GB28181 Restreamer - Complete Documentation

Overview

GB28181 Restreamer is a comprehensive video streaming solution that implements the GB28181 surveillance standard protocol. It enables streaming of video content from various sources (RTSP streams, video files, or live camera feeds) to GB28181-compatible platforms like WVP-Pro using SIP signaling and RTP streaming.

Key Features

- GB28181 Protocol Compliance: Full support for device registration, catalog queries, and media streaming
- Multiple Streaming Modes: File-based streaming, RTSP source streaming, and real-time frame processing
- Advanced Video Processing: Appsink/Appsrc pipeline for real-time frame manipulation
- · Time Series Recording: Recording management with historical video playback capabilities
- Flexible Configuration: JSON-based configuration with multiple preset options
- Robust Health Monitoring: Automatic stream recovery and health monitoring
- Multi-stream Support: Handle multiple concurrent video streams
- ARM Architecture Support: Optimized for ARM64 systems including Raspberry Pi

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System Requirements

Minimum Requirements

- Operating System: Ubuntu 20.04+, Debian 11+, or Raspberry Pi OS (64-bit)
- Architecture: x86_64 or ARM64 (aarch64)
- RAM: 2GB minimum, 4GB recommended
- Storage: 10GB free space minimum
- Network: Stable internet connection
- Python: 3.8 or higher

Required System Packages

```
# Core dependencies
sudo apt update
sudo apt install -y python3 python3-pip python3-dev
sudo apt install -y pkg-config libcairo2-dev gcc python3-dev libgirepository1.0-dev

# GStreamer dependencies
sudo apt install -y libgstreamer1.0-dev libgstreamer-plugins-base1.0-dev
sudo apt install -y libgstreamer-plugins-good1.0-dev libgstreamer-plugins-bad1.0-dev
sudo apt install -y gstreamer1.0-plugins-base gstreamer1.0-plugins-good
sudo apt install -y gstreamer1.0-plugins-bad gstreamer1.0-plugins-ugly
sudo apt install -y gstreamer1.0-libav gstreamer1.0-tools gstreamer1.0-x

# Additional video processing tools
sudo apt install -y ffmpeg v41-utils

# Development tools (if compiling from source)
sudo apt install -y build-essential cmake git
```

Installation

Quick Installation

- 1. Clone the Repository bash git clone https://github.com/your-org/gb28181-restreamer.git cd gb28181-restreamer
- 2. Install Python Dependencies bash pip3 install -r requirements.txt

- 3. Install PJSUA (See detailed instructions below)
- 4. Configure the Application bash cp config/config.json config/my_config.json # Edit config/my_config.json with your settings
- 5. Run the Application bash python3 src/main.py

PJSUA Setup

PJSUA is a critical component for SIP communication in GB28181. Follow these detailed instructions based on your system architecture.

For x86_64 Systems (Intel/AMD)

Option 1: Package Manager Installation (Recommended)

```
# Ubuntu/Debian
sudo apt update
sudo apt install -y pjsip-tools

# Verify installation
pjsua --help
```

Option 2: Build from Source

```
# Install dependencies
sudo apt install -y build-essential libasound2-dev

# Download and compile PJSIP
wget https://github.com/pjsip/pjproject/archive/2.14.1.tar.gz
tar -xzf 2.14.1.tar.gz
cd pjproject-2.14.1

# Configure and build
./configure --enable-shared --disable-video --disable-opencore-amr
make dep && make
sudo make install
sudo ldconfig

# Verify installation
pjsua --help
```

For ARM64 Systems (Raspberry Pi, ARM servers)

Installing PJSUA on ARM64

ARM64 systems often require building PJSUA from source due to limited package availability.

