# **Resiprocate SIP Stack**



### Contents

- Resiprocate Architecture
- Using Resiprocate Stack
- Using Resiprocate DUM
- Resiprocate Code Overview



# Architecture Overview of Resiprocate SIP Stack

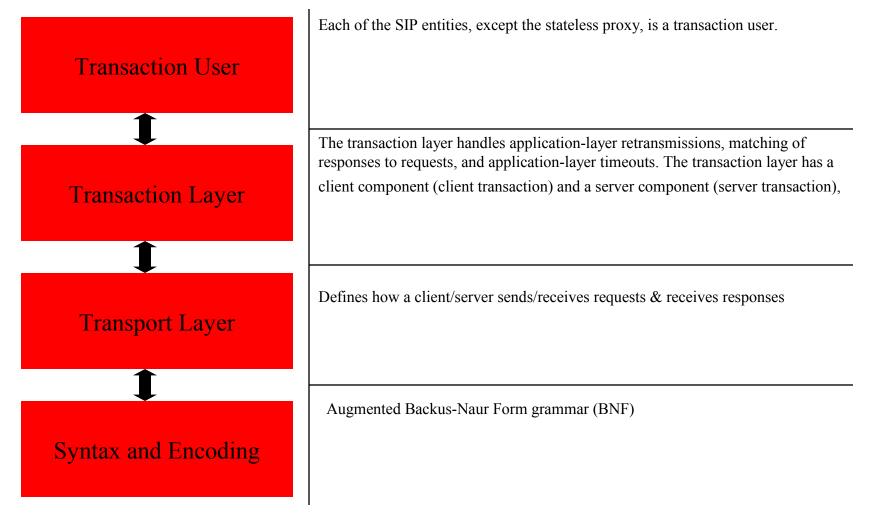


#### **ReSIProcate Stack**

- ReSIProcate is an object oriented SIP interface and stack implemented in C++. The ReSIProcate approach emphasizes consistency, type safety, and ease of use.
- The key goals are:
  - 3261 compliance
  - easy to program with
  - efficient (> 1000 transactions per second) (Platform Dependent)
  - excellent security implementation including TLS and S/MIME
  - Win32 and Linux/Unix support
  - usable by proxies, user agents and b2buas
  - object oriented interface

