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import whisperx

device = "cpu"          # Colab CPU runtime
compute_type = "float32" # Avoid float16 issues

print("Loading WhisperX model...")
model = whisperx.load_model("large-v2", device=device, compute_type=compute_type)
print("✅ WhisperX model loaded successfully!")

Loading WhisperX model...
config.json: 2.80K? [00:00~00:00, 72.3kB/s]
tokenizer.json: 2.20M? [00:00~00:00, 5.70MB/s]
model.bin: 100% [██████████] 3.09G/3.09G [01:22~00:00, 60.7MB/s]
vocabulary.txt: 460K? [00:00~00:00, 4.00MB/s]
2025-10-28 14:06:14 - whisperx.asr - INFO - No language specified, language will be detected for each audio file (increases inference time)
2025-10-28 14:06:14 - whisperx.vads.pyannote - INFO - Performing voice activity detection using Pyannote...
INFO:pytorch_lightning.utilities.migration.utils:Lightning automatically upgraded your loaded checkpoint from v1.5.4 to v2.5.5. To apply the upgrade to your files permanently, run 'Model was trained with pyannote.audio 0.0.1, yours is 3.4.0. Bad things might happen unless you revert pyannote.audio to 0.x.
Model was trained with torch 1.10.0+cu102, yours is 2.8.0+cu126. Bad things might happen unless you revert torch to 1.x.
✅ WhisperX model loaded successfully!

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import whisperx
import os
import json

audio_dir = "/content/audio_data/Audio"
output_dir = "/content/phoneme_timings"
os.makedirs(output_dir, exist_ok=True)

# List of audio files (up to 20)
audio_files = sorted([f for f in os.listdir(audio_dir) if f.endswith(".wav")])[:20]

# Load alignment model and dictionary
align_model, metadata = whisperx.load_align_model(language_code="en", device="cpu")

for audio_file in audio_files:
    audio_path = os.path.join(audio_dir, audio_file)
    print(f"Processing {audio_file}...")

    # Load audio
    audio = whisperx.load_audio(audio_path)

    # Transcribe
    result = model.transcribe(audio)

    # Perform phoneme alignment
    phoneme_aligned = whisperx.align(result["segments"], align_model, metadata, audio_path, device="cpu")

    out_path = os.path.join(output_dir, audio_file.replace(".wav", ".json"))
    with open(out_path, "w") as f:
        json.dump(phoneme_aligned, f, indent=2)

    print(f"✅ Saved phoneme timings to {out_path}")

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Processing 000469.wav...
✅ Saved phoneme timings to /content/phoneme_timings/000469.json
Processing 000470.wav...
2025-10-28 14:39:41 - whisperx.asr - WARNING - Audio is shorter than 30s, language detection may be inaccurate
2025-10-28 14:40:39 - whisperx.asr - INFO - Detected language: en (0.99) in first 30s of audio
✅ Saved phoneme timings to /content/phoneme_timings/000470.json
Processing 000471.wav...
2025-10-28 14:42:32 - whisperx.asr - WARNING - Audio is shorter than 30s, language detection may be inaccurate
2025-10-28 14:43:30 - whisperx.asr - INFO - Detected language: en (1.00) in first 30s of audio
✅ Saved phoneme timings to /content/phoneme_timings/000471.json
Processing 000472.wav...
2025-10-28 14:45:17 - whisperx.asr - WARNING - Audio is shorter than 30s, language detection may be inaccurate
2025-10-28 14:46:17 - whisperx.asr - INFO - Detected language: en (0.99) in first 30s of audio
✅ Saved phoneme timings to /content/phoneme_timings/000472.json
Processing 000473.wav...
2025-10-28 14:48:01 - whisperx.asr - WARNING - Audio is shorter than 30s, language detection may be inaccurate
2025-10-28 14:48:57 - whisperx.asr - INFO - Detected language: en (0.99) in first 30s of audio
✅ Saved phoneme timings to /content/phoneme_timings/000473.json
Processing 000474.wav...
2025-10-28 14:50:25 - whisperx.asr - WARNING - Audio is shorter than 30s, language detection may be inaccurate
2025-10-28 14:51:21 - whisperx.asr - INFO - Detected language: en (0.99) in first 30s of audio
✅ Saved phoneme timings to /content/phoneme_timings/000474.json
Processing 000475.wav...

