OBIVox System Architecture & Development Plan

System Overview

OBIVox is a bidirectional knowledge interface implementing real-time STT (Speech-to-Text) and TTS (Text-to-Speech) with phonetic structure optimization.

Core Architecture Components

1.	Codec	Conversion	Pipeline	Architecture
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```
graph TB
  subgraph Input Layer
    A1[Audio Input WAV/MP4/M4A]
    A2[FFmpeg Decoder]
    A3[Audio Normalization]
  end
  subgraph NLM Framework
    B1[Bottom-Up Processing]
    B2[Top-Down Processing]
    B3[Phonetic Analyzer]
    B4[Accent/Tone Detector]
  end
  subgraph Core Processing
    C1[ASR Engine < br/> Whisper/VOSK]
    C2[TTS Engine < br/>Coqui/VITS]
    C3[Confidence Validator]
    C4[Thread Pool Orchestrator]
  end
  subgraph ELF Linking Layer
    D1[OBIELF Loader]
    D2[Symbol Resolution]
    D3[Dynamic Binding]
    D4[Memory Mapping]
  end
  subgraph Output Layer
    E1[Audio Synthesis]
    E2[Text Generation]
    E3[Format Encoder]
  end
  A1 --> A2 --> A3 --> B1
  B1 --> B3 --> B4 --> C1
  B2 --> C2
  C1 --> C3 --> E2
  C2 --> C3 --> E1
  C4 --> D1 --> D2 --> D3 --> D4
```

Implementation Specification

Phase 1: Core Rust/C Implementation

```
// obivox_core.rs - Main system structure
use std::sync::{Arc, Mutex};
use std::collections::HashMap;
pub struct OBIVoxEngine {
  // Core components
  ffmpeg_handler: Arc<FFmpegProcessor>,
  nlm_framework: NLMFramework,
  codec_registry: HashMap<String, Box<dyn CodecHandler>>,
  elf_linker: OBIELFLinker,
pub struct NLMFramework {
  bottom_up: BottomUpProcessor,
  top_down: TopDownProcessor,
  phonetic_analyzer: PhoneticAnalyzer,
pub struct PhoneticAnalyzer {
  tone_detector: ToneDetector,
  pitch_analyzer: PitchAnalyzer,
  accent_classifier: AccentClassifier,
  // High-level features, not voice-dependent
  prosody_extractor: ProsodyExtractor,
pub trait CodecHandler: Send + Sync {
  fn encode(&self, input: &[u8]) -> Result<Vec<u8>, CodecError>;
  fn decode(&self, input: &[u8]) -> Result<Vec<u8>, CodecError>;
  fn get_confidence(&self) -> f32;
```

Phase 2: FFmpeg Integration Layer

```
// ffmpeg_bridge.c - FFmpeg C bindings
#include libavformat/avformat.h>
#include libavcodec/avcodec.h>
#include libswresample/swresample.h>
typedef struct {
  AVFormatContext *fmt_ctx;
  AVCodecContext *codec_ctx;
  SwrContext *swr_ctx;
  int sample_rate;
  int channels;
} OBIVoxFFmpegContext;
// Initialize FFmpeg pipeline
int obivox_ffmpeg_init(OBIVoxFFmpegContext **ctx, const char *input_path) {
  *ctx = malloc(sizeof(OBIVoxFFmpegContext));
  // Open input file
  if (avformat_open_input(&(*ctx)->fmt_ctx, input_path, NULL, NULL) < 0) {
     return OBIVOX_ERR_OPEN_FAILED;
  // Find stream info
  if (avformat_find_stream_info((*ctx)->fmt_ctx, NULL) < 0) {</pre>
     return OBIVOX_ERR_NO_STREAM;
  // Setup resampler for 16kHz mono
  (*ctx)->swr_ctx = swr_alloc_set_opts(NULL,
    AV_CH_LAYOUT_MONO, AV_SAMPLE_FMT_S16, 16000,
    (*ctx)->codec_ctx->channel_layout,
    (*ctx)->codec_ctx->sample_fmt,
     (*ctx)->codec_ctx->sample_rate, 0, NULL);
  return OBIVOX_SUCCESS;
```

Phase 3: System SDK & Plugin Architecture

rust

```
// plugin_system.rs - Plugin SDK interface
pub trait OBIVoxPlugin {
  fn init(&mut self) -> Result<(), PluginError>;
  fn process(&mut self, input: AudioBuffer) -> Result<AudioBuffer, PluginError>;
  fn get_metadata(&self) -> PluginMetadata;
// Platform-specific builders
pub struct PlatformBuilder {
  target_os: TargetOS,
impl PlatformBuilder {
  pub fn build_unix(&self) -> Result<Package, BuildError> {
    // Generate .deb/.rpm packages
     self.generate_systemd_service()?;
     self.compile_shared_libraries()?;
     Ok(Package::Unix(UnixPackage::new()))
  pub fn build_macos(&self) -> Result<Package, BuildError> {
    // Generate .dmg with code signing
     self.generate_launchd_plist()?;
     self.compile_frameworks()?;
     Ok(Package::MacOS(DMGPackage::new()))
  pub fn build_windows(&self) -> Result<Package, BuildError> {
     // Generate .msi installer
     self.generate_service_manifest()?;
     self.compile_dlls()?;
     Ok(Package::Windows(MSIPackage::new()))
```

Phase 4: ELF Linking Layer for Seamless Integration