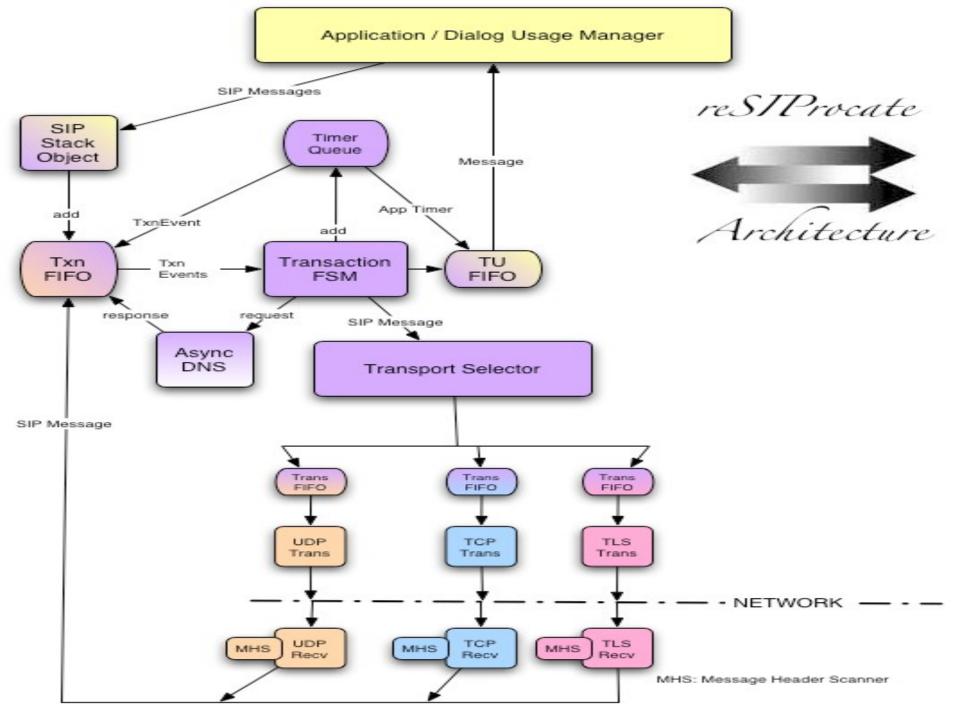


# SIP Stack Architecture (RFC 3261)



**Resiprocate Overview** 

#### www.resiprocate.org





# **Resiprocate Components**

#### • Entities

- Transaction User
- Sip Stack Objects
- Transaction FSM
- Transport Selector
- UDP/TCP/TLS Transport
- Message Header Scanner (MHS)
- Async DNS Utility

#### • FIFO

- TU FIFO
- Txn FIFO
- Trans FIFO
- Timer Queue



#### Transaction User

- To use the SIP Stack Transaction User has to be implemented.
- Dialog User Manager (DUM) is a Transaction User.
- DUM provides User Agent Functionality including handling of INVITE and SUBSCRIBE/NOTIFY dialogs, registration and instant messaging
- By using Transaction User Interface we can build applications like JSR 180, Customized B2BUA
- Mainly Provides Interface for Sending and Receiving SIP Messages

## SIP Stack Object

- Interface for accessing the SIP Stack functionalities
- Provides APIs for
  - Management of SIP Stack
  - Register/Unregister TUs
  - Adding new Transports
  - Sending/Receiving SIP Messages



### **Transaction FSM**

- Implements the SIP Stack Transaction State Machine
- Resolves the Domain Names for Routing

### Transport S elector

- The Transport Selector is primarily responsible for:
  - Determining the fully-specified destination for an outgoing SipMessage, given a (possibly) partially-specified destination.
  - Deciding which instance of class Transport the outgoing SipMessage is to be sent on.
  - If necessary, filling out any fields in the SipMessage that depend on 1) (ie, the host in the topmost Via, unspecified host parts in Contact header-field-values that refer to this UA, Record-Route header-field-values that depend on which transport the message is being sent on, etc)
  - Serializing the outgoing SipMessage, and passing the serialized form to the Transport instance chosen in second point.



# UDP/TCP/TLS Transport

• Sends/Receives the SIP Message to Network

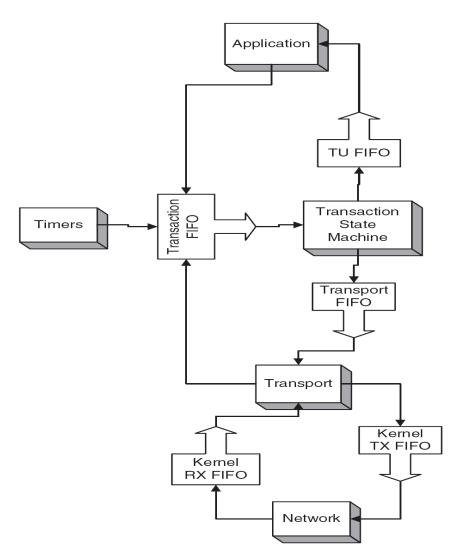
# MHS

- SipMessage parsing happens in phases. MsgHeaderScanner performs the first phase of parsing. MsgHeaderScanner is called from a transport. Framing of the message is an interaction between the transport and the scanner. The transport feeds the scanner consecutive chunks and the scanner reports when it has framed a message.
- The chunks containing the memory are handed back to the SipMessage for eventual deallocation. In addition, the scanner lexes the message and identifies the boundaries of header field values.
- Each header field value is parsed on demand. The instance of parser that is created for the header field value is determined by the header type accessed in the message



# FIFO

- FIFOs also known as queues, are a standard thread synchronization/buffering mechanism. FIFOs are the main way to move information between threads within reSIProcate. Since SIP is a message/event protocol, event FIFOs are a natural and error resistant mechanism for communicating among threads.
- There is a FIFO between every layer of reSIProcate.
- The following are the FIFOs present in Resiprocate
  - between the Transaction State Machine and the application (TU FIFO)
  - between the timers and the transaction state machine (transaction FIFO)
  - between the transaction state machine and the transports
  - between the transport and the network (kernel TX FIFO)
  - between the network and the transports (kernel RX FIFO)
- The actual FIFO is a template class (abstractFifo.hxx). The getNext method will wait until there is something in the FIFO.