Step 1: System Preparation

```
# Update system
sudo apt update && sudo apt upgrade -y

# Install build dependencies
sudo apt install -y build-essential
sudo apt install -y libasound2-dev libssl-dev
sudo apt install -y autoconf automake libtool
sudo apt install -y pkg-config cmake

# For Raspberry Pi specifically
sudo apt install -y libraspberrypi-dev
```

```
# Create build directory
mkdir -p ~/build
cd ~/build
# Download PJSIP source (use stable version)
wget https://github.com/pjsip/pjproject/archive/refs/tags/2.14.1.tar.gz
tar -xzf 2.14.1.tar.gz
cd pjproject-2.14.1
# Configure for ARM64
export CFLAGS="-02"
export CXXFLAGS="-02"
# Configure with ARM optimizations
./configure \
    --host=aarch64-linux-gnu \
    --enable-shared \
    --disable-video ∖
    --disable-opencore-amr \
    --disable-silk \
    --disable-opus \
    --disable-speex-codec \
    --disable-ilbc-codec \
    --disable-l16-codec \
    --disable-gsm-codec \
    --disable-g722-codec \
    --disable-g7221-codec \
    --disable-speex-aec \
    --enable-epoll \
    --prefix=/usr/local
# Build (this may take 20-30 minutes on Raspberry Pi)
make dep
make -j$(nproc)
sudo make install
# Update library path
echo '/usr/local/lib' | sudo tee /etc/ld.so.conf.d/pjsip.conf
sudo ldconfig
# Create symlink for easier access
sudo ln -sf /usr/local/bin/pjsua /usr/bin/pjsua
```

Step 3: Verify Installation

```
# Test PJSUA installation
pjsua --help

# Check library linking
ldd /usr/local/bin/pjsua
```

Step 4: Configure for GB28181

Using Docker (All Architectures)

If compilation fails, you can use Docker:

```
# Create a PJSUA container
docker run -it --rm \
  -v $(pwd):/workspace \
  --name pjsua-builder ∖
 ubuntu:22.04 bash
# Inside container
apt update && apt install -y build-essential wget
wget https://github.com/pjsip/pjproject/archive/2.14.1.tar.gz
tar -xzf 2.14.1.tar.gz
cd pjproject-2.14.1
./configure --enable-shared --disable-video
make dep && make
cp pjsip-apps/bin/pjsua-* /workspace/pjsua
exit
# Make executable
chmod +x pjsua
sudo mv pjsua /usr/local/bin/
```

Pre-built ARM64 Binaries

```
# Download pre-built binary (if available)
wget https://github.com/your-org/pjsua-arm64/releases/download/v2.14.1/pjsua-arm64
chmod +x pjsua-arm64
sudo mv pjsua-arm64 /usr/local/bin/pjsua
```

PJSUA Configuration for GB28181

Create a PJSUA configuration file for your GB28181 setup:

```
# Create PJSUA config directory
mkdir -p ~/.pjsua
# Create configuration file
cat > ~/.pjsua/gb28181.conf << EOF</pre>
--id sip:YOUR_DEVICE_ID@YOUR_SIP_SERVER:5060
--registrar sip:YOUR_SIP_SERVER:5060
--username YOUR_DEVICE_ID
--password YOUR_PASSWORD
--realm *
--local-port 5080
--auto-answer 200
--log-level 4
--app-log-level 4
--duration 0
--null-audio
--no-vad
EOF
```

Configuration

Basic Configuration

The main configuration file is <code>config/config.json</code> . Here's a complete example:

```
"sip": {
    "device_id": "81000000465001000001",
   "username": "81000000465001000001",
    "password": "admin123",
    "server": "your-gb28181-server.com",
    "port": 5060,
    "local_port": 5080,
    "transport": "udp"
  "local_sip": {
    "enabled": false,
    "port": 5060,
    "transport": "udp"
  },
  "stream_directory": "/path/to/recordings",
  "rtsp_sources": [
   {
      "id": 1,
      "name": "Camera 1",
      "url": "rtsp://192.168.1.100:554/stream1",
      "username": "admin",
      "password": "password123"
   }
  ],
  "srtp": {
    "key": "313233343536373839303132333435363132333435363738393031323334"
  },
  "logging": {
    "level": "INFO",
    "file": "./logs/gb28181-restreamer.log",
    "console": true
 },
  "pipeline": {
    "format": "RGB",
    "width": 1920,
    "height": 1080,
    "framerate": 25,
   "buffer_size": 33554432,
    "queue_size": 3000,
    "sync": false,
    "async": false
}
```

Configuration Parameters

SIP Configuration

- device_id: Your unique GB28181 device identifier (20 digits)
- username: SIP username (usually same as device_id)
- password: SIP password for authentication
- server: GB28181 platform server address
- port: SIP server port (usually 5060)
- local_port: Local SIP port (recommended: 5080)
- transport: Transport protocol (udp/tcp)

