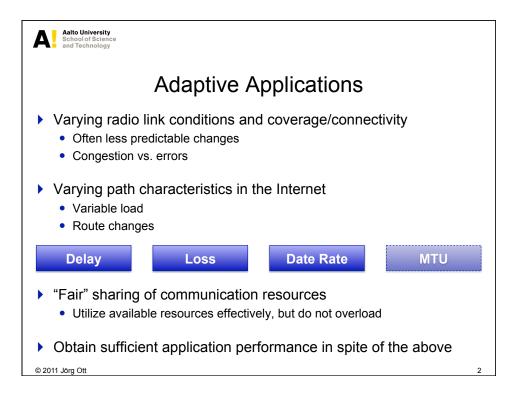
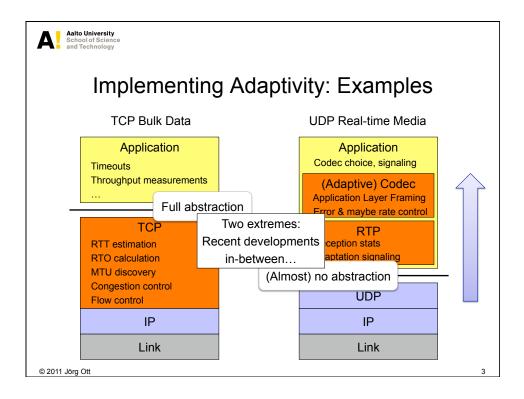
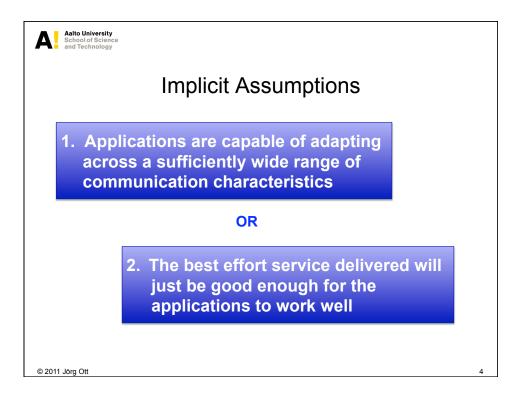
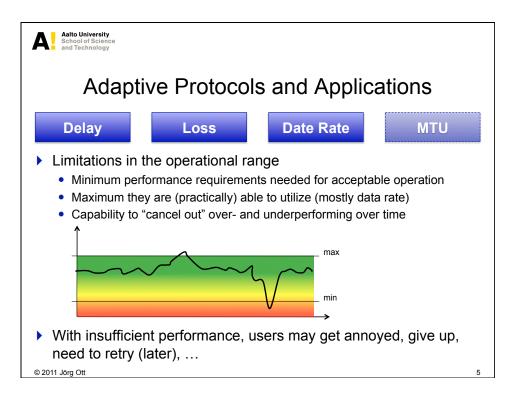
Aalto University School of Science and Technology	
Towards More Adaptive Voice Applications	
Jörg Ott <jorg.ott@aalto.fi></jorg.ott@aalto.fi>	
Md. Tarikul Islam, Cheng Luo Nagasai Panchakarla, Anssi Turkulainen	
13 October 2011	
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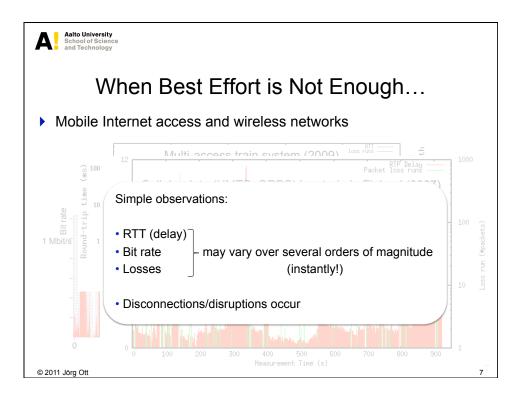


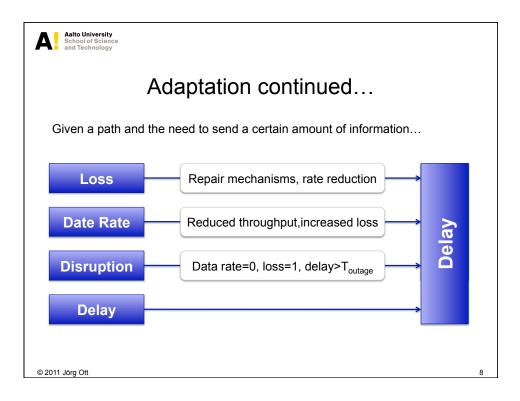


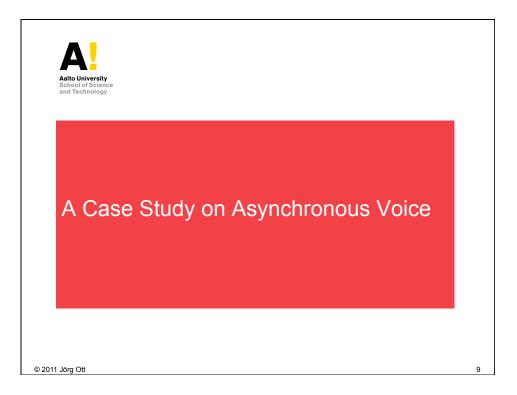




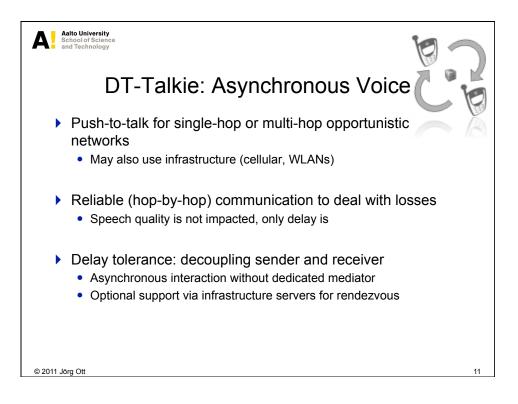
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Some Examples					
	Delay	Loss	Data rate	МТU	
Bulk data	Don't care as	long as TCP do (P2P ever	es not stall or dis h better)	connect	
Interactive web	RTT<300ms Interactivity = f(< 2% loss, delay) needs	100 kbit/s – 1 Mbit/s to be sufficient	1500 bytes ok	
Streaming	seconds Data rate = f(I	< 1% oss, delay) needs t	100 kbit/s – 100 Mbit/s to be sufficient	1500 bytes ok (could be larger)	
VoIP	< 200ms	< 5%	4 kbits – 100+ kbit/s	< 100s bytes	
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DTN-based Voice
 Plain and simple: record – send – forward – receive – playback Based upon user-indicated (button press) statements
User speech 1 2 3
Transmission 1 2 3
1 2 3 Subtleties: message size? • Semantic fragmentation (Application Layer Framing) • Keep talkspurts together ("MTU") • Good connectivity and short messages: interactive communication workable
User speech 1 2 3
Transmission 1 2 3
1 2 3
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