                subject label = f'unknown no face {unknown counter}' # Assign a
unique unknown label
                unknown counter += 1
            else:
                # Use the first face found
                current_face_encoding = face_encodings[0]
                match_found = False
                known labels = list(known faces.keys())
                if known labels:
                    known encodings = list(known faces.values())
                    # Increase tolerance slightly if needed, default is 0.6
                    matches = face recognition.compare faces(known encodings,
current_face_encoding, tolerance=0.6)
                    try:
                        first match index = matches.index(True)
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subject_label = known labels[first match index]
                         match found = True
                     except ValueError: # No True value found in matches
                         pass
                 if not match_found:
                     # If no match, add the face to known faces
                     subject label = f'subject {subject counter}'
                     known faces[subject label] = current face encoding
                     subject counter += 1
             subject mapping[trial id] = subject label
            cap.release()
        except Exception as e:
            print(f"Error processing video {video_path} for trial {trial_id}: {e}.
Assigning unknown subject.")
            subject label = f'unknown processing error {unknown counter}'
            unknown counter += 1
            subject_mapping[trial_id] = subject_label
            if 'cap' in locals() and cap.isOpened():
                 cap.release()
    print(f"Facial recognition complete. Identified {subject_counter - 1} unique
subjects and {unknown_counter - 1} videos needing unique IDs.")
    return subject mapping
# --- 2. Data Loading ---
# Using the refined function from pre-process-2 [cite: 9, 499] that reads .txt
transcriptions
def load_data(data_dir, annotation_file):
    """Loads and synchronizes annotation, transcription, and video data."""
    print("Loading data...")
    # Construct paths
    clip dirs = [
        os.path.join(data_dir, 'Clips', 'Deceptive'),
os.path.join(data_dir, 'Clips', 'Truthful')
    transcript dirs = [
        os.path.join(data_dir, 'Transcription', 'Deceptive'),
os.path.join(data_dir, 'Transcription', 'Truthful')
    # Initialize lists to store paths
    video_paths = []
    transcription paths = []
    for clip_dir in clip_dirs:
        if os.path.isdir(clip_dir):
             for filename in os.listdir(clip_dir):
                 if filename.endswith(".mp4"):
                     video_paths.append(os.path.join(clip_dir, filename))
        else:
            print(f"Warning: Clip directory not found: {clip_dir}")
    # Load transcription paths (looking for .txt files) [cite: 10, 499]
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for transcript dir in transcript dirs:
       if os.path.isdir(transcript dir):
           for filename in os.listdir(transcript_dir):
                if filename.endswith(".txt"): # Changed from .csv to .txt
                   transcription_paths.append(os.path.join(transcript_dir,
filename))
       else:
           print(f"Warning: Transcription directory not found: {transcript dir}")
   # Load annotations
   trv:
       annotations df = pd.read csv(annotation file)
   except FileNotFoundError:
       print(f"Error: Annotation file not found at {annotation_file}")
       return []
   # Create mapping dictionaries
   video path dict = {}
   for video path in video paths:
       video filename = os.path.basename(video path)
       # Extract trial id like 'trial lie 001' or 'trial truth 001'
       match = re.search(r"trial (truth|lie) (\d+)\.mp4", video filename)
       if match:
           trial id = f"trial {match.group(1)} {match.group(2)}"
           video path dict[trial id] = video path
   print(f"Found {len(video paths)} video files")
   print(f"Found {len(transcription paths)} transcription files")
   transcription dict = {}
   for transcript_path in transcription_paths:
       try:
           # Read the .txt file directly [cite: 11, 500]
           with open(transcript_path, 'r', encoding='utf-8') as f:
    transcription_text = f.read().strip()
           # Extract trial id from filename
           filename = os.path.basename(transcript path)
           trial_id = filename.replace('.txt', '') # Remove .txt extension
           # print(f"Reading transcription file: {transcript path}") # Optional
Debug
           transcription dict[trial id] = transcription text
           # print(f"Added transcription for {trial_id}") # Optional Debug
       except Exception as e:
           print(f"Error reading transcription file {transcript path}: {e}")
   # Synchronize data
   synchronized_data = []
   processed ids = set()
   if 'id' not in annotations df.columns or 'class' not in
annotations df.columns:
       print(f"Error: Annotation file missing 'id' or 'class' column.")
       return []
   trial_lie 001.mp4'
       trial id raw = str(row['id'])
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```
match = re.search(r"trial (truth|lie) (\d+)", trial id raw)
        if not match:
             print(f"Warning: Skipping annotation row with unexpected id format:
{trial_id_raw}")
             continue
        trial_id = f"trial_{match.group(1)}_{match.group(2)}"
        annotation label = row['class']
        if annotation label not in ['truthful', 'deceptive']:
            print(f"Warning: Skipping trial {trial_id} due to unexpected class
label: {annotation label}")
            continue
        if trial_id in processed_ids:
            print(f"Warning: Duplicate trial ID {trial_id} found in annotations.