# Using Resiprocate Stack



# Using Resiprocate Stack

- Transaction User Management
- SIP Stack Management
- SIP Message and Headers



### **Transaction User Management**

- To Create a new Transaction User the following needs to be done
  - Implement TransactionUser Interface
    - name() New Transaction User Name
  - Register Transaction User with the stack
- To send the SIP Message SIP Stack Object should be used and new Transaction User Reference should be passed.
- The Transaction User Class contains TU FIFO through which we can get the SIP Messages.
- Code Snippet



# SIP Stack Management

- The SIP Stack can be created as an Object inside the User created class
- Sip Stack Object provides Transaction Level APIs
- The following are the common APIs used
  - sipstack() Creates the SIP Stack
  - addTransport() adds Listening port
  - send() API call is used to send SIP Messages
  - buildFdSet() API builds the fdset of the Transport.
  - process() API call runs the SIP Stack. So it needs to be called periodically.
  - StackThread class can be used for doing the above functionalities.
  - The details of the APIs are present in the following location
  - Shutdown() API Call can be used to stop the stack



## Construction of SIP Message

- The Sip Message may be constructed by using Helper class present in the stack.
- The Helper Class provides static function calls
- The some of common API used
  - makeRequest()
  - makeResponse()
  - makeInvite()
  - makeCancel()
  - makeRegister()
  - makeSubscribe()
  - makeMessage()
  - makePublish()
  - **.**...
- Code Snippet



# Sip Message and Headers

- A central component of any SIP service is handling of SIP messages and their parts. SIP Messages are handled by SipMessage Class
- A SIP message consists of
  - Headers
  - Request/status line
  - Body.
- If we want to access/modify the headers, request/status line and body we have to use SipMessage class.
- SIP Headers can be grouped into two types
  - Single Instance
  - Multiple Instance
- SIP Headers may also consist of parameters
- Request Line and Status Line depend on the type of Message we have received that is whether Request or Response
- Body is the contents that is carried by SIP Message. For Example SDP.



### **Headers**

- Resiprocate Stack provides a uniform way of accessing SIP Headers. To access the headers following procedure has to be followed
  - Header RFC 3261- header access token. For From Header Token in "From"
  - In Resiporcate to access prefix "h\_". For From Header h\_From
  - For Multiple Instance header "s" will be appended to the RFC name.
  - For Example if RFC Name is Record Route then header access token is h\_RecordRoutes
  - header() overloaded method of SipMessage is used to access the header and assign value.
  - remove() overloaded method is used to remove the header
  - exists() overloaded method is used check the existence of the header
  - The Multiple Headers are accessed by stl fashion iteration.

#### Parameters

- Parameters are accessed from headers.
- p\_ should be used
- Example: const Data& tag = msg->header(h\_To).param(p\_tag); to access To tag
- param() overloaded method of SipMessage is used to access the header and assign value.
- remove() overloaded method is used to remove the parameter
- exists() overloaded method is used check the existence of the parameter



# Request/Status Line

- Special Types of Headers
- Accessed by h\_RequestLine and h\_StatusLine
- isRequest method and isResponse method
- RequestLine provides APIs like method(), uri(), getSipVersion()
- StatusLine provides responseCode(), statusCode(), getSipVersion()

# S ip Message Body

- getContents method returns Contents type
- Cast the Contents type to the required Content type found in Content-Type Header
- For Setting content type use setContents of SipMessage
- We can add new Content types and Parser without any modification to reSIP library.