Streaming Configuration

- stream_directory: Directory for storing/reading video files
- rtsp_sources: Array of RTSP source configurations
- pipeline: GStreamer pipeline parameters

Encoding Presets

The system includes predefined encoding presets in <code>config/streaming_presets.json</code>:

```
"presets": {
    "high_quality": {
     "width": 1920,
      "height": 1080,
      "framerate": 25,
      "bitrate": 4000,
      "keyframe_interval": 25
    },
    "medium_quality": {
      "width": 1280,
      "height": 720,
      "framerate": 15,
      "bitrate": 2000,
      "keyframe_interval": 15
    },
    "low_bandwidth": {
      "width": 640,
     "height": 480,
      "framerate": 10,
      "bitrate": 500,
      "keyframe_interval": 10
    },
    "mobile_optimized": {
      "width": 480,
      "height": 320,
      "framerate": 10,
      "bitrate": 200,
      "keyframe_interval": 10
    }
  }
}
```

Usage

Basic Streaming

1. Start the Application

```
python3 src/main.py
```

2. File-based Streaming

```
from media_streamer import MediaStreamer

config = {...} # Your configuration
streamer = MediaStreamer(config)

# Start streaming a video file
success = streamer.start_stream(
    video_path="/path/to/video.mp4",
    dest_ip="192.168.1.100",
    dest_port=9000,
    ssrc="1234567890"
)
```

3. RTSP Source Streaming

 $Configure\ RTSP\ sources\ in\ your\ config\ file\ and\ they\ will\ be\ automatically\ available\ as\ GB28181\ channels.$

Advanced Features

Appsink/Appsrc Frame Processing

The system supports real-time frame processing using GStreamer's appsink and appsrc elements:

```
def custom_frame_processor(frame, timestamp, stream_info):
    Custom frame processing function
    Args:
       frame (numpy.ndarray): Video frame in RGB format
       timestamp (float): Frame timestamp
       stream info (dict): Stream information
    Returns:
       tuple: (processed_frame, timestamp)
    # Example: Convert to grayscale
    gray = cv2.cvtColor(frame, cv2.COLOR_RGB2GRAY)
    processed = cv2.cvtColor(gray, cv2.COLOR_GRAY2RGB)
    return processed, timestamp
# Start stream with processing
streamer.start_stream_with_processing(
   video_path="/path/to/video.mp4",
   dest_ip="192.168.1.100",
   dest_port=9000,
    frame_processor_callback=custom_frame_processor
)
```

Available Frame Processors

- Grayscale Conversion python def grayscale_processor(frame, timestamp, stream_info): gray = cv2.cvtColor(frame, cv2.COLOR_RGB2GRAY) return cv2.cvtColor(gray, cv2.COLOR_GRAY2RGB), timestamp
- 2. Edge Detection python def edge_detection_processor(frame, timestamp, stream_info): gray = cv2.cvtColor(frame, cv2.COLOR_RGB2GRAY) edges = cv2.Canny(gray, 100, 200) return cv2.cvtColor(edges, cv2.COLOR_GRAY2RGB), timestamp
- 3. **Text Overlay** python def text_overlay_processor(frame, timestamp, stream_info): frame_copy = frame.copy() timestamp_str = time.strftime("%Y-%m-%d %H:%M:%S", time.localtime(timestamp)) cv2.putText(frame_copy, f"Time: {timestamp_str}", (20, 40), cv2.FONT_HERSHEY_SIMPLEX, 1, (0, 255, 0), 2) return frame_copy, timestamp

Testing Appsink/Appsrc Mode

Use the provided test script:

```
# Test basic frame processing
python3 test_appsink_appsrc.py --video /path/to/test.mp4 --processing grayscale

# Test with custom destination
python3 test_appsink_appsrc.py --video /path/to/test.mp4 --dest-ip 192.168.1.100 --dest-port 9000

# Test edge detection
python3 test_appsink_appsrc.py --video /path/to/test.mp4 --processing edge

# Test text overlay
python3 test_appsink_appsrc.py --video /path/to/test.mp4 --processing text
```