Skipping.")
            continue
        transcription_text = transcription_dict.get(trial_id)
        video path = video path dict.get(trial id)
        if transcription text is None:
            print(f"Warning: Transcription not found for trial {trial id}.
Skipping trial.")
            continue
        if video path is None:
            print(f"Warning: Video path not found for trial {trial id}. Skipping
trial.")
            continue
        label map = {'truthful': 0, 'deceptive': 1}
        numeric label = label map.get(annotation label)
        if numeric label is None: # Should not happen with the check above, but
            print(f"Warning: Could not map label '{annotation label}' for trial
{trial id}. Skipping trial.")
            continue
        synchronized_data.append({
            'annotation': numeric_label,
            'transcription': transcription_text,
            'video id': trial id,
            'video_path': video_path
        })
        processed ids.add(trial id)
    print(f"Data loading complete. Found {len(synchronized_data)} synchronized
trials.")
    return synchronized data
# --- 3. Feature Extraction ---
# 3.1 NLP Feature Extraction (BERT) - Unchanged from pre-process-2 [cite: 15, 503]
def extract nlp features(transcriptions):
```

```
Extracts BERT embeddings for a list of transcriptions. """
    print("Extracting NLP features (BERT)...")
    tokenizer = BertTokenizer.from pretrained(BERT MODEL NAME)
    model = BertModel.from_pretrained(BERT_MODEL_NAME).to(DEVICE)
    model.eval()
    nlp_features = []
    with torch.no grad():
        for i, text in enumerate(transcriptions):
            try:
                # Ensure text is a string
                text = str(text) if text is not None else ""
                if not text.strip(): # Handle empty strings
                    print(f"Warning: Empty transcription for item {i}. Using zero
vector.")
                    # Get expected hidden size from model config
                    hidden_size = model.config.hidden_size
                    sentence embedding = np.zeros((1, hidden size))
                else:
                    inputs = tokenizer(text, return_tensors='pt', truncation=True,
padding=True, max length=512).to(DEVICE) # Added max length
                    outputs = model(**inputs)
                    sentence embedding =
outputs.last_hidden_state.mean(dim=1).cpu().numpy()                         # (1, hidden_size)
                nlp features.append(sentence embedding)
            except Exception as e:
                print(f"Error extracting NLP features for item {i}: {e}. Using
zero vector.")
                hidden_size = model.config.hidden_size
                nlp_features.append(np.zeros((1, hidden_size)))
    print("NLP feature extraction complete.")
    # Ensure all features are arrays and handle potential shape issues before
stacking
    processed features = []
    target shape = None
    if nlp features: # Find target shape from the first valid feature
         for feat in nlp features:
             if isinstance(feat, np.ndarray):
                 target_shape = feat.shape
                 break
    if target shape is None and nlp features: # If no valid features found but
list not empty
        # Fallback: try to get shape from model config if possible
        try:
             target_shape = (1, model.config.hidden_size)
             print(f"Warning: Could not determine NLP target shape from features,
using default BERT hidden size: {target shape}")
        except Exception:
             print("Error: Cannot determine NLP feature shape. Returning empty
array.")
             return np.array([]) # Cannot proceed without a shape
    elif not nlp_features:
        print("Warning: No NLP features extracted.")
        return np.array([])
```

```
for feat in nlp features:
        if isinstance(feat, np.ndarray):
            if feat.shape == target_shape:
                processed features.append(feat)
                # If shape mismatch, pad or truncate (or use zeros as done in
exception handling)
                print(f"Warning: NLP feature shape mismatch ({feat.shape} vs
{target_shape}). Using zero vector of target shape.")
                processed_features.append(np.zeros(target_shape))
        else: # Should not happen if exceptions are caught, but as safeguard
            print(f"Warning: Non-array NLP feature found. Using zero vector of
target shape.")
            processed features.append(np.zeros(target shape))
    if not processed_features:
        return np.array([]) # Return empty array if no features were processed
    return np.vstack(processed features) # (num trials, hidden size)
# 3.2 Audio Feature Extraction (MFCCs) - Unchanged from pre-process-2 [cite: 18,
5067
def extract audio features(video paths):
    """ Extracts MFCC features from the audio track of video files. """
    print("Extracting Audio features (MFCCs)...")
    audio_features = []
temp_audio_dir = "temp_audio"
    if not os.path.exists(temp_audio_dir):
        os.makedirs(temp audio dir)
    num features = AUDIO N MFCC * 2 # Mean and Std Dev for each MFCC
    for i, video path in enumerate(video paths):
        # start_time = time.time() # Optional timer
        temp_audio_path = os.path.join(temp_audio_dir,
f"temp_{os.path.basename(video_path)}.wav") # More robust temp name
        feature_vector = np.zeros(num_features) # Default to zeros
        try:
            # Extract audio using moviepy
            with VideoFileClip(video path) as video clip:
                if video clip.audio is None:
                    print(f"Warning: Video {i} ({os.path.basename(video path)})
has no audio track. Using zeros.")
                else.