```
C C
```

```
// obielf_linker.c - Custom ELF linking for codec modules
#include <elf.h>
#include <dlfcn.h>
typedef struct {
  Elf64_Ehdr header;
  void* codec_symbols;
  void* memory_map;
} OBIELFContext;
// Dynamic codec loading with OBIELF format
int obielf_load_codec(const char* codec_path, OBIELFContext** ctx) {
  *ctx = malloc(sizeof(OBIELFContext));
  // Load ELF with custom sections
  void* handle = dlopen(codec_path, RTLD_LAZY | RTLD_LOCAL);
  // Resolve codec symbols
  (*ctx)->codec_symbols = dlsym(handle, "obivox_codec_table");
  // Map shared memory for zero-copy audio passing
  (*ctx)->memory_map = mmap(NULL, AUDIO_BUFFER_SIZE,
                  PROT_READ | PROT_WRITE,
                  MAP_SHARED | MAP_ANONYMOUS, -1, 0);
  return 0;
```

Platform-Specific SDK Integration

Linux Package (.deb/.rpm)

```
bash

# debian/control

Package: obivox

Version: 1.0.0

Architecture: amd64

Depends: libffmpeg4, libasound2

Description: Bidirectional audio-text processing system
```

macOS Framework (.dmg)

xml

```
<!-- Info.plist -->
<key>CFBundleIdentifier</key>
<string>org.obinexus.obivox</string>
<key>LSBackgroundOnly</key>
<true/>
```

Windows Service (.msi)

Build System Configuration

cmake		

```
# CMakeLists.txt
cmake_minimum_required(VERSION 3.16)
project(OBIVox LANGUAGES C CXX)
# Find dependencies
find_package(PkgConfig REQUIRED)
pkg_check_modules(FFMPEG REQUIRED
  libavformat libavcodec libswresample)
# Core library
add_library(obivox_core SHARED
  src/core/engine.c
  src/codecs/whisper_wrapper.c
  src/codecs/coqui_wrapper.c
  src/nlm/phonetic_analyzer.c
  src/elf/obielf_linker.c
# Platform-specific targets
if(UNIX AND NOT APPLE)
  add_custom_target(package_deb
    COMMAND ${CMAKE_COMMAND} -E cmake_echo_color
         --cyan "Building .deb package"
    COMMAND dpkg-buildpackage -b -uc
  )
elseif(APPLE)
  add_custom_target(package_dmg
    COMMAND ${CMAKE_COMMAND} -E cmake_echo_color
         --cyan "Building .dmg package"
    COMMAND hdiutil create -volname OBIVox
         -srcfolder ${CMAKE_BINARY_DIR}/OBIVox.app
         -ov OBIVox.dmg
elseif(WIN32)
  add_custom_target(package_msi
    COMMAND ${CMAKE_COMMAND} -E cmake_echo_color
         --cyan "Building .msi installer"
    COMMAND candle.exe obivox.wxs
    COMMAND light.exe -out OBIVox.msi obivox.wixobj
endif()
```

Command-Line Interface

OBIVox CLI Usage obivox --input lecture.mp4 --mode stt --output transcript.txt obivox --input book.txt --mode tts --voice en_US --output audiobook.m4a obivox --pipeline config.yaml --batch lectures/*.mp4 # Plugin management obivox-plugin install whisper-large obivox-plugin list obivox-plugin configure coqui-tts --voice jenny Development Roadmap Milestone 1: Core Implementation (Weeks 1-4) FFmpeg integration layer in C Basic STT/TTS pipeline in Rust Thread-safe task orchestration Confidence validation system Milestone 2: NLM Framework (Weeks 5-8)

Milestone 2: NLM Framework (Weeks 5-8) Bottom-up phonetic processing Top-down semantic analysis Tone/pitch/accent detection Prosody extraction Milestone 3: Platform SDKs (Weeks 9-12) Linux package generation (.deb/.rpm) macOS framework and .dmg Windows service and .msi Plugin system architecture Milestone 4: OBIELF Integration (Weeks 13-16) Custom ELF sections for codecs Dynamic symbol resolution Memory-mapped audio buffers

Testing Strategy

Zero-copy audio pipeline

rust

```
#[cfg(test)]
mod tests {
    use super::*;

#[test]
fn test_bidirectional_conversion() {
    let input = "Hello, OBIVox";
    let audio = text_to_speech(input).unwrap();
    let output = speech_to_text(audio).unwrap();
    assert_eq!(input.to_lowercase(), output.to_lowercase());
}

#[test]
fn test_confidence_threshold() {
    let low_quality_audio = generate_noisy_audio();
    let result = process_with_confidence(low_quality_audio);
    assert!(result.confidence < 0.6);
    assert!(result.requires_confirmation);
}
</pre>
```

Integration with OBINexus Ecosystem

The OBIVox system integrates seamlessly with:

- OBIELF: Custom ELF format for codec modules
- ObiCall: Zero-trust module invocation
- Self-Healing Architecture: Fault-tolerant audio processing
- Rift/GOSI toolchain: Compilation and linking

Performance Targets

- STT Latency: < 500ms for 10-second audio
- TTS Latency: < 200ms for 100-word text
- Memory Usage: < 512MB resident
- Thread Pool: 4-8 concurrent workers
- Confidence Threshold: > 0.85 for production