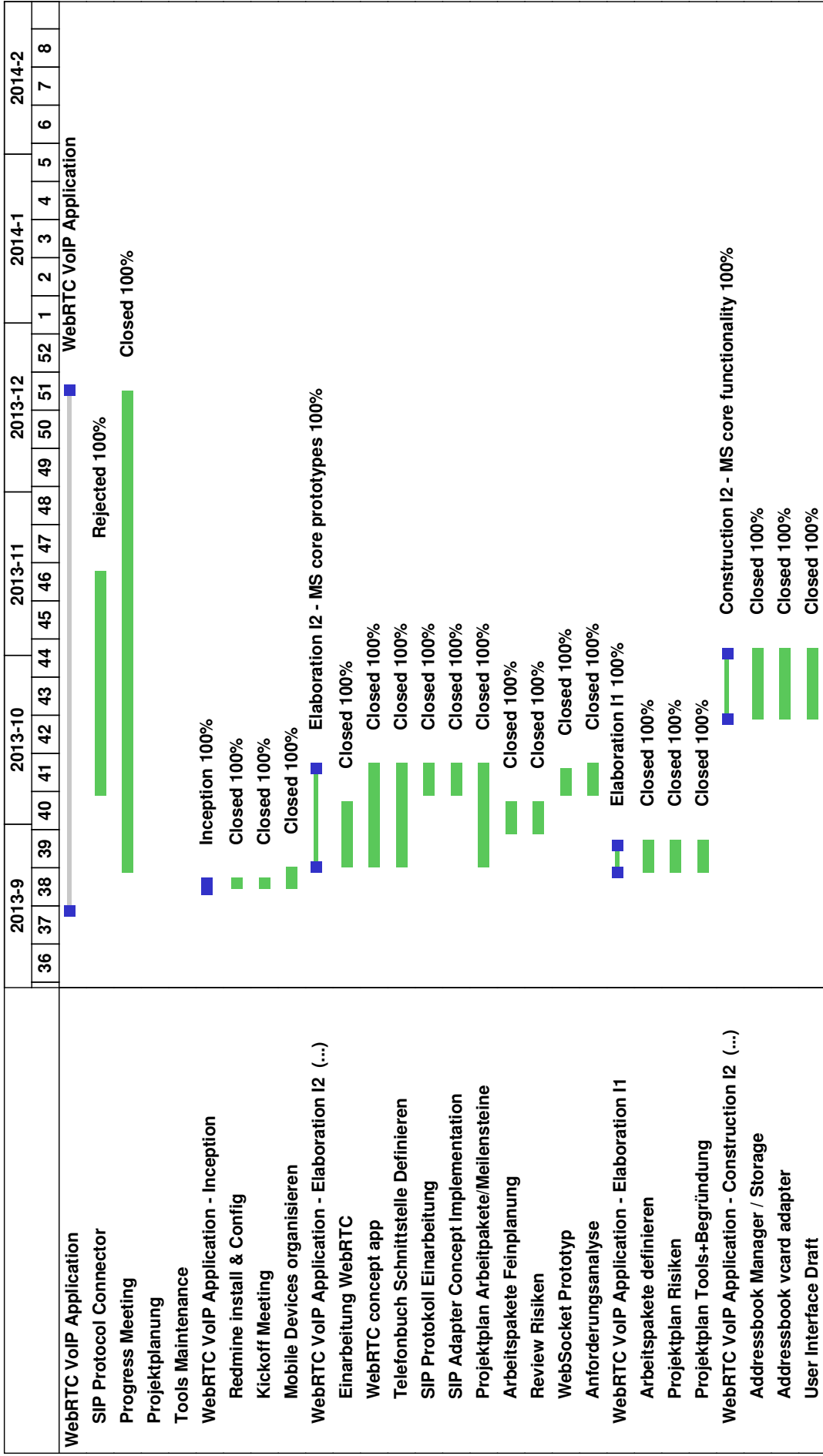




































WebRTC VoIP Application



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 100%
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, (...)	 Closed 100%
Rückblick Construction I3	 Closed 100%
internationalization	 Rejected 0%
refactor strict declaration	 Closed 100%
Timeout wenn Verbindung nicht korrekt (...)	 Closed 100%
Gesamtdokumentation	 Closed 100%
Review Dokumentation	 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)	 Closed 100%
WebRTC VoIP Application - Transition	 Closed 100%
Poster gestalten	 Transition 100%
Abgabe zusammenstellen	 Closed 100%
SA Abstract	 Closed 0%
Management Summary	 Closed 100%
CD für eprints.hsr.ch zusammenstellen	 Closed 100%
Rückblick Construction I4	 Closed 100%
Analyse über bestehende (WebRTC) VoIP (...)	 Closed 100%

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