                    # Specify logger=None to reduce console output
                    video_clip.audio.write_audiofile(temp_audio_path,
codec='pcm_s16le', logger=None)
                    # Load audio and extract MFCCs using librosa
                    # Load with native sample rate, librosa handles resampling if
needed by feature extraction
                    y, sr = librosa.load(temp_audio_path, sr=None)
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if len(y) > 0: # Check if audio signal is not empty
                        mfccs = librosa.feature.mfcc(y=y, sr=sr,
n mfcc=AUDIO N MFCC)
                        mfccs_mean = np.mean(mfccs, axis=1)
                        mfccs_std = np.std(mfccs, axis=1)
                        feature_vector = np.concatenate((mfccs_mean, mfccs_std))
                        print(f"Warning: Audio signal empty after extraction for
video {i}. Using zeros.")
        except Exception as e:
            print(f"Error extracting audio features for video {i}
({os.path.basename(video path)}): {e}. Using zeros.")
            # Ensure feature vector remains zeros
        finally:
            # Clean up temporary audio file
            if os.path.exists(temp audio path):
                    os.remove(temp audio path)
                except Exception as e rem: # Different variable name for exception
                    print(f"Warning: Could not remove temp audio file
{temp audio path}: {e rem}")
        audio features.append(feature vector)
        # Optional progress print:
        # print(f"Processed audio for video {i+1}/{len(video paths)} in
{time.time() - start time:.2f}s")
    # Clean up temp directory if empty
    try:
        if os.path.exists(temp_audio_dir) and not os.listdir(temp_audio_dir):
             os.rmdir(temp_audio_dir)
    except Exception as e_rem_dir:
        print(f"Warning: Could not remove temp audio directory {temp audio dir}:
{e rem dir}")
    print("Audio feature extraction complete.")
    return np.array(audio features) # (num trials, num audio features)
# 3.3 Visual Feature Extraction (ResNet) - Using sampling rate from version 1
[cite: 491]
def extract visual features(video paths):
    """ Extracts aggregated visual features using a pre-trained ResNet model. """
    print("Extracting Visual features (ResNet)..."
    # Load pre-trained ResNet model without the final classification layer
    vis model = models.resnet18(pretrained=True)
    vis_model = nn.Sequential(*list(vis_model.children())[:-1]) # Remove the fully
connected layer
    vis model = vis model.to(DEVICE)
    vis_model.eval()
    # Define image transformations appropriate for ResNet
    preprocess = transforms.Compose([
        transforms.Resize(256),
        transforms.CenterCrop(224),
        transforms.ToTensor(),
```

```
transforms.Normalize(mean=[0.485, 0.456, 0.406], std=[0.229, 0.224,
0.225]),
    ])
    visual_features = []
    num_visual_features = 512 # ResNet18 output size before FC layer
    with torch.no grad():
        for i, video path in enumerate(video paths):
            # start time = time.time() # Optional timer
            video feature vector = np.zeros(num visual features) # Default to
zeros
            try:
                cap = cv2.VideoCapture(video_path)
                if not cap.isOpened():
                    print(f"Warning: Could not open video {i}
({os.path.basename(video path)}). Using zeros.")
                    visual_features.append(video_feature_vector)
                    continue
                frame_count = int(cap.get(cv2.CAP_PROP_FRAME_COUNT))
                if frame count <= 0:</pre>
                    print(f"Warning: Video {i} ({os.path.basename(video_path)})
has no frames or invalid frame count. Using zeros.")
