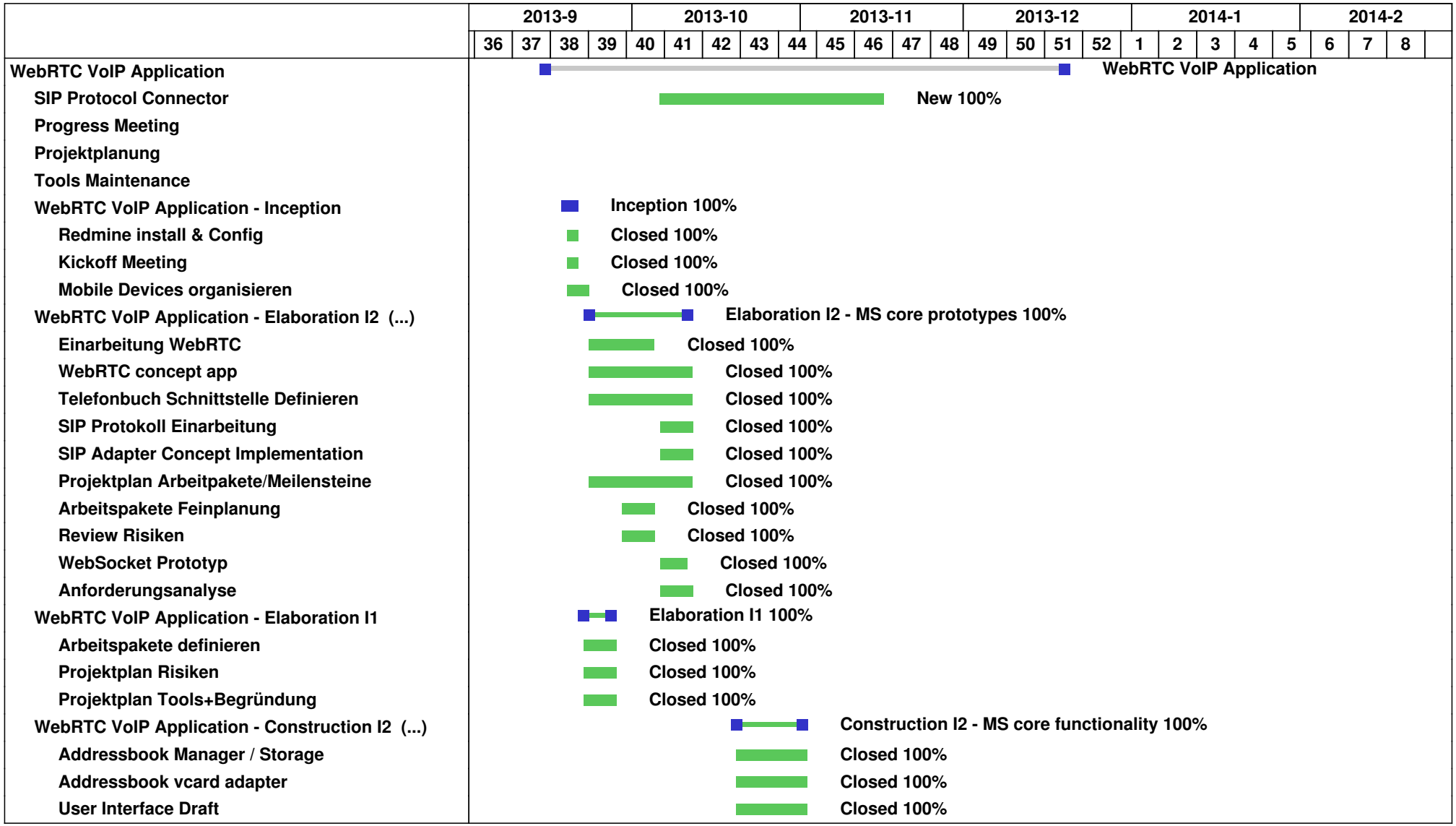
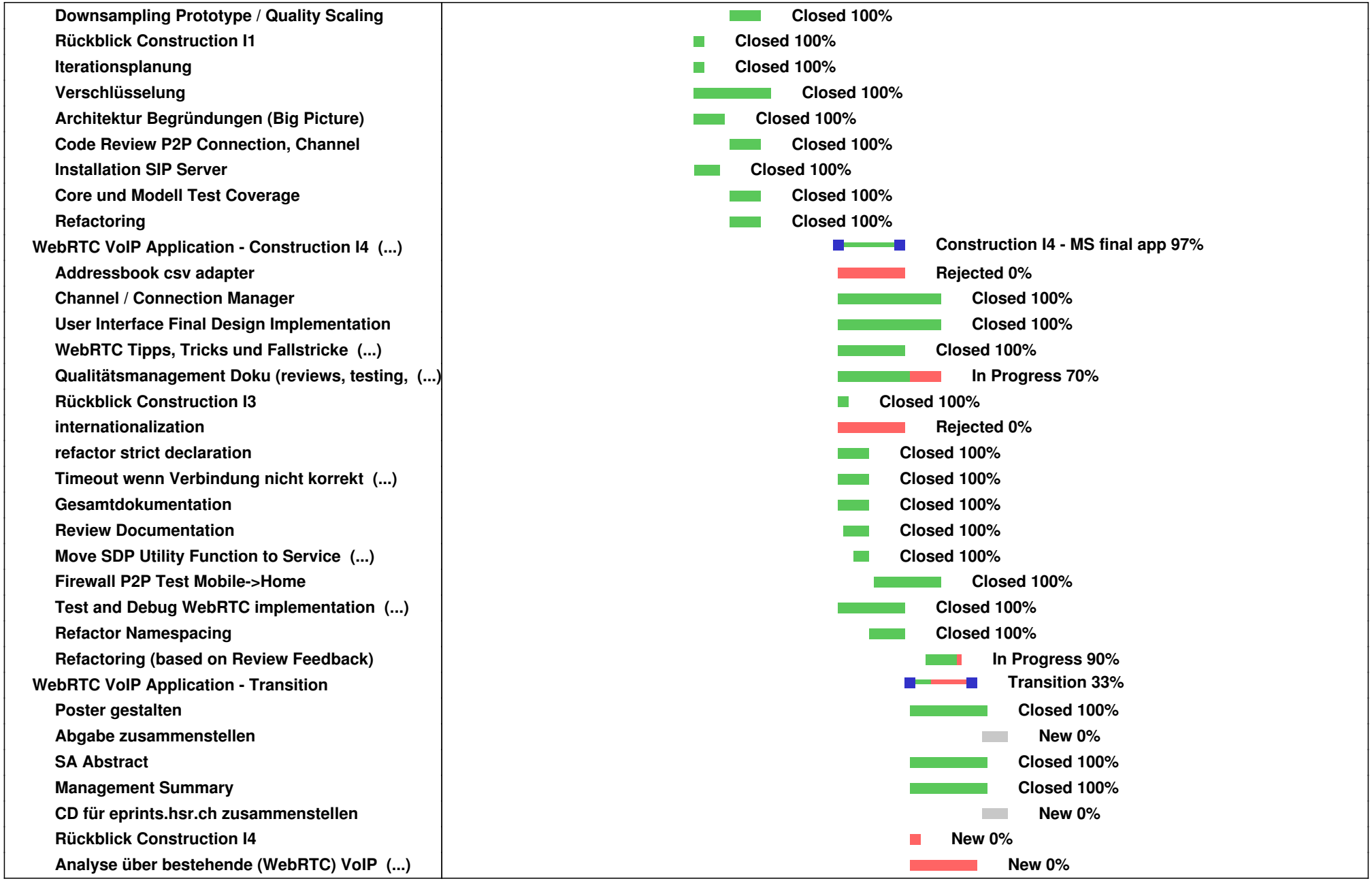
WebRTC VoIP Application




Downsampling Prototype / Quality Scaling
Closed 100%
Rückblick Construction I1
Closed 100%
Iterationsplanung
Closed 100%
Verschlüsselung
Closed 100%
Architektur Begründungen (Big Picture)
Closed 100%
Code Review P2P Connection, Channel
Closed 100%
Installation SIP Server
Closed 100%
Core und Modell Test Coverage
Closed 100%
Refactoring
Closed 100%
WebRTC VoIP Application - Construction I4 (...)
Construction I4 - MS final app 97%
Addressbook csv adapter
Rejected 0%
Channel / Connection Manager
Closed 100%
User Interface Final Design Implementation
Closed 100%
WebRTC Tipps, Tricks und Fallstricke (...)
Closed 100%
Qualitätsmanagement Doku (reviews, testing, (...)
In Progress 70%
Rückblick Construction I3
Closed 100%
internationalization
Rejected 0%
refactor strict declaration
Closed 100%
Timeout wenn Verbindung nicht korrekt (...)
Closed 100%
Gesamtdokumentation
Closed 100%
Review Documentation
Closed 100%
Move SDP Utility Function to Service (...)
Closed 100%
Firewall P2P Test Mobile->Home
Closed 100%
Test and Debug WebRTC implementation (...)
Closed 100%
Refactor Namespacing
Closed 100%
Refactoring (based on Review Feedback)