# **Using Resiprocate DUM**



# Dialog Usage Manager - DUM

- The Dialog Usage Manager makes writing user agents easy by hiding complex SIP specifics. DUM provides user agent functionality including the handling of INVITE and SUBSCRIBE/NOTIFY dialogs, registration, and instant messaging. With DUM we can create applications like softphones, back-to-back user agents, and load generators.
- DUM implements Transaction User
- DUM does the following functions
  - Implements Offer/Answer
  - Manages Profiles
  - Manages AppDialogSetFactory (Equivalent to Call)
  - Manages and stores Handlers, which are a way of referencing usages
  - Manages redirections (RedirectManager)
  - Manages client authentication (ClientAuthManager)
  - Manages server authentication (ServerAuthManager)
  - Interface to add new Transports
  - Manages handles to Usages (HandleManager)
  - Provides interfaces to create new sessions as a UAC (invite, subscription, publication, registration, pager, others)
  - Provides interfaces to find particular usages based on a DialogId.



# **DUM Application**

- **DUM Appliaction** 
  - Create Stack
  - Create DUM
  - Add Transports
  - Create Profile
  - Set Profile Options
  - Set Handlers
  - Start Process Loop
  - Application should implement Handlers
  - DUM provides API for constructing SipMessage
  - Handle are used to send/receive SipMessages within a Dialog



# **Dialog Usages**

- When Dialog is created they establish an association between endpoints within the Dialog. This association is known as Dialog Usage (http://www.ietf.org/internet-drafts/draft-ietf-sipping-dialogusage-06.txt)
  - A Dialog initiated by INVITE Request has an INVITE Usage
  - A Dialog initiated by SUBSCRIBE has an SUBSCRIBE Usage
- In Resiprocate Usage concept is used. There are two types of Usages
  - 1. DialogUsage
  - 2. NonDialogUsage

The Usage provides the APIs that are required to send/receive Requests/Responses inside a SIP Dialog.

- Some Example of Usages are
  - 1. ClientRegistration
  - 2. InviteSession
  - 3. ...
- Some of the API examples in InviteSession are as follows
  - 1. info()
  - 2. Refer()
  - 3. ...



#### resip::ClientSubscription resip::BaseSubscription resip::ServerSubscription resip::DialogUsage resip::InviteSession 🗲 resip::ClientInviteSession resip::ClientOutOfDialogReq resip::ServerInviteSession resip::AppDialog resip::ClientPagerMessage resip::Handled resip::AppDialogSet resip::ClientPublication resip::NonDialogUsage resip::BaseUsage resip::ClientRegistration resip::ServerPublication resip::ServerOutOfDialogReq resip::ServerPagerMessage resip::ServerRegistration

#### **Dialog Usage Inheritance**

How to use Dialog Usage?



### **DUM Handles**

- Handles are used to access the Usage Object.
- Usages are derived from Handled Class. Handles can point to objects which subclass Handled.
  - Reasons for Accessing Usages through Handles are
  - Usages might get deleted by the time application uses it
  - Once created Handle will continue to exists even if Handled Object gets deleted. So it can throw exceptions
- Smart Pointer
  - Keeps track of referenced Object
  - Throws exception if not found

Handle point to InviteSessions, ClientRegistration, ClientSubscription ...

• Code Example

# **DUM HandleManager**

- HandleManager keeps track of Handles.
- The Reference of Handles are stored in HandleManager
- DialogUsageManager is subclass of HandleManager
- This the way the DialogUsageManager and Handles (Usages) are linked.
- DialogUsageManager keeps track of Usages



# Handlers

- Handlers are essentially Callbacks needs to be implemented by the application.
- The Callbacks are called from SipStack when a Response or New Requests in the context of the Usages are received.
- The Handlers usually return
  - Handles to Usages
  - The SipMessage received
- Example of a Handler is InviteSessionHandler. Some of the Callback APIs are
  - virtual void onNewSession(ClientInviteSessionHandle, InviteSession::OfferAnswerType oat, const SipMessage& msg)=0;
  - virtual void onNewSession(ServerInviteSessionHandle, InviteSession::OfferAnswerType oat, const SipMessage& msg)=0;
- Application should implement the InviteSessionHandler class and the APIs
- setHandler method in DialogUsageManager should be used to set the Handlers.