                    visual_features.append(video feature vector)
                    cap.release()
                    continue
                # Sample frames evenly across the video - using
VISUAL_FRAMES_TO_SAMPLE = 70 [cite: 491]
                num_frames_to_sample = min(VISUAL_FRAMES_TO_SAMPLE, frame_count) #
Don't sample more than available
                if num_frames_to_sample > 0:
    frame_indices = np.linspace(0, frame_count - 1,
num_frames_to_sample, dtype=int)
                else:
                    frame indices = [] # Should not happen due to frame count
check, but safeguard
                frames data = []
                for frame index in frame indices:
                    cap.set(cv2.CAP_PROP_POS_FRAMES, frame_index)
                    ret, frame = cap.read()
                    if ret:
                         # Convert frame BGR -> RGB -> PIL Image -> Apply
transforms
                         frame_rgb = cv2.cvtColor(frame, cv2.COLOR_BGR2RGB)
                         img_pil = Image.fromarray(frame_rgb)
                         img_tensor = preprocess(img_pil).unsqueeze(0).to(DEVICE) #
Add batch dimension
                         frames data.append(img tensor)
                    # else: # Optional: Warn if specific frame fails
                         # print(f"Warning: Could not read frame {frame index} from
video {i}.")
                cap.release()
                if frames data:
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# Stack frame tensors and pass through the model
                    batch_tensor = torch.cat(frames data, dim=0)
                    frame outputs = vis model(batch tensor) # (num sampled frames,
num_visual_features, 1, 1)
                    frame_outputs = frame_outputs.squeeze() # Remove trailing 1s
> (num_sampled_frames, num visual features)
                    # Handle case where only one frame was sampled (squeeze might
remove batch dim)
                    if frame outputs.ndim == 1:
                        video feature vector = frame outputs.cpu().numpy()
                    elif frame outputs.ndim == 2:
                         # Aggregate features (e.g., mean pooling)
                         video feature vector = torch.mean(frame outputs,
dim=0).cpu().numpy()
                    else: # Should not happen
                         print(f"Warning: Unexpected visual feature dimension
{frame outputs.ndim} for video {i}. Using zeros.")
                         video feature vector = np.zeros(num visual features) #
Fallback
                else:
                    print(f"Warning: No frames could be processed for video {i}
({os.path.basename(video_path)}). Using zeros."
                    # video feature vector remains zeros
            except Exception as e:
                print(f"Error extracting visual features for video {i}
({os.path.basename(video_path)}): {e}. Using zeros.")
                if 'cap' in locals() and cap.isOpened():
                    cap.release()
            visual_features.append(video_feature_vector)
            # Optional progress print:
            # print(f"Processed visual for video {i+1}/{len(video paths)} in
{time.time() - start time:.2f}s")
    print("Visual feature extraction complete.")
    return np.array(visual features) # (num trials, num visual features)
# --- 4. Data Preparation for LOSO ---
# Unchanged from pre-process-2 [cite: 27, 515]
def prepare_loso(data, subject_mapping):
    Prepares data for LOSO cross-validation using the mandatory subject mapping.
    Returns filtered annotations, raw transcriptions, mapped subject IDs, and
valid indices.
    print("Preparing data for LOSO...")
    annotations = [item['annotation'] for item in data]
    transcriptions = [item['transcription'] for item in data] # Keep raw text
    video_ids = [item['video_id'] for item in data]
    # Map video_ids (trial_ids) to subject IDs using the facial recognition
mapping
    mapped subject ids = []
    valid indices = [] # Keep track of trials with successful subject mapping
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for idx, video id in enumerate(video ids):
        subject id = subject mapping.get(video id)
        if subject_id is None:
            print(f"Critical Warning: No subject mapping found for video_id
{video_id}. This trial will be skipped in prepare_loso.")
            # Ensure subject IDs are suitable for LOSO grouping (e.g., string or
int)
            # The current mapping creates strings like 'subject 1' or
 unknown no face 1' which is fine.
            mapped subject ids.append(subject id)
            valid indices.append(idx)
    if not valid indices:
         print("Error: No trials have valid subject mappings. Cannot proceed.")
         return np.array([]), [], [], []
    if len(valid indices) < len(data):</pre>
        print(f"Warning: {len(data) - len(valid indices)} trials were skipped due
to missing subject mapping.")
    # Filter data based on valid indices
    annotations filtered = np.array(annotations)[valid_indices]
    transcriptions_filtered = [transcriptions[i] for i in valid_indices]
# The mapped_subject_ids list already corresponds to the valid_indices
    # No, we need to filter the subject_ids based on the final mapping result.
Let's rebuild it.
    mapped subject ids filtered = [subject mapping.get(video ids[i]) for i in
valid indicesl
    print(f"Data preparation complete. {len(annotations_filtered)} trials ready
for LOSO.")