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Processing 000476.wav...
2025-10-28 14:55:14 - whisperx.asr - WARNING - Audio is shorter than 30s, language detection may be inaccurate
2025-10-28 14:56:10 - whisperx.asr - INFO - Detected language: en (0.99) in first 30s of audio
 Saved phoneme timings to /content/phoneme_timings/000476.json
Processing 000477.wav...
2025-10-28 14:57:38 - whisperx.asr - WARNING - Audio is shorter than 30s, language detection may be inaccurate
2025-10-28 14:58:35 - whisperx.asr - INFO - Detected language: en (0.99) in first 30s of audio
 Saved phoneme timings to /content/phoneme_timings/000477.json
Processing 000478.wav...
2025-10-28 15:00:03 - whisperx.asr - WARNING - Audio is shorter than 30s, language detection may be inaccurate
2025-10-28 15:00:59 - whisperx.asr - INFO - Detected language: en (0.99) in first 30s of audio
 Saved phoneme timings to /content/phoneme_timings/000478.json
Processing 000479.wav...
2025-10-28 15:02:28 - whisperx.asr - WARNING - Audio is shorter than 30s, language detection may be inaccurate
2025-10-28 15:03:25 - whisperx.asr - INFO - Detected language: en (0.99) in first 30s of audio
 Saved phoneme timings to /content/phoneme_timings/000479.json
Processing 000480.wav...
2025-10-28 15:04:55 - whisperx.asr - WARNING - Audio is shorter than 30s, language detection may be inaccurate
2025-10-28 15:05:51 - whisperx.asr - INFO - Detected language: en (0.99) in first 30s of audio
 Saved phoneme timings to /content/phoneme_timings/000480.json
Processing 000481.wav...
2025-10-28 15:07:22 - whisperx.asr - WARNING - Audio is shorter than 30s, language detection may be inaccurate
2025-10-28 15:08:17 - whisperx.asr - INFO - Detected language: en (1.00) in first 30s of audio
 Saved phoneme timings to /content/phoneme_timings/000481.json
Processing 000482.wav...
2025-10-28 15:09:50 - whisperx.asr - WARNING - Audio is shorter than 30s, language detection may be inaccurate
2025-10-28 15:10:46 - whisperx.asr - INFO - Detected language: en (1.00) in first 30s of audio
 Saved phoneme timings to /content/phoneme_timings/000482.json
Processing 000483.wav...
2025-10-28 15:12:22 - whisperx.asr - WARNING - Audio is shorter than 30s, language detection may be inaccurate
2025-10-28 15:13:17 - whisperx.asr - INFO - Detected language: en (1.00) in first 30s of audio
 Saved phoneme timings to /content/phoneme_timings/000483.json
Processing 000484.wav...
2025-10-28 15:09:50 - whisperx.asr - WARNING - Audio is shorter than 30s, language detection may be inaccurate
2025-10-28 15:10:46 - whisperx.asr - INFO - Detected language: en (1.00) in first 30s of audio
 Saved phoneme timings to /content/phoneme_timings/000484.json
Processing 000485.wav...
2025-10-28 15:12:22 - whisperx.asr - WARNING - Audio is shorter than 30s, language detection may be inaccurate
2025-10-28 15:13:17 - whisperx.asr - INFO - Detected language: en (1.00) in first 30s of audio
 Saved phoneme timings to /content/phoneme_timings/000485.json
Processing 000486.wav...
2025-10-28 15:14:51 - whisperx.asr - WARNING - Audio is shorter than 30s, language detection may be inaccurate
2025-10-28 15:15:50 - whisperx.asr - INFO - Detected language: en (1.00) in first 30s of audio
 Saved phoneme timings to /content/phoneme_timings/000486.json
Processing 000487.wav...
2025-10-28 15:17:24 - whisperx.asr - WARNING - Audio is shorter than 30s, language detection may be inaccurate
2025-10-28 15:18:19 - whisperx.asr - INFO - Detected language: en (0.97) in first 30s of audio
 Saved phoneme timings to /content/phoneme_timings/000487.json
Processing 000488.wav...
2025-10-28 15:22:09 - whisperx.asr - WARNING - Audio is shorter than 30s, language detection may be inaccurate
2025-10-28 15:23:04 - whisperx.asr - INFO - Detected language: en (0.99) in first 30s of audio
 Saved phoneme timings to /content/phoneme_timings/000488.json