Time Series Recording and Playback

The system supports GB28181 RecordInfo queries for historical video access:

```
from recording_manager import RecordingManager
# Initialize recording manager
config = \{...\}
recorder = RecordingManager(config)
# Query recordings by time range
recordings = recorder.query_recordings(
    device_id="81000000465001000001",
    start_time="20240101T000000Z",
    end_time="20240131T235959Z",
    recording_type="all"
)
# Start playback of a specific recording
for recording in recordings:
    streamer.start_recording_playback(
        recording_info=recording,
        dest_ip="192.168.1.100",
       dest_port=9000,
       start_timestamp=recording["timestamp"],
        end_timestamp=recording["timestamp"] + recording["duration"]
    )
```

Directory Structure for Recordings

Organize your recordings for optimal time series query performance:

Troubleshooting

Common Issues and Solutions

1. PJSUA Installation Issues

Problem: Cannot install pjsip-tools / pjsua

Solution for ARM64:

```
# If package installation fails, build from source
sudo apt install -y build-essential libasound2-dev
wget https://github.com/pjsip/pjproject/archive/2.14.1.tar.gz
tar -xzf 2.14.1.tar.gz
cd pjproject-2.14.1
./configure --enable-shared --disable-video
make dep && make
sudo make install
sudo ldconfig
```

Solution for older Ubuntu/Debian:

```
# Add universe repository
sudo add-apt-repository universe
sudo apt update
sudo apt install -y pjsip-tools
```

2. GStreamer Issues

Problem: Cannot read rtsp stream, Gstreamer error: Internal data stream error

Solutions:

- 1. Install additional GStreamer plugins: bash sudo apt install -y gstreamer1.0-plugins-bad gstreamer1.0-plugins-ugly sudo apt install -y gstreamer1.0-libav
- 2. Check RTSP source compatibility: bash # Test RTSP stream directly gst-launch-1.0 rtspsrc location="rtsp://your-camera-ip:554/stream" ! decodebin ! videoconvert ! autovideosink
- 3. Update pipeline configuration: json { "pipeline": { "format": "RGB", "width": 640, "height": 480, "framerate": 15, "buffer_size": 16777216, "sync": false } }

3. ARM Architecture Issues

Problem: RTSP server doesn't work on ARM (rtsp-simple-server is amd64)

Solution - Use MediaMTX (ARM64 compatible):

```
# Download ARM64 version of MediaMTX (successor to rtsp-simple-server)
wget https://github.com/bluenviron/mediamtx/releases/download/v1.5.0/mediamtx_v1.5.0_linux_arm64v8.tar.gz
tar -xzf mediamtx_v1.5.0_linux_arm64v8.tar.gz
chmod +x mediamtx
# Create configuration
cat > mediamtx.yml << EOF
rtspAddress: :8554
rtmpAddress: :1935
webrtcAddress: :8889
api: yes
apiAddress: :9997
paths:
 all:
    source: publisher
FOF
# Run MediaMTX
./mediamtx mediamtx.vml
```

Alternative - Use GStreamer RTSP Server:

```
# Install GStreamer RTSP server
sudo apt install -y libgstrtspserver-1.0-dev

# Use python gst-rtsp-server for ARM
pip3 install gst-rtsp-server
```

4. Registration Issues

Problem: WVP-pro device registration issues

Solutions:

- 1. Check SIP configuration: bash # Test SIP registration manually pjsua --id "sip:YOUR_DEVICE_ID@YOUR_SERVER:5060" \ --registrar "sip:YOUR_SERVER:5060" \ --username "YOUR_DEVICE_ID" \ --password "YOUR_PASSWORD" \ --duration 30
- 2. Verify network connectivity: "bash # Check if SIP server is reachable telnet YOUR_SERVER 5060

```
# Check UDP connectivity
nc -u YOUR_SERVER 5060
```

3. Check device ID format: - Must be exactly 20 digits - First 8 digits: Administrative region code - Next 2 digits: Device type (01 for camera) - Last 10 digits: Device serial number

5. Video Catalog Issues

Problem: Videos not showing in WVP catalog

Solutions:

- 1. Verify catalog response: bash # Check if catalog XML is generated ls -la catalog_response.xml cat catalog_response.xml
- 2. Check channel configuration: python # Ensure channels are properly configured { "rtsp_sources": [{ "id": 1, "name": "Camera 1",
 "url": "rtsp://192.168.1.100:554/stream1" }] }

6. Performance Issues on ARM

Problem: High CPU usage or poor performance on ARM devices

Solutions:

- 1. Optimize encoding settings: json { "pipeline": { "width": 640, "height": 480, "framerate": 15, "bitrate": 1000 } }
- 2. Enable hardware acceleration (Raspberry Pi): "bash # Enable GPU memory split sudo raspi-config # Advanced Options -> Memory Split -> 128

```
# Use hardware encoder
export GST_DEBUG=3
gst-inspect-1.0 | grep omx
```

3. Use lower quality presets: python encoder_params = { "width": 480, "height": 320, "framerate": 10, "bitrate": 200 }

Debugging and Logging

Enable Debug Logging

```
# In your configuration
{
    "logging": {
        "level": "DEBUG",
        "file": "./logs/debug.log",
        "console": true
    }
}
```

GStreamer Debug

```
# Set GStreamer debug level
export GST_DEBUG=3

# Debug specific elements
export GST_DEBUG=rtspsrc:5,rtpbin:5

# Run with debug
python3 src/main.py
```

PJSUA Debug

```
# Run with verbose logging
pjsua --log-level 5 --app-log-level 5 --log-file pjsua_debug.log
```

Network Debugging

```
# Monitor SIP traffic
sudo tcpdump -i any -n port 5060 -A

# Monitor RTP traffic
sudo tcpdump -i any -n portrange 10000-20000

# Check port usage
sudo netstat -tulpn | grep python3
```

API Reference

MediaStreamer Class

Methods

start_stream(video_path, dest_ip, dest_port, ssrc=None, encoder_params=None)

Start streaming a video file.

Parameters:

- video_path (str): Path to video file
- dest_ip (str): Destination IP address
- dest_port (int): Destination port
- ssrc (str, optional): RTP SSRC identifier
- encoder_params (dict, optional): Encoding parameters

Returns: bool - Success status

start_stream_with_processing(video_path, dest_ip, dest_port, frame_processor_callback=None, ssrc=None, encoder_params=None)

Start streaming with real-time frame processing.

Parameters:

- frame_processor_callback (function): Frame processing function
- Other parameters same as start_stream

Returns: bool - Success status

stop_stream(stream_id=None)

Stop a specific stream or all streams.

Parameters:

• stream_id (str, optional): Stream identifier

get_stream_status(stream_id=None)

Get status of streams.

Returns: dict - Stream status information

RecordingManager Class

Methods

query_recordings(device_id, start_time, end_time, recording_type=None, secrecy=None, max_results=100)

Query recordings by time range.

Parameters:

- device_id (str): Device identifier
- start_time (str): Start time (ISO format or GB28181 format)
- end_time (str): End time
- recording_type (str, optional): Recording type filter
- secrecy (str, optional): Secrecy level filter
- max_results (int): Maximum results to return

Returns: list - List of recording metadata

get_recording_stream_uri(recording_id)

Get streaming URI for a recording.

Parameters:

• recording_id (str): Recording identifier

Returns: str - Streaming URI

Performance Optimization

For ARM Devices

- 1. Use appropriate quality settings: json { "pipeline": { "width": 640, "height": 480, "framerate": 15, "bitrate": 1000 } }
- 2. Enable hardware acceleration (when available): bash # Check for hardware encoders gst-inspect-1.0 | grep -i "h264\|h265"
- 3. Optimize buffer sizes: json { "pipeline": { "buffer_size": 16777216, "queue_size": 1000 } }

Memory Management

- Monitor memory usage: htop or free -h
- Limit concurrent streams on low-memory devices
- Use appropriate buffer sizes for your hardware

Network Optimization

- Use UDP transport for better performance
- Configure appropriate RTP port ranges
- · Monitor network bandwidth usage

Support and Contributing

Getting Help

- 1. Check the troubleshooting section above
- 2. Review log files in ./logs/
- 3. Enable debug logging for detailed diagnostics
- 4. Check GStreamer installation: gst-inspect-1.0 --version

Contributing

- 1. Fork the repository
- 2. Create a feature branch
- 3. Make your changes
- 4. Test on both x86_64 and ARM64 architectures
- 5. Submit a pull request

Reporting Issues

When reporting issues, please include:

- System architecture (uname -a)
- Python version (python3 --version)
- GStreamer version (gst-inspect-1.0 --version)
- PJSUA version (pjsua --version)
- Complete error logs
- Configuration file (with sensitive data removed)

License

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Last Updated: May 2025 Version: 2.0.0 Compatible Architectures: x86_64, ARM64