    # Note: Audio/Visual features are extracted separately AFTER prepare loso
filters trials
    return annotations filtered, transcriptions filtered,
mapped subject ids filtered, valid indices
class MultimodalDeceptionModel(nn.Module):
    Simpler multimodal model for deception detection using NLP, Audio, and Visual
    (Based on the architecture from the first provided script).
    def __init__(self, nlp_input_size, audio_input_size, visual_input_size,
hidden_size, num_classes):
        super(MultimodalDeceptionModel, self).__init__()
        # Simple linear processors for each modality
        self.nlp_processor = nn.Linear(nlp_input_size, hidden_size)
        self.audio_processor = nn.Linear(audio_input_size, hidden_size)
        self.visual_processor = nn.Linear(visual_input_size, hidden_size)
        # Activation
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self.relu = nn.ReLU()
        # Fusion and Classifier layers
        self.fusion_dropout = nn.Dropout(0.5) # Dropout after fusion
        self.classifier = nn.Linear(hidden_size * 3, num_classes)
    def forward(self, nlp data, audio data, visual data):
        """ Forward pass processing and fusing features. """
        # Process each modality through its linear layer and activation
        nlp_processed = self.relu(self.nlp_processor(nlp_data))
        audio_processed = self.relu(self.audio_processor(audio_data))
        visual processed = self.relu(self.visual processor(visual data))
        # Fusion by concatenation
        fused features = torch.cat((nlp processed, audio processed,
visual processed), dim=1)
        # Apply dropout
        fused features = self.fusion dropout(fused features)
        # Final classification layer
        output = self.classifier(fused features)
        return output
# --- 6. Training and Evaluation ---
# Unchanged from pre-process-2 [cite: 31, 521]
def train evaluate(model,
                   nlp_train, audio_train, visual_train, labels_train,
                   nlp_test, audio_test, visual_test, labels_test,
                   optimizer, criterion, device, epoch):
    """ Trains and evaluates the multimodal model for one epoch. """
    # --- Training Phase ---
    model.train()
    optimizer.zero grad()
    # Move training data to device
    nlp train tensor = torch.FloatTensor(nlp train).to(device)
    audio_train_tensor = torch.FloatTensor(audio_train).to(device)
    visual train tensor = torch.FloatTensor(visual train).to(device)
    labels_train_tensor = torch.LongTensor(labels_train).to(device)
    # Forward pass (Training)
    outputs = model(nlp_train_tensor, audio_train_tensor, visual_train_tensor)
    loss = criterion(outputs, labels_train_tensor)
    # Backward pass and optimization
    loss.backward()
    optimizer.step()
    train_loss_item = loss.item() # Get loss value for reporting
    # --- Evaluation Phase ---
    model.eval()
    with torch.no grad():
        # Move test data to device
        nlp test tensor = torch.FloatTensor(nlp_test).to(device)
```

```
audio test tensor = torch.FloatTensor(audio test).to(device)
        visual test tensor = torch.FloatTensor(visual_test).to(device)
        labels test tensor = torch.LongTensor(labels test).to(device)
        outputs_test = model(nlp_test_tensor, audio_test_tensor,
visual test tensor)
        , predicted = torch.max(outputs test.data, 1)
        # Calculate metrics
        labels test cpu = labels test tensor.cpu().numpy()
        predicted cpu = predicted.cpu().numpy()
        accuracy = accuracy_score(labels_test_cpu, predicted_cpu)
        f1 = f1_score(labels_test_cpu, predicted_cpu, average='weighted',
zero_division=0) # Added zero_division
    return accuracy, f1, train loss item # Return loss from the training pass
# --- 7. Run LOSO Cross-Validation ---
# Checkpoint loading logic updated slightly to match the simpler model structure.
def run loso(annotations, nlp features, audio features, visual features,
subject ids,
             checkpoint dir="checkpoints", num epochs=50, learning rate=0.001,
hidden size=128):
    """ Runs LOSO cross-validation for the multimodal model. """
    print("Starting LOSO Cross-Validation...")
    loso = LeaveOneGroupOut()
    all accuracies = []
    all f1s = []
    all fold losses = [] # Store average loss of last epoch per fold
    num seeds = 3 # Number of random seeds for stability [cite: 34, 524]
    # Create checkpoint directory if it doesn't exist
    if not os.path.exists(checkpoint dir):
        os.makedirs(checkpoint_dir)
    num_classes = len(np.unique(annotations))
    if num classes < 2:</pre>
        print(f"Error: Only {num classes} unique class found. Cannot perform
classification.")
        return
    # Ensure all feature arrays have the same number of samples as
annotations/subject_ids
    n_samples = len(annotations)
    if not (len(nlp_features) == n_samples and len(audio_features) == n_samples
and len(visual_features) == n_samples and len(subject_ids) == n_samples):
         print("Error: Feature array lengths do not match annotation/subject ID
lengths after filtering!")