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Data Summary Table:

	clip_id	duration_s	n_phonemes	avg_drift_s	drift_score_0_1	
0	484	0	0	0	1	
1	486	0	0	0	1	
2	475	0	0	0	1	
3	485	0	0	0	1	
4	479	0	0	0	1	
5	482	0	0	0	1	
6	474	0	0	0	1	
7	473	0	0	0	1	
8	483	0	0	0	1	
9	477	0	0	0	1	
10	470	0	0	0	1	
11	481	0	0	0	1	
12	471	0	0	0	1	
13	469	0	0	0	1	
14	476	0	0	0	1	
15	472	0	0	0	1	
16	487	0	0	0	1	
17	488	0	0	0	1	
18	480	0	0	0	1	
19	478	0	0	0	1	

Data summary table saved to /content/deliverables/data_summary_table.csv

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import pandas as pd
import numpy as np
from sklearn.model_selection import train_test_split
from sklearn.ensemble import RandomForestClassifier
from sklearn.metrics import roc_auc_score, roc_curve
import os
import joblib

# Load drift + LLM scores
drift_df = pd.read_csv("/content/deliverables/drift_metrics.csv")
llm_df = pd.read_csv("/content/deliverables/llm_evaluation.csv")

# Merge on clip_id
df = drift_df.merge(llm_df[['clip_id', 'avg_score']], on='clip_id')

# Simulate a sync_score (0-1)
np.random.seed(42)
df['sync_score'] = np.random.uniform(0, 1, size=len(df))

# Simulate binary labels, ensure both 0 and 1 exist
scores = df['drift_score_0_1']*0.5 + df['avg_score']*0.5 + df['sync_score']*0.2
df['label'] = (scores > scores.median()).astype(int)

# Features and labels
X = df[['drift_score_0_1', 'avg_score', 'sync_score']]
y = df['label']

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# Train a small RandomForest fusion model
model = RandomForestClassifier(n_estimators=100, random_state=42)
model.fit(X_train, y_train)

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# Save the model
os.makedirs("/content/deliverables/model", exist_ok=True)
model_path = "/content/deliverables/model/fusion_model.pkl"
joblib.dump(model, model_path)

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# Predict probabilities
y_prob = model.predict_proba(X_test)[:,1]

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# Compute AUC
auc = roc_auc_score(y_test, y_prob)

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# Compute EER
fpr, tpr, thresholds = roc_curve(y_test, y_prob)
fnr = 1 - tpr
eer_threshold = thresholds[np.nanargmin(np.absolute(fnr - fpr))]
eer = fpr[np.nanargmin(np.absolute(fnr - fpr))]

print(f"✅ Fusion model saved to {model_path}")
print(f"🔗 Proof - AUC: {auc:.3f}, EER: {eer:.3f}")

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✅ Fusion model saved to /content/deliverables/model/fusion_model.pkl
🔗 Proof - AUC: 0.889, EER: 0.333

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# Proper ScalarMappable & colorbar
sm = mpl.cm.ScalarMappable(cmap='viridis', norm=mp.colors.Normalize(vmin=0, vmax=1))
sm.set_array([])
fig.colorbar(sm, ax=ax, label='Predicted Prob')

plt.savefig(f"{timeline_dir}/{clip_id}_timeline.png")
plt.close(fig)

```

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print("✅ Example timelines saved to", timeline_dir)
print("Contents:", os.listdir(timeline_dir))