resip::MasterProfile

### **Profiles**

- Profiles are used to set the properties of User Agent like Outbound Proxy, User Name, User Authentication parameters, etc
- The Profile Hierarchy as shown

#### resip::Profile

#### resip::UserProfile

- Profile is base class. Its as various settable properties such as RegistrationTime, MaxRegistrationTime, SubscriptionTime, outBoundProxy, etc.
- Three types of APIs
  - setXXXX
  - getXXXX
  - unsetXXX
- If a property is not set then the default value in BaseProfile is taken.
- The User Profiles handles the User related configurations sunce as AOR, Credentials etc
- MasterProfile handles SIP Capability related configurations such as method types, methods, options, mime types etc
- There are two models of Profile settings
  - Single Profile
  - Mutli Profile



# Dialogs

#### DialogSet

- Container class holding a set of dialogs initiated from a common requests
- Share the same Call-ID and the same from tag in the request that generated the dialog

#### Dialog

Container class holding SIP RFC dialog details

#### AppDialog

- An Application can associate user data with Dialog/DialogSet AppDialog, AppDialogSet and AppDialogSetFactory classes
- This type of association may be required when there are multiple dialogs the Application has to track.



# Logger Module

- The Logger Module can be used for debugging purpose
- Log::initialize() call used to initialize the Log Module
- The following options can be used to print the Logs
  - Cout Prints on Standard Output
  - Syslog Prints to Syslog
  - File Prints to User Given File or resiprocate.log
  - Cerr Prints to Standard Error Output
- The Log Levels that can be set are as follows
  - CRIT
  - Err
  - Warning
  - Info
  - Debug
  - Stack
  - StdErr



### **Code Structure**



# **Code Structure**

#### l rutil

Various protocol-independent utility classes, used by all the other modules.

#### Stack

- -Core SIP stack functionality, including message parsing, message synthesis, and transaction handling.
- Dum
  - User Agent functionality, including handling of INVITE and SUBSCRIBE/NOTIFY dialogs, registration, and instant messaging.

#### repro

- Flexible SIP proxy framework.
- API Definitions can be found at http://www.estacado.net/resip-dox/



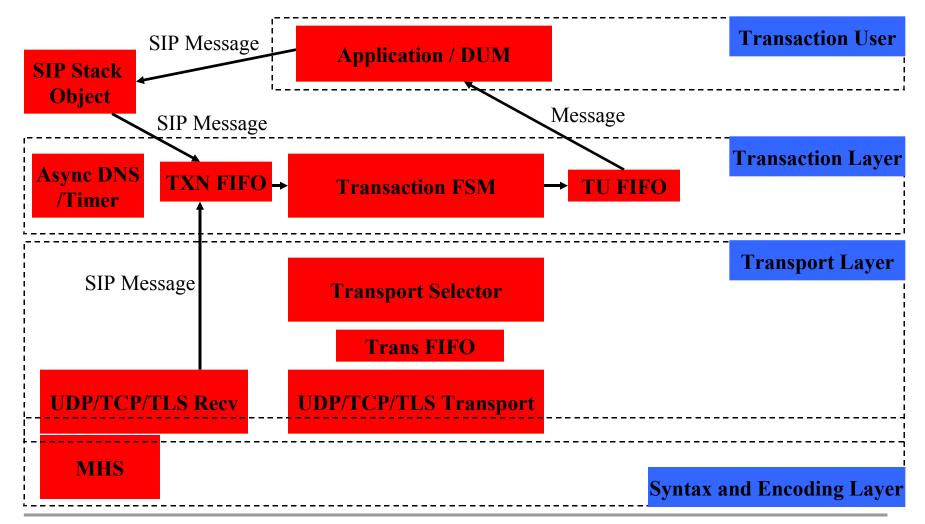
### Q&A



### **Backup Slides**



#### **ReSIProcate Stack Architecture**



Resiprocate Overview

www.resiprocate.org