         print(f"Lengths: Annotations={n_samples}, NLP={len(nlp_features)},
Audio={len(audio_features)}, Visual={len(visual_features)},
Subjects={len(subject_ids)}")
         return # Cannot proceed if lengths mismatch
```

```
fold num = 0
    n splits = loso.get n splits(groups=subject ids)
    for train_index, test_index in loso.split(X=nlp_features, y=annotations,
groups=subject_ids): # Provide X and y
        fold_num += 1
        fold accuracies seeds = []
        fold f1s seeds = []
        fold_last_epoch_losses_seeds = [] # Store loss of the last epoch for each
        # Identify the subject(s) being left out in this fold
        test subjects = np.unique(np.array(subject_ids)[test_index])
        test_subject_str = ', '.join(map(str, test_subjects)) # Handle multiple
subjects if group has >1 type
        print(f"\n--- Fold {fold_num}/{n_splits}: Testing on Subject(s)
{test_subject_str} ---")
        # Split data for this fold
        nlp train, nlp test = nlp features[train index], nlp features[test index]
        audio train, audio_test = audio_features[train_index],
audio features[test index]
        visual train, visual test = visual features[train index],
visual_features[test_index]
        labels_train, labels_test = annotations[train_index],
annotations[test index]
        # Check if train or test set is empty for this fold
        if len(labels train) == 0 or len(labels test) == 0:
            print(f"Warning: Skipping Fold {fold_num} due to empty train
({len(labels train)}) or test ({len(labels test)}) set.")
            continue
        # Get feature dimensions dynamically
            nlp dim = nlp train.shape[1]
            audio dim = audio train.shape[1]
            visual dim = visual train.shape[1]
        except IndexError:
             print(f"Error: Could not get feature dimensions in Fold {fold num}.
Skipping fold.")
             print(f"Shapes: NLP={nlp_train.shape}, Audio={audio_train.shape},
Visual={visual train.shape}")
             continue
        for seed in range(num_seeds):
            print(f" Seed { seed + 1}/{num_seeds}")
            torch.manual_seed(seed)
            np.random.seed(seed)
            if torch.cuda.is available():
                 torch.cuda.manual seed all(seed) # for multi-GPU
            # Model initialization, optimizer, and loss function
            # Using the simpler model architecture [cite: 519, 520]
            model = MultimodalDeceptionModel(
```

```
nlp input size=nlp dim,
                audio_input_size=audio_dim,
                visual_input_size=visual_dim,
                hidden_size=hidden_size,
                num classes=num classes
            ).to(DEVICE)
            optimizer = torch.optim.Adam(model.parameters(), lr=learning rate)
            criterion = nn.CrossEntropyLoss().to(DEVICE)
            # Define checkpoint file name
            checkpoint file = os.path.join(checkpoint dir,
f"fold {fold num} seed {seed + 1}.pth")
            # Check if checkpoint exists and load it
            start_epoch = 0
            if os.path.exists(checkpoint_file):
                try:
                    # Load checkpoint onto the correct device
                    checkpoint = torch.load(checkpoint file, map location=DEVICE)
                    model.load_state_dict(checkpoint['model_state_dict'])
                    optimizer.load state dict(checkpoint['optimizer state dict'])
                    start_epoch = checkpoint['epoch']
last_loss = checkpoint.get('loss', 'N/A') # Get last saved
loss if available
                    print(f"
                                Resuming training from checkpoint
{checkpoint_file} at epoch {start_epoch + 1} (Last saved loss: {last_loss})")
                except Exception as e:
                    print(f"
                               Warning: Could not load checkpoint
{checkpoint file}: {e}. Starting from scratch.")
                    start_epoch = 0 # Reset start epoch if loading failed
            best_f1_seed = -1.0 # Track best F1 for saving the best model state
for this seed
            last epoch loss this seed = float('nan') # Initialize loss for this
            for epoch in range(start_epoch, num_epochs):
                accuracy, f1, loss = train_evaluate(
                    model,
                    nlp_train, audio_train, visual_train, labels_train,
                    nlp_test, audio_test, visual_test, labels_test,
                    optimizer, criterion, DEVICE, epoch)
                last_epoch_loss_this_seed = loss # Store loss for the current
                # Optional: Save checkpoint based on best F1 for this seed
                if f1 > best_f1_seed:
                    best f1 seed = f1
                                    Epoch {epoch + 1}/{num_epochs} - Loss:
{loss:.4f}, Acc: {accuracy:.4f}, F1: {f1:.4f} (New best F1, saving checkpoint)") #
Verbose saving log
                    torch.save({
                        'epoch': epoch + 1, # Save the epoch number *after*
completion
                        'model_state_dict': model.state_dict(),
                         'optimizer_state_dict': optimizer.state_dict(),
                        'loss': loss,
```

```
'f1': f1, # Save F1 score in checkpoint
'accuracy': accuracy,
                        # Save model hyperparams used for this checkpoint
                        'hidden_size': hidden_size,
                        'learning_rate': learning_rate,
                        'nlp_input_size': nlp_dim,
                        'audio input size': audio dim,
                        'visual_input_size': visual_dim,
                        'num classes': num classes
                    }, checkpoint_file)
                # else: # Optional print for epochs that don't improve F1
                    # print(f"
                                    Epoch {epoch + 1}/{num epochs} - Loss:
{loss:.4f}, Acc: {accuracy:.4f}, F1: {f1:.4f}")
            # --- After training loop for the seed ---
evaluate
            final seed accuracy = float('nan')
            final seed f1 = float('nan')
            if os.path.exists(checkpoint file):
                    # print(f" Loading best checkpoint for seed {seed + 1} to
get final metrics...")