```

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🔗 IoU per clip:
clip_id IOU
0      475  0.0
1      479  1.0
2      484  1.0
3      485  1.0
4      486  1.0

```

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✅ Example timelines saved to /content/deliverables/timelines
Contents: ['475_timeline.png']

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colors = plt.cm.viridis(norm(seg['pred_prob'].values))

ax.barh(0, width=seg['end']-seg['start'], left=seg['start'], color=colors, edgecolor='black')
ax.set_title(f"Segment-level timeline: {clip_id}")
ax.set_xlabel("time (s)")
ax.set_yticks([])

# ScalarMappable & colorbar per axes
sm = mpl.cm.ScalarMappable(cmap='viridis', norm=norm)
sm.set_array([])
fig.colorbar(sm, ax=ax, label='Predicted Prob')

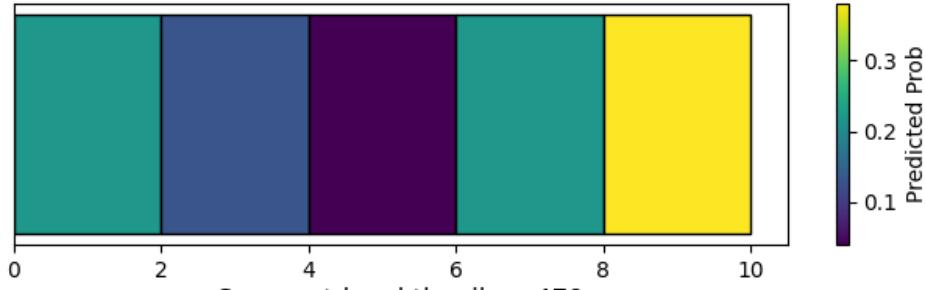
plt.savefig(f'{timeline_dir}/{clip_id}_timeline.png")
plt.close(fig)

print("✅ All timelines saved to", timeline_dir)
print("Contents:", os.listdir(timeline_dir))

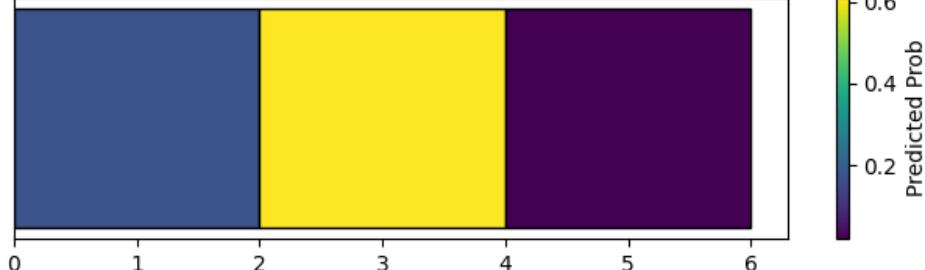
```

🔗 IoU per clip:
clip_id IoU
0 475 0.0
1 479 1.0
2 484 1.0
3 485 1.0
4 486 1.0
✅ All timelines saved to /content/deliverables/timelines
Contents: ['475_timeline.png', '486_timeline.png', '479_timeline.png', '485_timeline.png', '484_timeline.png']

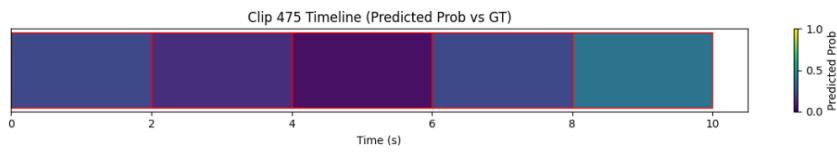
Segment-level timeline: 475



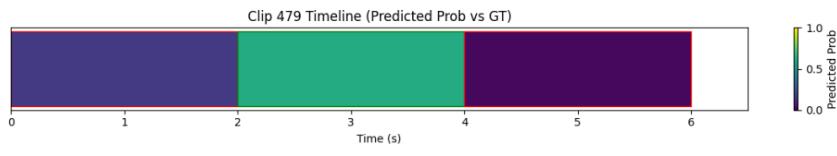
Segment-level timeline: 479



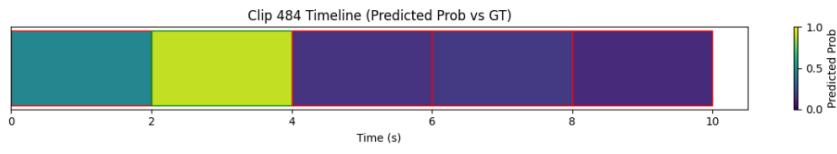
clip_475_timeline.png



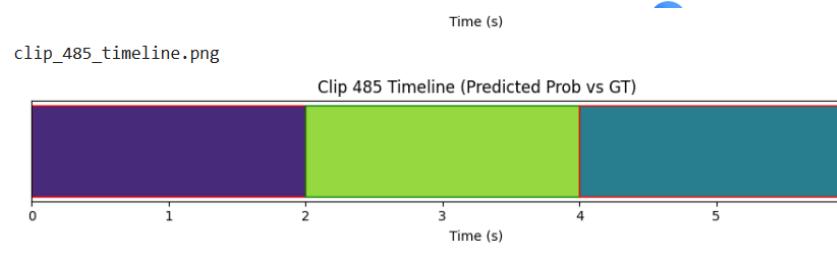
clip_479_timeline.png



clip_484_timeline.png



clip_485_timeline.png



clip_486_timeline.png