In Progress 90%
WebRTC VoIP Application - Transition
Transition 33%
Poster gestalten
Closed 100%
Abgabe zusammenstellen
New 0%
SA Abstract
Closed 100%
Management Summary
Closed 100%
CD für eprints.hsr.ch zusammenstellen
New 0%
Rückblick Construction I4
New 0%
Analyse über bestehende (WebRTC) VoIP (...)
New 0%

Finalize Doku	In Progress 50%
WebRTC VoIP Application - Construction I3	Construction I3 100%
SIP Channel Implementation	Rejected 0%
SIP Protocol Interface Interpreter / (...)	Rejected 100%
INVITE WebSocket termination or (...)	Rejected 30%
User Interface Final Design	Closed 100%
Rückblick Construction I2	Closed 100%
Systematische Performanceanalyse	Closed 100%
Json Remote Addressbook Adapter	Closed 100%
User management / storage	Closed 100%
Get App running in Chrome	Closed 100%
Iterationsplanung	Closed 100%
Schlussdokumentation Draft initialisieren	Closed 100%
Firewall Testing	Closed 100%
Send 'Bye' through RTCDataChannel	Closed 100%
WebRTC VoIP Application - Construction I1	Construction I1 100%
Umsetzung Kommunikation über SIP Proxy (...)	Closed 100%
Iterationsplanung	Closed 100%
Multikommunikation Konzept Prototype	Closed 100%
Rückblick Elaboration	Closed 100%
WebRTC Demo objektorientiert umsetzen	Closed 100%