                    checkpoint = torch.load(checkpoint_file, map_location=DEVICE)
                    model = MultimodalDeceptionModel(
                            nlp input size=checkpoint['nlp input size'], # Use
saved dims
                            audio input size=checkpoint['audio input size'],
                            visual input size=checkpoint['visual input size'],
                            hidden_size=checkpoint['hidden_size'],
                            num_classes=checkpoint['num_classes']
                            ).to(DEVICE)
                    model.load state dict(checkpoint['model state dict'])
                    model.eval() # Set model to evaluation mode
                    with torch.no grad():
                         nlp_test_tensor = torch.FloatTensor(nlp_test).to(DEVICE)
                         audio test tensor =
torch.FloatTensor(audio test).to(DEVICE)
                         visual test tensor =
torch.FloatTensor(visual_test).to(DEVICE)
                         labels_test_tensor =
torch.LongTensor(labels_test).to(DEVICE)
                         outputs_final_test = model(nlp_test_tensor,
audio_test_tensor, visual_test_tensor)
                         _, predicted_final = torch.max(outputs_final_test.data,
1)
                         labels_test_cpu = labels_test_tensor.cpu().numpy()
                         predicted_final_cpu = predicted_final.cpu().numpy()
                         final_seed_accuracy = accuracy_score(labels_test_cpu,
predicted final cpu)
                         final seed f1 = f1 score(labels test cpu,
predicted_final_cpu, average='weighted', zero_division=0)
                    print(f" Seed {seed + 1} Final Results (Best F1 Model) -
Accuracy: {final seed accuracy:.4f}, F1: {final seed f1:.4f}")
                 except Exception as e load final:
```

```
print(f"
                                 Warning: Could not load or evaluate best
checkpoint {checkpoint_file} for final metrics: {e_load_final}")
                      # Use metrics from the very last epoch as fallback
                      final_seed_accuracy = accuracy
                      final_seed_f1 = f1
                      print(f"
                                Using results from last epoch instead: Accuracy:
{final seed accuracy:.4f}, F1: {final seed f1:.4f}")
           else:
                 print(f"
                            Warning: No checkpoint found for seed {seed + 1}
after training. Cannot report final metrics.")
           # Store results (from the best model if loaded, else last epoch) for
           fold_accuracies_seeds.append(final seed accuracy)
           fold f1s seeds.append(final seed f1)
           fold last epoch losses seeds.append(last epoch loss this seed) # Still
store last epoch loss
       # --- After all seeds for the fold ---
       # Average metrics across seeds for this fold (handle potential NaNs if a
seed failed)
       if fold accuracies seeds: # Check if any seeds ran successfully
            avg_fold_accuracy = np.nanmean(fold_accuracies seeds) # Use_nanmean
            avg fold f1 = np.nanmean(fold f1s seeds)
           avg fold last epoch loss = np.nanmean(fold last epoch losses seeds)
           all_accuracies.append(avg_fold_accuracy)
           all_f1s.append(avg_fold_f1)
           all_fold_losses.append(avg_fold last epoch loss)
           print(f" Fold {fold num} Average (across seeds) - Accuracy:
{avg_fold_accuracy:.4f}, F1: {avg_fold_f1:.4f}, Last Epoch Loss:
{avg fold last epoch loss:.4f}")
       else:
            print(f" Fold {fold num} - No successful seed runs.")
   # --- After all folds ---
   # Overall results across folds (handle potential NaNs if a fold failed)
   if all accuracies:
       overall_avg_accuracy = np.nanmean(all_accuracies) # Use nanmean
       overall_avg_f1 = np.nanmean(all f1s)
       overall_avg_loss = np.nanmean(all_fold_losses)
       print(f"\n--- Overall LOSO Results (Avg. Across Folds & Seeds) ---")
       print(f"Overall Average Accuracy: {overall avg accuracy:.4f}")
       print(f"Overall Average F1-score: {overall avg f1:.4f}")
       print(f"Overall Average Last Epoch Loss: {overall avg loss:.4f}")
   else:
       print("\n--- No folds completed successfully. Cannot calculate overall
results. ---")
if name == " main ":
   start main time = time.time()
   # --- Configuration ---
```

```
# !!! ADJUST THESE PATHS TO YOUR DATASET LOCATION !!!
    # Using paths from the provided scripts [cite: 46, 534]
    data dir = 'Real-life Deception Detection 2016' # Example path, adjust as
needed
    annotation_file = os.path.join(data_dir, 'Annotation', 'All_Gestures_Deceptive
and Truthful.csv') # Construct path robustly
    checkpoint dir = "multimodal checkpoints simple" # Use a different dir for the
simple model checkpoints
    num epochs main = 50 # Adjust number of epochs if needed [cite: 46, 534]
    learning rate main = 0.001 # [cite: 46, 534]
    hidden size main = 128 # Hidden dimension for feature processing/fusion [cite:
46, 534]
    print(f"--- Starting Deception Detection Pipeline ---")
    print(f"Dataset Directory: {os.path.abspath(data_dir)}")
    print(f"Annotation File: {os.path.abspath(annotation_file)}")
    print(f"Checkpoints Directory: {os.path.abspath(checkpoint_dir)}")
    print(f"Device: {DEVICE}")
    print(f"Number of Epochs: {num epochs main}")
    print(f"Learning Rate: {learning rate main}")
    print(f"Hidden Size: {hidden size main}")
    print(f"Visual Frames to Sample: {VISUAL FRAMES TO SAMPLE}")
    print(f"Audio MFCCs: {AUDIO N MFCC}")
    print(f"------
    # --- Workflow ---
    # 1. Load Data (Paths, Annotations, Transcriptions)
    data = load data(data dir, annotation file)
    if not data:
        print("No data loaded. Exiting.")
        exit()
    # 2. Identify Subjects (Mandatory Facial Recognition) [cite: 535]
    subject_mapping = identify_subjects_facial_recognition(data)
   # print("Subject Mapping (Facial Recognition):", subject_mapping) # Optional
print for debugging
    # 3. Prepare Data for LOSO (Get filtered annotations, raw transcriptions,
mapped IDs) [cite: 535]
    annotations, transcriptions_raw, mapped_subject_ids, valid_indices =
prepare loso(data, subject mapping)
    if len(annotations) == 0:
        print("No valid trials remaining after preparing for LOSO. Exiting.")
        exit()
    # Filter original data list based on valid indices from prepare loso
    # This ensures feature extraction only happens for trials included in LOSO
    valid data = [data[i] for i in valid_indices]
    video_paths_valid = [item['video_path'] for item in valid_data]
    # Check if video_paths_valid is empty, which means no valid trials remained
    if not video_paths_valid:
         print("Error: No valid video paths found after LOSO preparation.
Exiting.")
         exit()
```

```
# 4. Extract Features for the valid trials [cite: 536]
    nlp_features = extract_nlp_features(transcriptions_raw) # Takes raw text list
    audio_features = extract_audio_features(video_paths_valid)
    visual_features = extract_visual_features(video_paths_valid)
    # Sanity check feature shapes before running LOSO
    print(f"Feature shapes after extraction: NLP={nlp features.shape},
Audio={audio features.shape}, Visual={visual features.shape}")
    if not (nlp features.shape[0] == audio features.shape[0] ==
visual features.shape[0] == len(annotations)):
        print("Error: Feature array lengths do not match number of annotations
after filtering and extraction. Exiting."
        print(f"Lengths: Annotations={len(annotations)},
NLP={nlp_features.shape[0]}, Audio={audio_features.shape[0]},
Visual={visual_features.shape[0]}")
        exit()
    if nlp features.size == 0 or audio features.size == 0 or visual features.size
== 0:
        print("Error: One or more feature sets are empty after extraction.
Exiting.")
        exit()
    run loso(annotations, nlp features, audio features, visual features,
mapped_subject_ids,
             checkpoint_dir=checkpoint_dir,
             num_epochs=num_epochs_main,
             learning_rate=learning_rate_main,
             hidden size=hidden size main)
    end main time = time.time()
    print(f"\nTotal execution time: {(end main time - start main time) / 60:.2f}
minutes")
    print("Multimodal training complete.")
```

OUTPUT:

--- Overall LOSO Results (Avg. Across Folds & Seeds) ---

Overall Average Accuracy: 0.9451

Overall Average F1-score: 0.9407

Overall Average Last Epoch Loss: 0.5255